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# Scalable Computing: Practice and Experience

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INTRODUCTION TO THE SPECIAL ISSUE ON ROLE OF SCALABLE COMPUTING AND DATA ANALYTICS IN EVOLUTION OF INTERNET OF THINGS

J.N. SWAMINATHAN∗, GOPI RAM† AND SUREKA LANKA‡

The evolution of Internet of Things has given way to a Smart World where there is an improved integration of devices, systems and processes in humans through all pervasive connectivity. Anytime, anywhere connection and transaction is the motto of the Internet of things which brings comfort to the users and sweeps the problem of physical boundary out of the way. Once it has come into the purview of developers, new areas have been identified and new applications have been introduced. Small wearables which can track your health to big automated vehicles which can move from one place to another self navigating without human intervention are the order of the day. This has also brought into existence a new technology called cloud, since with IoT comes a large number of devices connected to the internet continuously pumping data into the cloud for storage and processing. Another area benefited from the evolution of IoT is the wireless and wired connectivity through a wide range of connectivity standards.

As with any technology, it has also created a lot of concerns regarding the security, privacy and ethics. Data protection issues created by new technologies are a threat which has been recognized by developers, public and also the governing body long back. The complexity of the system arises because of the various sensors and technologies which clearly tell the pattern of the activities of the individual as well an organization making us threat prone. Moreover, the volume of the data in the cloud makes it too difficult to recognize the privacy requirement of the data or to segregate open data from private data. Data analytics is another technology which supposedly increases the opportunity of increasing business by studying this private data collected from IoT and exploring ways to monetize them. It also helps the individual by recognizing their priorities and narrowing their search. But the data collected are real world data and aggregation of this data in the cloud is an open invitation to the hackers to study about the behaviors of the individuals.

The special issues of Scalable Computing has attracted related to the Role of Scalable Computing and Data Analytics in Evolution of Internet of Things has attracted 28 submissions from which were selected 12. In what follows we present them shortly:

1. Murali et. al has executed on detecting brain tumor using thresholding and histogram techniques.
2. Suresh et. al has introduced an ensemble integrated security system with cross breed technique.
3. Shahidha Bhamu has done background modeling using a Q-Tree based foreground segmentation.
4. Agalya et. al has executed topological ordering signal selection techniques for iot based devices. The work has been funded under dst-inspire program.
5. Karthikeyan et. al has introduced secured identity based cryptosystem approach for intelligent routing protocol in vanet.
6. Indragandhi et.al has developed an application based ETLP scheme on heterogeneous multicore embedded system for real time image processing.
7. Putta Venkata Subbaiah et.al has done a study to analyze enhancement techniques on sound quality for bone conduction and air conduction speech processing.
8. Kuppusamy et.al has introduced a novel approach based on modified cycle generative adversarial networks for image steganography.

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‡Stamford International University, Bangkok, Thailand.
9. Jairam Naik et al. has executed minimizing deadline misses and total run-time with load balancing for a connected car systems in fog computing.
10. Antony Joseph Rajan et. al has proposed long and strong security using reputation and ecc for cloud assisted wireless sensor networks.
11. Pavithra et. al has detected and classified 2D and 3D hyper spectral image using enhanced harris corner detector.
12. Rengarajan et. al has proposed secured privacy-conserving authentication scheme using cuckoo filter in vanet.
13. Hema et. al has introduced logistics optimization in supply chain management using clustering algorithms.
14. Jayanthi et. al has executed acoustic feedback cancellation in efficient hearing aids using genetic algorithm.
15. Kusma et. al has analysed on deep learning methods for ECG based cardiovascular disease prediction.
16. Finally Rajalakshmi et. al has introduced a hybrid intrusion detection system for mobile adhoc networks using fbid protocol.

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BRAIN TUMOR DETECTION FROM MRI USING ADAPTIVE THRESHOLDING AND HISTOGRAM BASED TECHNIQUES

E. MURALI* AND K. MEENA†

Abstract. This paper depicts a computerized framework that can distinguish brain tumor and investigate the diverse highlights of the tumor. Brain tumor segmentation means to isolated the unique tumor tissues, for example, active cells, edema and necrotic center from ordinary mind tissues of WM, GM, and CSF. However, manual segmentation in magnetic resonance data is a time-consuming task. We present a method of automatic tumor segmentation in magnetic resonance images which consists of several steps. The recommended framework is helped by image processing based technique that gives improved precision rate of the cerebrum tumor location along with the computation of tumor measure. In this paper, the location of brain tumor from MRI is recognized utilizing adaptive thresholding with a level set and a morphological procedure with histogram. Automatic brain tumor stage is performed by using ensemble classification. Such phase classifies brain images into tumor and non-tumors using Feed Forwarded Artificial neural network based classifier. For test investigation, continuous MRI images gathered from 200 people are utilized. The rate of fruitful discovery through the proposed procedure is 97.32 percentage accurate.

Key words: MRI, Morphological, Thresholding, Brain tumor, Level set, Histogram

AMS subject classifications. 92B20, 68U10

1. Introduction. One of the necessary steps in most of the medical imaging analysis is to extract the boundary of a locality of our interest. Precise location is vital in cerebrum tumor identification. The level of exactness can be expanded through the use of computer aided system. Such framework can assist the radiologist with detecting brain tumor all the more properly. Magnetic resonance image have been extensively approved for imaging because of its competences are generating precise brain scan in less interval time, with changing contrasts, such as T1-weighted, T2-weighted and Flair. Respectively difference yields intensity dissimilarities in MRI scans. Cerebrum is a 3-D composite structure. Numerous views of brain images are axial, sagittal and coronal. In axial view the brain image was divided by a horizontal plane. The sagittal view, the brain image is divided into right and left part of the median plane and in coronal view, the brain is divided into ventral and dorsal by frontal plane. Figure 1.1 shows views of MRI brain scan.

Image enhancement techniques are used to develop the image feature for human perception. They are defined as methods of image processing such that, the result is more appropriate than the original image. Histogram equalization is a fundamental tool in image enhancement. It is likely to aid in perceptive of how they function on digital images. The equalization of intensity level method is an enhanced image with dynamic range, which will tend to have high difference. The increased in contrast is owing to the circumstance that the average intensity level of the equalized image is enhanced by the original. Overall, the increased in intensity is due to the reality that the average intensity level in the equalized image histogram is better than the original.

1.1. Related work. Mouli Laha [1] introduced an approach that embedded iterative thresholding in parallel to Otsu's global threshold. There is no need to set any limit value before the procedure. Similarly, the cropping skull stripped techniques from brain scan is efficient for three views. This method operates very well in the case of 2D.

K Sudharani and T.C Sarma [2] proposed an algorithm that calculates the threshold qualitatively and quantitatively by methods of the standard images. It diminishes misclassification errors where the insignificant
disparity inside each object by its own can’t ensure the required result. Sudipta Roy, Bhattacharyya D [3] suggested method is an automatic, multi-scale, brain tissue segmentation algorithm that reaches to very good acceptable result. Proposed methodology has the ability to detect irregularity in the brain lesion if any. In that paper, they proposed a level set-based minimization system for the variation segmentation method by iterative perceptions and mutable the speed function of level set.

Christopher Bowels, Chen Qin [4] proposed a model for brain tissues segmentation by using an image fusion algorithm, irrespective of primary pathology. They show that an seemingly good FLAIR image can be fused from the T1-weighted image of a subject, and the distinctions among this fake FLAIR and the true FLAIR can be coupled with validity from the true FLAIR to indicate tissue location. The drawback of the proposed model is that it needs both FLAIR and T1-weighted images and any important variations in T1-weighted gaining proprieties may negatively impact on performance. The researchers isolate the images into n-scale layers with completely separate visual categories [5] to address the linguistic gap between low level and high level features. For each layer, FDA was conducted and a multi scale histogram was finally constructed. Finally, the SVM classifier is used to classify. Using Gaussian function and down sampling, each image is divided into multi-scale. By using Gaussian pyramid, several distinctive histograms are constructed. The accuracy of the visual dictionary has been improved by extracting spatial features using sparse coding model. In the classification of medical images, non-negative scanty coding is performed to address the issue of optimization.

1.2. Proposed methodology. Figure 1.2 depicts the overall approach of the suggested system. In the frequency domain, smoothing is achieved by decreasing the parts of elevated frequency. The fundamental filtering model is

\[ G(u, v) = H(u, v) \cdot F(u, v) \]  

(1.1)
The image was filtered with Fourier transform $F(u,v)$ and filtered function is $H(u,v)$. From that point onward, gray level normalization is connected to change the scope of the pixel intensity. By utilizing anisotropic diffusion filter to diminish the image noise without evacuating critical parts of the image content, commonly edges, lines or different subtleties that is vital for the elucidation of the image. By using imsubtract, features are subtract from one image from another image, or to subtract a constant value from an image. Imsubtract subtracts each pixel value from the comparing pixel in the other input image in one of the input image and returns the result in the related pixel in the output image.

In adaptive thresholding modifications the threshold is dynamically over the image. Thresholding commonly takes a grayscale or colour image as an input and, in the least complex usage, yields a binary image speaking to the segmentation. The process of shape analysis consists of two main steps: (1) the extraction of image components of the target (e.g., area, boundary, network pattern and skeleton), (2) the description of the shape features (e.g., size, perimeter, circularity and compactness) and finally show the detected tumor area.

1.3. Organization of the paper. This paper is organized as follows. Section 1 describes the introduction, literature review and block diagram of the proposed method. Section 2 shows the various methodologies such as Histogram Method, Adaptive Threshold, Anisotropic Diffusion filtering and Level set method in detail. Section 3 presents the experimental results and discussion. Section 4 concludes the paper.

2. Methodology. The algorithm introduced in this section is performed by an automatic thresholding method in its place of physically changing the threshold for each image. The threshold range is done automatically dependent on the mean and standard deviation of every area among four sub regions.

2.1. Histogram Method. This technique is used to convert the input image into a gray image. A histogram equalization technique has been connected to this grayscale image to enhance the intensity image contract, which aids to recognize the brightest portion of the image. Before applying adaptive thresholding with ostu value 0.31, we shift just over white pixels into full white and others into dark ones. The image is converted to binary after that, morphological dilation and erosion with basic component of $[1;1;1]$ that has better segmentation was implemented at that point. The pixels less than 250 pixels have been expelled as this precisely portioned the districts of the tumor. At long last tumor area is recognized. The histogram strategy for ordering a pixel-by-pixel image characterizes single or multiple thresholds. A fundamental way to determining the threshold value $T$ is by studying the histogram for maximum values and fining the smallest point, typically between two consecutive maximum values of histograms. The statistics method will provide a good result, when a histogram is bi-modal. By equating the gray value of each pixel with the specified threshold $T$, a pixel can be classified into one or two classes. An image $f(x,y)$ may be split into two classes by a gray value limit $T$.

$$g(x,y) = \begin{cases} 1, & \text{if } f(x,y) > T \\ 0, & \text{if } f(x,y) \leq T \end{cases}$$ (2.1)

Here $g(x,y)$ is the segmented image with two binary values “1” and “0” and $T$ is the threshold assigned to the smallest point between two histogram peak values.

2.2. Adaptive Threshold. Thresholding is entitled adaptive since a different type of threshold is used for different regions in the image. Thresholding expect that the image has pixel values generally different from the background. This technique approves the threshold value $T$ to alternate depending on the image’s progressively different function characteristics. Threshold $T$ relies on the coordinated spatial $(x,y)$ itself. This technique is adapted with some optimization. Following is the outline of adaptive thresholding:

1. Binarizing the image with a single threshold $T$;
2. Thinning the threshold image;
3. Remove all branch points in the thin image;
4. All remaining endpoints are located in the analysis queue and used as an initial point for tracking;
5. Track the region with threshold $T$
6. If the region has passed, $T = T - 1$, go to 5

The adaptive thresholding algorithm used a recursive filter to determine the nearby weighted mean in the image just along the row, or set of rows. Here, to calculate the local mean, we use symmetrical 2D Gaussian
smoothing. This is slower yet gradually wide. Median filtration is an alternative to the mean and offers the option to use a fixed threshold relative to the mean / median.

Let \( f_s(n) \) is the sum of the \( s \) pixel values at point \( n \).

The resulting \( T(n) \) is either 1 or 0 depending on whether it is \( t \) percent darker than the previous \( s \) pixels average value.

\[
g(x, y) = \begin{cases} 
1, & \text{if } P_n < \left( \frac{f_s(n)}{s} \right) \left( \frac{100 - t}{100} \right) \\
0, & \text{otherwise}
\end{cases}
\] (2.2)

2.3. Anisotropic Diffusion filtering. In order to lighten the noise effects, noise reduction is frequently used before segmentation, characterization and recognition to expel or lessen the noise. Noise reduction is an important approach for image processing that has broad implementation in various areas [7, 8]. The key is to decrease the noise without breaking down the vital highlights in the image. In this way, noise decrease has two objectives. One is to expel the noise from the image, and the other is to save the imperative highlights, for example, the edges in the image. A uproarious image can frequently be shown as one of the two models depending on the type of noise: a linear model and a nonlinear model. A MR image is commonly demonstrated as a noise model.

The anisotropic diffusion filtering is a general scale space way to deal with edge detection presented by Perona and Malik. In their work they moved from the linear scale-space model which considers the filtered image \( I(x, y, t) \) and unique image \( I_0(x, y) \) implanted in a novel group of functions characterized by

\[
I(x, y, t) = I_0(x, y) \cdot G(x, y, t)
\] (2.3)

The simple equation of anisotropic diffusion as presented in [5] is

\[
\frac{\partial I(x, y, t)}{\partial t} = \text{div}[g(|\Delta I(x, y, t)||\Delta I(x, y, t)|)]
\] (2.4)

where \( t \) is the time parameter, the original image is \( I(x, y, 0) \), the gradient of the image version at the time \( t \), and the so-called conductivity function is \( g \).

2.4. Level set method. Interested object’s rough boundaries are divided by thresholding method. The extracted portion is regarded as a level-set method initialization [9]. Methods of level-sets rely on partial differential equation to the surfaces of model deformation. Level sets techniques rely on two key embedding techniques; first, the implantation of the surface as the zero level set of a greater velocity to this greater point set feature. A level set equation represents the assessment of the shape or surface. The solution to which this
partial differential equation tends is calculated iteratively by updating at each interval of moment, below is shown the overall form of the level set equation:

$$\frac{\partial \varphi}{\partial t} = -|\nabla \varphi| \cdot F$$

(2.5)

Here, $F$ is the frequency that defines the assessment of the level set. By using $F$, given a specific initialization of the level set function, we can guide the level set to the different areas or shapes. It is also essential to have an initial mask for the level set function, which can take the form of a two-dimensional square or any other closed form. In this article, threshold findings act as a seed for a level-set method and the final tumor region is acquired after a few iterations.

2.5. Feed Forward Artificial Neural network. Feed forward Artificial Neural Network One of the simplest feed forward neural networks (FFNN, consists of three layers: an input layer, hidden layer and output layer. In each layer there are one or more Processing Elements (PEs). PEs is meant to simulate the neurons in the brain and this is why they are often referred to as neurons or nodes. PE receives inputs from either the outside world or the previous layer. There are connections between the PEs in each layer that have a weight (parameter) associated with them. This weight is adjusted during training. Information only travels in the forward direction through the network - there are no feedback loops.

3. Results and discussion.

3.1. Performance evaluation. Table 3.1 gives the details of the tumor are present along with the location of it in the brain according the hemispheres of the brain.

The following performance metrics are used to evaluate the efficiency of the proposed algorithm.

- True Positive ($TP$) = the no. of images correctly identified as tumor
- True Negative ($TN$) = the no. of images correctly identified as healthy
- False Negative ($FN$) = the no. of images incorrectly identified as healthy.
- False Positive ($FP$) = the no. of images incorrectly identified as tumor.

| Table 3.1 |
| Tumor segmentation and its location |
|---|---|---|---|---|
| Input Image | Histogram Equalization | Tumor Outline | Brain Area (B) | Tumor Area (T) | Ratio(B/T) |
| ![Image](image1) | ![Image](image2) | ![Image](image3) | 12046 | 114 | 0.946 |
| ![Image](image4) | ![Image](image5) | ![Image](image6) | 24283 | 1435 | 5.90 |
| ![Image](image7) | ![Image](image8) | ![Image](image9) | 1523 | 98 | 0.639 |
| ![Image](image10) | ![Image](image11) | ![Image](image12) | 21652 | 1142 | 5.274 |
| ![Image](image13) | ![Image](image14) | ![Image](image15) | 32970 | 848 | 2.572 |
| ![Image](image16) | ![Image](image17) | ![Image](image18) | 126793 | 3368 | 12.570 |
**Fig. 3.1. Tumor detection using level set and bounding box with bias corrected image**

**Accuracy**: It is a measure of the closeness of the measurements to true value

\[
\text{Accuracy} = \frac{\sum (TP + TN)}{\sum (TP + TN + FN + FP)} \times 100
\]

(3.1)

**Precision**: It is definite as the accuracy as the combination of both actuality and exactness

\[
\text{Precision} = \frac{\sum (TP)}{\sum (TP + FP)} \times 100
\]

(3.2)

3.2. Results. In this module, the suggested structure is the detection of tumor slices in which the boundary box is situated around the tumor and acts as a seed for brain tumor segmentation (Figure 3.1). The algorithm set threshold and level set is used to obtain the accurate tumor.

Table 3.2 shows the segmentation of the image with delta value is 0.1429 and different threshold value for the different images.

<table>
<thead>
<tr>
<th>Subject</th>
<th>Density</th>
<th>Threshold</th>
</tr>
</thead>
<tbody>
<tr>
<td>P1</td>
<td>0.0745</td>
<td>0.3725</td>
</tr>
<tr>
<td>P2</td>
<td>0.0487</td>
<td>0.2196</td>
</tr>
<tr>
<td>P3</td>
<td>0.7941</td>
<td>0.2588</td>
</tr>
<tr>
<td>P4</td>
<td>0.0961</td>
<td>0.2471</td>
</tr>
<tr>
<td>...</td>
<td>...</td>
<td>...</td>
</tr>
<tr>
<td>P212</td>
<td>0.1669</td>
<td>0.1569</td>
</tr>
<tr>
<td>Mean</td>
<td>0.2869</td>
<td>0.2492</td>
</tr>
<tr>
<td>Std</td>
<td>0.0141</td>
<td>0.0123</td>
</tr>
</tbody>
</table>
Table 3.3

Performance of brain tumour segmentation

<table>
<thead>
<tr>
<th>Subject</th>
<th>TP</th>
<th>TN</th>
<th>FP</th>
<th>FN</th>
<th>Precision</th>
<th>Recall</th>
</tr>
</thead>
<tbody>
<tr>
<td>P1</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>P2</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>P3</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>P4</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
</tr>
<tr>
<td>P212</td>
<td>1</td>
<td>0</td>
<td>0</td>
<td>0</td>
<td>1</td>
<td>1</td>
</tr>
<tr>
<td>Total</td>
<td>212</td>
<td>182</td>
<td>25</td>
<td>0</td>
<td>5</td>
<td>99.52</td>
</tr>
</tbody>
</table>

Table 3.4

Performance of the proposed system

<table>
<thead>
<tr>
<th>Total Number of Images</th>
<th>TP</th>
<th>TN</th>
<th>FP</th>
<th>FN</th>
<th>Sensitivity</th>
<th>Specificity</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>212</td>
<td>182</td>
<td>25</td>
<td>0</td>
<td>5</td>
<td>97.76</td>
<td>100</td>
<td>97.32</td>
</tr>
</tbody>
</table>

Of the 187 images taken with the tumor, 182 images were evaluated by the framework efficiently. Other 5 images are incorrectly acknowledged without a tumor. The reason for this failure was that there is no clear difference in pixel intensity between the tumor region and the rest of the brain. The framework correctly recognized 25 images without a tumor as images without a 100 percent successful tumor. The outcome of the suggested framework is shown in Table 3.4.

As discussed above, segmentation of brain tumour plays an important role in diagnostic procedures. The accurate segmentation helps in clinical diagnostic, but also helps to increase the lifetime of the patient. Based on the conventional histogram equalization algorithm, the paper presented an adaptive gray level mapping algorithm which takes the entropy and visual effects as the target function. The selection rule of parameter in different conditions and the identification method of the image gray type are also presented. The result shows that the implemented method helps in detection of enhancing tumour as well as specifying tumour to the actual tumour region only.

4. Conclusion. In identifying tumor tissues in the medicine sector, this paper suggested an outstanding and innovative classification of brain image. The paper provided an adaptive gray level mapping algorithm based on the histogram equalization algorithm that takes entropy and visual effects as the target function. The enhanced images are reconfigured using histogram equalization to reconfigure their pixel levels in histogram techniques. Developing more suitable techniques for detecting malignant brain tumors that are tiny in size would be excellent. The real tumor size can also be calculated from the 3D image.

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AN ENSEMBLE INTEGRATED SECURITY SYSTEM  
WITH CROSS BREED ALGORITHM 

SURESH MUNDRU∗AND K. MEENA†

Abstract. Blockchain and IoT are two technologies are most widely popular in present scenario, but technologies are more complicated. The blockchain used to transforms storage and data analysis. In recent years, the blockchain is at the heart of computer technologies. It is a cryptographically secure distributed database technology for storing and transmitting information. Various attacks are done in many networks. Many research articles discussed about the security issues over the IoT based secure using block chain technology. In this paper, an Ensemble Integrated Security System (EISS) is introduced to improve the security for the heterogeneous network which consists of normal and abnormal nodes which is processed with the block chain, IoT. Results show the performance of the OUATH-2 and EISS algorithm.

Key words: Blockchain, Internet of things, Networks.

AMS subject classifications. 68M11, 68M10

1. Introduction. Security is most widely used in many applications. In routing protocols it is very important to secure the routing. If the WSN is integrated with IoT and blockchain it becomes more compatible for security. IoT is the fast-growing technology in the present world [1]. In 2015, i.e., around 20 years after the term was authored, the IEEE IoT Initiative discharged a report whose principle objective was to set up a benchmark meaning of the IoT, with regards to applications extending from little, confined frameworks obliged to a particular area, to enormous worldwide frameworks made out of complex sub-frameworks that are geologically circulated [2]. In this archive, we can discover an outline of the IoT’s design necessities, its empowering advancements, just as a brief meaning of the IoT as an “application space that incorporates distinctive innovative and social fields”. At its center, the IoT comprises of arranged items that sense and assemble information from their environment, which is then used to perform robotized capacities to help human clients. The IoT is still relentlessly developing around the world, on account of extending Internet and remote access, the presentation of wearable gadgets, the falling costs of installed PCs, the advancement of capacity innovation and distributed computing [3]. Today, the IoT pulls in a large number of research and modern interests. As time passes, littler and more intelligent gadgets are being executed in numerous IoT areas, including lodging, exactness agribusiness, foundation observing, individual medicinal services, and independent vehicles just to give some examples.

Blockchain is the technology which is used to improve the performance in terms of security. Users can use various private and public keys to solve the security issues to transfer the data. The most serious issue with IoT security is that “there is no most concerning issue [4] [5] [6].” IoT has more mind-boggling details than conventional data innovation (IT) framework. It is significantly more liable to comprise of different equipment and programming items. As indicated by Forrester senior investigator Merit Maxim, the three primary regions of IoT security are a gadget, arrange, and back-end, which can all be objective and we ought to be cautious about.

Providing security for the IoT and nodes in the Network. In this paper, an ensemble integrated security system (EISS) is introduced to provide security between the nodes. To improve the performance of the security

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1. S. Mundru, K. Meena

Fig. 1.1. Architecture Diagram for EISS

within the nodes which are integrated with IoT and blockchain. All the domains are integrated and called as EISS.

2. Problem Statement. The present problem is addressed in IOT devices is security. Many routing protocols have the problem with security and certainty. Many researches have been done on providing security and predictable issues with IOT and routing protocols. Blockchain is updated technology to improve the security in IOT and routing protocols.

3. Related Work. A basic security challenge of the IoT originates from its consistently extending edge. In an IoT arrange, hubs at the edge are potential purposes of disappointment where assaults, for example, Distributed Denial-of-Service (DDoS) can be propelled [7]. Inside the IoT edge, a lot of adulterated hubs and gadgets can act together to fall the IoT administration arrangement, as observed as of late in botnet attacks [8].

A main issue of disappointment not exclusively is a danger to accessibility, yet additionally to classification and approval [9]. A concentrated IoT does not give worked in ensures that the specialist co-op won’t abuse or alter clients IoT information. Besides, classification assaults emerge from personality parodying and dissecting directing and traffic data. In an information driven economy, ensures are important to anticipate misappropriation of IoT information.

IoT faces classification assaults that emerge from character parodying and examining steering and traffic data, just as uprightness assaults, for example, change assaults and Byzantine directing data assaults [10]. Information honesty in the incorporated IoT arrangement is tested by infusion assaults in applications where basic leadership depends on approaching information streams. IoT information modification, information burglary and personal time can bring about shifting degrees of misfortune. Guaranteeing security is vital in a framework where keen gadgets are required to connect self-ruling and take part in financial exchanges. Current security arrangements in the IoT are incorporated, including outsider security administrations. Utilizing blockchains for security strategy implementation and keeping up openly auditable record of IoT connections, without relying upon an outsider, can demonstrate to be profoundly beneficial to the IoT.

IoT frameworks produce huge volumes of information that require arrange network and power, preparing and capacity assets to change this information into significant data or administrations. Close to dependable availability and system versatility, digital security and information protection of are significant significance in utilizing IoT systems. Right now, unified engineering models broadly used to validate, approve and interface various hubs in an IoT organize. With the developing number of gadgets to many billions, incorporated frameworks will separate and bomb when they brought together server winds up inaccessible. Decentralized IoT design was proposed to understand this issue, wherein it moves away a portion of the system handling
assignments to the edge [11]. For example, in haze registering models, a portion of the basic activities that used to be handled by cloud servers are currently relegated to be performed by IoT centers or haze [12]. Distributed (P2P) engineering gives another arrangement, where neighbouring gadgets legitimately connect with one another in lattices to distinguish, verify and trade data without utilizing any incorporated hub or operator between them [13].

4. **Blockchain.** The blockchain, consists of a chain of blocks. In every block, the data structure is allowed to blockchain to save the transactions done on every block which is linked to the chain by cryptography. In blockchain there are basic fundamental attributes such as Saved, transparent, and decentralized [14]. Every transaction in blockchain is safely and communicate with each other based on the trust-less method, i.e there is no need to believe another device and third parties. Especially in this paper, the blockchain technology is used to save every data into the each block and uses the encryption and decryption with the key. It is very powerful to use the algorithm for security.

5. **An Ensemble Integrated Security System (EISS).** This section explains about the functionality of the Encryption and decryption algorithm and how the IoT and blockchain are integrated in this. RSA is a very efficient and fast encryption algorithm that is used for securing data with the public-key. In this scenario, RSA is used to provide security at every node which is integrated with blockchain and generates a key for every block. Maintaining the secret keys at the block level is very difficult. The key generation is also very fast for every block and this also maintains the large data at every block. At the network setup, the integration of RSA and blockchain is implemented with the efficient network setup and security. The total no of nodes within the network is based on the data allowed at every node. The integrated system is implemented at network.

The integration of RSA and Block chain at every network is formalized by:

**KeyGen:**

\[ (E,q,a,b,G,n,h; d,Q) \]

where

- \( E \) is variable with elliptic curve

\[ y^2 = x^3 + ax + \frac{b}{F_q} \]  

- \( q \) is prime

\[ 2^{256} - 2^{32} - 2^9 - 2^8 - 2^7 - 2^6 - 2^4 - 1 \]  

while \( a,b : a = 0, b = 7 \)

- \( G,n \): consider random base point in \( E \) with prime order \( n \).
- \( h \): hash, instantiated with SHA1
- Signing key: \( d = [1,n-1] \)
- Verification key: \( Q = dG \in E \)
- \( \text{Sign}(d : m) : (r,s) \in F^2 \)

\[ \text{Verify}(Q; r,s) : (s,s \in [1,n-1]) \text{ and } (v = r), \text{where } v = \text{the x-coordinate of point} \]

\[ s = k^{-1}(hm + d \cdot r) \mod n \]

\[ a_j < 2^{j(n+1)} \mod (n+1), j = 1, \ldots, 2^y \]

\[ h(a_1) \oplus h(a_2) \oplus \ldots \oplus h(a_{2^y}) = 0 \]

where \( h \) is the Blake2b hash function.
Table 6.1
The performance of the OAuth2 with various data sizes (kb)

<table>
<thead>
<tr>
<th>File size (Kb)</th>
<th>Key Type</th>
<th>Encryption Time (MSec)</th>
<th>Decryption Time (MSec)</th>
</tr>
</thead>
<tbody>
<tr>
<td>10</td>
<td>64Bit</td>
<td>0.987</td>
<td>0.9878</td>
</tr>
<tr>
<td>20</td>
<td>64Bit</td>
<td>1.343</td>
<td>1.234</td>
</tr>
<tr>
<td>30</td>
<td>64Bit</td>
<td>3.432</td>
<td>2.542</td>
</tr>
<tr>
<td>40</td>
<td>64Bit</td>
<td>4.321</td>
<td>3.766</td>
</tr>
</tbody>
</table>

Table 6.2
The performance of the EISS with various data sizes (kb)

<table>
<thead>
<tr>
<th>File size (Kb)</th>
<th>Key Type</th>
<th>Encryption Time (MSec)</th>
<th>Decryption Time (MSec)</th>
</tr>
</thead>
<tbody>
<tr>
<td>10</td>
<td>256Bit</td>
<td>0.789</td>
<td>0.678</td>
</tr>
<tr>
<td>20</td>
<td>256Bit</td>
<td>0.897</td>
<td>0.789</td>
</tr>
<tr>
<td>30</td>
<td>256Bit</td>
<td>1.231</td>
<td>0.987</td>
</tr>
<tr>
<td>40</td>
<td>256Bit</td>
<td>1.341</td>
<td>1.987</td>
</tr>
</tbody>
</table>

Table 6.3
Overall performance in terms of security and accuracy

<table>
<thead>
<tr>
<th>Parameters</th>
<th>OAuth2</th>
<th>EISS</th>
</tr>
</thead>
<tbody>
<tr>
<td>Encryption Time (MS)</td>
<td>2.231</td>
<td>1.098</td>
</tr>
<tr>
<td>Decryption Time (MS)</td>
<td>2.432</td>
<td>1.567</td>
</tr>
<tr>
<td>Accuracy</td>
<td>78%</td>
<td>97%</td>
</tr>
</tbody>
</table>

Figure 6.1
Variations in the system

6. Evolution Results. The experiments are done in UBUNTU operating system, NS3 is used to develop the proposed EISS. To maintain the system speed and performance the compatibility if the implementation is needed. The Processor is I3 or I5. The simulation parameters are such as encryption time, decryption time and accuracy.

Table 6.1–6.3 show the overall performance in terms of security and accuracy. Based on the system performance the result may get variations (Figure 6.1).

7. Conclusion. This paper, mainly focus on providing the security for the routing protocol and data transfer nodes within the network with the integration of blockchain technology to the nodes present in the network and IoT is used to monitor the data transmission between the nodes. With the integration of blockchain
An Ensemble Integrated Security System with Cross Breed Algorithm

and IoT the security is provided very highly to transfer the data within the nodes. To access the data between the nodes the private and public keys are generated with RSA algorithm. According to the EISS the three parameters are calculated to improve the performance of the security and accuracy.

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BACKGROUND MODELLING USING A Q-TREE BASED FOREGROUND SEGMENTATION

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Abstract. Background modelling is an empirical part in the procedure of foreground mining of idle and moving objects. The foreground object detection has become a challenging phenomenon due to intermittent objects, intensity variation, image artefact and dynamic background in the video analysis and video surveillance applications. In the video surveillances application, a large amount of data is getting processed by everyday basis. Thus it needs an efficient background modelling technique which could process those larger sets of data which promotes effective foreground detection. In this paper, we presented a renewed background modelling method for foreground segmentation. The main objective of the work is to perform the foreground extraction only in the intended region of interest using proposed Q-Tree algorithm. At most all the present techniques consider their updates to the pixels of the entire frame which may result in inefficient foreground detection with a quick update to slow moving objects; The proposed method contract these defect by extracting the foreground object by controlling the region of interest (the region only where the background subtraction is to be performed) and thereby reducing the false positive and false negative. The extensive experimental results and the evaluation parameters of the proposed approach with the state of art method were compared against the most recent background subtraction approaches. Moreover, we use challenge change detection dataset and the efficiency of our method is analyzed in different environmental conditions (indoor, outdoor) from the CDnet2014 dataset and additional real time videos. The experimental results were satisfactorily verified the strengths and weakness of proposed method against the existing state-of-the-art background modelling methods.

Key words: Background modelling, dynamic background, foreground detection, background subtraction, Quad Tree segmentation

AMS subject classifications. 68U10, 94A08

1. Introduction. The noteworthy increase in the number of cameras being used for the purpose of video analysis and surveillance is due to low cost of higher resolution cameras. However, some issues always hold back users related to the surveillance of the captured videos, for example the cost will be more as we maintain the recorded data a long duration. Thus it is required for a motion detection algorithm which forms basic working of background subtraction, which can further build a background model to which the current frame is compared.

The background subtraction algorithm is therefore required to discriminate foreground from background. The static part of the scene here is background and the moving objects are the foreground. While a static background model is enough to analyses short videos in a controlled indoor environment, it is insufficient for most practical cases; Therefore, a more refined model is required to handle dynamic background, background with shadow and illumination change. Moreover, the motion detection is on the preliminary step in any of the application related to surveillance. For example, finding the region where moving object is detected might be made available for the further analysis or processing of unattended objects, ghost removal, human monitoring, traffic control, etc. Thus, background subtraction (BS) is the normal approach to detect motion in videos. The procedure is termed as foreground segmentation which involves frame comparison, update to background model and foreground segmentation. This procedure produces a binary mask.

Background modeling [18, 19], is a process of considering the background under different condition. A good algorithm should correctly detect the moving object, and at the same time remove the shadow. Besides, it is also important that it must exactly extract the foreground having same color as the background. A "universal" background subtraction algorithm has to handle the following critical situations:

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• Noise in the image due to sudden and gradual lightning variation
• Detecting slow moving object as well intermittent objects where the non-static object stays for a period of time as the static object.
• Movement of objects in the background as well multiple foreground objects.
• Handling shadow region.

It is complicated to achieve these goals with a simple background subtraction, by design, that promotes at the pixel level for improved productivity and it is not easy to analyze significant change patterns to produce better results.

The BS algorithm performs the task in two stages: background modelling and foreground extraction. The modeling is all about the measured design of the background demonstration and a technique engaged to accommodate the background changes at different timing instant. The object detection refers to the method engaged for pixel classification; generally, it is a threshold process where the present value x of the active pixel is compared to the same pixel in the B(x) background model. Conferring to the calculated difference in pixel value, x may be classified as foreground when the calculated difference is greater than threshold and when it is relatively smaller the pixel x is considered to be background. Furthermore, this procedure indirectly involves a mechanism using feedback among the modeling and the foreground detection modules. It is also a critical phase in reducing the misclassifications and increasing the accuracy in foreground segmentation. This problem invades an open choice between various methods for updating the background pixels.

In this paper, we present an adaptive region based background modeling method for the efficient foreground segmentation. The proposed method detects the primary region into which background subtraction to be performed. Region adjustments are regulated by monitoring: 1) moving objects in the background; 2) similarity between observed foreground and background models; 3) instable region based on regular fluctuations; and 4) the transmission of changing brightness or so called illumination changes. These four regulations are used as algorithm framing guide for an efficient foreground segmentation technique. The rest of the paper is organized as follows: In Section II, we comprehensively review the literature of background modeling techniques. This work gives an idea about the other frameworks developed for the purpose. Some of the methods are implemented and results are taken for comparison with our routine. Section III labels our method and specifies the routines used in our major research work: the background model initialization process, finding the region of interest and the update policy. Section IV deliberates the metric parameters used, and the experimental results along with comparisons and execution performance. We show that performing background subtraction in the specified region alone in its simplified form is equivalent the other techniques that are more sophisticated. The conclusion is given in Section V.

2. Literature Review. Ample count of background modeling methods have been offered over the past decades to identify or to fragment the foreground objects in a video stream. They commonly follow the general procedure of considering the initial frame or the mean of first few frames as background model, and then it finds the pixel based difference between the the current frame with the background model to sense the foreground objects. Various proposed modeling methods either parametric or Non-parametric were categorized into pixel-based, region or block based, and hybrid. One of the popular parametric methods which is also pixel-based is the Gaussian model. Wren et al. [1], which is considered to be first pixel based background modelling method.

However, a simple Gaussian function is not suitable to detect slow moving objects and frames with dynamic background due to lack of update mechanism to the background model [2]. Movements of tree leaves and water waves are considered to be dynamic background [3],and Stauffer and Grimson [4, 5] recommended a model that works on every pixel of the frame called the Gaussian mixture model (GMM). It is the procedure which works with mixture of K-mean Gaussians functions. As an improvement to above method, EM-based algorithm offered where it introduces a way to initialize the factor through online in the modelling of background, which is left to be time consuming. Adaptive Gaussian was proposed by Zivkovic,[30, 6] who anticipate an update to the parameters and D.S.Lee [7] has proposed a mechanism to improve the combining rate without disturbing the gaussian [5] stability as an adaptive learning rate. Chien et al. [49] throws a threshold based factor for the object detection. Here, they consider the camera noise and set to zero mean in gaussian distribution since it is the only factor that affects the threshold.
Shimada et al. [8] introduces a varying component to control the GMM in order to increase the accuracy and to reduce the computational time. Also, Oliver et al. [10] introduced a background modelling method based on Bayesian method which uses the previous information of data and keeps gathering proof from the data. Considering the Non-Parametric category in background modeling methods, Maddalena et al. [18] proposed a nonparametric algorithm based artificial neural networks with self-organization (SOBS). Kim et al. [19, 20] proposed a method of codebook which performs background modelling that initializes codeword and stores the statistics of codeword used in the codebook. Barnich et al. [11, 12] projected a pixel-based algorithm from the category nonparametric named ViBe. It detects the moving object as foreground using a the pattern of sampling intensity and random selection strategy. Wang et al. [42] suggested a technique (SACON) using manipulative sample consensus based on every pixel of the statistical model. The model detects the foreground object by exploiting both color and motion information. The superior performance to many other state-of-the-art methods is considered to be ViBe and is considered to be the exact method that detects the background changes in every frames [13]. The further study about the ViBe method was conceded by Droogenbroeck and Paquot [14] and they promote the performance enhancement by enhancing the ViBe with some considered additional constraints. Another such pixel based non parametric method proposed by Hofmann et al. [15] is pixel based adaptive segmenter (PBAS) method. The method keeps track of history of recent time observed pixel values and these are applied as background model to segment the foreground object. Although, the result of above discussed state-of-art methods projects well the outlines of foreground objects, i.e. the shape of the object and Also, they are affected by various factors like noise, dynamic background and lighting changes, etc without any difficulty.

As an another approach to Pixel-based background modelling there exist the Region-based methods. It uses the inter-pixel relations to enhance its efficiency and segments the images into regions and performs foreground objects identification from the segmented image regions. Kernel Estimation Density (KDE) projected by El-gammal et al. [9, 28] offered a novel method by building a nonparametric background model. Seki et al. [50] applied a different technique of background modelling by applying image variation co-occurrence in the defined image regions. A heuristic algorithm distinguishes the foreground object from the background based on block matching and similar method was proposed by Russell et al. [51] to . In which they compared fixed size of background regions with the input frames. In order to provide solution to the dynamic background modeling challenges in outdoor settings, Eng et al. [52] offered a random regular arrangements and image regions with pre filters, by using color space to detect foregrounds in the CIELab. Furthermore the method is featured by color and texture based analysis which receives a attractive attention among region based methods. An another background modelling method proposed by Heikkila et al. [53] engaged a technique called local binary pattern (LBP) , which involves the discriminative texture feature of the image for modeling the background. In the above model, LBP histogram is built for partially overlapping regions and the same will be compared against the LBP histogram of every incoming frames regions. Liu et al. [54] proposed a method to resolve the challenge of illumination change. It is also an efficient background modeling method for foreground segmentation based on binary descriptor and proven to adapt under continuously varying illumination condition. In addition, another method based on binary descriptors was proposed by Huang et al. [55] as a sample based background modelling a replacement to parametric methods. In divergence to pixel-based foreground segmentation methods, these region-based segmentation methods can obtain only coarse shapes of foreground objects but can reduce the misclassification to a greater extent, thus reducing the false correlations.

3. Proposed Architecture for Q-tree based Intrusion Detection System. Here we have proposed an architecture for an Intrusion detection and prevention system. The Fig. 4.1 portrays the flow in the architecture. The flow includes the data acquisition from the camera attached in the surveillance area is then forwarded to the intrusion detection module which incorporate the proposed Q-Tree based foreground segmentation. Once the intrusion is detected in the surveillance area the intimation will be given to the provided webserver which further gets forwarded to the user by the webserver.

The data acquisition model collects the video data from the camera in the area under surveillance region and is given to the IDM(Intrusion detection model) which in turn incorporates our proposed approach using the concept of “Q-tree Image segmentation(QIS)”. In first subsection, we presented preprocessing technique, second subsection explains the Q-Tree Decomposition, third subsection focuses on the routine to find Region
of Interest based on the quadruple image segmentation for the input frame sequence (QTIS) and its matrix representation, fourth subsection depicts the classifier and background update policy.

The background modeling methods adopted in the literature [1, 2, 9, 33] etc. uses various parameter based as well non-parameter based background modeling which does its operation to every pixel in the frame. In this paper, we present a different technique through quadruple based image segmentation from where the foreground segmentation is done only to the specific Q-Tree node in the background frame.

The given video stream is considered as the input and it is divided into frames and the defined function Init_BGM() initializes the background model based on the average of first $n$ input frames. It is followed by Q-Tree representation of the defined background model. Then for the input frames the Q-Tree representation is done as well as the matrix child and threshold of size $2n \times 2n$ will also get updated. Fig. 3.2 Depicts the flow involved in Adaptive intrusion detection system.

3.1. Preprocessing. The first step in the Adaptive intrusion detection system is the preprocessing which includes framing, background initialization, and Q-tree representation of the background model.
1. Framing
The video stream is considered as the input and is divided into the frames suitable for further processing and is stored in the local database for the foreground identification and Intrusion detection.

2. Background Initialization
It is the first step but considered to be an important step in the process of background subtraction. There are various well know methods available to model the initial background. Few such is assigning the average illumination of first \( n \) frames, or assigning the very first frame directly as BGM, or using the image order to approximate the background model, and the frame average over the time. Although the average of background frame that updates at an instance of time \( t_n \) is familiar with many systems, it is not good enough to handle shadow and ghost problems. Here, we adopt a modest and proficient way of initializing the background where the median of first \( n \) frames are considered. When considering the median of first \( n \) frames, it also reflects the illumination change. At the initial time \( t_0 \), we set the initial background as the median of first \( n \) frames.

\[
\text{Init}_{-}\text{BGM} = \text{Median}(f_k)
\]

where \( k \) varies from 1, 2, \ldots, \( n \) at the timing instance \( t_0 \). The initial background plays a vital role in the process since the foreground detection that happens at timing instance \( t_j \) depends on the background frame detected at \( t_0 \), where \( j \) is closer to \( n \).

3. \textit{Q-Tree representation of the Background model}

The initial frame which is set as the background model is taken and is given with the \textit{Q-Tree} representation. The background image is divided into regions with varying information. This is accomplished by the \textit{Q-Tree} decomposition technique. Such decomposition is powerful as it gives homogeneous regions which in further gets constructed as a tree with each parent having maximum of four children. Each child corresponds to a quadrant of the segmented image frame as shown in Fig. 3.3.

The tree representation is also given with a coding procedure where 1 is given for the node with a child i.e. if node is a parent and 0 for child node which do not further have a child. So for the above tree in Fig. 3.3(c), the \textit{Q-Tree} code (\textit{QTC}) is given as 1-0000.

3.2. \textit{Q-Tree Decomposition and its Representation.}
Assuming the size of image array is \( m \times n \), the entire image array is represented by quadruple tree. The \textit{Q-tree} will have four child representing the four quadrants of the array and each of the child will be subdivided in a recursive manner. The procedure terminates the \textit{Q-Tree} with leaves with homogeneous regions or until the specified level. These leaves correspond to the unique regions of the image segments.

Here the root of the tree corresponds to the entire image. When \textit{Q-tree} iteration takes place at the first iteration, the image is segmented into four quadrants. Thus, resulting the tree with four homogeneous leaves. The iterations can be further continued until desired level or desired pixels are segmented. A sustainable condition is set by finding the pixel intensity difference and must be lesser than the threshold. The difference between high intensity pixel and low intensity pixel is calculated within the node of the decomposed quad tree and is ensured against the sustainability condition. Each level in the decomposed tree is given with \( 2^n \times 2^n \) nodes for the given integer \( n \), where \( n \) corresponds to the number of iterations. For example, at 0 split the whole image is represented as the singe node. At 1st split the image is segmented into \( 21 \times 21 \) nodes which represents the four quadrants of the parent image frame.

Each node has left and right segments denoted by \( l, r \), and top, bottom segments denoted by \( t, b \), respectively. So each image is denoted by four disjoint set as nodes and are represented by \( tl, tr, bl, br \) in the clockwise order.
Also for \( n \) splits, the total number of nodes in the tree is given by formula,

\[
\text{No. of nodes in the tree} = 4^{n-1} - 1.
\]

Since the scope of the research work is restricted to intrusion detection, we limit the value of \( n \) to 2. So that maximum of two levels are possible and the level 0,1,2 corresponds to whole image, image with 4 quadrants and image with 16 regions respectively.

Q-Tree decomposition is performed by considering the input image frame of size \( m \times n \). At each iteration, it divides the image frame into four quadrants and assign it as four children to the parent node. The same gets iterated for every child node until desired pixels are selected. At each iteration it updates the array \( QTC \) and \( QTT \) which corresponds to Q-Tree code and Q-Tree node threshold, respectively.

\[
QTC_{i}[2^m] [2^n] = \begin{cases} 
1 & \text{for parent node} \\
0 & \text{for child node} 
\end{cases} \quad \text{for all } 0 \leq m \leq 2, 0 \leq n \leq 2. \tag{3.1}
\]

\[
QTT_{i}[2^m] [2^n] = \{ \text{adaptive threshold of the specific node} \}, \quad \text{for } 1 \leq m \leq 2m, 1 \leq n \leq 2n. \tag{3.2}
\]

In the above equation \( QTC_i \) is the array which holds vale 1 for parent node and 0 for child node. Thus \( QTC_0 \) holds 0 since there is no splitting, \( QTC_1 \) holds 4 values in a two dimensional array \([21]\), \( QTC_2 \) holds 16 values as it corresponds to \([22]\), \( 4 \times 4 \) matrix array. Similarly \( QTT_i \) corresponds to the respective adaptive threshold of the node. \( QTC_i \) has been given with a third value 2 when every pixel in the region is a foreground pixel. The algorithm for Q-Tree decomposition is as follows.

**Algorithm: Q-Tree Decomposition**

**Input:** Input Frames

**Output:** Q-Tree segmented image frame with \( QTC \) and \( QTT \)

**Step 1:** Ensure \( I \) is the Image frame

**Step 2:** Check whether the dimension of \( I \) is \( m \times n \)

**Step 3:** Perform the Quad tree decomposition of \( I \)

**Step 4:** Update the array \( QTC \) and \( QTT \), for every iteration \( i \) according eqs (3.1) and (3.2)

**Step 5:** Calculate the pixel intensity \( P_i \) as follows

\[
P_i(x, y) = \frac{1}{4} \sum_{m=0}^{1} \sum_{n=0}^{1} x_i - 1(2x + m, 2y + n)
\]

**Step 6:** for \( m, n = 0, 1, 2 \) if \( P_{i-1}(2x + m, 2y + n) \) are leaves go to step 5

**Step 7:** Check equation in 5-6 for sustainability condition

\[
|P_i(x, y) - P_{i-1}(2x + m, 2y + n)| < T
\]

If true, set as leaf

Otherwise set as node

**Step 8:** continue step 5 for next \( x, y \)

**End procedure**

3.3. Finding ROI In Background Model. Most of the state of art background modeling methods keep only single background frame for foreground segmentation, in which the result of foreground detection consumes the whole frame. This procedure is termed as FBM (frame-based background modeling). The inadequacy of the background modeling process opens a large room for the scholars to post their ideas. Many times the background modelling faces a problem where the part of the generated background may be inaccurate because of the problems like slow moving objects, image ghost and the shadows. The next problem is that the accuracy of background update may get worse as the frame distance increases, most cases this varies with frame compression rate. To solve these complications, we come out with a region-based background model (RBM).

The enhancements that are noticed with the proposed method are at spatial level where foreground generation and segmentation mechanism performed on the basis of ROI but not entire frame. The other improvement is, in temporal level, where background generation and modeling are performed for every frame but not a whole video sequence.

Some of the literature discussed are the block based methods like Heikkila et.al.[71] proposed a method that models the scattering of block texture in a frame using the local binary pixel outline method. Moreover, in
these block-based methods [71, 72, 73] there is a recurring problem of dealing with the entire frame. Although they consider the frame as a gathering of blocks, they perform their operation to entire block segments and reassemble those blocks back to frames.

To avoid these problems, we offered an algorithm to find the region of interest in the background to which the foreground segmentation is constrained. This approach presents two main ideas where it works on a Q-Tree of the background model. QTC is the code given for the regions in the image segments. For the point \( P \) on the frame \( f \), the coordinates are given as \( P_{i,j} \) and the point \( B \) on the background is given as \( B_{i,j} \) and the pixel intensity difference is given as \( D_{i,j} \). When there is the change in intensity i.e. \( |D_{i,j}| \neq 0 \), then it is classified as foreground pixel. On the other hand, if there is no change in the intensity where \( |D_{i,j}| = 0 \), then QTC code is set as follows.

Procedure: Find_ROI()

Input : Q-tree of frame \( f \) in the video stream

Output : Segmented foreground.

Algorithm:

Step 1: Read \( QTC_i[2^m][2^n], QTT_i[2^m][2^n] \);

Step 2: check for \( QTC_i[2^m][2^n] \); for \( i = 0, 1, 2; \)

Step 3: If code=0, the region has no intrusion;

Not a ROI

Classify every pixel in the region as background pixel;

Step 4: If code=1, the region has some intrusion;

Is a ROI

Collect coordinates within the region and calculate change in Intensity

\[ D_{i,j} = |P_{i,j} - B_{i,j}| \]

if \( |D_{i,j}| \neq 0 \) then

classify as foreground,

else

classify as background;

end if

Step 5: Goto step 2 for the next \( i; \)

End procedure

The above procedure is used to check for the intensity change which corresponds to the foreground appearance, illumination change and many other issues.

3.4. Foreground Segmentation And Background Update. Once the ROI is determined, the next step is to perform the foreground segmentation. The colossal of the existing method includes ViBe, PBAS, Word consensus model. The ViBe set the first frame as background and foreground segmentation is done using the fixed threshold. This restriction holds back the adaptability of the method. PBAS, the non-parametric background model integrates the ideas of state-of-art foreground detection methods and works in a controlled environment. Here we follow the basic idea of ViBe, PBAS with some variations to provide an efficient segmentation. Fixed threshold of ViBe is replaced by an adaptive threshold, and the first frame as background model is over laced by the median of first \( n \) frames. To update the background model, we use the adjustable learning rate concept from the PBAS with a random scheme and dynamic threshold is used to identify the foreground. The algorithm has two important factors: \( T(x), L(x) \) which corresponds to the dynamic threshold and learning rate for background update The background model \( B_i(x) \) for the pixel \( x \) is defined as \( B(x) \).

\[ B(x) = \{ B_1(x), \cdots, B_k(x), \cdots B_n(x) \}, \text{ where } 1 < k < n \]

Pixel \( x \) is considered to be the foreground pixel when the given condition holds true

\[ f(x) = \begin{cases} 1 & \text{if dist}((Ix), (Bx)) < T(x) \\ 0 & \text{otherwise} \end{cases} \] (3.3)
The above formula illustrates that \( f(x) \) is set to 1, i.e. the pixel is classified as foreground, when the distance between the current frame and the background frame is lesser than the dynamic threshold. Otherwise, set with 0 means the pixel \( x \) is the background pixel.

The dynamic threshold changes its value at each frame over the time instance \( t_n \), for \( n \)th frame. It is calculated by the formula as suggested in the literature [74].

\[
\Delta T = \alpha \frac{1}{w \cdot h} \sum_{i=0}^{n-1} |I(x) - B(x)|
\]

Since it changes with time, it is considered to be dynamic and the pixel we use the notation \( \Delta T, \alpha \) is the coefficient called inhibitory coefficient, whose reference value is 2. \( w \cdot h \) represents the width and height of the frame which in turn represents the size of the frame. The algorithmic representation is given below.

**Algorithm:** Pixel classification

**Input:** Input frame, ROIB

**Output:** Classified Foreground object

**Procedure:**

for \( zi \in ROIB \) for every row in the ROI

for \( zi \in 1, 2, \cdots , w \) for every pixel in the row do

\begin{align*}
&\text{calculate difference between the pixel at background frame (at time } t-1 \text{) and current frame (at time } t) . \\
&\text{calculate } Dist(B_{t-1}(xy), I_t(xy)); \\
&\text{end for}
\end{align*}

\begin{align*}
&Dt(x) = \min_{y \in 0, 1, 2, \cdots , w-1} \text{ Dist}(B_{t-1}(xy), I_t(xy)) \\
&\text{Distance at time instance } t \\
&\text{Update } T(x) \text{ using the equation (3.4)} \\
&\text{Classify the pixel} \\
&\text{If } D_t(xy) < T(xy) \text{ then} \\
&\quad F(xy) = 1; \\
&\text{Else} \\
&\quad F(xy) = 0; \\
&\text{End if}
\end{align*}

End for

End procedure.

The above procedure is used to perform the background subtraction only to the region of interest predicted in the previous step. The procedure is repeated for every line in the ROI and for every pixel \( x \) of the line the distance is calculated, then compared with dynamic threshold, \( QT_T, [2^m][2^n] \). If the intensity difference is lesser than the threshold then it is classified as foreground otherwise classified as background.

As proposed in ViBe, the update mechanism used is the conservative type where it never includes the pixel to the background which belongs to the foreground. Thereby, avoids the misclassification and keeps the false positive minimum.

4. Experimental Analysis. Hereby, we discuss the quantitative and qualitative performance evaluation of the proposed technique. For which we presented the foreground segmented image as qualitative evaluation. Then, parameter based metrics evaluation compared with the results of five other recent state-of-art techniques.

4.1. Evaluation metrics and the Test Dataset. The video sequences of the Change Detection Challenge 2014 dataset [75] has been used to assess the execution performance of the proposed approach both as a qualitative and quantitative evaluation. Since it is confirmed in many of the works, it is clear that traditional datasets are cramped and lacks challenging video streams. Moreover, standard methods have outperformed and no longer face the challenges of current video surveillances application.

The Change Detection Challenge 2014 dataset has been promised with 53 videos of 11 categories including the states of both outdoor as well indoor videos. These videos have been prearranged under categories Base-
line, dynamic background, camera jitter, Intermittent object motion, shadows, thermal signatures, challenging weather, low frame-rate, night videos, PTZ capture and air turbulence. The baseline is considered to be an important category that includes the following challenges: background motion, isolated shadows, having uncontrolled objects and ghost. The intermittent object motion is the category that has the video with moving object stays static for some time and again started to move. To measure the accuracy, we calculated few metrics namely F-measure, percentage of correctness.

Classification (PCC) and Matthew’s correlation coefficient (MCC) metrics are referred in the literatures F-Measure[SOIR], PCC[word consensus model], MCC and are calculated using terms true positive (TP), true negative (TN), false positive (FP), false negative (FN). To calculate the metrics it is assumed like foreground as positive and background as negative. When it is correctly detected then true, otherwise it is false. Thus, True positive is the number of pixels where foreground pixels are correctly detected as the foreground, false positive depicts the number of background pixels wrongly detected as the foreground, True negative represents background pixels correctly detected as the background and False negative represents the number of foreground pixels wrongly detected as background. These metrics in turn are used to calculate precision and recall.

Precision is the foreground prediction which is measured as the ratio of correctly classified foreground pixels to the total number of pixels classified as foreground pixels which include the misclassification.

\[ \text{Precision} = \frac{TP}{TP + FP} \] (4.1)

The precision is calculated since it gives the ability of the method to identify foreground pixels (considering misclassified positive class). The Recall measures the detection rate and is calculated as the ratio of correctly classified foreground pixels to the total number of foreground pixels in the ground truth (i.e. ratio of true positives to actual foreground pixels in the test image).

\[ \text{Recall} = \frac{TP}{TP + FN} \] (4.2)

The above two metrics namely precision and Recall are used in calculating the evaluation metrics like F-measure, PCC, MCC which are used to assess the quality of a method; As just, the method is considered if it has high Recall values, without losing Precision. F-Measure is the metric obtained as the weighted harmonic average of the values recall and precision. It is the metric used in most of the datasets to rank the methods. It is defined as follows:

\[ F_{\text{Measure}} = 2 \cdot \frac{\text{Recall} \cdot \text{Precision}}{\text{Recall} + \text{Precision}} \] (4.3)

The metric percentage of correct classification (PCC) is also used to determine whether both the background and foreground is classified correctly. For an effective foreground segmentation, PCC value must hold high. The PCC value is calculated using the below given equation:

\[ \text{PCC} = \frac{TP + TN}{TP + TN + FP + FN} \] (4.4)

The last metric used for evaluating the performance is Matthew’s Correlation Coefficient(MCC) scores. As stated in the literature [2], it is found to assess the overall performance of unbalanced binary classification, into which the background subtraction comes in. it is calculated using the formula

\[ MCC = \frac{(TP \times TN) - (FP \times FN)}{\sqrt{(TP + FN)(TP + FP)(TN + FN)(TN + FP)}} \] (4.5)

4.2. Results of challenge change detection 2014 dataset.. We compared our method to six other recent foreground segmentation methods, the adaptive background modeling(ABMFS)[78], the word consensus model(PAWCS) [77], the Gaussian mixture model (GMM) [6], the sample consensus background model (SACON) [17], ViBe [76], the pixel-based adaptive segmentation (PBAS) [79], the boosted Gaussian model
General Architecture of the intrusion detection system.

(BMOG), and multimode background subtraction method (UMBS). Gaussian Mixture Model is a pixel-based parametric method and BMOG are pixel and blob-based parametric method. ABMFS and PAWCS are pixel-based nonparametric techniques, and are the two top performing foreground detection methods reported [20]. We integrate the region based background subtraction to the basics of pixel based adaptive segmentation method (PBAS).

For the execution result, the code available in OpenCV Library for GMM is used [6]. The codes for PBAS, ABMFS, PAWCS were provided by the authors and the parameters are also set as suggested by the authors respectively. The output of BMOG, GMM, UMBS and SubSENSE are presented as posted in the CDnet2014 dataset by the authors. Some of the reference values are taken directly from the papers.

Here we have taken only the video categories of baseline, Intermittent object and shadow. Since our work is restricted to the Closed room Intrusion Detection Technique we consider only indoor videos from the dataset and qualitative cum quantitative comparisons are done, proposed method obtained agreeable results for the different videos including shadow video. The other foreground detection methods used for comparison are all executed for the frame level operations Wherein the proposed method is operated only in the region of interest focused by Q-Tree code and seems to have acceptable foreground segmentation neared to other state of art methods. However, most segmentation methods classify some of the background pixels as foreground. The proposed method possibly obtains nearly correct foreground objects.
Table 4.1

Average quantitative values of evaluation metrics from change detection dataset

<table>
<thead>
<tr>
<th>Scenarios</th>
<th>Percentage of correctness classification</th>
<th>Recall</th>
<th>Precision</th>
<th>F-measure</th>
</tr>
</thead>
<tbody>
<tr>
<td>Bad Weather</td>
<td>98.5</td>
<td>0.604</td>
<td>0.315</td>
<td>0.411</td>
</tr>
<tr>
<td>Baseline</td>
<td>96.0</td>
<td>0.901</td>
<td>0.849</td>
<td>0.874</td>
</tr>
<tr>
<td>Camera Jitter</td>
<td>93.8</td>
<td>0.791</td>
<td>0.359</td>
<td>0.495</td>
</tr>
<tr>
<td>Dynamic Background</td>
<td>99.1</td>
<td>0.655</td>
<td>0.200</td>
<td>0.317</td>
</tr>
<tr>
<td>Intermittent motion object</td>
<td>97.5</td>
<td>0.815</td>
<td>0.835</td>
<td>0.824</td>
</tr>
<tr>
<td>Low frame</td>
<td>89.7</td>
<td>0.640</td>
<td>0.645</td>
<td>0.642</td>
</tr>
<tr>
<td>Night Video</td>
<td>92.0</td>
<td>0.802</td>
<td>0.502</td>
<td>0.616</td>
</tr>
<tr>
<td>PTZ</td>
<td>84.5</td>
<td>0.725</td>
<td>0.287</td>
<td>0.410</td>
</tr>
<tr>
<td>Shadow</td>
<td>96.8</td>
<td>0.914</td>
<td>0.563</td>
<td>0.649</td>
</tr>
<tr>
<td>Thermal</td>
<td>98.5</td>
<td>0.865</td>
<td>0.623</td>
<td>0.727</td>
</tr>
<tr>
<td>Turbulence</td>
<td>97.4</td>
<td>0.710</td>
<td>0.559</td>
<td>0.625</td>
</tr>
<tr>
<td>Overall</td>
<td>94.9</td>
<td>0.763</td>
<td>0.516</td>
<td>0.615</td>
</tr>
</tbody>
</table>

Another quantitative comparative approach on experimental result on the Change Detection Challenge dataset is presented were we broadly test the proposed method. Here, the scenario considered includes Baseline, Intermittent motion objects, and Shadows. The other category usage is suppressed due to the scope of the research.

The qualitative analysis is given above with representation of actual frame, its ground truth and the executed output. Table 4.1 depicts the average values of the metrics Recall, Precision, F-measure and percentage of correctness classification. It projects the average value from the CDnet14 dataset for all the categories of videos in the open dataset. These average metric values are used for comparison of the proposed algorithm. Though at some places the proposed algorithm can not achieve remarkable enhancement, it goes above the average from the dataset.

Table 4.2 presents evaluation metrics F-Measure score of different video criteria considered for our method in comparison with six other high performing methods on the CDnet 2014 dataset. The values of top three performing methods are given in red, blue and yellow color shade. Though our method performed best only for the videos pedestrian and copy machine, it is clearly visible that the performance of other criteria office, parking, bus station is also evaluated to very near of well performed methods The proposed modeling method could acclimate to speedy changes caused by slow background changes, as well to illumination variation and static objects since it can execute more number of frames per second. Though, the performance is not good for the category sofa, the overall scores proven to be the best performance. For the same the graphical representation is given in Fig.4.1.

Table 4.3 also shows in detail the PCC metric values in the compared state of specific criteria Baseline, Intermittent motion object and Shadows respectively. It could be proven that our method obtained higher scores in every criteria at least for one video frame sequences. The top three values are represented with bold red, blue and yellow respectively. Table 4.4. Also represent the MCC( Mathew’s Correlation Coefficient) with its graphical representation in Fig.4.3

5. Conclusion and Future Work. In this paper, we proposed an adaptive and efficient background modeling technique that exploits region based background subtraction. It also provides quick response to the real time scenario with data inconsistency. Furthermore, we have introduced “Q-Tree based foreground segmentation” algorithm to model the background that selects the specific region in a faster manner and promotes efficient foreground segmentation. This selection mechanism works well in most situations with varying conditions. The experimental result shows that the overall performance of proposed method is better than the entire benchmark systems considered. The method was proposed to identify the moving object in the region of interest, and to perform pixel based background subtraction for foreground detection. Satisfactorily,
Table 4.2
Comparitive F-Measure values of the selected methods from CDnet14

<table>
<thead>
<tr>
<th>Category</th>
<th>Video</th>
<th>Baseline</th>
<th>Intermittent object</th>
<th>Shadow</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>proposed</td>
<td></td>
<td>0.981</td>
<td>0.959</td>
<td>0.768</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.819</td>
<td>0.725</td>
<td>0.914</td>
</tr>
<tr>
<td>ABMFS</td>
<td></td>
<td>0.960</td>
<td>0.656</td>
<td>0.482</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.698</td>
<td>0.632</td>
<td>0.640</td>
</tr>
<tr>
<td>PAWCS</td>
<td></td>
<td>0.954</td>
<td>0.963</td>
<td>0.4447</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.672</td>
<td>0.769</td>
<td>0.807</td>
</tr>
<tr>
<td>GMM</td>
<td></td>
<td>0.966</td>
<td>0.972</td>
<td>0.618</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.819</td>
<td>0.725</td>
<td>0.914</td>
</tr>
<tr>
<td>BMOG</td>
<td></td>
<td>0.960</td>
<td>0.656</td>
<td>0.482</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.698</td>
<td>0.632</td>
<td>0.640</td>
</tr>
<tr>
<td>SUBSENSE</td>
<td></td>
<td>0.938</td>
<td>0.965</td>
<td>0.4447</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.672</td>
<td>0.769</td>
<td>0.807</td>
</tr>
<tr>
<td>UMMBS</td>
<td></td>
<td>0.946</td>
<td>0.937</td>
<td>0.819</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.656</td>
<td>0.482</td>
<td>0.652</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.698</td>
<td>0.632</td>
<td>0.640</td>
</tr>
<tr>
<td></td>
<td></td>
<td>0.672</td>
<td>0.769</td>
<td>0.807</td>
</tr>
</tbody>
</table>

Table 4.3
Comparitive PCC values of the selected methods from CDnet14

<table>
<thead>
<tr>
<th>Category</th>
<th>Video</th>
<th>Baseline</th>
<th>Intermittent object</th>
<th>Shadow</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>proposed</td>
<td></td>
<td>99.962</td>
<td>99.433</td>
<td>97.932</td>
</tr>
<tr>
<td></td>
<td></td>
<td>94.507</td>
<td>98.747</td>
<td>98.339</td>
</tr>
<tr>
<td>ABMFS</td>
<td></td>
<td>99.916</td>
<td>99.616</td>
<td>97.323</td>
</tr>
<tr>
<td></td>
<td></td>
<td>95.605</td>
<td>97.068</td>
<td>96.113</td>
</tr>
<tr>
<td>PAWCS</td>
<td></td>
<td>99.892</td>
<td>99.166</td>
<td>94.507</td>
</tr>
<tr>
<td></td>
<td></td>
<td>96.332</td>
<td>95.605</td>
<td>95.729</td>
</tr>
<tr>
<td>GMM</td>
<td></td>
<td>99.919</td>
<td>95.510</td>
<td>96.168</td>
</tr>
<tr>
<td></td>
<td></td>
<td>95.603</td>
<td>97.068</td>
<td>96.113</td>
</tr>
<tr>
<td>BMOG</td>
<td></td>
<td>99.833</td>
<td>95.510</td>
<td>97.222</td>
</tr>
<tr>
<td></td>
<td></td>
<td>96.168</td>
<td>97.222</td>
<td>95.729</td>
</tr>
<tr>
<td>SUBSENSE</td>
<td></td>
<td>99.912</td>
<td>99.491</td>
<td>95.335</td>
</tr>
<tr>
<td></td>
<td></td>
<td>94.456</td>
<td>98.115</td>
<td>99.047</td>
</tr>
<tr>
<td>UMMBS</td>
<td></td>
<td>99.871</td>
<td>99.517</td>
<td>95.335</td>
</tr>
<tr>
<td></td>
<td></td>
<td>95.216</td>
<td>98.115</td>
<td>97.675</td>
</tr>
</tbody>
</table>

The proposed method performed the task of foreground segmentation in the identified region of interest with a reduced mis-classification which results in better scores for the metrics considered. Thus, the quantitative performance is achieved. However, method shows lack in performance for the criteria shadow elimination, Baseline of CDnet2014. We will attempt to solve these issues using adaptive threshold for specific region based on the learning rate to the specific region.

6. Acknowledgement. We thank the team who made the website (www.changedetection.net) available and for providing the resource to test and compare our method.

REFERENCES

Table 4.4
Comparative MCC values of the selected methods from CDnet14

<table>
<thead>
<tr>
<th>Method / Video</th>
<th>Baseline</th>
<th>Intermittent object</th>
<th>Shadow</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Pedestrian</td>
<td>Office</td>
<td>Parking</td>
</tr>
<tr>
<td>proposed</td>
<td>0.981</td>
<td>0.956</td>
<td>0.857</td>
</tr>
<tr>
<td>ABMFS</td>
<td>0.956</td>
<td>0.970</td>
<td>0.591</td>
</tr>
<tr>
<td>PAWCS</td>
<td>0.946</td>
<td>0.934</td>
<td>0.806</td>
</tr>
<tr>
<td>GMM</td>
<td>0.960</td>
<td>0.672</td>
<td>0.459</td>
</tr>
<tr>
<td>BMOG</td>
<td>0.921</td>
<td>0.611</td>
<td>0.698</td>
</tr>
<tr>
<td>SUBSENSE</td>
<td>0.954</td>
<td>0.960</td>
<td>0.516</td>
</tr>
<tr>
<td>UMMBS</td>
<td>0.938</td>
<td>0.962</td>
<td>0.650</td>
</tr>
</tbody>
</table>

Fig. 4.2. Graphical representation of PCC

Fig. 4.3. Graphical representation of MCC.


Background Modelling Using a Q-Tree Based Foreground Segmentation


TOPOLOGICAL ORDERING SIGNAL SELECTION TECHNIQUE FOR INTERNET OF THINGS BASED DEVICES USING COMBINATIONAL GATE FOR VISIBILITY ENHANCEMENT

AGALYA RAJENDRAN∗AND MUTHAIAH RAJAPPA†

Abstract. In modern System on Chip (SoC) design consist of intelligence of products, Internet of Things (IoT) based devices, mobile phones, laptops, servers etc. This shrinking market reduces the design automation validation process. Signal selection is the most effective and challenging technique in post-silicon validation and debug. The vital problem prevailing in this method is that it has limited observability and controllability due to the minimum number of storage space in the trace buffer. This tends to select the signals prudently in order to maximize the state reconstruction. To identify the trace signals, signal restoration is the extensive metric that has been used so far. Topology-based restoration method is proposed here to minimize the error detection latency which helps to select the trace signals with minimum error or even errorless. This method aid to detect more number of errors within limited number of clock cycles than the restoration only selection techniques.

Key words: Design bugs, Error detection, Internet of Things (IoT), Restorability, Topology, Trace signal, validation

AMS subject classifications. 94A12

1. Introduction. Recent System-on-Chip (SoC) design studies specifies the extensive usage of SoCs in the electronic industry with increasing complexity in design circuits and its influence to meet the demands in ever growing technologies from various applications containing Internet of Things (IoT). Since there is a growing in the technology it is difficult to meet the functional correctness of the circuits during pre-silicon validation. The escaped errors (both functional and electrical errors) must be captured during posts-silicon validation. Fig. 1.1 represents the key components in the design industry.

The main challenge in post-silicon validation (psv) is that it has limited observability and controllability of internal signals after the first silicon is available [1]. Because of poor observability of internal signals, it is very difficult to debug the design and electrical errors in this atmosphere is laborious [2, 3]. Electronic design automation (EDA) tools are used to automate the product development cycle and validation to achieve the anticipated results with minimum cost and time that ensures the efficiency. Controllability defines the ability of moving an internal signals from input to output. On-chip buffers are used to control these signals. Compare to several methods, scan chain is the most effective technique to enhance both the observability and controllability of internal signal states for the determination of manufacturing. However, it is very less effective during chip execution of debugging and validation [4]. At that time of verification, it need to halt the process to readout the each state of flip-flops. Few signal states are stored in the trace buffer within a limited no. of clock cycles by incorporating it to the on-chip to attain the non-destructive state of internal signals. Only small set of signals can be traced due to the constraint of area overhead. There is a difficulty in the selection of correct signals since they are selected at the design stage. For smaller design circuit, it is easy to recommend few signals based on the designers own knowledge but for larger design circuit like automated signal selection, it is difficult. Hence, restoration of signals helps to increase the visibility of untraced signals from the value of traced flip-flops [5]. This is the most common method for finding the signals automatically. From [6, 7, 8, 9], none of the papers has discussed about the effectiveness of detecting or localizing the errors.

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This paper is focusing on the efficacy of bugs and latency that present in the traced signals that may affect observability and controllability enhancements. These effects can be observed at the gate-level selection. An alternative way is proposed at this level of selection that considers a path along the traced gates and their topological sort rather than considering flip-flops only. By this way, it reduces the error detection latency and increases the erroneous responses from the traced gates.

The remaining paper is organized by research background and how this work motivates us are described in Section II. In Section III and IV, the authors proposed the signal section techniques and their findings based on the topological sorting respectively. In section V they have conclude their research work.

2. Research Background and Motivation.

2.1. Background. The state restoration (SRR) can be done either by forward restoration or backward restoration of unknown signal states that are traced during silicon debug [1]. This helps to decide which signal are to be traced in an automatic manner to increase the rate of post-silicon validation part. It can be defined as

\[
SRR = \frac{\text{Number of states restored} + \text{traced signals}}{\text{Number of states traced}}
\]  

(2.1)

The visibility of internal signals increases if the state restoration value increases. It may be helpful to find the root cause of internal design bugs when debugging. There is a wide variety of selection techniques were existed and it may cause a serious issue while debugging. One such factor is that assuming all the signals are having equal importance in the selection process and it is really unrealistic from the point of debugging. But in reality, which internal signals are ought to be observed that varies between design structure and design feature. Existing methods like [10], attempted to link the trace signal selection method with debugging, error detection and localization. But their methods were not appropriate to find the location of random errors. If few flip-flops are traced, it can affect the other flip-flop when the bit flip is detected if they consider different restoration scenarios [12]. However in [11], they critically analyzed about the state restoration but they strongly suggest the usage of assertion coverage to sort out the trace signal problems. This method is restricted to one type of error model and their signal selection methodology is commendable for missing error models. So there is lack in standardizing the error models in psv is the biggest weakness for the evaluation of various signal selection algorithms and design bugs are categorized from few of reported works [13]. Some bugs were missed out at the gate level abstraction like exchange of wires, inverters, etc. This is the starting point to propose this paper by taking the above said bugs as our motivating example. The bugs in the RTL, translated into gate level manifestations. In our experiment we consider different instances to represent a single design bug.
2.2. Motivation. To elucidate the irrelevances between the actual error localization and the signal selection algorithms, we simulated the gate-level description of Fig. 2.1 [6]. For example, here we are simulating 2 different types of errors. Primarily, haphazard gate replacement is consider as type 1 bugs and exchange of wires between the circuit is consider as type 2 bugs. As we know that some bugs are not identifiable at the pre-silicon stage under some circumstances. For example, if the design has two input Ex-OR gate that can be replaced by two input OR gate but the only identification of replacement of the gate is ’11’. This is the place where we can fail to identify the state at pre-silicon stage when the corresponding state is activated rarely in the state machine. Similarly, type 2 bugs are also justified in this section.

A quick identification of a bug location is observed and its latency report is shown in table 2.1 and 2.2 based on gate replacement before Flip-flop ‘C’ and exchange of wires between flip-flops ‘C’ and ‘F’ respectively. Here, we experiment 4 number of iterations by injecting an error. ’NO’ represents either the error doesn’t propagate through the flip-flop or cannot be observed (Not observed). In paper [6], they consider AC and AB and in [5, 9] they consider C&F are the combination to achieve high restoration. If SRR helps to find out the erroneous response then these combinations can capture all kind of errors as soon as possible. But in the above table it is evident that the gates which helps to get high SRR value will not observe the errors at all times. If two signals gives high restoration value but if one is traced and the other is restored fully then it will be ineffective for error identification. Because tracing of one correlated signal is ineffective then the restored states were also

<table>
<thead>
<tr>
<th>Errors</th>
<th>Flipsflops</th>
<th>A</th>
<th>B</th>
<th>C</th>
<th>D</th>
<th>E</th>
<th>F</th>
<th>G</th>
<th>H</th>
</tr>
</thead>
<tbody>
<tr>
<td>Error1</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>3</td>
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</tr>
<tr>
<td>Error2</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>3</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>4</td>
<td></td>
</tr>
<tr>
<td>Error3</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>2</td>
<td>3</td>
<td>3</td>
<td>4</td>
<td>4</td>
<td></td>
</tr>
<tr>
<td>Error4</td>
<td>NO</td>
<td>NO</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td>4</td>
<td>1</td>
<td>5</td>
<td></td>
</tr>
<tr>
<td>Error5</td>
<td>NO</td>
<td>NO</td>
<td>2</td>
<td>4</td>
<td>3</td>
<td>4</td>
<td>1</td>
<td>5</td>
<td></td>
</tr>
<tr>
<td>Error6</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Errors</th>
<th>Flipsflops</th>
<th>A</th>
<th>B</th>
<th>C</th>
<th>D</th>
<th>E</th>
<th>F</th>
<th>G</th>
<th>H</th>
</tr>
</thead>
<tbody>
<tr>
<td>Error1</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>2</td>
<td>NO</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td></td>
</tr>
<tr>
<td>Error2</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>NO</td>
<td>2</td>
<td>2</td>
<td>3</td>
<td>4</td>
<td></td>
</tr>
<tr>
<td>Error3</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>2</td>
<td>2</td>
<td></td>
</tr>
<tr>
<td>Error4</td>
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<td>3</td>
<td>NO</td>
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<td>2</td>
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</tr>
<tr>
<td>Error5</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>NO</td>
<td>3</td>
<td>NO</td>
<td>2</td>
<td>NO</td>
<td></td>
</tr>
<tr>
<td>Error6</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>NO</td>
<td>3</td>
<td>NO</td>
<td>NO</td>
<td>2</td>
<td></td>
</tr>
<tr>
<td>Error7</td>
<td>NO</td>
<td>NO</td>
<td>1</td>
<td>3</td>
<td>NO</td>
<td>NO</td>
<td>NO</td>
<td>2</td>
<td></td>
</tr>
</tbody>
</table>
follows the same tendency.

As mentioned earlier, restoration is the only criteria that fails to address the debug and validation in an appropriate manner. In [9], some of the restored states does not illustrate for the erroneous design. Sometimes these are different from the actual state that demonstrates in Fig. 2.1 Type 1 bugs are performed to introduce the design bugs. If the trace buffer width is 2, then C and F are the selected to trace the signal combinations in [5]. Here Table 2.3 shows the buggy design of eight restored states of flip-flops. The highlighted position indicates the conflict value of actual response of design bugs. This is done for 8 cycles, at least one of values of each cycle have the conflict values instead of actual restoration value. This is point we consider to propose this system with the following points since it is undesirable for circuit misbehavior in comparison to actual response. One is the multiple occurrences of false restoration which can affect the root cause of the problem and can increase the debug time. And the second is, the incorrect values may lead to wrong path when try to localizing the error during backtracing.


To improve the signal observability, restoration plays a major role since the untraced signals are restored from the traced signal states. In these approaches, flip-flops are not considered for error propagation whereas it is purely depends on the presence and absence of combinational paths in the circuit between the flip-flops (internal signals). So we adventure this principle in our proposed methodology by allowing circuit topology.

At first, we create an S-graph to examine the error propagation from one gate to the other since the actual bugs are present on the combinational path is the main issue. However the reported work says it is a tedious process hence they have used flip-flops instead of gates present on the path. Therefore the proposed work concentrates on the error propagation through the gates using topological sort is the main concept to reduce the complexity. So that it reduces the time and identifies the errors that propagates through the combinational path. While creating an S-graph, gates are represented as nodes and the flip-flops between the gates are represented as path with directed weighted edges. To optimize the error detection and latency, we are ordering the gates in design structure (Algorithm 1). The traced signals must capture the bugs/errors within limited number of clock cycles, so that this method can ensure the error detection latency is curtailed.

It is not necessary to trace the neighborhood of primary output, it can be done at the primary output with high latency. The vicinity of primary input (PI) can be ignored from a set of traced signals that can observe erroneous behavior through flip-flops. After performing the topological sort (S-graph), the list of gates are pruned until the elements are compared with the width of the trace buffer. The main intension behind the above sorting is to select the gate that can capture higher number of erroneous behavior as early as possible. But this method has major limitation that the topological order can be done only on acyclic graph but the real circuits have more number of cycles. Because of this reasons, group of gates forming cycles into single region and then it becomes a directed acyclic graph. This definitely causes implications on bug propagation in order to bug detection. Hence, this aids to give rudimentary solution incurred by limited topological order. A node in a graph (SG) has definite no.of inputs and outputs. The weights are calculated between the edges of two gates along the path. SG” indicates the conversion of building SG by grouping the nodes as acyclic graph which forms a cycles into a single node. For larger cycles this conversion of SG” can break the cycle. To select the particular node we need to find out which node has more priority that gives higher visibility. The outline of the procedure is given in Algorithm 2.
Table 3.1
Algorithm 1

1. Create S-graph (Input: Design; Output: Graph (SG))
   
   Total gates (G) = count the total no. of gates
2. for each gate Gi do
   
   Gates_remaining = Gates_total − Gi
3. for each gate Gj in G_remaining do
   
   node ← 0;
4. if path exists between Gj to Gi then
   
   node ← node + 1; end if
5. edge weight = node
6. node ← 0;
7. link Gi and Gj using edge weight
8. end for
9. SG' = directed weight of the design graph
10. SG = Convert SG''

Table 3.2
Algorithm 2

1. Priority node allocation (Input: Graph (SG); Output: priority list)
2. For each node Ni of graph do
3. Edge_input = no. of incoming edges of each node
4. Edge_output = no. of outgoing edges of each node
5. Weight_input = incoming edge weight
6. Weight_output = outgoing edge weight
7. Notch = Edge_input (Σ Weight_input) + Edge_output (Σ Weight_output)
8. List = priority of each node
9. end for

Table 3.3
Algorithm 3

1. Find pruning_factor pf (Input: Graph (SG); Trace_buffer bandwidth (TB), list of priority node values; Output: Pruning factor)
2. Graph_list = Topological order (SG)
3. Temp_list = Graph_list
4. Pruning_factor= first_pf
5. While Temp_list > TB do
6. Temp_list' = pruned factor of Temp_list
7. Total_priority node values = Temp_list'
8. Number = no. of Temp_list' entries
9. Avg_value = Total_priority node values/number
10. Store pruning_factor and avg_score in a list
11. Pf = pf+pf_increment
12. Temp_list = Temp_list'
13. end while
14. pf = pf of highest avg_score

The node values are calculated and it assigns the priority to each gates for the persistence of ranking with
regards to error collection. These are calculated using how many number of connections are there from one flip-
flop to the other. By multiplying the number of incoming and outgoing edges with their corresponding
weights are summed up to known the total number of connections present. After sorting of gates in the
topological order these should be cut down from the both the ends of primary input and primary output. The pruning
factor can be find by using Algorithm 3.

Here pruning factor (pf) signifies the node which are detached from the list of S-graph. It plays a major role
for the selection of trace signal. The proposed fine-grained algorithm find out the optimal solution that begins
with complete topological order depends on first_pf and pf_increment. The remaining list is represented
as Temp_list’, the priority node value calculation has done for the Temp_list’ and avg_score that obtained
Table 3.4
Algorithm 4

1. Trace_SS (Input: Graph_list, pf, list of priority nodal values, TB_Bandwidth; Output: FF_list)
2. Node value = value of each node from the priority list
3. Graph_list' = graph_list based on pf
4. FF_list' = sort graph_list by corresponding node value
5. FF_list = FF_list' entries based on TB_Bandwidth

Fig. 3.1. Directed Acyclic Graph (DAG)

during calculation. Here pf incrimination is based on the both pf and pf_increment. At last, highest value of pruning factor of avg_score has chosen as pf. These are all based on the window size ($\omega$) of the trace buffer. Based on the percentage of the size of the window, the topological order and the priority list; the arrangements were done and it is shown in Algorithm 4. For example, the pruning factor of 80% and 60% are 0.1 and 0.2 respectively. There is no pruning value for 100%. The topological sorting should be done before pruning factor findings. With the help of both priority assignment and pruning factor, this proposed technique can find more number of bugs within limited clock cycles. To select the signals, the signal selection are done based on the topological order. The remaining nodes of pruning order are arranged using priority node values. Based on the width of the trace buffer the signal selections are done with highest node value.

Based on the above four algorithms (table 3.1-3.4), the erroneous response can be find out easily. Consider a design having 8-combinational gates and its directed acyclic graph is shown in Fig.3.1. This graph contains 8-nodes and its reported results are shown in this section. Based on the priority, the node values are assigned and its corresponding edge values from G1 to G8 are 16, 4, 12, 9, 3, 32, 11 and 14 respectively. These values are arranged in decreasing order to compute the procedure. The pruning factor decides at the previous step is represented as the highest node value after detachment of gates at both the ends. Gate values and its ordering are represented below:

\[
\text{Gates}(G1 - G8) \rightarrow 1, 2, 3, 4, 5, 6, 7, 8
\]

\[
\text{Values} \rightarrow 16, 4, 12, 9, 3, 32, 11, 14
\]

Table 3.5
Topological order (TO) based on priority of node values

<table>
<thead>
<tr>
<th>Window size (%)</th>
<th>TO</th>
<th>By its priority node value (descending order)</th>
</tr>
</thead>
<tbody>
<tr>
<td>100</td>
<td>1, 3, 7, 2, 4, 5, 6, 8</td>
<td>6, 1, 8, 3, 7, 4, 2, 5</td>
</tr>
<tr>
<td>80</td>
<td>1, 3, 7, 2, 4, 5, 60</td>
<td>1, 8, 3, 7, 4, 2, 5</td>
</tr>
<tr>
<td>60</td>
<td>3, 7, 2, 4, 5</td>
<td>1, 3, 7, 4, 5</td>
</tr>
</tbody>
</table>

Employing the distributed trace buffer concept rather than using a single trace buffer that helps to debug the modules efficiently. This method can be applied to any larger circuit. For better understanding the authors has taken the ISCAS ’89 benchmark circuit to represent it.
4. Experimental Results. To evaluate the trace signal selection, some parameters should be identified. They are: State Restorability (SR), Error Detection Latency (EDL) and Non-Detected Bugs after injecting (NDB). The restoration principle can be applied to any type of signal selection like [5, 6, 7, 8]. However, these techniques were insufficient to increase the visibility of internal signal states than the proposed system. To estimate the weakened state of restoration, the above mentioned parameters are used to derive from the following equation.

\[ \Delta SR = SR\{SRR \text{ based Trace } \_SS\} - SR\{obtained \text{ Trace } \_SS\} \quad (4.1) \]

\[ \Delta EDL = EDL\{SRR \text{ based Trace } \_SS\} - EDL\{obtained \text{ Trace } \_SS\} \quad (4.2) \]

\[ \Delta NDB = NDB\{SRR \text{ based Trace } \_SS\} - NDB\{obtained \text{ Trace } \_SS\} \quad (4.3) \]

To evaluate the diminished states, the above mentioned parameters are used. \( \Delta SR \) shows the difference between state restorations for the selected set of trace signals. The difference of error detection latency and the number of errors are reported by \( \Delta EDL \) and \( \Delta NDB \). These three parameters are evaluated more than thousand times on type 1 bug by injecting an error. The signals are selected using either by restoration method or by our proposed topological method. Table 4.1 shows the evaluation of larger benchmark circuits of ISCAS’89 using different parameters as follows. Since the restoration and error restoration approach incompetent, the principle operation of state restoration is applied on the above mentioned parameters.

<table>
<thead>
<tr>
<th>Benchmark Circuit</th>
<th>No. of FFO</th>
<th>SR</th>
<th>EDL</th>
<th>NDB</th>
</tr>
</thead>
<tbody>
<tr>
<td>s38584</td>
<td>1426</td>
<td>0.906</td>
<td>3.86</td>
<td>152</td>
</tr>
<tr>
<td>s38417</td>
<td>1636</td>
<td>-0.263</td>
<td>5.28</td>
<td>491</td>
</tr>
<tr>
<td>s35932</td>
<td>1728</td>
<td>-0.01</td>
<td>39.4</td>
<td>223</td>
</tr>
</tbody>
</table>

There is an alternative method to find out the efficacy of trace signal selection using this proposed technique using different bandwidth of trace buffer. These can be calculated using combinational score value by summing the EDL and NDB for different window sizes (\( \omega \)). The negative sign indicates that this method has higher restorability than existing methods.

\[ A(\omega_i) = EDL_i / EDL_i' + NDB_i / NDB_i' \]  

(4.4)

These can be applied to various modules of [6, 9]. To search the highest value the pruning factor should be searched iteratively. It helps to find out the optimal solution from the pruning factor. As mentioned in previous section, it is applicable to distributed modules. If the flip-flop of window size (\( \omega \)) is traced, then the proposed system can detect the errors in limited cycles. It can be normalized with the help of pruning factor with different window size. The proposed topological sorting method can be applied to automatically generate various learning paths for the mapping analysis concept.

Fig.4.1 shows the error detection ratio between the three larger benchmark circuits. The error detection is defined by

\[ \text{Error detection ratio} = \frac{\text{No. detected errors}}{\text{No. errors detectable}} \]  

(4.5)

Consider the circuit is divided into different regions and each region has one error-prone and we have introduced 50 random errors to the active regions and the error density is directly proportional to region size. Here, we executed two kinds of simulation one with an ideal case and the other with the erroneous signals. We have pragmatic our algorithm to various active regions.
5. Conclusion. Trace buffers are used to store the traced value and to find out the root cause of errors during post-silicon validation and debug. State restoration technique fails to find out the errors present in the traced signals whereas the proposed technique identifies the useful signals that can detect errors. To avoid the problem of false restoration, we can consider the region-separation method. So that we can ensure the selection of signals regardless of topological order.

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REFERENCES


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SECURED IDENTITY BASED CRYPTOSYSTEM APPROACH FOR INTELLIGENT ROUTING PROTOCOL IN VANET

A KARTHIKEYAN *, P G KUPPUSAMY †, AND IRAJ S AMIRI * ‡

Abstract. Vehicular specially appointed systems named as Vehicular Ad-hoc NETwork (VANET) have been raising dependent on the condition-of-art advancements in remote and system communication. The message confirmations among vehicles and infrastructure are fundamental for the VANET security. The genuine personality of vehicles ought not to be uncovered, yet which is just detectable by approved nodes. Existing arrangements either depend vigorously on a carefully designed hardware or cannot fulfill the security necessity. Secured Identity Based Cryptosystem Approach (SIDBC) for intelligent routing protocol is proposed for better results since implementing a secured network for traffic forecasting and efficient routing in dynamically changing environment. Polynomial key generation is utilized for generating identity based pseudonym keys for each and every node that comes under the system. This keying process protects the node from malignant node from passing false information. The assessment output demonstrates that the planned method is more effective than past schemes since it is free pairing and it fulfills security and protection prerequisites.

Key words: Data rate, Vehicular Ad hoc Networks, Trust Authority, Message Authentication, Polynomial Keys.

AMS subject classifications. 68M15

1. Introduction. VANET can be described in simple manner i.e. the vehicles that moves in the roadways are interconnected virtually by passing messages among them. It acts as a kind of Mobile Ad-hoc Networks (MANET) that comprises of mobile nodes and operates in a decentralized way. Infrastructure condition in which, moving devices can be in connection continuously. It is fundamentally formed by those elements that deal with the traffic or offer an outside administration. On one hand, producers are once in a while considered inside the VANET model. A few unique functions are rising in VANETs that incorporates well-being related applications to make more secured driving, portable business and other data benefits that will illuminate drivers about any kind regarding clog, driving perils, mishaps, roads turned traffic jams.

The paper organised is as follows section describes a formal introduction of VANET, section 2 depicts overview of VANET, section 3 describes the proposed security identity based algorithm, section 4 deals with the result analysis of the proposed and existing mechanism and section 5 provides the conclusion of this proposed methodology.

2. Related Works. IVC comes under the operating system of VANETs which turn out to be admired in current years [1]. The vehicles function here as mobile nodes and they are connected together wirelessly. The principle contrast is that versatile switches which manufacture the system are vehicles like cars or trucks. Many academic communities are working for enabling new applications with respect to VANETs [2, 3] for reducing road hazards. Significant role of the method is to enhance the road safety measures.

Several safeties related approaches have been undertaken in VANET environments to enable secure information transmission among vehicles. Identity aware system that has main impacts in reducing accidents and increasing traffic control services were discussed, here the information is shared among them in a timely manner. Public Key Algorithms (PKA) was proposed [4, 5, 6] for providing safety measurements during information

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exchange in Ad-hoc networks. Symmetric keys are generated by the system and master keys are loaded as per arbitrary distribution.

To protect the road environments from false information and privacy of the vehicles a provably secure batch-verification method was proposed. Privacy preserving mechanism [7, 8] for secured vehicular communication includes priori and posteriori counter measures for identifying the number of anonymous vehicles present in RSU. Based on threshold value the priori and posteriori messages can be set and the batch of messages can be verified in a single turn. Scalable Robust Authentication protocol for secured vehicular network was proposed [9], here the group signatures is employed for each RSU rather than centralized authority. RSU preserves and manages the on fly group communication to the up hold range. Anonymous messages passed by the vehicles can be instantaneously verified by the other vehicles that present in the same RSU. If received message resulted to be duplicate then the message originator identity is declared to every vehicle. Identity based security scheme was proposed to achieve privacy in information sharing with fundamental security needs like confidentiality, non-repudiation and authentication. Here privacy preserving defense mechanism [10] was proposed to identify misbehaving vehicles using network authorities. A pseudo-identity-based scheme [11] for conditional anonymity with integrity and authentication in a VANET was proposed and it provides conditional anonymity to guarantee the protection of an honest vehicle’s real identity, unless malicious activities are detected. Certificates are not required in this mechanism since identity based cryptosystem is employed for authenticating the vehicles. Efficient Privacy-Preserving Data-Forwarding Scheme (EPPDFS) [12] was proposed to examine the security aspects in VANET and to increase the user privacy efficiency. Lite Certificate Authority (CA) based Public Key (PK) cryptosystem and route-generate encryption schemes are proposed in EPPDFS. Secure and efficient authentication scheme (SEAS) [13] with unsupervised anomaly detection. Here, certificateless authentication method is deployed for conditional privacy preserving, along with the Chinese remainder theorem for efficient group key distribution and dynamic updating [16]. A cryptographic primitive scheme [15] was proposed for public updating of reputation score based on the Boneh-Boyen-Shacham short group signature scheme. This allows private reputation score retrieval without a secure channel.

3. Proposed Algorithm. SIDBC technique for intelligent routing protocol is proposed; here each task is carried out by different agent allotted for the particular task. Then the agents are cooperatively interacted to form a reliable network structure. Road Side Units (RSU’s) are placed in the infrastructure and On Board Units (OBU’s) are placed in the vehicles and the communication takes place with the help of this devices. Regional Transport Authority provides security keys for vehicles and RSU’s. Routing the packets securely and efficiently with the help of cooperative multi-agent mechanism greatly improves the network throughput [15].

3.1. IDC Authentication Offer. Initially each node registers to the Trust Authority (TA) and security analysis is made for every node that participating in the network. Identity based Cryptosystem (IDC) is used to provide secured environment during data transmission among nodes. IDC allows PK to be derived from its communal identity. Regional Transport Authority (RTA) establishes trust relationship between the nodes using pair-wise keys. RTA generates public/private keys for each and every node that entered into the network. Each RSU is connected to RTA and pre-established keys are generated for trusted region. RSU generates pseudonym for every node that entered into the region using Polynomial Key Generation (PKG). This pseudonym is used to identify the malicious vehicles that access the lane for privacy-conserving authentication and secure communication.

Trust Authority creates pseudonym for every node as given below:

\[
T_A = PK(id)_n \| [T_p] \| C_n
\]  

(3.1)

where,

- \(T_p\) → Present Time
- \(PK(id)_n\) → PK identity of node N
- \(C_n\) → Code name of every node

The source vehicle broadcast the route request message to their neighbor vehicle up to the communication range or the bounded cluster region. Control Request message consist of this parameters such as

\[
CR = \langle ID_s, T, HD, I_N, Nf, SIG(Id_d, T) \rangle
\]  

(3.2)
where,

- $Id_s \rightarrow$ Src node ID
- $T \rightarrow$ Timestamp
- $HD \rightarrow$ Node’s codename
- $IN \rightarrow$ Invite the node
- $Nf \rightarrow$ Node’s Freshness
- $SIG(Id_d, T) \rightarrow$ Signature (key) for sink node with Timestamp.

CH or the vehicle node that present in the cluster region accepts this control message during the present time gap. Later the nodes launch reply message back to the source via intermediate vehicles through SA. Control Reply message contains,

$$ C_{RP} \leftarrow ID_d, P_s, T, connect, SIG(P_s, T) $$

3.2. Key Generation Process. When the authenticity of the hub is demonstrated with their sign confirmation, then the declarations are affirmed from CA. In the event that another hub is participated in the system and set as a switch, at that point the hub requires its substantial character. Without authentication the hub can’t process the data further. Two connection keys ate gotten from the pr-evaluated values (eg., $L_k$ and $Q_v$) and it is produced from the Node ID. When the authenticity of the node is demonstrated ($L_k$, $Q_v$ and $Sidi$) at that point the two hubs can impart. In key concern stage, nodes are approved for keeping it from trickery and pantomime assaults. Here the node A make sense of the node B open key utilizing its certU and CA key’s QCA. The private key ‘KAB’ calculation at together closures is by all accounts indistinguishable just when the endorsements are designated by the legitimate CA. The irregular key generator produces the mystery key Mk for the message encryption process. Each message comprises of basic verification of message authenticated code key ‘K’ and this key shield the message from the change by the malignant nodes.

3.3. Data Processing Process. Base station initiates authentication process once the request message received; the node is verified for legal access by using sign authentication system. Once the signing process finished the certificate is provided for the legitimate nodes also the communication encoding process gets done. Source node sends the encrypted message using hash algorithm and the encryption process commonly includes general authentication key. Then the EMK is sent to the destination with the valid token $CA, Sidi(K)$ added in the header field of the encrypted message.

$$ EM_k = M \parallel L_k, Q_v, Sidi, K $$

4. Simulation Result Analysis. The results demonstrate various types of reproduction parameters utilized in the proposed secured information of convention reenactment. Proposed scheme is analyzed using the parameters like Delivery Rates of data packet (PDR), Average delay and Throughput.

- Delivery Rate of data packet

Delivery Rate of data packets can be computed by taking the ratio between the sum of number of packets that reached over the sink node and the sum of packets sent by the source vehicle. It is obtained from the equation (4.1) below.

$$ PDR = \frac{\sum PktsDelivered}{Time} $$
Set $N$ number of nodes
Registered with BS
Create Sign for each registered node $N_i$ using PKG
Set source node ‘$A$’ and sink node ‘$B$’
Check $NI_d$ for node verification
Set pseudonym key $Y_n$
If $NI_d$(key) matches then provide nodes with certificate ‘CertA’
Do node validation $L_k, Q_v, Sidi$
Create Private Key for legitimate nodes $L_k, Q_v, Sidi, K$
Apply keying function $M || L_k, Q_v, Sidi, K$
Send secured data from $A$ to $B$

$E_{PDR}$ - PDR referred for existing protocol
$P_{PDR}$ - PDR referred for proposed protocol

Here $PktDelivered$ is the number of packets arrived to destination and $PktSent$ is the number of packets sent by the vehicle. The proposed method SIDBC have better delivery rates of packets at the destination compared to the existing system SEAS and it is shown in figure 4.1.

- **Throughput**
  It is described with their rate of effective transmission of data.
  \[
  \text{Throughput} = \left( \frac{\text{PacketsReceived} \cdot 8}{\text{Delay in ms}} \right) \text{ kbps}
  \]  
  \[\text{(4.2)}\]
  The throughput of the proposed SIDBC gives better result compared to the conventional SEAS method. When delivery rates increases then throughput also increases gradually (figure 4.2).

- **Delay**
  It is characterized as the time contrast between the present data packets received and the received time of past packets.
  \[
  \text{Delay} = \frac{\sum_{n} \text{Pkt recvd Time} - \text{Pkt send Time}}{n}
  \]  
  \[\text{(4.3)}\]
  From the figure 4.3, it is clearly shows that the average delay of the proposed SIDBC scheme has consumed lesser time delay for transferring the packets compared to the SEAS (E-Dly.tr) protocol.
Fig. 4.1. Delivery Rates of packet

Fig. 4.2. Throughput

Fig. 4.3. Delay
5. Conclusion. A Secured Identity Based Cryptosystem Approach for intelligent routing protocol is proposed for better results since implementing a secured network for traffic forecasting and efficient routing in dynamically changing environment is a challenging task. So as to ensure that it can fulfill message confirmation necessity, existential enforceability of fundamental signed key against adaptively selected message attack. Polynomial key generation is utilized for generating identity based polynomial keys for each and every node comes under the network. The evaluation results show better efficiency rate compared to previous methods as shown in the result analysis throughput and delivery rates.

REFERENCES


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AN APPLICATION BASED EFFICIENT THREAD LEVEL PARALLELISM SCHEME ON HETEROGENEOUS MULTICORE EMBEDDED SYSTEM FOR REAL TIME IMAGE PROCESSING

K. INDRAGANDHI∗AND P.K. JAWAHAR†

Abstract. The recent advent of the embedded devices is equipped with multicore processor as it significantly improves the system performance. In order to utilize all the core in multicore processor in an efficient manner, application programs need to be parallelized. An efficient thread level parallelism (ETLP) scheme is proposed in this paper and uses computationally intensive edge detection algorithm for evaluation. Edge detection is the important process in various real time applications namely vehicle detection in traffic control, medical image processing etc. The main objective of ETLP scheme is to reduce the execution time and increase the CPU core utilization. The performance of ETLP scheme is evaluated with basic edge detection scheme (BEDS) for different image size. The experimental results reveal that the proposed ETLP scheme achieves efficiency of 49% and 72% for the image size $300 \times 256$ and $1024 \times 1024$ respectively. Furthermore an ETLP scheme reducing 66% execution time for image size $1024 \times 1024$ when compared with BEDS.

Key words: Heterogeneous, Multicore Processor, Image processing, Edge Detection, CPU utilization

AMS subject classifications. 68W10, 94A08

1. Introduction. Multicore processor plays a drastic effect in the digital world. In earlier days, single core processors were in market in which performance was achieved by increasing the frequency and transistors count per chip. The dynamic power dissipation ($P_d$) of CMOS chip is given in equation (1.1):

$$P_d = CV^2 f$$

(1.1)

where $C$ is the capacitance, $V$ is the supply voltage, $f$ is the frequency. From the equation 1, it was observed that the frequency is directly proportional to the power dissipation. When we increase the frequency of single core processor to get good performance, the chip will dissipate more power which in turn increases the heat [1]. In 1990, Processor performance was increased by 60% per year but from 2000 to 2004, it was reduced to 40% per year with only 20% increase in performance. In a single core processor, multi tasking was based on time slice which will degrade the performance if more number of multiple tasks wants to be executed parallelly [2]. Compared to many high frequency single core processor, multicore processor chip will dissipate less heat since it has simpler CPU core. Multicore processor means many cores are fabricated in a single socket in which each core operating in same or different frequency. Performance is achieved in multicore chip by allocating the task parallelly on all cores. Parallel processing is easily achieved in multicore chip compare to multiple single core chips with low cost. The main feature of multicore chip is that tremendous increase in performance by increasing the number of cores instead of increasing the frequency [3]. To improve the multicore CPU performance, three factors namely parallelism granularity, incorrect programming model and language compilers to be tuned [10].

Nowadays, most of the embedded applications are parallel processing application. To improve the performance, an application must be multithreaded. But multithreading an application is a time consuming work. Programmers concentrate mostly on the efficiency of an application. Since most of the popular programming languages support the sequential code, the programmers prefer to write an application in sequential manner. Using the automatic parallelization tool eg. Par4All, Cetus, TRACO etc., the sequential code is converted into

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parallel application. If the code is too complex, the compiler may wrongly parallelize the code which may increase the execution time [7]. For a last decade, Computer vision field has a drastic growth but the need for real time image processing is not fulfilled by conventional sequential computing. These leads to parallel processing high performance computing. The image processing and signal processing programmers can benefit dramatically from the advances in multicore processor, by modifying single threaded code to multithreaded code [5]. Xin Gao [18] proposed a post processing schemes for vehicle detection in wide area aerial imagery, the proposed scheme gives better performance for this kind applications.

In this paper, we propose an Efficient Thread Level Parallelism (ETLP) scheme to utilize the CPU cores efficiently in an embedded device. ETLP scheme is applied based on the application. ETLP scheme provides very good efficiency for multithreaded application.

2. Related Work. Hahn Kim et al. [4] provides a survey of software technologies for image and signal processing on multicore architectures. The programmer need not to aware of the complexity of multicore architectures, instead of that, they can focus on application algorithms. Parallel programming languages, high level languages, general purpose programming on GPUs and middleware libraries were described and their efficiencies are compared.

Marungo et al. [8] presented a extended version of OpenMP parallel programming model for a multi-processor system on chip embedded system. The task were scheduled among the accelerators and host using OpenCL and OpenMP run time environment. The algorithm was evaluated in STMicroelectronics STHORM development board using standard benchmark applications. The results shows 30x speed up using OpenMP compared to OpenCL.

S. Mittal and J. Vetter [9] surveyed the heterogeneous computing using CPU and GPU processor. The performance of heterogeneous computing technologies are reviewed at algorithm, programming, compiler, runtime and application levels.

T. Singh et al. [11] developed the tools to measure performance and CPU core utilization for multicore architecture. The quick sort with different set of data elements are executed serially and parallelly in dual and quad core CPU. The compared results are represented as graph using those tools and exploiting more parallelism provide better CPU core utilization.

A. Goyal et al. [12] presented a comparative study of edge detection algorithm using sobel, prewitt and canny filter on satellite images using OpenACC, OpenMP, MPI and hybrid OpenMP/MPI model in multicore architecture. The comparative results showed that the edge detection algorithm using OpenACC achieved greatest speedup over other models.

S. Bernabé et al. [13] developed a new parallel un mixing chain for hyper spectral images sensing in multicore processors. The optimization were achieved using OpenMP and Intel Math Kernel Library for real time image processing on CPU-GPU architecture. The results reveal better performance compared to GPU architecture.

A. Monteria et al. [14] developed a small, low cost and lightweight convolution neural network (CNN) – embedded device using Raspberry Pi 3 to find faults in structural health monitoring system. The case study were evaluated experimentally using piezoelectric patches glued over an aluminum plate and results reveal 100% effective hit rate.

G. Tagliavini et al. [15] proposed a fine grained parallelism using a parallel programming model OpenMP for embedded programmable many core accelerators. The run time environment (RTE) design were implemented in Kalray MPPA 256 and the results showed an average speed up of 12× for real benchmarks compared with traditional scheduling of tasks.

A. Nadjaran et al. [16] presented CLOUDS-Pi platform, a low cost software defined cloud data center for research purpose using raspberry pi embedded computers integrated with OpenVSwitch to develop a network of open flow switches.

W. Kwedlo et al. [19] proposed four approaches namely Drake’s, Elkan’s, Annulus, and Yinyang algorithms to minimize the unnecessary distance calculation in K-means clustering algorithm. In this paper a hybrid MPI/OpenMP programming models are used to parallelize these algorithms. The results of proposed method showed efficiency in computing time compared to Lloyd’s algorithm.

M.K. Pekturk et al. [20] developed a spectral angle mapper and matched filter algorithm using OpenMP
programming model for online remote sensing applications. The experimental result reveals that 3-4 times & 16-19 times improvement for offline and online data processing.

3. Proposed Method. The main idea of proposed Efficient Thread Level Parallelism (ETLP) scheme is to exploit the full processing power of multicore processor by using parallel programming model for multithreaded application. The proposed system implemented in raspberry pi 3 embedded device for image processing application to increase the performance and CPU core utilization. In our method edge detection application is chosen for evaluation. The ETLP scheme is achieved using parallel programming model OpenMP. The OpenMP has a runtime library routines, a set of pre-processor directives and environment variables. The ETLP scheme is implemented based on the multithreaded application. The programmer want to identify which section of application source code can be multithreaded and single threaded, based on that, we want to use the appropriate OpenMP pragmas, runtime library routines, etc., The proposed scheme is evaluated for edge detection process using sobel, prewitt and scharr operator for different image size and compared with Basic Edge Detection Scheme (BEDS). BEDS represents edge detection execution in sequential approach. The proposed ETLP scheme shows the better performance, efficiency and CPU core utilization compared to BEDS.

4. Parallel programming model – OpenMP. A single core processor execute only one task at a time in a sequence manner and it may take hours or days to execute complex application. A multicore processor can execute many task parallel which help to solve complex application with less time and cost compared to single core processor. OpenMP is an powerful application programming interface, those API's will support more functionalities needed for parallel programming .Manual parallel execution is achieved by using the parallel programming model. The programming model will act as a bridge between software and hardware. Some of the parallel programming models are Pthread, MPI, OpenMP, CUDA etc. In the proposed method, OpenMP parallel programming model is used to improvise the performance of embedded device. OpenMP is designed for shared memory based multicore architecture.

OpenMP provide explicit full control to parallelize the code for the programmers. Using threads, OpenMP parallelize the code. The thread is a small unit of processing, scheduled among the cores. OpenMP follow fork-join model. From Semr Aydin et al. [17], a fork is the master thread begin executing serial region while encountered the parallel region, it creates group of independent threads and all the threads are executed parallel.

Once the parallel execution over, the results are synchronized and joined as a master thread to execute serial region of the code as shown in Fig.. 4.1. The OpenMP API are available in C and FORTRAN language. In the proposed method C language based API is used. The general C structure of OpenMP is shown in Fig. 4.2.

OpenMP directives are selected based on various parts of source code. Before implementing the OpenMP directives, analyse the code carefully, because improper selection of OpenMP directives may degrade the performance and execution time may get increased.

5. Image Processing – Edge detection. An image is a 2-D function $f(x, y)$, where $f$ is the intensity and $(x, y)$ are coordinates of pixel. If an image is finite and discrete, then it is known as digital image. Digital image processing helps to exploit the digital image in the computer system. Image segmentation is the part of

![Fig. 4.1. Fork – join model of openMP](image-url)
#include <omp.h>
main ()  {
    serial execution code
    parallel region begin
    Fork a threads as group.
    #pragma omp parallel
    {
        All the threads execute parallel
        Run-time Library calls
        Other OpenMP directives
        ....
        ....
        All threads are joined as a master thread
    }
    Resume serial execution of code
}

Fig. 4.2. General C structure of openMP

Table 5.1

<table>
<thead>
<tr>
<th>Direction</th>
<th>Sobel Operator</th>
<th>Prewitt Operator</th>
<th>Scharr Operator</th>
</tr>
</thead>
<tbody>
<tr>
<td>Horizontal</td>
<td>+1 0 -1</td>
<td>+1 0 -1</td>
<td>+3 0 -3</td>
</tr>
<tr>
<td></td>
<td>+2 0 -2</td>
<td>+1 0 -1</td>
<td>+10 0 -10</td>
</tr>
<tr>
<td></td>
<td>+1 0 -1</td>
<td>+1 0 -1</td>
<td>+3 0 -3</td>
</tr>
<tr>
<td>Vertical</td>
<td>+1 +2 +1</td>
<td>+1 + +1</td>
<td>+3 +10 +3</td>
</tr>
<tr>
<td></td>
<td>0 0 0</td>
<td>0 0 0</td>
<td>0 0 0</td>
</tr>
<tr>
<td></td>
<td>-1 -2 -1</td>
<td>-1 -1 -1</td>
<td>-3 -10 -3</td>
</tr>
</tbody>
</table>

Edge detection operators

image processing. Image segmentation guides to extract the useful information from the image. It is typically used to detect the edges, curves and boundaries of object in an image. Edge detection is one of the essential processes of image segmentation. Edges are substantial changes of intensity in an image. The boundary between two different segments in an image is called as edges. The four basic steps of edge detection are smoothing, enhancement, detection and localization. Edge detection is done by two methods namely first order derivative and second order derivative. Edge points in an image can be detected by finding the minima and maxima of first derivative method or detecting zero crossing of second derivative method. Spatial kernel operator is used to detect the edges in horizontal and vertical direction of an image.

In the proposed ETLP scheme, Edge detection is carried out by gradient method of first order derivative using sobel, prewitt and scharr operator. To generate the gradient of an image, convolute the input image with 3X3 kernel matrix of particular operator. Table 5.1 represents the kernel matrix of sobel, prewitt, scharr operator for horizontal and vertical edge detection.

\[
\text{Horizontal derivative approximation (} G_x \text{)} = 3 \times 3 \text{ kernel} \ast f(x, y) \tag{5.1}
\]

\[
\text{Vertical derivative approximation (} G_y \text{)} = 3 \times 3 \text{ kernel} \ast f(x, y) \tag{5.2}
\]

\[
\text{Gradient magnitude (} G \text{)} = \sqrt{G_x^2 + G_y^2} \tag{5.3}
\]

where \( f(x, y) \) represent input image and \( \ast \) stand for 1-D convolution operation

6. Hardware description. Raspberry Pi 3 is a single board heterogeneous embedded device, which includes BCM2837 Broadcom processor, 1.2GHz quad core 64- bit ARM Cortex A53 processor, maximum stressed power consumption of 1W and 1GB RAM [14]. Since it is a quad core processor , four tasks can
run parallelly. By principle, the performance is four times greater than the original speed, but in real time it is difficult to make use of full potential of multicore processor. To utilize the full processing power of quad core processor, the proposed ETLP scheme uses the parallel programming model OpenMP. The proposed method helps to improve the execution speed by using software optimization in a raspberry pi 3 embedded devices. Raspbian is the primary operating system for the Raspberry Pi family boards. To implement parallel programming model, CPU architecture should be well known. Fig. 6.1 shows Raspberry Pi 3 CPU architecture.

![CPU architecture details of Raspberry Pi 3](image)

Fig. 6.1. CPU architecture details of Raspberry Pi 3

7. ETLP scheme. The edge detection algorithm is written in C language and compiled in GCC compiler of raspbian operating system. Before implementing ETLP, analyze the application source code and identify the private variable, shared variable and critical section. Loop level parallelism is achieved only if there is no loop dependence. The main focus is to increase the performance without affecting the expected result. So we should use the OpenMP cautiously. The pseudo code for BEDS is given in Fig. 7.1. In the code Step 4 and 5 has nested four for loops and it is a time consuming tasks. The critical section, shared and private variables are mentioned in Fig. 7.1. Race condition is one of the major issues in thread level parallelism. Race condition will occur when multiple threads try to access the same shared resource, these leads to deadlock. Critical section is a piece of code, which access the shared variables. Only one thread is allowed to enter the critical section for execution at a time. Threads are get synchronized at this point. The kernel operator is Sobel, Prewitt or Scharr.

Fig. 7.2 is the pseudo code of ETLP scheme. Based on the code analysis of Fig. 7.1, the OpenMP pragma, run time libraries are used in the code. Collapse clause instructs the compiler to collapse the first two for loops. The collapsed for loops should not have any instruction between them. Raspberry Pi3 is having a quad core processor, so threads are equally distributed among the cores using OpenMP dynamic scheduling mechanism. In a dynamic schedule, based on the workload the iterations are distributed among the threads at runtime. In the proposed method, dynamic scheduling with chunk size 50 is used, representing 50 iterations are distributed among the threads dynamically based on the availability of core during run time.

The proposed ETLP scheme is evaluated with sobel, prewitt and scharr operator for different image size. Four different image size of $300 \times 246$, $512 \times 512$, $720 \times 526$ and $1024 \times 1024$ are selected for evaluation. Fig. 7.3 shows the original image and edge detected image using BEDS and ETLP scheme. Fig 7.3.a is the original image of size $720 \times 526$, Fig 7.3.b and c are edge detected image using BEDS and ETLP scheme respectively. From the figure it is observed that the edge detected output is same for BEDS and ETLP scheme.

The edge detection application is executed three times and an average time was calculated for evaluation of proposed method. Different sizes of images are used to highlight performance of ETLP scheme. Table 7.1 represents the time taken for different image size using sobel, prewitt and scharr operator.

Fig. 7.4 Shows the execution time comparison for BEDS and ETLP scheme of different image size. Figs. 7.4.a, b and c are comparison of execution time using Sobel, Prewitt and Scharr operator respectively. From the Fig. 7.4.a it is observed that for image size $300 \times 246$ the execution time is 0.33 s for BEDS and 0.166 s for ETLP scheme, nearly 50% execution time is reduced by the proposed method. Similarly for image size
Algorithm 1: Basic Edge Detection Scheme (BEDS)

1. Input image <- IP [a] [b]
2. Width<- w, Height<- h, Maximum<-M & Minimum<-m
3. 3X3 Kernel for horizontal direction <-KX [c]
d & vertical direction <-KY[c][d]
4. Calculating the gradient value(G)  //G,Gx,Gy are private variable and M & m are shared
   for a<- 1 to w{
      for b<- 1 to h{
         for c<- -1 to <=1{
            for d<- -1 to <=1{
               Gx +<- KX(d+1)(c + 1) * IP(a + c)(b + d)
               Gy +<- KY(d+1)(c + 1) * IP(a + c)(b + d))
               G<- square root of ((Gx * Gx) + (Gy *Gy));
               if G is less than m then , m <- G                        //critical section
               if G  is greater than M) then, M<- G;}}
   G<- square root of ((Gx * Gx) + (Gy *Gy))
   S<- maximum brightness * (G-m)/(M-m)
   OM[a-1][b-1]<-S}}

Algorithm 2: ETLP scheme

1. Include OpenMP header file
2. omp_set_num_threads(4) <- To utilize the quad core processor
3. Start time<-omp_get_wtime();
4. Input image <- IP[a] [b]
5. Width<- w, Height<- h, Maximum<-M & Minimum<-m
6. 3X3 Kernel for horizontal direction<-KX [c][d] & vertical direction <-KY[c][d]
7. Calculating the gradient value
   #pragma omp parallel for private(G,Gx,Gy,c,d) shared(M,m) collapse(2)
   schedule (dynamic,50)
   for a<- 1 to w{
      for b<- 1 to h{
         for c<- -1 to <=1{
            for d<- -1 to <=1{
               Gx +<- KX(d+1)(c + 1) * IP(a + c)(b + d)
               Gy +<- KY(d+1)(c + 1) * IP(a + c)(b + d))
               G<- square root of ((Gx * Gx) + (Gy *Gy));
               #pragma omp critical
               (if G is less than m ,then  m <-G
                if G  is greater than M ,then M<- G;}}
   G<- square root of ((Gx * Gx) + (Gy *Gy))
   S<- maximum brightness * (G-m)/(M-m)
   OM[a-1][b-1]<-S}}
8. Output image<- OM [a] [b]
   #pragma omp parallel for private(G,Gx,Gy,c,d) shared(OM) collapse(2)
   schedule (dynamic,50)
   for a<- 1 to w(
   for b<- 1 to h{
   for c<- -1 to <=1{
   for d<- -1 to <=1{
   Gx +<- KX(d+1)(c + 1) * IP(a + c)(b + d)
   Gy +<- KY(d+1)(c + 1) * IP(a + c)(b + d))
   G<- square root of ((Gx * Gx) + (Gy *Gy))
   S<- maximum brightness * (G-m)(M-m) OM[a-1][b-1]<-S}}
9. End time<-omp_get_wtime();
10. Execution time <- End time- Start time

Fig. 7.1. Code structure of BEDS

Fig. 7.2. Code structure for ETLP scheme

1024 × 1024 the execution time is 4.679 s for BEDS and 1.712 s for ETLP scheme, nearly 63% execution time is reduced. Thus the proposed method gives better performance for large amount of data.

The performance of proposed method is evaluated based on speed up, performance improvement and efficiency using execution time for sequential and parallel execution [6]. The expression for speedup, performance improvement and efficiency are given in equations (7.1), (7.2) and (7.3).

\[
\text{Speedup} = \frac{\text{Execution of Sequential method (s)}}{\text{Execution of Parallel method (s)}}
\]  

(7.1)
An Application based Efficient Thread Level Parallelism Scheme on Heterogeneous Multicore Embedded System

Fig. 7.3. Comparison of edge detection output using BEDS and ETLP scheme: (a) Original image 720 × 526; (b) Edge detected with BEDS; (c) Edge detected with ETLP scheme.

Table 7.1

<table>
<thead>
<tr>
<th>Image Size</th>
<th>Iteration</th>
<th>Sobel BEDS</th>
<th>Sobel ETLP</th>
<th>Prewitt BEDS</th>
<th>Prewitt ETLP</th>
<th>Scharr BEDS</th>
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<td>1.711</td>
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<td></td>
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<td>1.149</td>
<td>0.464</td>
<td>1.216</td>
<td>0.461</td>
</tr>
<tr>
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<td>0.199</td>
</tr>
<tr>
<td></td>
<td>2</td>
<td>0.376</td>
<td>0.172</td>
<td>0.315</td>
<td>0.169</td>
<td>0.343</td>
<td>0.151</td>
</tr>
<tr>
<td></td>
<td>3</td>
<td>0.344</td>
<td>0.163</td>
<td>0.312</td>
<td>0.17</td>
<td>0.345</td>
<td>0.172</td>
</tr>
<tr>
<td></td>
<td>Average</td>
<td>0.355</td>
<td>0.166</td>
<td>0.313</td>
<td>0.165</td>
<td>0.343</td>
<td>0.174</td>
</tr>
</tbody>
</table>

Fig. 7.4. Comparison of execution time for BEDS and ETLP scheme using different operators and different image size: (a) Sobel operator; (b) Prewitt operator; (c) Scharr operator.
In our proposed method, Raspberry Pi 3 embedded board is used which consists of quad core processor. In equation 7.3, number of processor is four. Fig. 7.5 shows the performance evaluation of proposed method. Figs. 7.5.a, b and c show the speed up, performance improvement and efficiency respectively for different image size and different operator. From Fig. 7.5.c, it is observed that the efficiency of proposed method using sobel, prewitt and scharr operator for the image size $1024 \times 1024$ are 68.33%, 70.45% and 72% respectively. Table 7.2 is the performance evaluation of image size $1024 \times 1024$.

The main objective of proposed method is not only increase the efficiency also increase the individual CPU core utilization. Fig. 7.6 shows the comparison of CPU core utilization for various image size using different operator. Figs. 7.6.a, b, c and d shows the CPU core utilization for image size $300 \times 246$, $512 \times 512$, $720 \times 256$ and $1024 \times 1024$. Consider Fig 7.6.b, sobel operator based BEDS method, the CPU0,CPU1,CPU2 & CPU3 utilization is 39%, 2%, 2% & 0% respectively. Even though the processor having four CPUs, only CPU0
Table 7.2
Sample performance evolution of ETLP scheme for image size 1024X1024

<table>
<thead>
<tr>
<th>Edge detection operator</th>
<th>Speed up</th>
<th>Performance Improvement</th>
<th>Efficiency</th>
</tr>
</thead>
<tbody>
<tr>
<td>Sobel</td>
<td>2.73</td>
<td>63.411</td>
<td>68.33</td>
</tr>
<tr>
<td>Prewitt</td>
<td>2.82</td>
<td>64.51</td>
<td>70.45</td>
</tr>
<tr>
<td>Scharr</td>
<td>2.88</td>
<td>65.27</td>
<td>72.00</td>
</tr>
</tbody>
</table>

**Fig. 7.6.** Comparison of CPU core utilization of BEDS and ETLP scheme for different image size and operator: (a) CPU utilization for image size 300×246; (b) CPU utilization for image size 512×512; (c) CPU utilization for image size 720×256; (d) CPU utilization for image size 1024×1024

is utilized more compare to others and CPU3 is in idle state. In sobel operator based ETLP scheme, the CPU0, CPU1, CPU2 & CPU3 utilization is 13%, 11%, 12% & 6% respectively. Thus the proposed method utilized the CPU core efficiently by distributing the thread among the CPU core.

8. Conclusion and Future work. The main ultimate goal of many embedded device is, how to increase the performance of the application without degrading the expected outcome. The life time of multicore processor may reduce if the threads are not evenly distributed among the CPU core. The proposed ETLP scheme shows better performance compare to traditional method of execution. Multimedia applications based embedded
device get more advantages since the most of the operations are multithreaded. The proposed ETLP scheme is implemented in quad core and evaluated for different image size. From the experimental results, it is concluded that the proposed method shows the gradual increase in performance and CPU cores are evenly utilized for small to large image size. The performance of proposed ETLP scheme can be further extended for larger parallel data and many CPU core architecture.

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A STUDY TO ANALYZE ENHANCEMENT TECHNIQUES ON SOUND QUALITY FOR BONE CONDUCTION AND AIR CONDUCTION SPEECH PROCESSING

PUTTA VENKATA SUBBAIAH* AND HIMA DEEPTHI V.†

Abstract. In recent years, enhancing speech signal for communications become more complex due to the environmental noises for military and navel applications. Many researchers analyzed and developed complex algorithms and expressed ways to enhance speech signal for both Bone Conducted Speech (BCS) and Air Conducted Speech (ACS) processing techniques. BCS signal is vigorous for outer environmental noises and with low quality signal. ACS signal is sensitive for environmental noises but rigid in amplitude and high frequencies. Here we have reviewed and investigated various articles to explore their ideas in speech processing for enhancing intelligibility of sound quality.

Key words: Bone Conducted speech, Air Conducted speech, Sound quality, Signal Enhance, Speech intelligibility, Cochleae implant, Deep neural network, Linear prediction, Denoising.

AMS subject classifications. 92B20, 94A12

1. Introduction. Speech is primary tool and form to communicate by human beings. Few issues make the speech to restrict in outer environmental areas. Researchers utilized some hearing aids and speakers to overcome from the restrictions. There are two types of ways to collect the speech data from humans. Air Conducted Speech (ACS) signals have high efficiency, collected through speakers and sensitive with outer environmental noises. Bone Conducted Speech (BCS) signals have high amplitude, robust to the outer environmental issues by restricted with high frequency voice signals.

To enhance the intelligibility of speech in noise environment, automatic speech recognition systems are introduced with the help of complex algorithms to reduce or cancel the low level noises, but ineffective for high level of noises due to algorithm and equipment limitations. To overcome this microphones are utilized to record the speech signals through the bone conducted speech [BCS] process.

There are 85000 of deaf people can not sense any sounds with normal hearing aids. Normal human hearing range should be less than 24 KHz. Speech processing researchers fall on various bone conducted speech techniques and developed Ultrasonic Bone Conduction techniques, can be used to hear higher frequencies above 24 KHz. Several methods of BCS signals retrieving can be done by cochleae Implant (CI) i.e., a microphone will be placed behind skin by surgery. Several hearing aids rapidly explored, came to exist for BC signal retrieving. On implanting the hearing aids under skin, the performance of microphone in CI should not be attenuated by acoustic damping effect. Additionally due to the vibrations arrived of various body parts like mouth, skull and other the unexpected biological signals may generated and effects the performance of hearing aids microphone. To enhance the speech signal various conventional methods, fully implantable hearing devices (FIHDs) are developed for sensorineural hearing impaired people. Now a days a few approximately 5% of people are going for CI methods to embed in the skin, to outperform this Bone Conducted Ultrasonic hearing Aids are used for profoundly deaf sense people. Wearable electronic devices are used for military and navel applications to overcome from outer noises for ACS signals recording. ACS can go for high frequency values but sensitive with environmental noises but have better performance for high frequencies.

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This paper explores various methodologies developed by researchers to enhance in speech processing techniques by comparing or by combining the BC Speech signal and AC Speech signal. The scope of paper is as follows. Section 1 deals with basic information and ideology of speech processing issues, Section 2 with a study on researcher’s propagation to enhance speech processing techniques, section 3 with a discussion on escalations obtained from Section 2, Section 4 analyzing all sections.

2. Brief review on approaches. To recover the profoundly deaf sense of auditory sense the 5% or less of them utilizes the hearing aid Cochlea Implant [CI] for its unfinished concert and a surgery needed to embed behind skin. According, Nakagawa et al.[1] to overcome this Bone conducted ultrasonic hearing aid developed for profoundly deaf sense audition is utilized without surgery with 20% of words were able to extricate successfully with further developments in epoch assembly confidently, and approximately 47% of profoundly deaf themes were not able to recognize the sounds through this aid. According to Dong-Wook Kim et al. [11], Cochlea aid’s performance mainly effected by various body vibrations causing biological noise done while eating, playing and other. This can be reduced by restructuring the microphone and signal processing of FIHDs.

According to Yishan Luo et al. [5], the comparison of restoring speech signals of normal hearing people to the profoundly deaf people, Bone conducted ultrasonic hearing aid (BCUHA) will perform very efficiently by comparing using Finite Difference Time Domain with the restoration signals with Conventional bone conducted hearing aid (BCHA) and Cochlea implant. The conventional BCHA have limited output energy and dampening effect in frequency response, and BCHA can’t adopt contralateral cochlea unilaterally so Bilateral BCHA should be used. Where in as BCUHA unilateral is sufficient and will have great significance to stimulate both cochlea.

According to Aminaka et al. [12], amyotrophic lateral sclerosis patients suffering from ear-stacking-syndrome have difficulty in auditory stimuli perception, by removing the obstructed auditory AC channel employing bone conduction auditory BCI for testing a few Morse-code auditory bone conducted patterns for speller collected from BCI 2000 bundle. 80% of test results through online spelling interfacing explores the technique is efficient.

Shunsuke Ishimitsu expressed in his book [25] as in noisy of marine engine systems for body conducted speech with small repetitions of utterance through the effectiveness of adaptive processing, above 50% of recognition rates at 98dB SPL were successfully achieved with in one utterance of speech recognition rate of higher value of 95% were attained. In a few experiments on healthy people retrieved speech, high frequency of formant information of speech signals were retrieved and possible applied for conversion and to estimate natural speech retrieval for speech disorders people.

According to Yamasaki et al. [9], speech processing technology is used for digital speech coding, spoken language, text to speech, automatic speech recognition (ASR). Information can also be extracted from speech. Yamasaki et al. proposed two new techniques for robust bone conducted speech in speaker authentication systems in highly noise environments. These proposed transformations methods shows improved equal error rate in conventional speech transformation.

According to Satoki Ogiso et al. [14], the characteristics of BC sound generated in human head to be studied through computer interfaced simulation in Finite Element Method (FEM), results the sound wave propagates in form of longitudinal and curving wave via skull of approximatively 1500 M/S and 260 M/S. The propagation curve wave in the skull is very slower in speed with high amplitude than of longitudinal wave. According to Zhang et al. [21], perfect frequency estimation in noisy environment is done by coalescing weighted auto correlation function (WACF) with Cepstrum (CEP) method for enhancing the speech in ARS and BCS in white noise environment. At variable SNR conditions, to improve the pitch signal in white noise environment the compared, clear voice ACS to the BCS signal is utilized. The proposed CEP scheme for female speech value shows better performance with lower error rate at high SNRs than ACF and WACF schemes. The experimental conditions shown in Table 2.1 chosen for 4 male and 4 female speech at same time with BC microphone were recorded are shown below.

According to Shimamura et al. [2], by passing speaker output of bone conducted speech signal a long term spectra reconstructed filter design, the arena of bone conducted and reconstructed speech from normal signal could be highly utilized. By passing normal speech signal through linear process with a transfer function of bone conduction can be obtained, to identify the properties of numerous sounds, spectral shapes by removing the dissimilarities. In Noisy environments, speech enrichment is done by using Bone conduction, to reconstruct
Table 2.1
The experimental conditions

<table>
<thead>
<tr>
<th></th>
<th>Male 4, Female 4</th>
</tr>
</thead>
<tbody>
<tr>
<td>Speaker</td>
<td>Male 4, Female 4</td>
</tr>
<tr>
<td>Flame Shift</td>
<td>64</td>
</tr>
<tr>
<td>Flame Length</td>
<td>256</td>
</tr>
<tr>
<td>Sampling Frequency</td>
<td>12 kHz</td>
</tr>
<tr>
<td>FFT Point</td>
<td>1024</td>
</tr>
<tr>
<td>Bone - Microphone</td>
<td>Temco HG - 17</td>
</tr>
<tr>
<td>Air - Microphone</td>
<td>BOSH ECM - 17</td>
</tr>
<tr>
<td>SNRs (db)</td>
<td>∞, 10, 5, 0, 5</td>
</tr>
</tbody>
</table>

Table 2.2
Comparison Table of MSE for FIR and FLANN recovered signals

<table>
<thead>
<tr>
<th>SNR (dB)</th>
<th>-3</th>
<th>-1</th>
<th>0</th>
<th>1</th>
<th>10</th>
<th>100</th>
</tr>
</thead>
<tbody>
<tr>
<td>Female</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>FLANN</td>
<td>1.8103</td>
<td>1.8023</td>
<td>1.7801</td>
<td>1.7702</td>
<td>1.7604</td>
<td>1.7528</td>
</tr>
<tr>
<td>FIR</td>
<td>1.9481</td>
<td>1.9287</td>
<td>1.9128</td>
<td>1.9063</td>
<td>1.8922</td>
<td>1.8895</td>
</tr>
<tr>
<td>Male</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>FLANN</td>
<td>1.7806</td>
<td>1.6774</td>
<td>1.5883</td>
<td>1.4962</td>
<td>1.3142</td>
<td>1.2936</td>
</tr>
<tr>
<td>FIR</td>
<td>1.9754</td>
<td>1.7980</td>
<td>1.7336</td>
<td>1.6871</td>
<td>1.5143</td>
<td>1.4898</td>
</tr>
</tbody>
</table>

the bone conducted speech signal from speaker a linear phase impulse response filter will be utilized effectively. The designed linear phase FIR filter improves the quality of bone conducted speech with speech enhancement method, at around filter length 64 can be tolerably utilized without effected by filter length [3].

According to Ran Xaio et al. [19], artificial neural network based linear and non-linear adaptive noise cancellers (ANC) is utilized to improve the speech de-no semantic for BCS and ACS signals. At noisy conditions, high frequency components are recovered by the proposed ANC and outperforms its links for FIR filter, FLANN and other. Watanabe et al. [20, ?]

Singh et al. [24], by regenerating sounds from Mel frequency ceptrum [MFC] coefficients, the mitigated frequency components have no effect, to overcome this effect and to improve of low frequency components of BCS signal, introduced a wavelet transform for the removal of noise from same speech by utilizing anti phase high frequency components in speech.

In noise environments like live military scenarios, ships and other, the air conducted speech will be highly distorted if the outer noise is very harsh. To overcome this a wearable bone conducted speech system was implemented and signals were collected from various skull positions which have very high frequency components and compared with air conducted speech signals. According to Boyan Haung et al. [22], in order to enhance the Bone and Air Conducted speech signal, employing FIR filter alone may not shows better performance, using trigonometric expansions of FLANN filter combined together be the better signal enhance. The proposed scheme have smaller MSE with recovered speech signal and clean speech signals this shown in Table 2.2. This can be achieved by taking largely effected ACS signal as primary with BCS as reference signal [17]. The experimental results for recovered signal from FIR and FLANN filter are compared each with clean signal, noisy signal and bond conducted is shown in Figure 2.1.

According to Thang Tat Vu et al. [4], due to its stability against highly noisy environment, bone conducted speech is highly advantageous however to mend the quality of restoration speech signal of bone conducted method, LP-Linear Perdition is used to improve the voice quality. The proposed linear perdition (LP) based blind restoration is completely distinguished and compared with various models by measuring their intelligibility of bone conducted speech with LSD, MCD and LCD, JWITs- Japanese word intelligibility tests and Vietnamese word intelligibility, tests proven better performance and lower degradation of signal in conduction of bone transduction [6].

According to Tsuge et al. [7], the speech data collected from the transducers for recognition through Bone conducted and Air conducted speech, by comparison proves that performance of AC is higher than BC speech. By combining the both AC Speech and BC Speech, similarities are calculated and noise will be eliminated certainly, results reduced equal error rate of BC speech by 71.7% and AC speech 16%.
According to Rahman et al. [8, 10], the speech intelligibility can be enhanced with no distortion is employed by improper noise estimation, BC speech is healthy to ambient noise by improving speech tainted low frequency band noise. Using time domain filtering the BC speech can be performed efficiently for noise destruction at lower signal to noise ratios. Regularly speech travel in two formants, speech directly to air from mouth, i.e. air conducted speech and another is bone conducted speech travel through vocal tract and skull bone, subjective by vocal tract shapes and pathways can be apprehended by bone conducted microphone on talkers head. The second formant have very sensitive amplitude with higher value than the second formant [13].

According to Minghi Li et al. [16], for clear speech formants in short time Fourier transform domain, geometric harmonics and Laplacian pyramid were utilized and analyzed by log spectral measure and spectrograms.

According to Rachel E. Bouserhal et al. [18], in order to improve the hearing and radio communication experience for workers in noisy environments requires high in signal to noise ratio with high bandwidth. By investigating the three uniquely signals received from three microphones placed at outer the ear, inside ear [occluded ear], and in front of mouth using Gaussian mixture approach low and high frequency signals varies at signal to noise ratio with extended bandwidth. According to Steven L. Grant et al. [15], by coupling bone vibrators and bone conduction microphones will enhances the transmitted signal to noise ratio at ambient air-born noise issues. Echo cancellation performed effectively and short by 10Ms in path is linear enough through bone conduction echo path. Double talk control is designed in echo path canceller and tests outperforms in quality.

According to Irwansyah et al. [23], the cross-talk should compensate when sound transfers from transducers, wave moves from either left ear or right ear unservingly to both ears of cochlea. It also increases the ability of sound sense in bone conduction headphone. This compensation filter can be assumed by otoacoustic emission (OAE), can be done by placing a microphone at outer ear canal, anti-sound produced by it using transfer functions results by filter-x least mean square algorithm reduces the prevised loudness.

3. Analysis. On reconstruction of speech signal BCS and ACS have different restrictions, ACS signals have high amplitude and contains outer noises. By the same time BCS are rigid to outer noises with low quality of retrieval signal. So most of the researchers concentrated on BCS signal processing techniques to enhance the signal by utilizing wavelets, deep neural networks and filters. Some researchers utilized ACS signal and BCS signal by combining and measured the retrieval signal strength. They found some crosstalk signal, need
to compensate, utilized OAE to enhance reconstructed signal.

Artificial neural networks used for improve speech de-no semantic in ACS speech and BCS speech by noise cancellers. Some study says voice transferring from vocal card should contains some vibrations occurred by skull, mouth and other may have impact on BCS. The speech wave travelling from mouth to outer air have two formants, longitudinal with lower amplitude and curve wave with slow travelling nature. For ACS, the signal to noise ratio should be high by increasing the band width utilizing gaussian mixture approach. ACS can be measured.

4. Conclusion. Firstly we investigated on various methods of speech enhancement techniques and their escalations caused due to current scenarios. On both BCS signal and ACS signal quality of the speech can be retrieved by removing noises, utilizing filters or cancellers from outer environments noises and inner vibrations. Combining the both signals have performed well than by measuring the individual Bone and Air conducted. BCS signal have high stability strength and rigid against ambient noise.

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A NOVEL APPROACH BASED ON MODIFIED CYCLE GENERATIVE ADVERSARIAL NETWORKS FOR IMAGE STEGANOGRAPHY

P.G. KUPPUSAMY *, K.C. RAMYA †, S. SHEEBA RANI †, M. SIVARAM ‡, AND VIGNESWARAN DHASARATHAN ¶∥∗∗

Abstract. Image steganography aims at hiding information in a cover medium in an imperceptible way. While traditional steganography methods used invisible inks and microdots, digital world started using images and video files for hiding the secret content in it. Steganalysis is a closely related field for detecting hidden information in these multimedia files. There are many steganography algorithms implemented and tested but most of them fail during Steganalysis. To overcome this issue, in this paper, we are proposing to use generative adversarial networks for image steganography which include discriminative models to identify steganography image during training stage and that helps us to reduce the error rate later during Steganalysis. The proposed modified cycle Generative Adversarial Networks (Mod Cycle GAN) algorithm is tested using the USC-SIPI database and the experimentation results were better when compared with the algorithms in the literature. Because the discriminator block evaluates the image authenticity, we could modify the embedding algorithm until the discriminator could not identify the change made and thereby increasing the robustness.

Key words: Generative Adversarial Network (GAN), PSNR, Steganalysis, LSB replacement, Cryptography, generator, discriminator.

AMS subject classifications. 94A60

1. Introduction. Steganography deals with embedding data in some cover medium. The content that is getting hidden should be not visible both perceptually as well as statistically. The embedding capacity also needs to be as high as possible with best robustness as well. Fig. 1.1 depicts a basic image steganography model.

The steganography process involves a cover image and a stego-key to hide the secret data in to the cover medium and produces a stego image as output. There are many algorithms discussed in the literature including those in spatial domain and transformed spectral domain. Though image steganography has many advantages compared to cryptography as they do not attract attention directly as in the case of the latter, they are still not very robust with the present day algorithms. There are many challenges including those through the Steganalysis methods that can identify the presence of secret data. This includes identification of visual attacks, statistical attacks and structural changes through different Steganalysis approaches. To overcome these issues, we propose a new methodology which includes inbuilt discriminator network to identify the above mentioned attacks during the initial stages of steganography process before producing the output stego image. Generative Adversarial Networks aims to generate data from zero and through reinforcement learning, the learning rate is higher. Since the generator alone can produce or create only some random noise, we need a mechanism to provide feedback to the generator. The discriminator portion of the GAN model will provide inputs to the generator model on what and how to create. So, when we have some secret data to be hided in a cover image, the discriminator will provide feedback to the generator on the best possible way to hide the data in an imperceptible way. We use real and generated images together for GAN’s to learn initially on what is real and

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what is fake as shown in Fig 1.2 and the proposed algorithm is tested using the USC-SIPI image dataset and the results are encouraging.

2. Related Work. Steganography, the method of concealing the secret data like an image or a text within another plain image in the most imperceptible manner possible. Modern steganography techniques exists as early as from 1985 by concealing messages in image files, chaffing and winnowing, pictures hided in video materials, changing the elements order in a set and so on. R. Chandramouli and N. Memon presented a study method that uses LSB technique for image steganography [1] through which large capacity of around 25% of the compressed data size were obtained at minimal degradation of image quality. J. Spaulding, M. N. Shirazi and E. Kawaguchi discussed about information hiding through various techniques [2]. They have discussed various applications of information hiding along with concealing techniques in this paper. Abbas Cheddad, Joan Condell, Kevin Curran and Paul Mc Kevitt also made a survey much later [3] and analysed the methods at that point of time. The common standards of image steganography and guidelines drawn from the literature are discussed in depths.

Wojciech Mazurczyk, Wendzel, Sebastian Zander, Houmansadr and Szczypiorski, detailed about information hiding in communication networks, the fundamentals, mechanisms, applications, and countermeasures [10] in their Wiley published IEEE book. This book covers the network steganography in detail along with traffic types, network flow and steganography counter measures. Prosanta Gope, Anil Kumar and Gaurav Luthra, have brought out the image steganography in compressed domain [4] with encryption technique. They have proposed an enhanced JPEG image steganography methodology added with a suitable encryption technique for achieving better results in data hiding mechanism.

Anjali and Umesh have again used wavelets for a secure skin tone based image steganography [5]. They describe on how Discrete Wavelet Transform (DWT) can outperform the Discrete Cosine Transform (DCT) when the secret message is hided on one of the high frequency DWT sub band. M.Vijay and V.VigneshKumar brought in image steganography method using integer wavelet transform [6]. They have used wavelets to hide secret data in the source image and found that higher embedding capacity with lower distortion rate can be
A Novel Approach Based on Modified Cycle Generative Adversarial Networks for Image Steganography

Deepika Kulshreshtha and Kamal Kumar in their research have used Discrete Cosine Transform for high capacity and lossless steganography [9]. Their results indicate that the mean square error almost reaches zero even after embedding the message on to the source image. Though there are multiple techniques discussed in the literature on both spatial and transformed domain, most of them fail during Steganalysis and the need for robust image steganography algorithm still remains. We started looking in the direction of techniques that has inbuilt Steganalysis methods to overcome these challenges and that made us to use Generative Adversarial Networks with inbuilt discriminative networks for robust image steganography.

Ian J. Goodfellow and his team first proposed a framework for generative model using an adversarial process [10] that includes both a generator network and discriminator network in it. They proposed a new framework for finding generative model through an adversarial process. They had two models defined in their architecture, a generative model for data distribution and a discriminative model for estimating the output data. Their results were outstanding when tested with various data sets. Their framework can also be further extended for a conditional generative model, learned approximate inference, semi-supervised learning and efficiency improvements. They have used MNIST data set for training adversarial nets and also the Toronto Face Database, and CIFAR.

Denis Volkonskiy, Ivan Nazarov, Boris Borisenko and Evgeny Burnaev have discussed about detection of payloads through binary classifier [11] and also proposed a new model based on deep convolutional generative adversarial networks. Their research work also showcased the viability of the adversarial framework and the challenges involved in generative modeling. Kevin Alex Zhang, Alfredo Cuesta-Infante, Lei Xu and Kalyan Veeramachaneni have used GANs for high capacity image steganography [18]. Their approach could achieve payloads of 4.4 bpp and is found to be efficient when tested on multiple data sets.

Ru Zhang, Shiqi Dong and Jiunyi Liu further used these Generative Adversarial Networks [15] for image steganography. They had three major contributions in their work including a) hiding data only on the luminance channel of the grey image, b) Presence of GAN architecture strengthen their security and c) construction of a mixed loss function which helps to generate more realistic stego images. Though they have proposed a robust method of image steganography, it can be applied only on stego images that are lossless otherwise the secret data portions gets lost. To overcome these short comings and to implement a robust algorithm, we proposed to use a modified cycle GAN approach which gives better results even in case of slightly lossy images.

3. Generative Adversarial Networks in image steganography. For hiding the secret data in images, we need to select the right set of images for training and testing purposes. There are many times where do not have sufficient data to create a model. Generative Adversarial Networks (GAN’s) can learn about the data and to synthesize or generate never before seen data to augmented data set. So, they can be considered as an unsupervised learning technique or semi-supervised learning.

While machine learning helps in the study of statistical models and algorithms that systems use to perform a function without explicit instructions, they require training before the task can be accomplished. Supervised, semi-supervised and unsupervised machine learning algorithms are the different types present in the literature while the GAN’s fall under semi-supervised learning methodology where the model uses a small amount of labelled data along with large quantity of unlabelled data for training.

Adversarial can be considered as a competition between two components including the generator and the discriminator. The generator replicates real data to produce fake data while the discriminator helps to distinguish between the real data and the fake generated data. The discriminator loss will then be passed as feedback to generator to produce a new set of data as an objective function. We make use of this principle in our image steganography technique to improve the security by minimizing the error between the stego images and natural images until the discriminator can no longer identify the hidden data.

GANs were first introduced by Ian Goodfellow and his team at the University of Montreal and their applications are huge as they can mimic any data distribution. Fig 3.1 below depicts the basic GAN:

The first step to train a GAN is to define the problem which in our case is to hide the secret data on the cover image. We define the GAN architecture next describing the tasks of the generator and the discriminator. The discriminator model then needs to be trained for classifying the real and fake data. The training here uses both the real images and the fake images produced by the generator and labelled as fake. We then train the
generator using the loss of the discriminator as the objective function. The generator parameters are fine-tuned on every iteration to maximize the loss of discriminator. The generator and discriminator training is continued in loop until the discriminator cannot classify between the real and fake data as shown in Fig 3.2. Once the training is complete, we synthesize data from generator and the system is ready for live testing.

The discriminator uses cross entropy loss as we are working on a classification model and also they give a better performance than misclassification rate and MSE. The predictions are also better with entropy loss as compared to other techniques. Learning also does not get stalled at any stage of the training with this method. The cross entropy loss between the true distribution $p$ and the estimated distribution $q$ is given by:

$$H(p, q) = - \sum p_i \log(q_i)$$  \hspace{1cm} (3.1)

The discriminator used here is just a binary classifier and the output can hence be either true or false. So equation (1) gets reduced to:

$$H(p, q) = -(1 - p_i) \log(1 - q_i) - p_i \log(q_i)$$  \hspace{1cm} (3.2)

Equation 3.2 represents a loss for single sample. The overall loss for $N$ samples can be computed as well. Considering that almost half our samples come from the real images while the other half are just imaginary or fake, we can write the predicted output of discriminator as follows:

$$E_x P_{\text{generator} \left[D(x)\right] = 0}$$  \hspace{1cm} (3.3)

$$E_x P_{\text{data} \left[1 - D(x)\right] = 0}$$  \hspace{1cm} (3.4)

There are many different types of GAN’s available including the basic GAN, deep convolutional GAN, conditional GAN, Info GAN, Wasserstein GAN, Attention GAN and cycle GAN. We discuss in detail about cycle GAN and our proposed algorithm in the coming sections.

4. Proposed Cycle GAN Method. Image to image translation aims at learning the mapping between an input source image and the output image using multiple training data sets. There are cases or applications where paired training data sets are not available. So, there is a need for an efficient approach to produce an image with data hided in it using a source image. Let the source image be represented by $X$ while the output stego image be represented by $Y$ and the goal here would be to map $G : X \rightarrow Y$ in such a way that images from $G(X)$ is indistinguishable from $Y$ using an adversarial technique. The transfer happens without one-to-one mapping between images from source to output in the training data set. Cycle GAN is capable of achieving
A Novel Approach Based on Modified Cycle Generative Adversarial Networks for Image Steganography

The need for pairing is eliminated due to the two step transformation: a) mapping it to the target domain and b) back to the source image. This mapping or hiding the data is performed using the generator. The quality of the newly generated image is then improved by using the discriminator network.

Both the generator and discriminator are trained together and over a period of time, the generator produces samples using the underlying distribution while the discriminator tries to guess randomly. It is a kind of graphics problem in which the goal is to learn the mapping of an input image to output. There are cases or exceptions during which the adversarial loss alone is not sufficient enough to confirm that the learned function can map an input to desired output. So, there is a necessity of consistency required and hence cycle-consistent GANs were required to overcome this issue.

Cycle GANs work by first taking an input image and fed to the generator to produce a new image which in turn is fed to another generator to make sure that the new image matches to the one in the original domain without much deviation. So, it contains 3 parts namely the encoder, the transformer and the decoder. The encoder takes care of feature extraction from the input image, the transformer in turn uses these features and combines them together while the decoder uses deconvolution methods for reverse encoding.

If the discriminator outputs $D(x)$ predicting the chance of $x$ being a real image, then the objective is maximum likelihood of the data observed. This objective is written as follows:

$$\max_D V(D) = \mathbb{E}_{x \sim p_{data}} \log D(x) + \mathbb{E}_{z \sim p(z)} \log (1 - D(G(z)))$$

Similarly the objective function on the generator side is defined as follows:

$$\min_G V(G) = \mathbb{E}_{z \sim p(z)} \log(1 - D(G(z)))$$

Here $G$ needs to be optimized to the extent that can fool the discriminator.

4.1. Proposed Algorithm (Modified cycle GAN for data hiding and producing stego image).

The USC-SIPI dataset which is a collection of digitized images for supporting research in image analysis, segmentation, processing and machine vision is used for image steganography as well in our work. Most of the images in the data set used for training are color images. In order to embed the secret data in to the image set, we prefer to use only the luminance part of the image. For this purpose, we first convert the RGB images into grey image and pull out the luma part from it as follows:

$$\begin{bmatrix} Y \\ Cb \\ Cr \end{bmatrix} = \begin{bmatrix} 0.256 & 0.504 & 0.098 \\ -0.148 & -0.292 & 0.441 \\ 0.441 & -0.369 & -0.071 \end{bmatrix} \begin{bmatrix} R \\ G \\ B \end{bmatrix} + \begin{bmatrix} 16 \\ 128 \\ 128 \end{bmatrix}$$

Here $Y$ represents the luminance information of the image while $Cb$ and $Cr$ represent the color. We embed the secret bits in the LSB’s of this $Y$ plane and not on the color data as color distortions are human perceptible easily while brightness changes are less visible. Also, the other bits in the pixel are adjusted during every iteration of the discriminator output in order to produce less error thereby at some point of time the discriminator will not be able to identify the secret data present in the source image.

The hiding capacity of this modified cycle consistent GAN algorithm depends on the number of training data set and the learning rate. The number of bits that gets adjusted in the identified pixel values also improves the proposed techniques embedding capacity. The steps are listed as follows:

1. The color images are converted into grey scale images and the LSB bits of the $Y$ plane pixel are substituted with the secret bits.
2. The newly generated image is fed as input to the discriminator along with the original images.
3. The discriminator takes both the sets of inputs then returns the output as 0 or 1.
4. The generator and discriminator neural networks run in competition during the training phase in loop. The discriminator is first trained (forward propagated) when the generator is idle and vice versa.
5. During every iteration, the bits that are present in the $Y$ plane before the hidden bits are adjusted according to the feedback from the discriminator.
If $d$ is the pixel value, then $d_1$ represents the value of last $n$ bits while $d_2$ shows value of hidden bits. If $(d_1 \sim d_2) \leq (2^n)/2$, then no adjustments are required for that pixel else it requires modification during every iteration. The data hiding process using this modified cycle GAN algorithm is depicted in Fig 4.1.

Cross entropy is the discriminator loss function. One of the frequently encountered problem during training is the gradient saturation problem where the gradients are very small for any type of learning. A poorly designed generator loss function can lead to this issue and we have addressed this problem in our research work. The generators as seen in fig 5 above does the density estimation from the stego data to real data and then provides it as input to the discriminator. We wanted the discriminator to maximize the log probability in prediction and we achieve this by adjusting the MSB pixel values along with LSB values based on the class output from the discriminator. Due to this additional substitution, the pixels gets normalized and the noise is not observed either visually or through any steg analysis algorithms. Cycle GAN learns to perform translations of images without the need for explicit pairs. In other words, there is no need for one-to-one image pairs. Once the training is completed, the generator present in the GAN model should be able to produce stego images that the discriminator or any steg-analysis algorithm will not be able to identify. Through this cycle GAN approach, we are able to define two types of loss:

1. Adversarial loss;
2. Cycle-consistency loss.

The adversarial loss matches the distribution of stego images and the ones in the target domain. The cycle consistent loss adds stability to the training process and helps in producing a more robust system as compared to the traditional GAN networks. The proposed bit manipulation technology overall improves the embedding capacity and the robustness.

5. Results and discussions. Steganography is the science that works towards transmitting the secret data with the help of a suitable carrier normally a multimedia transporter like image, audio or video files. In our work as detailed in the previous sections, we have used image carrier to hide the secret data using GAN algorithm. In order to train and test the proposed modified cycle GAN algorithm, we have used the USC-SIPI database provided by the signal and image processing institute of the Southern California University. The data sets were first published in 1977 and are updated on a periodic basis. The image details and the results are detailed in the coming sections.

The dataset is organized across various types based on the characteristics of the image. They are of type having different textures, aerials, miscellaneous and sequences. The user has to identify the images for reading and access them within programs. The image size in the database varies between 256 * 256 to 512 * 512 and 1024 * 1024. The images present are either color images or grey images. Some of them are black and white images and are 8 bit depth while color images are of 24 bit depth. Fig 5.1 gives the glimpse of the same. The data set used in this work is originally present in TIFF format.
As a first step of our programming, we have to first uncompress the TIFF image. A sample c program is also available for this purpose along with the data set. But we have used MATLAB which is a high level programming language and simple to use for machine learning and image processing applications. While the grey images can be used as such without any conversion, in case of RGB color images, we convert the color images into grey scale image as shown in figure 5.2 below before proceeding with data hiding techniques using the proposed algorithm. We then select the source image and the secret data for training and image steganography.

We evaluate our proposed GAN method based image steganography through secrecy (difficulty in identifying the stego images), capacity (size of hiding message), and robustness (PSNR comparison). In terms of security, the methods discussed in the literature uses GAN to learn a modification probability matrix have higher visual effects because they depend on the original carrier and the output image quality decides the security. In our method, the key space is too small and the quality does not deter much. In terms of capacity, there is a big gap between the cover synthesis steganography and traditional cover modification steganography.

We embed the secret bits into the source image and provide it as input to the discriminator to get its feedback. Based on the feedback, the other bits in every pixel of the y plane is adjusted and provided as input again to the discriminator. The generator loss function using different methods are compared in fig 5.3. The discriminator takes both the source image and the stego image and tries to classify which is real and which has hidden data in it. So, there is a double feedback loop and this framework hence resists all types
of steganalysis attacks. 10% of the data set is considered for Steganalysis purposes. The training set from the database is denoted by \( A \), the testing set by \( B \) and the proposed steganography algorithm by \( \text{stego}(x) \). We embed secret information to the data set and the training set becomes \( A + \text{stego}(A) \) while the testing set becomes \( B + \text{stego}(B) \). The accuracy of the steganalysis trained using real data set is represented in Table 5.1.

Both SGAN and cyclic GAN are capable of reducing the detection accuracy of any Steganalysis method but our proposed modified cyclic GAN algorithm is capable of giving even more robust results and with a better embedding capacity. Table 5.2 shows the capacities of various non-modification methods and its seen that the proposed method is better here as well. Table 5.3 represents the PSNR values observed across different image steganography algorithms. Even this indicates that the proposed algorithm presents state-of-the-art results as compared with the other techniques in the literature.

Fig 5.4 below represents the Steganalysis detection values comparison across different methods. It is evident that with the proposed method, the chances of stego image detection is almost impossible whereas with the regular LSB substitution method it is quite easy to identify. Also Figure 8 shows the different generator loss functions where minimax outweighs the other two methods.

It is always possible for the distributions derived from training to overlap as shown in Fig 5.5. In such scenarios, the feature space decision boundaries that are proportional to the covariance presented by each class take a quadratic form and decide the output class. So, our proposed classification method has less chances of failing as compared to other methods.
A Novel Approach Based on Modified Cycle Generative Adversarial Networks for Image Steganography

Table 5.3

<table>
<thead>
<tr>
<th>Different Methods in Image Steganography</th>
<th>PSNR value in dB across different images</th>
</tr>
</thead>
<tbody>
<tr>
<td>DCT</td>
<td>Lena 47.11  Barbara 46.9  Boat 46.73</td>
</tr>
<tr>
<td>LSB Technique</td>
<td>Lena 53.54  Barbara 53.65  Boat 52.96</td>
</tr>
<tr>
<td>Regular DWT</td>
<td>Lena 46.09  Barbara 46.43  Boat 47.72</td>
</tr>
<tr>
<td>Proposed Method</td>
<td>Lena 64.7  Barbara 58.3  Boat 60.1</td>
</tr>
</tbody>
</table>

Fig. 5.4. Steganalysis detection values comparison across different methods

Fig. 5.5. Discriminator Output classifications

6. Conclusion and future directions. Steganography and also cryptography aims at protecting information from unwanted parties but both the methods have their issues to be resolved. We have studied about various existing image steganographic algorithm present in the literature and came up with a new approach of modified cycle GAN for robust image steganography. The PSNR values witnessed across different techniques illustrates undoubtedly the benefits of the proposed method. Moreover, the stego images produced through the proposed algorithm has got better visual quality as compared to the other methods thereby making the stego image not to be discovered easily. We have also tested our algorithm on the MNIST database [13] containing over 60,000 examples for training and 10,000 samples for testing. The classifier model produced great accuracy
both on the real images as well as on the steganographic images produced by the proposed method. There are both merits and demerits to the proposed framework. Though the system has improved payload capacity, better hiding technique and integrity check for issues encountered in LSB method, the system cannot be overly trained and that can lead to the Helvetica scenario which corresponds to mode collapse in case of generative adversarial networks. Also, the proposed algorithm can be optimized for even better and improved results in terms of capacity and robustness. We aim to address these issues in our next research paper.

7. Acknowledgement. The authors also would like to express their sincere thanks to Prof. Dr. Truong Kham Nguyen, Division of Computational Physics, Institute for Computational Science, Faculty of Electrical and Electronics Engineering, Ho Chi Minh University, Ho Chi Minh City, Vietnam for giving his valuable suggestion, comments and support to complete this work as effective.

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MINIMIZING DEADLINE MISSES AND TOTAL RUN-TIME WITH LOAD BALANCING FOR A CONNECTED CAR SYSTEMS IN FOG COMPUTING

K. JAI RAM NAIK∗ AND D. HANUMANTH NAIK†

Abstract. Cloud computing helps in providing the applications with a few number of resources that are used to unload the tasks. But there are certain applications like coordinated lane change assistance which are helpful in cars that connects to internet has strict time constraints, and it may not be possible to get the job done just by unloading the tasks to the cloud. Fog computing helps in reducing the latency i.e the computation is now done in local fog servers instead of remote datacentres and these fog servers are connected to the nearby distance to clients. To achieve better timing performance in fog computing load balancing in these fog servers is to be performed in an efficient manner.

The challenges in the proposed application includes the number of tasks are high, client mobility and heterogeneous nature of fog servers. We use mobility patterns of connected cars and load balancing is done periodically among fog servers. The task model presented here in this paper solves scheduling problem and this is done at the server level and not on the device level. And at last, we present an optimization problem formulation for balancing the load and for reducing the misses in deadline, also the time required for running the task in these cars will be minimized with the help of fog computing. It also performs better than some common algorithms such as active monitoring, weighted round robin and throttled load balancer.

Key words: fog computing, connected car, offloading, mobility prediction, task set, load balancing, deadline miss, link rate, optimization

AMS subject classifications. 68M14, 90C26

1. Introduction. In cloud computing the distance which is present in between the datacentres and the clients is high as compared to that of fog computing. It is difficult to run some time sensitive applications in cloud as the latency that cloud computing posseses is very high [1]. Fog computing processes the applications closer to the users within the proximity distance [2]. Cloud computing servers are placed in the datacentre which are centralized whereas in fog computing, the servers are distributed to the edge. Fog computing is a distributed computing paradigm that provides the users with cloud services at the edge of a network. This helps in the reduction of traffic on the backbone of network.

But there are certain issues in fog computing while considering the deadline misses and runtime which includes resource allocation and load balancing [3]. To explore the problems in fog computing such as load balancing, we here use the motivating application such as connected cars. Fog computing benefits the connected cars as coordinated lane change assistance in connected cars uses the local fog servers for data processing and responds quickly [4, 5, 6].

Load balancing in fog computing is performed among the fog servers to reduce deadline misses as well as runtime in the applications like connected cars. This helps in improving the performance of connected cars by providing with the accurate results. While load balancing is an issue to be performed with the fog servers, we here in this work develop a task model as well as optimization problem formulation for dealing with issues like load balancing and resource allocation. The result obtained with this technique outperforms common scheduling algorithms.

Fog computing characteristics are the followings:

1. The nature of the fog servers are heterogeneous that is each fog server will be having different connectivity and computation capability.

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2. The scale of the fog computing systems is relatively small as compared to a datacentre. This helps in more precise approaches in dealing with the applications.

Here in fog computing the client’s position is not static and mobility of the client is also a feature of fog computing.

Connected car is a car that has internet access, and also equipped with wireless local area network. This helps the car and provides with internet access, and data can be shared, with other devices both in and outside of the vehicle. These connected cars ensure the safety in cars. They help in reducing the accidents.

The relationship between the server and the clients in fog computing depends on their instantaneous locations. The information of client and their mobility is used to perform load balancing. And load balancing is done by predicting the travel patterns of the car and allocating the resources that fits the server to the best and this helps us to reduce processing time and deadline misses can also be abandoned.

We first develop the algorithm for mobility prediction and propose a model that is the resemblance of task model and finally we implement optimization problem for load balancing that is useful in connected cars. The scheduling is done at the server level. In the task model that is proposed we develop an algorithm that requires no knowledge of how the tasks are being scheduled earlier. It is tedious process to obtain the information regarding the scheduling of each task because the traffic present in the network increases more rapidly and the problem size increases. The final step in the workflow diagram is we perform comparison of our proposed work with the common scheduling algorithms like throttled load balancer. Here we analyse the performance of the optimization problem.

The work scope is high in the connected as well as the autonomous vehicles. It helps in connecting the mobile devices with the vehicles and providing numerous advantage for a user. Fog computing has many applications like traffic control system, health care systems, machinery related systems, video streaming system, smart city and smart home.

Further development is in the progress related to this fog computing.

In this work we will follow the flow of the given solution in the particular order. We will first review existing work that is already there in the load balancing for mobile clients and is developed with the help of fog computing. This is presented in the Literature review section 2.

The proposed system is developed is described in Section 3. It consists of Linear mobility prediction algorithm and it gives the accurate results and shows that online task scheduling can benefit the end user. In this chapter we also present our task model as well as the optimization problem formulation.

In Section 4 result is analysed with the common scheduling algorithms that is throttled load balancer. Conclusions and Future scope are being discussed in Section 5.

2. Related Work. The work proposed by Chen et al [7] is related to connected cars which has numerous advantages with the help of fog computing. Here, a prediction algorithm is proposed which locates a car location and load balancing is done at server level to minimize deadline misses and runtime.
Minimizing Deadline Misses and Total Run-time with Load Balancing for a Connected Car Systems in Fog Computing

The work presented by Oueis et al [8] is that, constructing two methods which helps in optimizing and in minimizing total transmitting power and total time required for computation. The tasks here are partitioned and distributed among the servers and they are hard deadline constrained, but in the work that we implement further is constrained to soft deadlines and are indivisible.

Zeng et al [9] work is related to the formulation of linear program and it helps in minimizing the average task finishing time but, in our work, our target is the avoidance of deadline misses. Hong et al [10] proposes a model which takes workload for dynamic scaling of the fog system to provide them with available number of resources. In our work we impose deadline constraints and placing of only one task for a fog servers. This is the difference between their and our work.

Takayuki et al [11] uses the routing of smart cars as an application and offloads them to the local servers. Here the energy constraints are taken into account and formulation for implementing optimizing problem is performed. We impose the number of tasks that fit to the server and also constraints on resources, while in the work of Takayuki this is not considered.

Hong et al [12] uses the patterns of the cars and predicts its location to which it is travelling and forward the tasks to local server the fog server that is in the range of obtained location so that it processes the tasks without delay. In this work the focus is only on single client and resource and capacity constraints are not imposed whereas in our work the system is multiclient and it has various constraints on the servers.

Wang et al [13] uses facial recognising system that runs on the handheld devices. They treat the tasks of these devices as a graph and then solves which particular element is matching with other. The difference in their work and our work is that we impose constraints on time and it is multiuser model while their work is not.

Li et al [14] solves the problem of partitioning the tasks for remote cloud and local servers and then allocation of resources based on the information obtained on partitioning. Their work includes optimization formulation for minimizing the finishing time, processing cost and bandwidth. But the disadvantage of their work is they do not include deadline misses.

3. Predicting the Pattern of Mobility.

3.1. Model for connected car. Most of work on predicting the movement as well as direction in vehicles are done using the methods Markov Chain [15] or probability distributions [16]. Here we present a simple linear model which gives high accuracy in predicting their locations.

The example model that we present is as given in Fig 3.1. Hexagon represents wireless coverage area of fog server; Rectangle represents fog server; and Circles represents connected cars.

In the model presented above there are three different fog servers and the connected cars which are shown as circle are counted as clients. The fog servers manage the cars to the range upto which it can handle in wireless connection. Here we apply 802.11p [18] protocol in between the tasks and servers, Wireless Access in Vehicular Environments (WAVE) [17].The range is up to 500 metres. The hexagon shape is applicable in implementation of system of network in wireless environment [19].
Table 2.1
Tabular representation of existing works

<table>
<thead>
<tr>
<th>Authors</th>
<th>Work proposed</th>
<th>Advantages</th>
<th>Limitations</th>
</tr>
</thead>
<tbody>
<tr>
<td>Oueis et al.</td>
<td>Optimization problem</td>
<td>Minimize total transmitting power and computation time</td>
<td>Tasks are partitioned and constrained to hard deadlines</td>
</tr>
<tr>
<td>Hong et al.</td>
<td>Programming model for dynamic scaling</td>
<td>Scaling fog devices to provide enough resources</td>
<td>Deadline constraints are not considered</td>
</tr>
<tr>
<td>Takayuki et al.</td>
<td>Routing for smart cars</td>
<td>Minimizes aggregated task finishing time</td>
<td>Capacity and resource constraint on server is not included</td>
</tr>
<tr>
<td>Hong et al.</td>
<td>Car’s mobility pattern to predict future location</td>
<td>Processing speed is increased</td>
<td>System focuses on only a single client</td>
</tr>
<tr>
<td>Li et al.</td>
<td>Allocation of resources based on partitioning</td>
<td>Minimizes finishing time and cost of processing</td>
<td>System does not allow any deadline misses</td>
</tr>
</tbody>
</table>

3.2. Prediction for pattern of linear mobility. Here the algorithm for predicting linear mobility is proposed that identifies the cars future location with the help of its previous location. We assume that a car updates its location for theta seconds and if the updating of cars location at a particular time \( t \), the fog server gets the timestamp and cars previous location to calculate the direction and the speed of the car. So, the updated location of this system will be its position at timestamp \( t + \theta \). If the location obtained is beyond the range of fog server, then the algorithm will identify which nearest server that is closest to the predicted location of the car.

The GPS data of taxis in Rome, Italy [20] is compared with the algorithm we implemented. We divide the region into seven fog nodes with each node consisting of radius 500 metres. With the help of this prediction algorithm we notify that to which of the six servers the car is heading towards leaving the middle server. We get an accuracy of 94.90%.
Here in this algorithm we see that after calculating the speed and the direction of the car we predict that to which server location the car is travelling to and we allocate the resources prior to its arrival.

In Figure 3.4 the lines given in blue represent correct prediction and the lines in red colour are the incorrect prediction. So this is the result obtained considering the GPS data of taxis.

The obtained accuracy tells us that how the prediction algorithm we implemented is giving the results. It is the simpler algorithm as mentioned with markov chain and probabilistic distributions.

4. Load Balancing.

4.1. Model for the tasks in the system. In applications like coordinated lane change assistance, the data must be ready for the car when it is connected to the new fog server. With the help of mobility prediction algorithm, we try to figure out to which server the car is travelling and perform load balancing before the car reaches the location to reduce the runtime.

There are fog server, cars and remote cloud server present in this task model. And there is interconnection between them. In task model there are three local fog servers and one cloud server. Cars are connected with the local fog servers that are in the specific range given by the hexagonal region.

The workload of a car is treated as a single task. These tasks are either processed by the servers or passed to nearest servers with high capability.

In the task model (Fig 4.1) five tasks are assigned to server A, there are four tasks in server B and three tasks in server C and cloud server is initially passed with no tasks.
Fig. 4.2. Task model in fog computing graphically

Table 4.1

<table>
<thead>
<tr>
<th>Variables used inbetween servers and tasks</th>
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<tbody>
<tr>
<td>( m )</td>
</tr>
<tr>
<td>( K_i )</td>
</tr>
<tr>
<td>( L_{i,j} )</td>
</tr>
<tr>
<td>( P_j )</td>
</tr>
<tr>
<td>( y )</td>
</tr>
<tr>
<td>( n_j )</td>
</tr>
<tr>
<td>( D )</td>
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<td>( t )</td>
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<table>
<thead>
<tr>
<th>Optimizing variables</th>
</tr>
</thead>
<tbody>
<tr>
<td>( k_{i,j} )</td>
</tr>
</tbody>
</table>

In the model shown in graphical format in Figure 4.1:

\( k_{i,j} = \text{Total task distribution from one server to another,} \)

\( L_{i,j} = \text{Link rate from server \( i \) to server \( j \)} \)

The local servers in the task model are interconnected and these servers can distribute the tasks among the other servers or the remote cloud. This implies that the servers in the task model can offload as well as accept the tasks from other servers. All the servers are allotted with various constraints and resources.

In Figure 4.2, server \( i \) offloads its tasks to the accepting server \( j \). These tasks are further processed by the server \( j \).

The fog nodes are placed in between the inbetween the cloud and the edge devices. The fog servers are placed as in the Figure 4.2 and these servers are static. So the total number of servers here in the system is 8 which includes the cloud server.

Table 4.1 shows all the variables that are being used in this model.

From the above table \( m \) represents the servers count in the model, \( K_i \) is tasks initially assigned to server \( i \), \( L_{i,j} \) speed of data transmission present inbetween the server \( i \) and \( j \) (tasks/sec).

Consider the task model example in which \( m = 4 \), and \( KA = 5, KB = 4, KC = 3 \) and \( Kcloud = 0 \), \( P_j \) is the capacity of server \( j \), \( y \) is cycles number required for CPU that consume bandwidth equally, \( n_j \), frequency of CPU at the accepting server. \( D \) is total tasks deadline and \( t \) is the time for processing a server. Here \( t \) is required because load balancing can be done based on prediction algorithm in which periodic calculation is
done. Each time when the prediction is done load balancing is to be done for sure. The condition here is that time taken to process the total tasks is to be less than the time that will remain that is $D - t$, and this time is helpful for implementing the prediction algorithm and balancing the load. $k_{i,j}$ is the optimization variable that is to be solved and it tells that total tasks distribution number from one server to another.

The drawback of this model is it is difficult to predict the scheduling of tasks by the server that are being sent by different servers and their arriving pattern of these tasks is also bit complex. So, it is difficult in calculating the complete time the task is executed. The solution is to balance the load not at device level but by making the tasks as one large aggregated task it can be resolved. By doing this it becomes easy to calculate the time taken to complete one aggregate task at each server in contrast with how the server is scheduling its tasks. Figure 4.3 shows the aggregated task example.

**Aggregate task transmission time at server $j$:**

$$
Turnaround time: \sum_{i=1}^{m} \frac{k_{ij} \cdot y}{n_j}
$$

(4.1)

Time taken for offloading server $i$ for transmitting the tasks.

$(k_{ij})$ to accepting server is division between $k_{i,j}$ and the link rate $(L_{ij})$. The total time taken for all the combination of tasks at the accepting server is total sum of time required for transmission. The equation is given in equation (4.1).

**Aggregate task turnaround time at server $j$:**

$$
Turnaround time: \sum_{i=1}^{m} \frac{k_{ij} \cdot y}{n_j}
$$

(4.2)

The processing time required for the accepting server $j$ for all the tasks that are being shared by other servers.

**Aggregate task completion time at server $j$:**

$$
Completion time: \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \sum_{i=1}^{m} \frac{k_{ij} \cdot y}{n_j}
$$

(4.3)

Completion time is known as the time taken for server $j$ to complete its aggregate tasks. It is the addition of turnaround time and transmission time.

**Aggregate task lateness server $j$:**

$$
Lateness at server j: \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \sum_{i=1}^{m} \frac{k_{ij} \cdot y}{n_j} \right) - D
$$

(4.4)

Lateness is calculated by subtracting the deadline of a task from its completion time.
4.2. Problem formulation. Here in the problem of load balancing the final result is minimizing the misses in deadline and runtime. The objective function is

\[
\text{Min} : \sum_{j=1}^{m} \left[ \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D + v \cdot \max \left( 0, \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D \right) \right]
\]  

(4.5)

The goal for optimizing is to minimize total runtime and deadline misses. Minimizing the term lateness of completing the aggregated tasks is reflected in minimizing the total runtime in the objective function.

If we minimize the second term \((v \cdot \max \left( 0, \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D \right))\) then it minimizes the misses in deadline. \(v\) is the weighing factors so that more emphasis is put to minimize the deadline misses and runtime. From the objective function above the term \(v \cdot \max \left( 0, \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D \right)\) is taken as \(v \cdot x_j\).

Therefore, the objective function now becomes:

\[
\text{Min} : \sum_{j=1}^{m} \left[ \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D + v \cdot x_j \right]
\]  

(4.6)

The constraints subjected to the above given objective function are:

\[
\sum_{i=1}^{m} k_{ij} \cdot y \leq n_j \cdot t \quad \text{for } j \in \{1, 2, \ldots m\}
\]  

(4.7)

\[
\sum_{i=1}^{m} k_{ij} \leq P_j \quad \text{for } j \in \{1, 2, \ldots m\}
\]  

(4.8)

\[
\sum_{j=1}^{m} k_{ij} = K_i \quad \text{for } i \in \{1, 2, \ldots m\}
\]  

(4.9)

\[
k_{ij} \in Z^+ \quad \text{for } i \in \{1, 2, \ldots m\}, \text{ for } j \in \{1, 2, \ldots m\}
\]  

(4.10)

\[
x_j \geq 0 \quad \text{for } j \in \{1, 2, \ldots m\}
\]  

(4.11)

\[
x_j \geq \left( \sum_{i=1}^{m} \frac{k_{ij}}{L_{ij}} + \frac{k_{ij} \cdot y}{n_j} \right) - D \quad \text{for } j \in \{1, 2, \ldots m\}
\]  

(4.12)

Equation (4.7) states that there are there sufficient number of cycles for CPU are present for offloading tasks. Constraint Equation (4.8) is about that all tasks loaded to server will be not exceeding its capability to hold the tasks. Equation (4.9) will make sure that the tasks are distributed and are processed. [Eq-10]states that the tasks are not divided to subtasks. Here in this methodology we will not be partitioning the tasks i.e the tasks will remain itself and not gets divided. We solve the optimization function using lp_solve.

We have implemented the above given constraints and objective function with the help of IBM ILOG CPLEX optimization studio. We solve for \(k_{ij}\) that is the task distribution number from one server that offloads to another server that accepts.
5. Results and discussion.

5.1. Experimental results. The simulation in this experiment adopts a system that consists of a cloud server and the fog servers are seven. The simulation parameter values are given in the table 5.1.

The taskset generation is randomly done by selecting number for Ki from a range of integers. Different loadings can be created for the system by varying the integers. Optimizing principle is applied to every taskset that is created newly.

The experiments performed are implemented on machine that supports AMD operation with quadcore of dual nature that has the specification of 2.3 GHz and 16GB memory. The time taken to execute both optimizing principle and predicting algorithm on this machine is below 0.01 seconds. It satisfies the remaining time \( D - t \) which has a value of 0.02 and here the experiments are run below that time which is in the range of overhead budget.

5.2. Comparison between deadline misses and total runtime. Here we vary the values of weighing parameter. In the objective function from equation (4.6) a higher \( v \) value results in lesser misses in deadline with high runtime.

The result obtained by varying \( v \) is shown in below graphs. From the below graphs we see that as the value of \( v \) increases, the deadline misses count decreases and the total runtime increases.

From the above equation (4.6) as the value of \( v \) (weighing parameter) increases then the term total runtime value gets reduced and the value which represents penalty in tardiness i.e. the deadline misses value gets reduced.

5.3. Comparison with throttled load balancing algorithm. We perform comparison of our optimization result with commonly used load balancing algorithm throttled load balancer [23].

In throttled load balancing the task is processed by the server which is best suitable, the server with lowest

<table>
<thead>
<tr>
<th>Table 5.1</th>
<th>System parameters for simulation</th>
</tr>
</thead>
<tbody>
<tr>
<td>( m )</td>
<td>Total servers</td>
</tr>
<tr>
<td>( y )</td>
<td>CPU cycles of a task (M cycles)</td>
</tr>
<tr>
<td>( D )</td>
<td>Total tasks deadline (sec)</td>
</tr>
<tr>
<td>( t )</td>
<td>Time to process a task at each server (sec)</td>
</tr>
<tr>
<td>( K_i )</td>
<td>Tasks initially assigned for a server</td>
</tr>
<tr>
<td>( L_{i,j} )</td>
<td>Speed of transmission of tasks in between the servers, for ( i \neq j ) (tasks/sec)</td>
</tr>
<tr>
<td>( L_{i,j} )</td>
<td>Speed of transmission of tasks in between the servers, for ( i = j ) (tasks/sec)</td>
</tr>
<tr>
<td>( P_j )</td>
<td>server ( j ) capability to hold the tasks (tasks)</td>
</tr>
<tr>
<td>( n_j )</td>
<td>CPU frequency (MHz)</td>
</tr>
<tr>
<td>( K_{cloud} )</td>
<td>Initial tasks of cloud</td>
</tr>
<tr>
<td>( L_{cloud} )</td>
<td>Link rate present in between the server and cloud</td>
</tr>
<tr>
<td>( P_{cloud} )</td>
<td>Capacity of cloud</td>
</tr>
<tr>
<td>( n_{cloud} )</td>
<td>Cloud server frequency of CPU</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Table 5.2</th>
<th>Workloads</th>
</tr>
</thead>
<tbody>
<tr>
<td>Range for ( Ki )</td>
<td>[10,26] [10,28] [10,30] [10,32] [10,34]</td>
</tr>
<tr>
<td>Total tasks</td>
<td>12687 13472 14143 14739 15051</td>
</tr>
</tbody>
</table>
completion time is assigned with the new task that did not reach its maximum capacity. The cloud server is used only when the local servers are not available. Our optimization outperforms throttled load balancer by almost 50% considering deadline misses as a parameter.

As the value of the number of tasks increases then the performance percentage of optimized as compared to throttled load balancer get reduced to 25%. The reason for this is that if the load of a system increases then the options get reduced for task distribution. So, the performance of optimized algorithm outperforms 25% which is reduced half as the number of tasks increases. But the optimized performs better than the common scheduling algorithm.

In Figure 5.4 the locations of the taxi drivers are plotted and with the given datasets [20] and verified if
the user is travelling to the predicted server or not.

6. Conclusion and future work. We performed scheduling model in fog for the connected cars. This model schedules the tasks at server level instead of device level. The calculation of the completion time of a task at device level will be a stochastic process as it is difficult when the tasks are high in number and the arrival order of the tasks is also difficult to predict. So, we combine the tasks into one aggregated task at server level and perform load balancing.

Optimization load balancing formulation is also discussed for reducing the total runtime and minimizing the deadline misses. We showed that our optimization performs better than common scheduling algorithms like throttled load balancer.

Fog and edge computing are the emerging techniques for data processing in IOT. It also helps in reducing latency and network congestion thus reducing the traffic at the backend cloud so that the processing speedup is done.

There is a scope to make the cars more updated as well as other application which can benefit from fog computing.

Fog computing is expanding to provide services in connected cars. There is high potential in the work related to fog computing and the technology being implemented in connected cars.

For simulation purposes we use fogsim which is inbuilt with libraries and basic functionalities to execute the applications which helps users with numerous advantages.

Fog computing has many applications like Traffic control system, Health care systems machinery related systems, video streaming system, smart city and smart home.

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LONG AND STRONG SECURITY USING REPUTATION AND ECC FOR CLOUD ASSISTED WIRELESS SENSOR NETWORKS

ANTONY JOSEPH RAJAN D * AND NAGANATHAN E R †

Abstract. Wireless sensor network plays a significant role in the construction of smart cities, and the social network includes the Internet of Things, etc. In general, networks are most vulnerable of all the wireless devices due to the massive damage caused by disrupting these networks. Hence the nodes present in the network should get validated for its reputation. Therefore a Long and Strong Security mechanism with two-level checks is proposed here. Level 1 check includes verifying node reputation value and level 2 check includes Elliptical curve cryptography (ECC). Each sensor node sends the public master key to the cloud and secretly stored in the sensor node. Before data transmission, every node checks the master key, and if the master key is a match, then it transmits the data to the next hop. This process is continued until the source reaches the destination in the network.

Key words: Node Reputation, Grade Factor, Elliptical Curve Cryptography, Wireless Sensor Network.

AMS subject classifications. 94C12

1. Introduction. Many ways are available for implementing security measures, and among them, the most commonly used technique is cryptography. The cryptographic methods usually have three operations to be carried out for providing security. The activities include conversion of simple passage to cipher (secret code) text, identifying the value of secret key used and processing the algorithm of an understandable or straightforward transcript [17].

From the key variations, bi-cryptosystem are generated, namely symmetric key cryptosystem (private) and asymmetric key cryptosystem (public). If a similar key is used for both decrypting and encrypting the text or communication, then it is said to symmetric key cryptosystem. If a pair of keys such as private and public key utilized for decrypting and encrypting process, then this process is said to be asymmetric key cryptosystem. There were several scheme and algorithms projected to public-key cryptography as its initiation. Some example techniques for public-key cryptography are Diffie Hellman, Secured hash algorithm, RSA, ECC, etc. Prime numbers are used to generate keys for securely transferring data [2]. The public key is utilized for encrypting simple transcript or to confirm a digital mark, and the private key is utilized for decrypting secret code transcript or to generate a digital mark. Prime based ECC key generation is implemented for providing security measures and Further sections of this paper are organized in the following way. Section 2 gives a detailed description of existing security algorithms. CC based Secure Data Encryption scheme analysis is explained in detail. In section 4 the results are analyzed based on the presentation projected and conventional systems. In the final section, the conclusion of the work is included with the applications.

2. Related Works. Most of the works have been carried out for providing security in the field of wireless communication using symmetric and asymmetric key cryptosystems. The security issues are reduced in maximum by using the existing security mechanisms. Some of the secured mechanisms are discussed here in detail as follows;

The packets are forwarded to the node which is located away from and attack other nodes that are present in the network. This attacker node itself present in the network to attack the other nodes by assuming itself as a node that is present in the shortest path. To protect the network from malicious nodes i.e. node replication attack the secured transmission and communication approach [3] was proposed.

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Localized Encryption and Authentication Protocol [5] is a Key organization protocol in which four various keys are generated such as personal key, pair-wise key, cluster key and set, each key was created for a different type of purposes. The keys are established and updated in a timely manner among each other so that the involvement of base station reduced, in turn, energy consumption got reduced.

Elliptic Curve based asymmetric cryptosystem was implemented [4] by Muhammad Hammad Ahmed et al., in which ECC is used for its smaller key size. This greatly reduces the computational cost. The elliptic curve field was defined over the prime number, so that complexity in mathematical computations gets diminished. The maximum key length used for ECC is 384 for providing security. The double-and-add algorithm is used for point multiplication, and computation delays were compensated using 32 block size. A secure node authentication scheme [6] was implemented for protecting the data from unsecured routing. The sensor nodes collect data and shares over the communication link that shared data should be accessed only when the nodes are authenticated. The BS programs the signatures with a certificate authority, and once the nodes are verified that they are legitimate nodes, then the nodes can pass the information. Symmetric calculations offer secrecy, whereas satisfying the supremacy and memory requirements of sensor networks [7]. Here each sensor nodes uses a single key for encrypting and decrypting process and hence, it is challenging to provide authenticity, and there are no appropriate key switch mechanisms. The utilization of a single key in every hub of a system is a security hazard as one traded off hub can take a chance with the security of the entire system.

Prime Field Arithmetic for ECC [8] was proposed for high-level security mechanism using Optimal Prime Fields (OPF). The OPF is the general form of prime designed in the form of \( p = u.2^K + v \). Primes with low hamming weight are considered for reducing the execution time in consideration of hybrid technique Montgomery multiplication. Combinatorial Design of key distribution mechanisms [9] was proposed to fix the key size and which keys to be used for secured nodes and to create secured wireless links among them. This combinatorial approach uses two factors such as Balanced Deficient Block Devise (BDBD) and Global Quadrangles (GQ) are mapped for obtaining efficient keys for distribution among the network. However, the length of the key path gets varied according to the size of the network. Trajectory privacy-preserving framework was proposed [10] to analyze the threat models with various background information as well as to calculate the efficiency of the trajectory privacy. By considering the time factor, the theoretical mix-zone model is applied to preserving data by defining a data-sensitive area. Public key cryptography mechanism is used in the Secured Data Discovery and Dissemination based Hash (SDDDH) scheme [11]. Here one-way hash cryptographic function was applied for data security process. The base station generates a signature packet in each round of data distribution, and every sensor nodes receive a signature packet for node verification. Trickle algorithm is applied for packet verification, and puzzle keys are added for encryption of messages. However, this Merkley hash tree algorithm consumes more energy for packet encryption and verification. Secure communication for Industrial Internet of Things [12, 13] was proposed by applying a user validation protocol along with seclusion fortification for Industrial IoT was proposed. Security of the proposed scheme is proved under a random oracle model is used to prove the security of the IoT system and other security measures of the protocol. The mechanism [14] provides cryptanalysis of system based on ECC, the cryptanalysis consequence guides to intend a resolution to overcome the issues. It provides a robust hash based conditional P2 authentication and probabilistic key swap protocol that results lightweight and computes low overheads during the entities involved. Anonymous ECC-based self-certified two-factor key management scheme [15] was proposed that offers the required security features. This mechanism has the ability to produce a secure channel during the communication of secure sensors (cluster members) and access point (cluster head) [16].

3. Proposed Method – Long and Strong Security (L&SS) using two-level checks. The node undergoes two levels of security check during the transmission of information to the other node. Level 1 Node Reputation Check (L1-NRC) check includes the elimination of malicious node from the communication process by assigning a duplicate address for the routing nodes in the system. Level 2 Elliptic Curve Cryptography (L2-ECC) check adds the illegitimacy of the node for completely eliminating the node from the communication process.

3.1. Level 1-NRC. The nodes present in the network are categorized into malevolent and customary class. Malevolent nodes are selfish behaviour nodes, and customary nodes perform routine operations. These nodes are classified on behalf of reputation values of each node that is determined based on the grade calculation.
The grade factor \( (G_f) \) calculation of nodes includes their energy level and the energy threshold level. GH nodes fall under Normal Set Nodes (NSN), and GL nodes fall under Malevolent Set Nodes (MSN). The grade factor calculation is done based on the total number of processed control messages as per required data transmissions.

\[
G_f(n) = \frac{\text{No. of RREQ processed}(n)}{\text{No. of RREP processed}(n)}
\]  

\((GL < 35 < GH)\)

The nodes are fallen under the category of secured nodes present in the routing path provided with a private key and certificate. To establish trust between the normal set of nodes and malignant nodes, a signed certificate is provided by a certificate authority. Using a Hash function, fake Cryptographically Generated Addresses (CGA) is created for each node. Here, CGA uses a hash extension method, which has security parameter 'sec' that linearly scales the number of bits used in the hash extension by imposing 16xsec many bits to zero in the hash value denoted by a hashing algorithm.

Dummy address is created for each node present in the route starts from source to destination. The dummy address is created for all the nodes such as source node, relay node and the destined node. This dummy address significantly reduces the passive attacks and also provides more security to the system. Besides, the dummy address generation ensures that the malicious observer cannot identify the original identity of the nodes.

Dummy CGA Generation: Address generation for legitimate nodes are not expected to exceed 2S. Besides the time required to generate address is considered. Therefore the cost of address generation is \( G_T \). The time value \( T \) is set constant for

\[
G_T = 2^x + T
\]

The source hub 3 generates dummy address by using \( G_T \) formulae \((23+1)\), and the temporary node address of 3 is 9. The same procedure is done for all the intermediate nodes present in the route until it reaches to destination. Then the private key is assigned for all the intermediate nodes. If the key matches then the nodes processed for L2 ECC check. This process makes the node reputation value stronger.

**Algorithm for L1-NRC:**

- **Begin** Procedure
- **Set** Src & Dest node
- **While** Src not in range of Dest do
- **Compute** \( G_f(n) \)
- **Select** NSN nodes
- **Assign** CGA for all nodes
- Checks for \( M_N \) in the routes
- **If** \( M_N \) present in route nodes do
Create dummy CGA for all nodes

**Assign** private key for all the routing nodes

**If** key matches process L2 check

**Else** label the node as \( M_N \)

**End** Procedure

### 3.2. Level 2- ECC.

Elliptical curve cryptography is a well-known process for providing security in wireless networks. Each sensor node sends the public master key to the cloud and secretly stored in the sensor node once the NRC is done. Before data transmission, every node checks the master key, and if the master key is a match, then it transmits the data to the next hop. This process is continued until the set of transmission between source and destination is done in the network.

The corroboration action of this level-2 check is to bring out that the directing node is malevolent to all other nodes present in the routing so that the node can be excluded from the communication process. It comes beneath level 2 (L2) action of the proposed mechanism. To incorporate this, there is a need to confirm that a node is illegitimate before broadcasting its label 'malicious' to all nodes. Therefore, the node that fails the NRC test is further examined using the elliptical curve cryptography technique using the Weierstrass Elliptic function. Consider the coordinate points of the initiator node or source and the Next node \( N_i \) are \( P \) and \( Q \), respectively; there is another point \( R \) that creates a straight line as demonstrated in figure 3.2.

The Weierstrass Elliptic Curve function is defined in 3.3,

\[
x^2 = y^3 + by + c
\]  

\[P + Q = R \quad \text{where} \quad P \neq Q \quad \forall P, Q \in E
\]  

where \((x_P, y_P)\) and \((x_Q, y_Q), (x_R, y_R)\) are the coordinates of points \( P, Q \) and \( R \) that forms an elliptic curve. Consequently, the coordinates can be attained through the subsequent expressions,

\[x_R = \alpha^2 - x_P - x_Q\]  

\[y_R = \alpha(x_P - x_R) - y_P\]

with \((y_Q - y_P)/(x_Q - x_P)\).

The commutative property of this function states that,

\[P + Q = Q + P\]

![Fig. 3.2. Nodes Representation with Elliptic Curve](image)
Table 4.1  
Simulation Parameters

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>Simulation Area</td>
<td>1000m x 1000m</td>
</tr>
<tr>
<td>Number of Nodes</td>
<td>100</td>
</tr>
<tr>
<td>Channel</td>
<td>WirelessPhy</td>
</tr>
<tr>
<td>MAC</td>
<td>802.15.4</td>
</tr>
<tr>
<td>Radio Propagation Type</td>
<td>Two Ray-Ground Type</td>
</tr>
<tr>
<td>Antenna Type</td>
<td>Omni Directional</td>
</tr>
<tr>
<td>Traffic Models</td>
<td>CBR</td>
</tr>
<tr>
<td>CBR Interval</td>
<td>1.0 ms</td>
</tr>
<tr>
<td>Simulation Time</td>
<td>50 sec</td>
</tr>
<tr>
<td>Node Communication Range</td>
<td>100m</td>
</tr>
</tbody>
</table>

The algorithm shows that the node estimates the reputation value and sends it for L2 verification. Meanwhile, the forward secret $FS_K EY$ is calculated using equation 3.8 at the source end. The next node replies to the $L3_R EQ$ with $R_K EY$ using the equation 3.9.

$$FS_K EY = P + Q$$  \hspace{1cm} (3.8)

$$BS_K EY = Q + P$$  \hspace{1cm} (3.9)

The source node S compares its FSKEY with the BSKEY to conclude that the directing node N to the destination is a malignant node (MN). Therefore the computed FSKEY is not equivalent as BSKEY then the node is published as malicious to all other nodes and the next nearest node is considered for communication. The algorithm 2 is a part of the main algorithm 1 and is described below.

Algorithm 2
1: Level-2_ECC_Check()
2: S sends $L2_R EQ$ to check directing nodes N
3: S calculates $FS_K EY \triangleleft P+Q$
4: D replies with $BS_K EY \triangleleft Q+P$
5: If ($BS_K EY \neq FS_K EY$)
6: Remove $M_N$ from the neighbour list
7: Label N as $M_N$ & Broadcast to all nodes
8: Return 1
9: Else
10: Set node as Normal node
11: Return 0
12: end if
13: end

4. Results and Discussions. The proposed work of L&SS is examined with the Network Simulator (NS2). The imitation of the L&SS method has 100 nodes arrange in the simulation area 1000 × 1000 as revealed in Table 4.1. Every node accepts the signal from all direction through using the Omnidirectional antenna. By using a Constant Bit Rate (CBR) traffic model, the traffic is handled. The parameters used for analyzing the performance of the proposed and existing schemes are packet loss rate, packet received rate, average delay, leftover energy, and throughput.

5. Parameter Analysis. WSN implementation of many security schemes affects the QoS of the system. The metrics considered for the system evaluation are delivery rates, throughput, delay and detection rate. There is a tradeoff between QoS and security in this scheme.
5.1. Delivery Rate. The delivery rate of data from the source point to the destination point is identified to determine the total amount of packets received effectively by the destination point. DR is estimated by using equation 5.1

\[
PDR = \frac{\text{Total number of packets received successfully}}{\text{Total number of packets sent}}
\]  \hspace{1cm} (5.1)

Figure 5.1 illustrates the delivery rate of packets for the proposed L&SS and the conventional method such as SDDDH. The L&SS method achieves greater PDR compared to other existing mechanisms since the node ID verification process prevents the intermediate nodes from malicious attacks.

Delay

Average delay is estimated by analyzing the time difference of sent packets and received packets. The delay estimation includes nodal processing and queuing time evaluation during transmission and reception of data, and the calculation is given in next equation:

\[
\text{Delay} = \frac{\text{Packets received time}-\text{Packets sent time}}{\text{Total time}}
\]  \hspace{1cm} (5.2)

Delays of proposed (PDly.tr) and existing (Dly.tr) schemes are measured and plotted in figure 5.2. It can be observed that the delay of the proposed method is bare minimum since the route is filtered and the intermediate nodes are verified. The decrease in delay reflects the efficiency of network routing.
Throughput

Throughput is commonly evaluated for analyzing the overall network performance, and it is defined as the rate of successful delivery of packets to the destination over a preferred set of connections. Throughput calculation is given in the equation

$$\text{ThroughPut} = \text{Total Pkts Received} \times \text{Pkts Size} \times 100$$  \hspace{2cm} (5.3)

The simulation analysis proves that L&SS method has higher throughput compared to the SDDDH mechanism.

Detection

The detection rate for both proposed and existing routing protocols are examined. Figure 6 shows that the detection rate for both conventional and proposed schemes.

$$\text{Detection Rate} = \frac{D_{mn}}{T_{mn}}$$  \hspace{2cm} (5.4)

where $D_{mn}$ is the number of malicious nodes detected by one or more normal nodes, $T_{mn}$ is the total number of malicious nodes.

6. Conclusion. Long and Strong Security mechanism with two-level checks is proposed here. Grade factor keys are generated for determining node reputation values as well as for node verification process. Besides, dummy address creation actions are performed to improve security measures. The proposed L&SS
simultaneously reduces energy consumption among nodes, and the routes are undetectable by malicious nodes since it uses dummy CGA for the transmission process. The data transmitted over the route are strongly encrypted with the help of public-key cryptosystems. Level 1 check includes verifying node reputation value and level 2 check includes Elliptical curve cryptography (ECC). Each sensor node sends a public master key to the cloud and secretly stored in the sensor node. Before data transmission, every node checks the master key, and if the master key is a match, then it transmits the data to the next hop. However, the computational cost makes the system quite complicated but offers strong security, and the number of packets outcome at the receiver is quite high. The performance of the network is analyzed and the proposed system has 35.7% greater efficiency compared to the conventional in terms of data delivery rate.

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DETECTION AND CLASSIFICATION OF 2D AND 3D HYPER SPECTRAL IMAGE USING ENHANCED HARRIS CORNER DETECTOR

S. PAVITHRA †, A. KARTHIKEYAN †, AND P.M. ANU ‡

Abstract. Image classification and visualization is a challenging task in hyper spectral imaging system. To overcome this issue, here the proposed algorithm incorporates normalized correlation into active corner point of an image representation structure to perform hasty recognition by matching algorithm. Matching algorithms can be of two major categories, based on correlation and based on its features based on correlation and on its feature detection. Proposed algorithms often ignore issues related to scale and orientation and also those to be determined during the localization step. The task of localization involves finding the right region within the search image and passing this region to the verification process. A Harris corner detector is an advanced approach to detect and extract a huge number of corner points in the input image. We integrate all the extracted corner points into a possible task to locate candidate regions in input image. In terms of detection and classification the proposed method has got better result.

Key words: Harris corner detector, pyramid Correlation, Sub Pixel Mapping, Auto-correlation matrix.

AMS subject classifications. 68U10

1. Introduction. In 2D and 3D reconstruction from hyper spectral images has seldom been addressed in the literature. This is a challenging problem because 3D models reconstructed from different spectral bands demonstrate different properties [16]. Hyper spectral picture contains an abundance of information, however deciphering them requires a comprehension of precisely what properties of ground materials are attempting to estimate, and how they recognize with the estimations really made by the hyper spectral sensor. The instruments which are used in creation of Hyper spectral pictures called imaging spectrometers. The progression of these brain boggling sensors has incorporated the converging of two related however unmistakable advancements. Spectroscopy is the examination of light that is released by or reflected from assets and its assortment in vivacity with wavelength and spectroscopy and the remote imaging of earth and planetary surfaces. Detecting Corner points are considered important features for feature extraction and detection. Corner detection is a low-level image processing technique that is widely used in different computer vision applications.

Fifteen years prior just phantom remote detecting specialists had entry to hyper spectral images or programming tools to exploit such pictures. Over the previous decade hyper spectral picture investigation has developed into a standout among the most capable and quickest developing innovations in the field of remote detecting. The improvement of hyper spectral sensors and programming to break down the subsequent image information. Nowadays, developed build up area recognition in light of nearby invariant components has uncovered promising results. Be that as it may, it needs some format building pictures for preparing and in this way experiences a high figuring multifaceted nature and memory necessity. In their later, a more straightforward technique is utilized. Be that as it may, since it exclusively relies on upon nearby elements for acknowledgment, it can regularly be excessively feeble of a sign, making it impossible to dependably identify the developed districts in complex satellite picture. For this experiment of Harris corner evaluation we used a 3.2 eight core processor with 16 GB of memory with a NVIDIA GeForce GTX970 GPU [19].

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2. Materials and Methods. In this paper, we have introduce an unsupervised method toward all the while finding developed regions from various high-determination satellite pictures. The proposed strategy incorporates two noteworthy segments. In the first strategy a probability capacity based way to deal with concentrate hopeful developed locales, in which an enhanced Harris operation is proposed. In the second strategy spectrum grouping and chart cut-based unsupervised clustering calculation are adopted and territory recognition. Hyper otherworldly remote detecting innovation has progressed essentially in the previous two decades in these techniques. Based on the extensive tests, the proposed approach demonstrates the following main advantages over the previous works. First, it can simultaneously detect built-up regions from multiple images, and our experiment results show that the performance of built-up area detection can be further improved by exploring the cues drawn from multiple images. Second, the entire process is highly automatic and requires no human interaction. Taking after an underlying arbitrary assignment of delicate pixel extents to rigid sub pixel double classes, the calculation works in a progression of emphases, the calculation step contain three stages [3].

In Hyper Geometrical, Statistical, and Sparse Regression-Based method introduced by Jose M. Bioucas - Hyper ghastly instruments secure electromagnetic vitality scattered inside of their ground quick field view in several otherworldly channels with high ghastly resolution [4]. A new sub-pixel mapping calculation in light of a Backpropagation neural network system with a perception model by Liangpei Zhang. The blended pixel is a typical issue in remote detecting order. Despite the fact that the structure of these pixels for various classes can be evaluated with a pixel un-blending demonstrate, the yield gives no sign of how such classes are circulated spatially inside of these pixels [5]. Joining pixel swapping and forming techniques to improve super-determination mapping by Yuan-Fong Su, Giles M. Foody-Combining super-determination strategies can build the exactness with which the state of articles might be portrayed from symbolism. This is shown with two ways to deal with brushing the molding and pixel swapping techniques for super-determination mapping for parallel order applications. [6] super-determination land spread example expectation utilizing a Hop field neural system by A.J Tatema, H.G. Lewis-Landscape design speaks to a key variable in administration and comprehension of nature, an in addition driving numerous ecological models. Remote detecting can be utilized to give data on the spatial example of area spread elements, however investigation and arrangement of such symbolism experiences the issue of class blending inside of pixels.

In like manner four methods for highlights can be utilized as a part of picture handling systems: low-level visual features, local features, local-global features and biologically inspired features [11-12]. Even though these features classification are always based on the assumption that all pixels in the image are pure pixels, which is unreasonable in remotely sensed images. In [13] Fisher suggested that there are four types of mixed pixels in remotely sensed images as follows in the upcoming section.

1. In convoluted sub pixel – the overall size of the 2D and 3D object is smaller than the size of the pixel.
2. Identify the Region Of Interest (ROI) – we identify the boundary value of an image along with the sizes of two or more land-cover classes on the ground value may be larger than the sizes of the pixel value, but parts of their boundaries lie in a sub-pixel.
3. Inter grade method – a pixel value assign a space for a transition from a cluster of one class to another class.
4. Sub-Pixels: the width of a land-cover is thinner and length of the land-cover class may be longer than a pixel. Many sub pixel mapping methods have been proposed to tackle the different types of mixed pixels [15].

The proposed strategies is explained briefly about pixel-based techniques. These strategies attempt to measure the likelihood of every pixel to have a place with the conceivable classes by utilizing measurement measures construct just in light of ghastly properties. The (MaxVar) Maximum Likelihood Classification method stays a standout among the most well known strategies for RSI characterization. The likelihood of every pixel to have a place with each of the characterized classes in MaxVer figure.

2.1. Sub Pixel Mapping. The key issues present in sub pixel mapping are the means by which to decide an ideal sub pixel dispersion of every class in a pixel. A coarser division picture is acquired by a phantom un-mixing method, as the information picture and every pixel is separated into sub pixels, where S speaks to the scale component [9-10]. The quantity of sub pixels for every area spread class is then dictated by the
plenties of the portion picture. Fig. 2.1 delineates the rule of sub pixel mapping and depicts a basic case with two classes. In picture, as appeared in Fig. 2.1(a), assuming the scale division is represented as \( S \) and the value is 4, a coarse pixel is isolated into 16 (\( 4 \times 4 \)) sub pixels, and 0.5 in the portion picture implies that 8 (\( 16 \times 0.5 \)) sub pixels fit in with area spread class 1. Fig. 2.1(b) and 2.1(c) depict two possible dispersion of sub pixels, and the previous is better than the last mentioned and uses that data to dole out the class with the most astounding likelihood. The upgrades in sensor advances have expanded the openness to high-determination pictures. Thus, new methodologies have been created to improve an utilization of the accessible information.

In the review of RSI characterization until 2005 is displayed in. Those works discuss about the difficulties and portrays every one of the strides that create the characterization process. Different arrangement strategies are introduced and gathered by scientific classification. All the more as of late another pattern can be watched. Numerous studies consider data encoded in locales [16]. In this strategy, districts were ordered by utilizing Mahalanobis separation and Support Vector Machines (SVM). In pattern recognition SVM is a class of algorithm made a locale based order strategy for high determination pictures that abused two approaches: In Lee et al., [15] MaxVer with area means and MaxVer with Gaussian Density Function were adopted and interpreted. Both mechanism shoes preferable results over pixel-based classifiers. Yu et al [16] likewise proposed a strategy to order RSI taking into account areas. The picture division and characterization were performed by utilizing fractal systems and non-parametric K-Nearest Neighbor individually. Another late work in this exploration territory was produced by Katartzis et al [17], who proposed an area based RSI arrangement technique that uses Hierarchical Markov Models.


3.1. Harris Corner Detector. The main challenges present in the Harris Corners Detector (HCD) are used to identify corners points present in the input image [18]. The proposed technique is used to identify vertical and horizontal edges with help of Sobel detector. Those edge detectors are then used to reduce the effect of noise present in the image. The subsequent edges distinguished esteems are then all things considered to frame a vitality delineate contains valleys and tops. The pinnacles show the nearness of a corner point without loss of all-inclusive statement and also we have chosen different gray scale 2D and 3D image as a data set to generate the output. Let this image be given by \( I \). we will assume a grayscale 2-dimensional image is used. Let this image be given by \( I \). Consider taking an image patch over the area \((u,v)\) and shifting it by \((x,y)\). The weighted sum of squared differences (SSD) between these two patches, denoted \( S \), is given by equation 3.1.

\[
S(x,y) = \sum_{u} \sum_{v} w(u,v)(I(u + x + y) - I(u,v))
\]  

(3.1)

**Algorithm**

1. First it calculates the range from X and Y derivatives of an image

\[
I(x) = G_x^\sigma * I
\]  

(3.2)

\[
I(y) = G_y^\sigma * I
\]  

(3.3)
2. It produces the products of derivatives at every pixel

\[ I_{x^2} = I_x I_x \]  
\[ I_{y^2} = I_y I_y \]  
\[ I_{xy} = I_x I_y \]  
\[ I_{xy} = I_x I_y \]  

3. It generates the sum of the products of derivatives at each pixel

\[ S_{x^2} = G_{\sigma_1} I_{x^2} \]  
\[ S_{y^2} = G_{\sigma_1} I_{y^2} \]  
\[ S_{xy} = G_{\sigma_1} I_{xy} \]  

4. Identify the matrix function at each pixel of \( H(x,y) \)

\[ H(x,y) = \begin{bmatrix} S_{x^2}(x,y) & S_{xy}(x,y) \\ S_{xy}(x,y) & S_{y^2}(x,y) \end{bmatrix} \]  

5. It responds the detector at each pixel threshold where

\[ R = \text{DET}(H) - K(\text{Trace}(H))^2 \]  

6. Threshold on value of \( R \), Here we adopted the faster Non-maximal Suppression to the gradient magnitude. In canny approach the edge direction is reduced to any one of the four directions.

To complete this task for a given point, its gradient is compared with that point of its 3x3 neighborhood.

If CM (candidate magnitude > neighborhood)
  Then
    The edge strength is maintained
  Else
    Discarded

3.1.1. Corners as Interest Points. There are several applications requiring to relating two or more pictures keeping in mind. The aim is to concentrate extract interested corner point from the image. For example, in case of two dynamic edges in a video gathering taken from a moving camera can be associated to concentrate data with respect to the profundity of articles in the earth and the pace of the camera. The best power strategy for contrasting each pixel in the two pictures is computationally restrictive for the dominant part of uses. Instinctively, one can picture relating two pictures by coordinating just areas in the picture that are somehow fascinating.

A Number Of Criteria To Identify A Corner Detector (Figure 3.1)

1. At first All "true corners" should be detected and then

![Sample image with corner types of L, Y, T, Arrow, and X-junction](image)
2. No "false corners" should be detected.
3. All Corner points in the input image should be localized well.
4. Detector is supposed to have a high repeatability rate.
5. Detector is supposed to be robust.
6. It is computationally more efficient.

**Steps to extract corner points** (Figure 3.2)
- It Extract the interest points with the proposed algorithm
- It compare the two points with cross-correlation
- Verification will be done with the fundamental matrix.

### 3.2. Determining Auto-correlation Matrix.
It confine the structure of the local neighbourhood and gauge based on eigenvalues of $M$:

$$M = (\Delta x \Delta y) \left[ \frac{\sum_{(x_k y_k) \in W} (I_x(x_k, y_k))^2}{\sum_{(x_k y_k) \in W} I_x(x_k, y_k) I_y(x_k, y_k)} \right] \frac{\Delta x}{\Delta y}$$

(3.12)

The image has been calculated the interest point based on the eigenvalues of $M$ with help of equation 3.12:
- 2 strong eigenvalues => interest point
- 1 strong eigenvalue => contour
- 0 eigenvalue => uniform region

The values of the pixels is compared with sum-square difference (SSD) of square neighbourhood about the points. This is the form of SSD measure:

$$SSD = \sum \sum (W_{1, i, j} - W_{2, i, j})^2$$

(3.13)

The experimental results and performance of the image classification are carried out based on the following result, initially the all the images done by preprocessing. The parameter that we use in the Harris algorithms was a Gaussian window with $W = 3 \times 3$ kernel and $\sigma = 0.3$ as the baseline algorithms. For the proposed technique, the pruning technique is first applied to the entire image in order to select the corner candidate set. Next, the corner measure of the corresponding baseline algorithm is applied to extract the final corners (Figures 4.1 and 4.2).

### 5. Conclusion.
Detecting and classification of an 2D and 3D hyper spectral image, corner point extraction using enhanced Harris corner detector and sub-pixel mapping method has been adopted in this paper. The proposed technique concentrate on various hyper spectral pictures as dataset and otherworldly unmixing to meet with the genuine task; and also the ability to normalize and extract the interesting features with corner points in the input image by using enhanced Harris corner detector. By adding former information to obtain a better sub-pixel mapping result. The conventional algorithms always utilize the fraction image obtained by the classification image sub-sampling method as the input information. In Harris corner detector, the original image sub-sampling is used to reduce the size of the image with original values and extract the interest point.
The proposed method was created to perform good accuracy but it requires more computational time. Therefore, in future it is suggested that the research should be carried out in the field of video mosaicing and other real-time hyper-spectral images using back propagation with neural network.
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SPCACF: SECURED PRIVACY-CONSERVING AUTHENTICATION SCHEME USING CUCKOO FILTER IN VANET

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Abstract. Providing security for vehicular communication is essential since the network is vulnerable to severe attacks. The message authentication between vehicles and pavement units are essential for the purpose of security. Messages that passed between the vehicles should be encrypted and verified before the vehicle nodes could be trusted. The original identity of nodes can only be traceable by authorized parties and should not be exposed at any cause. However authentication between vehicles during message transformation does not guarantees message authentication rate accurately. To address these issues, the SPCACF scheme is proposed which is based on software devoid of relying on any particular hardware. Binary search algorithm is added in partial with Cuckoo Filter to achieve higher accomplishment than the preceding schemes in the batch wise verification phase. In order to guarantee that it can assure message authentication constraint, existential enforceability of underlying signature against adaptively chosen-message attack is proved under the positive filter pool method.

Key words: Message Filters, Vehicular Ad hoc Networks, Authentication, wireless.

AMS subject classifications. 68M15

1. Introduction. The systems that interconnect vehicles on road are called Vehicular Adhoc NETwork (VANET). "A portable impromptu system comprises of versatile nodes that associate themselves in a decentralized, self-sorting out and may likewise set up multi-hop paths. In the event that versatile nodes (cars) are movable, this is called VANET". "The fundamental focus of research in VANETs is the upgrades of vehicle security by methods for Inter Vehicular Correspondence (IVC)". Several different applications are emerging in VANETs [1-3]. These 3 applications incorporate security applications to make driving a lot more secure, portable trade and other data benefits that will illuminate drivers about any sort regarding blockage, driving risks, mishaps, roads turned parking lots. VANET technology uses moving vehicles as nodes in a network to create a mobile network. VANET turns each taking an interest vehicle into a remote switch or hub, permitting cars around 100 to 300 mts from one another to connect and make a system with a wide range. As vehicles drop out of the sign range and drop out of the system, different vehicles can participate, connecting vehicles to each other with the goal that a versatile system is made [4]. It is assessed that the main frameworks that will incorporate this innovation are police and fire vehicles to communicate with one another for security related purposes.

2. Related Works. VANETs had updated in several ways for faster communication among vehicles. The main difference between MANET and VANET is that the MANET has mobile routers that act as nodes for transmission of message and those routers builds VANET system using vehicles.

In network communication, a topology is a typically schematic depiction of the arrangement of a system, including its nodes and interfacing lines. There are two different ways of characterizing system geometry: the physical topology and the logical (or sign) topology. The physical topology of a system is the real geometric design of workstations. Logical (or signal) topology alludes to the idea of the ways the sign pursue from node to node. The vehicle shares their sensed message and location information for changing the traffic flow and controlling it in a futuristic manner.

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For reliable communications localization is a major challenge in achieving reliable communication in WSN. Sensor node position is termed as localization and non-linear version of Extended Kalman filtering deals with the case governed by stochastic degree of difference equations. This filter have consistency issues. Efficient algorithm for sensor locality enables to estimate the location with high accuracy. Particle swarm optimization assisted EKF for localization in WSN. A few distinct applications are rising as to vehicular communications that are mainly based on the fingerprint recognition for secured data communication [5-7]. For instance, applications for more secure driving, data administrations to educate driver’s business benefits in the vehicle region. Government, organizations, and the scholastic networks are taking a shot at empowering new applications for VANETs. A primary objective of VANETs is to build street security by the utilization of remote correspondences. To accomplish these objectives, vehicles go about as sensors and advise each other about irregular and conceivably destructive conditions like mishap, car influxes and coats. Vehicular systems intend to look like specially appointed systems on account of their quickly changing topology [8, 9] by applying filter techniques. For various regions there exist trust agents in order to provide secured keys for secured communication. There are numerous elements engaged with a VANET settlement and sending. Despite the fact that by far most of VANET fundamental tasks in those systems performed with various substances [10]. Selection mechanism was proposed to identify the Line of Sight and Non-Line of Sight conditions. Here hybrid EKF and H-infinity filter is used to look up the accuracy. Finally we use linear least square algorithm to estimate the location [11].

3. SPCACF - Description.

3.1. Node Creation and Configuration. Node creation is only the production of the nodes wirelessly in the system that is chosen. Node configuration system basically comprises of characterizing the diverse hub attributes before making them. They may comprise of the kind of tending to structure utilized in the recreation for characterizing the system segments of portable nodes.

3.2. Neighbor node Discovery. Each wireless node finds its neighboring nodes, the ones that are within its transmitting range. Neighbor discovery in a WSN is the determination of all nodes with which a given node may communicate directly. On the contrary, neighbor discovery reveals all possible paths between any two nodes in a network. Neighbor discovery is achieved by the conditions specified in this paper: using distance and the number of hops conditions.

3.3. Initial handshaking and Message signing. When vehicle enters into new Road Side Unit (RSU) then initial progress of handshaking should be carried out. Through the use of RSU the vehicles substantiate itself with trust authority. Trust authority is the main tool to recognize the nodes original distinctiveness or ID. Trust authority will pass message to roadside unit to allow it to verify the vehicle’s signature even if it uses pseudo identity for message signing. Trust authority has two filters such as positive filter pool and negative filter pool and it generates secret key with the vehicle. RSU will forward shared secret from the positive filter pool to the automobile. Each vehicle generates secret integer when moving to each new RSU.

3.4. Group key signing and verification. RSU utilizes this module to check a lot of messages without the bilinear matching tasks in a group mode. Creating a notification communicate message utilizing cuckoo filter is presented along with working instance of invalid signatures in the batch verification is clarified in this scheme.

4. SPCACF Algorithm - Description. Secured privacy-conserving authentication model for VANETs along with Cuckoo Filter is used to improve the network security features. Cuckoo Filter along with binary search methods which accomplish advanced progress rate than the earlier schemes in the batch verification stage. Binary search consists of positive filter pool and negative filter pool for verification purpose. The vehicles should get paired before them starting communication.
Let $V_i$ be the source vehicle and $V_j$ be the sink vehicle. $V_i$ computes fingerprint signature and sends to the node $V_j$, now $V_j$ computes fingerprint signature and sends back to the node $V_i$. $V_i$ checks for the positive and negative filters on the binary search basis. If the fingerprint signature of both nodes $F_i$ and $F_j$ falls under positive filters and also get matched, then the set of data transmission is done and the query is passed to some other vehicles present in the road side units.

The generated signature by the source node falls under the positive filter pool and it contains some integers with it. Also the negative filter pool consists of some integer representation indicates malicious fingerprint signature. If the sink’s signature falls under the positive filter pool, then the source can send the requested information ($M_i$, $\sigma_i$) to the destination. But the signature falls under the negative filter pool then the source vehicle $V_i$ needs to re-confirm with the $V_j$. The data cannot be processed since the node $V_j$ marked as malicious vehicle.

Let $X_{j,i} = (ID_j \parallel T_j \parallel M_j)$ (4.1)

Algorithm SPCACF

Require: $X_{j,i} <- (ID_j \parallel T_j \parallel M_j)$.

1:Vi computes a fingerprint $f_j$ <- Fingerprint$(x_j)$
2:Vi queries Cuckoo Filters with query.
3:Vi checks novelty of Ti.
4:While $T_i - T_j \leq \Delta T$ Do
5:Vi checks $f_j$ against Cuckoo Filters;
6:if $f_j$ is in positive Filters then
7:Vi checks $f_j$ against negative Filters
8:if $f_j$ is not in negative Filters then
9:Vi send $(M_i, \sigma_i)$ to $V_j$; split;
10:end if
11:else
12:if $f_j$ presents in negative Filters then
13:Vi re-verification required; split;
14:else
15:if $f_j$ present in negative Filters then
16:Vi discards $V_j$; break;
17:else
18:end if
19:end if
20:else
21:Vi wait for subsequent relay; split;
22:end if
23:end while
24:end

5. Results and Discussions. System performance is evaluated by tracing the files of an output through generated Xgraph files of version NS-2.35. The events occurred in the network animator window are testimony into trace files while accomplishing record process. The events like packet delivery rate, lost rate and residual energy of the different schemes considered in the proposed work and conventional scheme can be traced out using the trace files.

Delivery Rate: $DR$ is defined as the proportion of total data packets established by the sink to total sent packets by source in multiplication with the number of receivers. Delivery rate is calculated by the equation 5.1:

$$PDR = \frac{\text{Total Pack Received}}{\text{Total Pack Send}}$$ (5.1)

Loss Rate The Loss Rate is the proportion of packets that failed while reaching the receiver to the sum of data packets sent by the sender. The LR is calculated by Equation 5.2:

$$PLR = \frac{\text{Total pack Dropped}}{\text{Total pack Sent}}$$ (5.2)
Energy Consideration: WSN comprises of sensor nodes fixed over a geological region for observing physical conditions like temperature, mugginess, vibrations, seismic occasions, etc. Normally, a sensor node is a little device that incorporates three essential parts: a detecting subsystem for information obtaining from the physical encompassing condition, a processing subsystem for nearby information preparing and capacity, and a wireless correspondence subsystem for information transmission. What’s more, a power source supplies the vitality required by the device to play out the modified undertaking.

Residual energy can be computed from the initial energy of the node. The difference between the initial energy level and the current energy level gives the remaining energy of the node. Figure 5.3 shows residual energy for both proposed and existing method.

6. Conclusion. Secured privacy-conserving authentication scheme with cuckoo filter utilized for both between vehicles communications and vehicle to roadside communications. Positive and negative filter pool is established for verifying the integer keys by applying binary searching process. This improves the efficiency of the proposed scheme through batch validation process. The performance analysis results show that the proposed scheme has better efficiency when compared with existing schemes. Internet of Things can be included for the
further development of application in the case of achieving secured circumstances.

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LOGISTICS OPTIMIZATION IN SUPPLY CHAIN MANAGEMENT USING CLUSTERING ALGORITHMS

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Abstract. Today’s business environment, survival and making profit in market are the prime requirement for any enterprise due to competitive environment. Innovation and staying updated are commonly identified two key parameters for achieving success and profit in business. Considerably supply chain management is also accountable for profit. As a measure to maximize the profit, supply chain process is to be streamlined and optimized. Appropriate grouping of various suppliers for the benefit of shipment cost reduction is proposed. Data relating to appropriate attributes of supplier logistics are collected. A methodology is proposed to optimize the supplier logistics using clustering algorithm. In the proposed methodology data preprocessing, clustering and validation process have been carried out. The Z-score normalization is used to normalize the data, which converts the data to uniform scales for improving the clustering performance. By employing Hierarchical and K-means clustering algorithms the supplier logistics are grouped and performance of each method is evaluated and presented. The supplier logistics data from different country is experimented. Outcome of this work can help the buyers to select the cost effective supplier for their business requirements.

Key words: Supply Chain Management (SCM), Supplier logistics, Optimization, Hierarchical clustering, K-means clustering

AMS subject classifications. 90B06, 62H30

1. Introduction. Researchers have focused on most of the processes of manufacturing supply chains. More attention has been shown on both optimization and effectiveness improvement of supply chain. Gradual increase in cost for raw material, manufacturing, logistics, distribution and inventory is the main reason to focus attention on supply chain. Apart from the cost involved, time is an important criterion challenging the effectiveness of a supply chain. Two basic process of supply chain are inventory control and production planning process and logistics and distribution process. These two processes holds the responsibility of the manufacturing process design and management. First process includes scheduling and acquisition, design and scheduling of manufacturing process and design and control of material handling. Second process completely governs logistics and transportation of the products, which counts the quantity of product to be shipped from warehouse to retailers. Inventory management, shipment of products and delivering final product are major role of this process. Designing and analysis of supply chain are categorized into four namely 1. Analytical model (deterministic) which will be using specified and known variables. 2. Analytical model (Stochastic), with at least one unknown variable which is assumed to follow any one probability distribution. 3. Economic models 4. Simulation models

The main objective of supply chain management (SCM) is to maximize the total profit of the manufacturing facilities. Associated constraints are resource and production constraints and logical consistency constraints. Logical consistency constraints include demand limits, availability, variable non-negativity and feasibility. As a result, it is suggested to improve the effectiveness of SCM of a business. Instead of concentrating the entire supply chain activities it is suggested to choose a particular part and optimize it. As a continuation concentrate on all individual areas of supply chain and make the entire system as optimized one. In this work an algorithm based on clustering approach is developed to group the suppliers involved in the business. Grouped suppliers

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1. Raw data collection of supplier
2. Normalize the supplier data using Z-score normalization
4. Error calculation for intra and inter clusters and reassignment of elements among the clusters
5. Predict the accuracy of the modified K-means algorithm

Fig. 3.1. Clustering Process in SCM

are ranked in clusters. Based on the first cluster elements it is decided to be the best suppliers to be preferred first in placing the order and also the transportation route decision. The Z-score normalization is used to normalize the raw data. Hierarchical and K-means algorithms are used to cluster the supplier.

Introduction is followed by literature review, proposed methodology, results and discussion and the conclusion. Future scope for research are also pointed out.

2. Literature Survey. A detailed study on performance metrics of SCM was made [1]. Literature review on modeling a supply chain was made along with a new research agenda in SSM [2]. Quality measurement in SCM was carried out to ensure the right quantity of right product in right time [3]. An optimization based on network was proposed [5]. By using a special structure optimal inventory position was found in a reverse logistic system with a special structure [6]. Using agent oriented software architecture, the agent oriented SCM was built [7]. For the purpose of selecting and benchmarking potential supplier’s artificial neural network and hybrid care based reasoning was used [8]. Inventory model was proposed with multiple supply chains options. It ensures their concrete contribution to SCM [9]. SC modeling using simulation was explained. It dealt the requirements for simulation modeling and issues of SCM. A complete casual forecasting systems using GA in SCM was illustrated. Forecasting in SCM is carried out by using a computerized system [10]. The author proposed metaheuristics based approach for SCM [11]. The author presented quantitative model using clustering for SCM [12]. Various types existing typologies in the supply chain risk management since 2000 used for analysis. Idea was proposed to from new typology [13]. Triangulrisation Clustering Algorithm and petri net were used to check vulnerabilities in supply chain networks and assessed disruption mitigation strategies in SCM [14]. In order to model, monitor and manage SCM a unified framework was proposed [15]. By conducting a questionnaire survey a framework was proposed to improve the supply chain sustainability in Indian e-waste management system [16]. Based on the existing supplier strength reconfiguration of supply base of buying firms was done [17]. Linkage between SCM and porters cluster theory was demonstrated [18]. In a different way clusters were inserted into global value chains in order to disable and enable the upgrading efforts in local level [19]. A frame work using dynamic nonlinear multi attribute decision model was developed [20]. In order to perform dynamic analyses of SCM data management a system based on information technology was developed [21]. Cost involved in total SCM was reduced by the authors with a new approach [22]. Suppliers were clustered using EM algorithms and DEA calculated the internal operation efficiency and AHP was applied to assess the importance of the external functions in a SCM [23]. A set of manufacturing firms were clustered using various clustering methods like fuzzy c means, Gaussian mixture model, self-organized map, k means and hierarchical clustering [24]. By using branch and bound algorithm route was found. Customers were clustered using random clustering, K-means and K-medoids algorithm [25].

3. Proposed Methodology. The main objective of proposed algorithms is to cluster the supplier and to reduce the overall all expenses of the SCM. The optimization will improve the performance and profit of the SCM. The proposed methodology is shown in Fig 3.1.

3.1. Normalization of data. In supply chain management, all the supplier data are in different scale. If the clustering is performed with raw data, then the accuracy of the data may be reduced. The solution to improve the accuracy of the data is to convert all into uniform scale. The Z-score normalization is used to convert all data into uniform scale. The Z-score normalization formula is given in Equation 3.1:

$$Z = \frac{X - \mu}{\sigma}$$

(3.1)
where $Z$ is normalized value of corresponding data value. $X$ is supplier logistic values of corresponding feature. $\mu$ is the mean of the corresponding features in the supplier logistics. In $Z$ score normalization, the mean is zero and standard deviation is one. $\sigma$ is standard deviation of the supplier logistics feature. The standard deviation is calculated using Equation 3.2:

$$
\sigma = \sqrt{(\Sigma(X - \overline{X})/n)}
$$

where $X$ is Supplier logistics values of corresponding feature. $\overline{X}$ is the mean of the feature. $n$ is total number of supplier logistics data sample. The $Z$ score is calculated using value of the supplier logistics data sample, standard mean and standard deviation.

### 3.2. Clustering using K-means algorithms.

K-means clustering is unsupervised algorithm. The data are grouped based on feature similarity. The input of the algorithm is data set and number clusters. The normalized data is given as input. K-means algorithm has following three steps which are Initial centroid selection, distance calculation and assignment of elements in the cluster and calculation of cluster center and assignment of elements.

#### 3.2.1. Initial centroid selection.

The initial centroid of the cluster is calculated using formula shown in Equation 3.3:

$$
D = (Kmeans[sum(i) - min])/((max - min)) + 1
$$

where $D$ is the centroid of the cluster. $sum(i)$ is the sum of row elements for the given problem. $max$ is the maximum values and $min$ is the minimum values of $sum(i)$ respectively.

#### 3.2.2. Distance calculation and assignment of elements in the cluster.

The distance is calculated between data point and centroid of each cluster using formula shown in Equation 3.4:

$$
D(i,l) = \sqrt{\sum_{i=1}^{n} \sum_{j=1}^{k} (L_k - Y_j)^2}
$$

where $D(i,l)$ is distance between cluster ($L$) and $i$th data point of the cluster. $L_k$ is the centroid of the $k$th cluster. $Y_j$ is the $j$th data point of the cluster. After calculation of distance, data point is allocated to one cluster which has the minimum distance from that data point.

#### 3.2.3. Calculation cluster center and assignment of elements.

After allocation of each data point to the cluster, the cluster center is calculated using formula shown in Equation 3.5:

$$
C_i = \sqrt{\sum_{i=1}^{n} x_i / n}
$$

where $C_i$ is the center of the cluster. $x_i$ is the value of the features in the cluster. $n$ is the total elements in the cluster. After that, the distance is calculated between remaining data points and centroid of the cluster. The assignment of data points to cluster is done. The center calculation and assignment procedure is repeated until all the elements are assigned to clusters.

#### 3.2.4. Distance calculation and assignment of elements in the cluster.

The sum Squared Error (SSE) between each observation and its group’s center is calculated. It is used as a measure of variation within a cluster. If SSE is equal to 0 then all the cases within the cluster are identical. The SSE is calculated using formula shown Equation 3.6:

$$
SSE = \sum_{i=1}^{p} \sum_{q \in O_i} (q - C_i)^2
$$
where \( p \) is the total number of clusters, \( O \) is the number of objects in a cluster, \( C \) is the center point of a cluster. The SSE is calculated for each cluster. Minimum SSE will give more accuracy. To minimize the SSE is shown Equation 3.7, the center point is calculated using formula shown in Equation 3.8:

\[
SSE = \min_{i=1}^{k} \sum_{q \in O_i} (q - C_i)^2 \quad (3.7)
\]

\[
C_i = \frac{\sum_{x \in C_i} x}{|C_i|} \quad (3.8)
\]

where \( C_i \) is the center point of the cluster. \( X \) is the element of the cluster. \(|C_i|\) is the number center points of the cluster. The center point is calculated using data points. Which data point gives minimum SSE with center point then that data point is moved that cluster. It minimizes the error within the cluster then it will reduce the SSE.

### 3.3. Clustering using Hierarchical clustering algorithm.
In hierarchical clustering is used group the data in hierarchical manner. There are two types of hierarchical clustering namely agglomerative and divisive. In proposed, the complete link and average link agglomerative clustering is used to cluster the supplier data.

#### Agglomerative hierarchical clustering

1) Start with the independent cluster with level 0 and sequence is zero
2) If the clustering method is single link the the minimum distance between elements of each cluster, say pair \((r),(s)\) is calculated using Equation 3.9:

\[
SLD = \min \{distance(r,s) : r \in C1, s \in C2\} \quad (3.9)
\]

where \( SLD \) is single link minimum distance between cluster points. \( r \) and \( s \) are data points. \( C1 \) and \( C2 \) are clusters. If the clustering method is complete then the maximum distance between elements of each cluster, say pair \((r),(s)\) is calculated using Equation 3.10.

\[
CLD = \max \{distance(r,s) : r \in C1, s \in C2\} \quad (3.10)
\]

where \( CLD \) is complete link maximum distance between cluster points. \( r \) and \( s \) are data points. \( C1 \) and \( C2 \) are clusters.
3) Join cluster \((r)\) and \((s)\) combined to form new cluster in the next level clustering. The sequence number is incremented by one.
4) Remove the old distance matrix value of \( r \) and \( s \) and update new distance matrix. The new cluster distance is calculated using Equation 3.11:

\[
distance(k,(r,s)) = \min \{distance([k),(r)], distance([k),(s)])\} \quad (3.11)
\]

where \( k \) is old cluster points. \( r \) and \( s \) are new cluster data points.
5) All points are merged into one cluster then stop the iteration, otherwise repeat step 2 to step 5.

### 4. Results and discussion.
The general goods of supplier logistic data are taken from the internet for experimentation. The goods are imported from foreign countries. The parameters consider for clustering are distance in KM, transportation cost and time to taken to transport. The total observation in the data set is 240. The suppliers are grouped based on the distance, time and cost using hierarchical clustering and K-means clustering. In hierarchical clustering with complete and average link is used to cluster the supplier data. The buyer can select the optimal supplier to import the goods from foreign countries based on the clustering results.

Various methods such as silhouette method (Fig. 4.1), gap statistics (Fig.4.2) and sum square (Fig.4.3) methods are used to find the number ideal number of cluster for hierarchical and K-means clustering. The analyses of methods concluded that the ideal number of cluster is 3. The range of silhouette chart is shown in Fig.4.4. It shows that reasonable structure has been formed.
4.1. Hierarchical Clustering. In Hierarchical clustering, complete link and average link is used for clustering. The aggregation is presented in Fig. 4.5. In complete link clustering with normalized value, the aggregate means shows that the distance is positive value in cluster 1 and 2. So, it contributed more for clustering. But in cluster three cost contributed more than distance and time. Similar results for original data and average link also. Both complete linkage and average linkage are given the same result but complete linkage improves 20% distance than the average linkage. The performances show that K-means and hierarchical clustering algorithm improves the distance, cost and time 20%, 12% and 10% respectively.

4.2. K-means clustering. The performance of K-means algorithm with different K is shown Fig. 4.6.

5. Conclusion. The clustering algorithm is proposed and implemented to optimize the shipping cost from the supplier. The supplier logistic data is taken for clustering. The Z-score normalization has been used for normalizing data. It converted the data into uniform scale. The K-means and Hierarchical clustering algorithm with complete link and average link for clustering is implemented and evaluated. Outcome of this work can
Fig. 4.3. Total width Sum of Square

Fig. 4.4. Range of Silhouette Chart

Fig. 4.5. Performance aggregation of Hierarchical clustering
help the buyers to select the cost effective supplier for their business requirements. In future, more parameters taken for clustering to optimize the supply chain management. The clustering algorithm can be used for all divisions of supply chain management. In future mean-shift clustering algorithm and expectation-maximization algorithm will be used for clustering.

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ACOUSTIC FEEDBACK CANCELLATION IN EFFICIENT HEARING AIDS USING GENETIC ALGORITHM

G. JAYANTHI∗ AND LATHA PARTHIBAN †

Abstract. Many people are distracted from the normal life style, because of the hearing loss they have. Most of them do not use the hearing aids due to various discomforts in wearing it. The main and the foremost problem available in it is; the device introduces unpleasant whistling sounds, caused by the changing environmental noise, which is faced by the user daily. This paper describes the development of an algorithm, which focuses on the adaptive feedback cancellation, that improves the listening effort of the user. The genetic algorithm is one of the computational technique, that is used in enhancing the above features. The performance can also be compared with other comprehensive analysis methods, to evaluate its standards.

Key words: Hearing aids, Acoustic feedback, Feedback cancellation, Genetic algorithm

AMS subject classifications. 68W25

1. Introduction. People with hearing loss, struggle to lead a normal healthy life, is considered to be 13% of world’s population. Among them, hardly 20% will ever purchase a hearing aid. While of 32%, may not wear hearing aids, though it is purchased. This is due to irritation and unpleasant whistling, caused by the changing environmental noise, when the user enters into more noisy background. He is forced to hear a sudden and unpleasant sound under which the hearing becomes more complicated and restricts them from moving into a noisy background.

This irregular use of hearing aids leads to a variety of problems. The hearing aid system used for the hearing impaired people should be more natural, humanized and personalized as they likely to prefer the most suitable scheme for their own hearing compensation strategy [2].

Based on the degree of hearing threshold (Mild, Moderate and Severe), the type of loss (Conductive or Sensori-neural) is determined. The hearing thresholds are measured at each octave as 250/500/1K/2K/4K/8K in a standard audiogram and the hearing loss is evaluated using pure tone audiometry [3]. Hence the hearing loss is characterized by the high thresholds in an audiogram is found out using the electro-acoustic characteristics, shown in Figure 1.1. To match the audiogram at high frequencies, large number of bands are to be used, which can be formed as the membership functions by the trial and error process, which is dealt later.

Hearing aid is an assist device, can be fitted in anyone or both the ears, which aids in hearing. The inner structure of the hearing aid is shown in Figure 1.2b. The hearing aid style is chosen with a particular degree of hearing loss, that is been fitted to users . Among them, the bimural hearing aids offer more flexibility, contains all basic functions and much affordable. While the open-fit hearing aids have modified ear tips. The Receiver-in-canal is one of the open-fit hearing aids, in which the receiver is separated from the housing and kept at the ear tip (Figure 1.2.a).

2. Occlusion Effect. The occlusion effect is of occluding earmold or shell which leads to the unnatural, hollow sensation when the user is talking. Similarly annoying sensation of amplification called amplclusion. The utilization of vents of appreciable size is the most common and popular recommendation for hearing aids, dealing with the occlusion effect. Large vents are used to shunt a portion of low-frequency signals to the environment, enabling to remove all or part of the disturbing low-frequency own-voice elements. Whenever

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occlusion effect problems occur, the vent shortening technique is recommended for use [8]. The occlusion effect is the perception of unnaturalness of the user’s own voice, which makes them feel like talking in the barrel.

3. Acoustic Feedback. Feedback is a most common problem, which occurs in hearing aids as whistling or howling sounds. Normally the sound traverses from the microphone to the receiver, crossing the processing circuit Figure 3.1. When this processed sound after reaching the receiver, reverts back to the the microphone, primarily through air, which creates the feedback path. This path leads to the hearing of unnecessary sounds and makes the hearer uncomfortable. These may occur sometimes through the bone and the soft tissues that
Feedback that occurs in ear is known for Acoustic Feedback, which occurs when the high frequency tonal sounds emitted by hearing aids. It happens only when the amplification factor of the forward path signals are much greater than the attenuation factor in the feedback path, by which the hearing aid is lead into oscillation. When the amplification factor is set just closer or below the attenuation factor, a sub-ocillatory feedback is generated. This includes the peaks and valleys in the frequency response of the hearing aid, which degrades the speech quality and intelligibility.

The feedback effect is cancelled in many hearing aids, by reducing the frequency-specific gain when the output level exceeds the frequency-specific threshold. Also this happens when the closed loop gain exceeds unity for the corresponding frequency, at that particular time. A popular LMS filtering is used to cancel the signal components with specific correlation properties.

The main sources of feedback can be due to talking, chewing or yawning, etc. In multichannel hearing aids, a wide range dynamic compression method is used. Figure 3.2. This is done by providing more gain for lower level sounds and minimum gain for higher level sounds. But still the feedback limits the amount of gain that can be provided by a hearing device. Additionally, the presence of feedback in a hearing aid limits the style of wearing, which makes the user to avoid wearing the device, particularly those who want to conceal their hearing loss.

The general methods to reduce acoustic feedback are: (i) Reducing the gain of forward path at one or more high frequency bands in which feedback occurs (ii) Increasing the attenuation level in the feedback path. This is done by decreasing the vent size to an extent.

4. Literature Survey. Soon-Suck Jarng has divided the noise reduction algorithm into two parts. One is to estimate the environmental noise and the other is to enhance the amplifier to improve voice audibility over the estimated noise. A noise spectrum is calculated from the measured signal during the absence of speech activity, using spectral subtraction method. In this paper, it is found that whenever the noise spectrum is estimated to be too steep, some musical noise is generated and if it is estimated to be too low, the input noise speech is distorted [16].

A.H.Kparsi proposed a novel noise power spectral density estimator for binaural hearing aids operating in a diffuse noise field environment which can work for colored noise in a diffuse noise field. Although it does not require a voice activity detection, the noise power spectrum can be estimated irrespective of the speech

---

**Fig. 3.2. Dynamic Feedback Cancellation**
activity. No noise tracking latency is experienced and the direction of arrival of the source speech signal can be arbitrary. The proposed noise estimator can be combined with any hearing aid noise reduction technique, where the accuracy of the noise estimation can be critical to achieve a satisfactory de-noising performance. However the method requires a binaural system, which requires access to left and right noisy signals [17].

Y.F.Chiang has introduced a novel pitch based formant estimation algorithm and an architecture for the adaptive feedback cancellation for the hearing aid applications. The system consists of a peak and harmonic detection in the forward path, includes the decorrelation filter coefficients update and an adaptive feedback cancellation in backward path with a pitch based detector in the forward path. The pitch based formant estimation has four orders lower complexity and the simulation results achieve similar perceptual evaluation speech quality and added stable gain. Still the method provides slight degradation in the speech quality without voice activity detector. It also shows small and high frequency artifacts in the beginning of each sentence [18].

Some techniques project on the adaptive gain reduction algorithm method (i), in which the amount of gain reduction is variable, depending on the magnitude of feedback signal. This is only for low level signals and not for all input signals, which in turn, the gain of the desired signals are reduced. Also from method (ii), reducing the size of the vent may lead to occlusion effect, which may introduce another major problem.

The alternate method for adaptive feedback cancellation is the usage of notch filters. Whenever a whistling signal is detected, the algorithm generates sharp notch filter in order to suppress it. The notch filters are introduced at the feedback frequencies with the center frequency matching the peak frequency of feedback signal.

In a circuit, when only a few number of notch filters are generated by the algorithm to control the distortion in the acoustic environment, a frequency-hopping artifact is generated, as there is always an uncovered whistling feedback signal [12]. The notch filter method generally limits the gain reduction at the feedback frequencies.

Many companies offer several types of products from analog hearing devices (which has only fixed gain amplifier) to the digital based signal processing devices. Whenever there is a spectral overlap between the signal and noise or if the band occupied by the noise is unknown or varies with time, the use of conventional filters would lead to unacceptable distortion of the desired signal.

5. System Design. In many instances, the filter characteristics needed to be variable, adapted to the signal characteristics or to be altered intelligently. In such cases, the coefficients of the filter must be varied and cannot be specified in advance. The feedback cancellation method preferred is through monitoring the transfer function of the feedback path. The transfer function of the signal is found, which is generated similar to the feedback path. Then the generated signal is subtracted from the output signal of the microphone circuit, so that the algorithm modifies the characteristics of the digital filter circuit. In such a way, the transfer function should have the similar frequency-amplitude-phase relationship corresponds to the feedback path.

A low-level noise signal is injected into the feedback path of the algorithm to estimate the transfer function of the corresponding signal, which is processed in multiple stages. Hence an adaptive filter is used in addition to a fixed filter. Initially, the noise signal that is sent to the receiver circuit of the hearing aid, which is then cross-correlated with the source output. This in turn estimates the transfer function of the feedback path. Thus the modelling of the feedback signal generates the cancellation signal. Generally a howl detector is used to monitor the changes in the feedback path above a particular frequency, refer Figure 5.1. A combination of adaptive and fixed filters are used to reduce the dynamic component of the feedback signal. A linear adaptive filter is chosen to control the variable parameters of the transfer function and those parameters can be adjusted by a suitable optimization algorithm.

The two major types of linear adaptive filters are (i) Least Mean Square(LMS) filter - reduces Mean Square Error while the input signals are stochastic. Under stationary environment, the error characteristics of the LMS filter has constant shape and orientation and makes the filter characteristics to converge at or near the optimum point. The signal statistics (mean, variance and autocorrelation) are changed, once the weight values are converged, leads to readjustment of filter characteristics to the new optimum value.

Under non-stationary environment, the orientation and curvature of the error characteristics keep changing and make the filter characteristics to be converged slowly. This shows that the algorithm performs only for the seeking of the minimum point of the surface but tracking of changing position is also important to improve the performance of filter. This clears out that the LMS filtering is not suitable for real time or on-line filtering [13].
The Recursive Least Squares (RLS) filter minimizes the Weighted Linear Least Squares Cost Function. Also the input signals are deterministic and extremely Fast Convergence. The RLS filter is sensitive to computer round-off errors and eventually leads to instability.

5.1. Power Spectrum Estimation. The input noise voltages and the feedback signals generated in the hearing aid devices, can be modeled as random processes as most of the natural phenomena are characterized in terms of statistical averages. Random processes can be characterized by the estimation of spectral characteristics of signals [14].

Since the fluctuations in the signals are very random, the statistic characteristics of those signals are adopted. To characterize the random signals in time domain, the auto-correlation function is determined to yield the appropriate statistical average. The power density spectrum is obtained from the fourier transform of autocorrelation function, in order to transform the signals from the time domain to the frequency domain.

The estimation of power density spectrum of a signal from the observation of signal over a finite time interval is the basic problem to be considered as the short duration estimation limits the quality of power spectrum estimate. For statistically stationary signals, estimate is better for longer data record.

But when the signal statistics are non-stationary, it is difficult to select an arbitrarily long data record for the spectrum estimation. Therefore selecting a data record, as short as possible, allows to resolve the spectral characteristics of different signal components that have closely spaced frequency components.

To limit the duration of impulse response, windows are used in the FIR filter design. To say, to compute the spectrum of the signal, for the finite-duration sequence of $x(n)$ is $0 \leq n \leq N - 1$.

Convolution function smooths the spectrum $X(f)$, only when the window spectrum $W(f)$ is relatively narrower than $X(f)$. Rectangular windows smooth time-domain cut-off in addition they have side lobe leakage. Blackman window reduces lobe leakage but spectral width increased to 50% as it should be closely spaced spectra for a better estimate. That is, smooth time-domain windows reduces leakage, which in turn decreases the frequency resolution.

The finite-energy signals possess a fourier transform and are characterised in spectral domain by their energy density spectrum. Since the stationary random processes possess finite average power and they are characterised by the power density spectrum.

The leakage and frequency resolution problems provide the motivation for the power spectrum estimate methods based on non-parametric methods. This makes no assumption of data generation and hence the estimate of power spectrum is obtained either through averaging or smoothing operations, which is performed directly on the periodogram or on the autocorrelation.

Bartlett method is a windowing technique based on averaging periodogram, used for reducing variance in the periodogram. The N-point sequence is sub-divided into L non-overlapping segments for each segment length is $M$. 

![Fig. 5.1. Adaptive Feedback Algorithm](image)
Table 5.1
Comparison of Welch with Bartlett

<table>
<thead>
<tr>
<th></th>
<th>Bartlett PSE</th>
<th>Welch PSE</th>
</tr>
</thead>
<tbody>
<tr>
<td>Quality factor</td>
<td>( Q_B = 1.0560 )</td>
<td>( Q_W = 0.7488 ), for no overlap</td>
</tr>
<tr>
<td></td>
<td>( = 1.3344 ), for 50% overlap</td>
<td></td>
</tr>
<tr>
<td>FFT length</td>
<td>( M = 15.0 )</td>
<td>( M = 21.3333 )</td>
</tr>
<tr>
<td>No. Of FFTs</td>
<td>( \frac{N}{M} = 1.0656 )</td>
<td>( \frac{N}{M} = 1.4976 )</td>
</tr>
<tr>
<td>No. of computation</td>
<td>( \frac{N}{2} \log_2 \frac{M}{2} = 468.864 )</td>
<td>( N \log_2 \frac{M}{2} = 666.8278 )</td>
</tr>
</tbody>
</table>

The periodogram is computed as:

\[
P_{XX}^{(i)}(f) = \frac{1}{M} \left| \sum_{n=0}^{M-1} x_i(n) e^{-j2\pi f_n} \right|^2 \quad i = 0, 1, \cdots, L - 1
\]  
(5.1)

Hence the Bartlett power spectrum estimate is obtained by averaging the periodograms for the \( L \) segments,

\[
P_{XX}^B(f) = \frac{1}{L} \sum_{i=0}^{K-1} P_{XX}^{(i)}(f)
\]  
(5.2)

A window is obtained by reducing the length of data from \( N \) points to \( M = N/L \), and the spectral width is increased by a factor of \( L \) and frequency resolution is decreased by factor \( L \), in turn variance is reduced. Bartlett power spectrum estimate is asymptotically unbiased.

Welch method is based on averaging modified periodogram, originally derived from Bartlett window method. Two modifications made in Bartlett technique are: (i) Welch method allows the data segments to overlap, which is not done in Bartlett method. (ii) It windows the data segments prior to compute the periodogram, which results in Modified periodogram.

\[
P_{XX}^{(i)}(f) = \frac{1}{MU} \left| \sum_{n=0}^{M-1} x_i(n) w(n) e^{-j2\pi f_n} \right|^2 \quad i = 0, 1, \cdots, L - 1
\]  
(5.3)

Hence in Bartlett method, the power spectrum estimate is computed by averaging the periodograms for the \( L \) segments,

\[
P_{XX}^W(f) = \frac{1}{L} \left| P_{XX}^{(i)}(f) \right|
\]  
(5.4)

Overlapping of the data segments can be varied for more or less than 50\%, in order to improve the relevant characteristics of the estimate [14]. On comparing the quality of Bartlett and Welch power spectrum estimates are given in Table 5.1.

The results obtained from Welch method is comparatively better than the Bartlett estimate. However the differences in performance are relatively small. The quality factor depends on the product of data length \( N \) and the frequency resolution \( \Delta f \). From these results, it is concluded that Welch method requires a little more computational power than the Bartlett method. However the differences in performance are relatively small [14].

Characterization of the ambient noise has been carried out using the Power Spectrum Welch method (Figure 5.2). Simulation was carried out by sending the input signal with the noise data collected from the acoustic environment and used for denoising purpose.

The power spectrum estimation methods provide no idea about the data generation, when it is non-parametric. Bartlett windowing technique is one of the method, used to derive the power spectral estimation of a given signal. It provides smooth magnitude response in both passband and stopband, compared to other windowing techniques. Under which, the Welch method uses averaging modified periodogram technique.
6. Data Processing. The DSP based modern hearing aids have numerous parameters and fitting it, is a challenge for the audiologist after a comprehensive analysis. In order to select the algorithm that uses fewer resources, is implemented on the hearing device, based on low cost DSP. The vital feature of the assistive device, is to improve the signal to noise (SNR) under any type of background noises. The another important factor is to determine the memory requirements for the noise reduction methods to be used [6]. The DSP based hearing aids, functions for voice compression, noise reduction and dynamic equalization tasks to produce better results in hearing operations.

The digital processors process the incoming sounds received by a person with a particular hearing aid. The main features of a HA of a family are: (i) soft sound amplifier, which allows the user to perceive soft sounds related to weak or distant speech; (ii) Audibility extender is a vital function, which enables the user to hear sound, which is spectrally placed into the band where he has a problem. Also it is necessary to improve the convenience of the user. Hence the SNR ratio of the output signal is maximized, so that the sounds perceived by the person can be heard more naturally [6]. The hearing aids undergo classification of sounds signals and adaptation of signals among the noisy or quiet environments.

The filter coefficients of the FIR filter are obtained by Genetic Algorithm (GA) using the optimization procedure. The error signal is obtained by comparing the desired and the actual magnitude response, which is then minimized by GA. Also it is used to customize the specific subband in a particular octave in the hearing range of the hearing aid user.

7. Generic Operators. Genetic Algorithm (GA) is a class of probabilistic optimization algorithm and uses the concepts of ‘Natural selection’ and ‘Genetic Inheritance’, which is inspired by the biological evolution process. GA is simple, robust and generates Membership Functions. Hence it is more suitable for the complicated problems with little or unknown solution [4]. The flowchart of Genetic algorithm is shown in Figure 7.1.

The acoustic feedback available in the hearing circuit, is controlled or cancelled by varying the parameters, for fitting the system using genetic algorithm. To integrate the system of GA with perceptual feedback, an efficient search can be performed in a perceptual space is described by Eric A.Duant, etal [5]. This focuses on the selected features to be programmed on the DSP, to limit the computational facilities effectively. The algorithm parameters are used to control the processed incoming sound by the multitude of algorithm, which is then presented to the user.

The synthesizing of digital IIR filters using the real coded genetic algorithm (RCGA), is based on the real number representation of the chromosomes. The usage of real numbers allow us to extend the search domain by reducing the precision which has also the capacity to fine tune the solutions. Therefore comparing to binary coded strings, real numbers have more benefits in RCGA [7].

The RCGA is applied to solve the synthesis of digital filter whose magnitude response matches the human audiogram. The IIR filter exhibits significantly better performances than the FIR filter in approximating the
required magnitude characteristic and in minimizing the resources necessary for the implementation [7]. The genetic algorithm is used with restricted search, is described by Enrique Alexandre [1], is explored for the above mentioned feature selection. The restricted search driven by the implemented GA performed better than both sequential methods and unconstrained genetic algorithms.

In general, optimization problems can be classified under (i) Combinatorial optimization problems in which Ant Colony optimization technique is used and (ii) Continuous optimization problems in which Particle Swarm Intelligence is followed. The firefly algorithm is an another nature-inspired algorithms in GA, which is based on the concept of selecting the mating pairs under the point of attractiveness.

The GAs include the natural selection, mutation and other biological concepts, which can be applied to non-biological optimum based search problem. It generally starts from an initial population and using the evolutionary mechanism, the new members are produced by selection, crossover and mutation, shown in Figure 7.1.

The selection process is best performed by using Roulette wheel (RW) method, which is the most preferred fitness technique. The chromosomes that occupy larger area of the wheel and has more chances to be selected. It provides the area which is proportional to the fitness value. In RW selection, if the best chromosomes has the fitness of about 90%, compared to other chromosomes, then all other will get very few chances to be selected. After performing selection, the new population can be obtained.

One or two genes are changed randomly by the mutation operator randomly with a predetermined probability. It yields insignificant results, when the mutation probability is low and it conducts the divergence of algorithm, when it is too high. Thus fixing the probability of mutation is considered as crucial. Concluding the experiments only with crossover operator than the mutation is of course more effective.

The three different types of cross-over operators is shown through:
(i) single point crossover: the binary string value is copied from the beginning of the chromosome to the crossover point of one parent and the remaining is copied from the second parent. (Figure 7.2.a);
(ii) linear or uniform crossover: the binary bits are randomly copied from first or the second parent (Figure 7.2.b)
(iii) intermediate crossover: the binary string is copied from the beginning of chromosome of one parent to the
first crossover point, and the another part from the first to the second crossover point is copied from the second parent while the remaining string is copied from the parent on (Figure 7.2.c).

Therefore, in one generation, the total population fitness is changed and the same process is carried out again and again, until a stopping criterion is met. GA is based on Darwin’s theory: the best ones should survive and propagate, based on their fitness. GA is used to learn or tune all the parameters of a network and its chromosomes encodes all the parameters of the model.

8. Simulation Results. The files correspond to speech in noise and speech in silence are considered under diverse environments. The data base contains certain number of files for training, validation and testing respectively. The speech signals are recorded from sources, with variable reverberational grades. Noise signals from stationary and non-stationary environments are collected and fed as a noise source. After following the perceptual criteria, signals with low SNRs are considered as noise and higher SNRs as clean speech signals [15].

To get more precise results, the input signal is divided into number of frames, with no overlap and with 50% overlap between the adjacent frames. The the modified periodogram using welch method is computed. Hence the mean and the variance are calculated in order to mitigate the values. The speech signal mixed with noise signal (here bus horn sound is considered as noise) is shown in Figure 8.1(a) and the noise is removed from which the clean speech is obtained, is shown in Figure 8.1(b). Matlab software is used to test the speech in noise and speech in silence.

The GA solver algorithm iterates the above steps, until a maximum number of generations are reached or till the lowest error value remains unchanged for a given number of iterations. The values obtained are found to be sufficient for the algorithm to converge properly.

The mixed signal (speech and the background noise signals) is given to the processor as input wave files. The algorithm which is used to reduce the noise level is stored in the specific processor (TMS320C6745) using Code

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**Fig. 7.2.** Types of Crossover (a) Single Point Crossover (b) Linear Crossover (c) Intermediate Crossover

**Fig. 8.1.** (a) Speech with Bus Background Noise (b) Speech without Bus Background Noise
Composer Studio (CCS). Based on this noise level, the volume of the hearing aid will be automatically adjusted by using GA solver in optimization toolbox. The processor will provide the desired result (speech signal) based on the input given to it. The fitness function can be created and minimized using genetic algorithm through the global optimization toolbox. The GA solver assumes one input $x$, where $x$ is a row vector elements as that of number of variables in the problem.

The value of the function is computed by the fitness function and it returns that scalar value in its one return argument $b$. It is minimized using GA by passing in a function that handles fitness function and also by mentioning the number of variables present in that problem. When there is average change in fitness value less than the options, then the optimization is terminated. The $x$ value returned by the solver is the best point in the final population in genetic algorithm as shown in Figure 8.2.

To vectorize the fitness function, the GA solver is passed at one point of it. This is to speed up the process, by taking set of points and returning set of function values. Therefore in the fitness function, the vectored version of matrix $x$ with an arbitrary number of points as that of the rows of $x$ and it returns the column vector $y$ with the same number of rows $x$. When the average change in fitness value is less than the options provided, then the optimization is terminated and the fitness graph is shown in Figure 8.3.

9. Conclusion. The main intent of the paper is to address the issues related to Acoustic feedback effect in digital hearing aids and the possible methods available for cancellation. On the suitable database implementation, to evaluate the proposed algorithm, optimization tool is implemented to test the speech in noise and speech in silence in the acoustic environment. This paper presents an efficient sound signal separating from noise signal and the algorithm with very low computational cost for the speech in background noise for the
hearing aid user. The genetic algorithm is useful in optimizing the bit distribution and executes by variation in the bit rate[9]. In order to improve the effectiveness of the proposed algorithm, the performance can also be validated by uniform distribution of bits technique. Improving the speech intelligibility, increases the comfort of the user and allows him to lead a normal life.

The design and the results obtained from the matlab simulation show the accuracy of hearing loss compensation and find the optimal coefficients. The proposed algorithm yields high levels of performance for different noises and SNR conditions but still it is not much sensitive to changes in the noise level. Tuning the algorithm with speechrecorded in more diverse noise environments. Optimizing the DSP compatible ANSI-C code of the algorithm to ensure faster execution time. A certain number of listeners are required to justify the recordings containing the processing results to be tested, based on the features like Quality, Comprehension difficulty and Audibility of clipping.

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ANALYSIS ON DEEP LEARNING METHODS FOR ECG BASED CARDIOVASCULAR DISEASE PREDICTION

KUSUMA S.* AND DIVYA UDAYAN J.†

Abstract. The cardiovascular related diseases can however be controlled through earlier detection as well as risk evaluation and prediction. In this paper the application of deep learning methods for CVD diagnosis using ECG is addressed and also discussed the deep learning with Python. A detailed analysis of related articles has been conducted. The results indicate that convolutional neural networks are the most widely used deep learning technique in the CVD diagnosis. This research paper looks into the advantages of deep learning approaches that can be brought by developing a framework that can enhance prediction of heart related diseases using ECG.

Key words: Deep learning, Python, CVD, ECG.

AMS subject classifications. 68T05

1. Introduction. According to statistics by the World Health Organization, each year 17.9 million deaths occur worldwide, all as a result of Cardiovascular Diseases [1]. Being a major concern requiring urgent attention, numerous methods and researches have been carried out in order to establish the most favorable approaches in this very crucial area of medicine. Most of these cardiovascular related diseases can however be controlled through earlier detection as well as risk evaluation and prediction. As the health systems revolve, the amount of data generated from the field is expanding vigorously. The data obtained is then used to derive more knowledge in the field as well as improvise different and innovative approaches necessary for the improvement of health services provided.

In what can be termed as the information age, research on the methods that can be used in the timely diagnosis are extensive. Forecasting is important because it eases the control of the rate and probability of someone suffering from the disease and at the same time makes it possible to reinforce the necessary measures in order to control the disease. Risk prediction of heart diseases is important because it aids in decision making on lifestyle choices and the pros and cons of some habits such as smoking, obesity and alcohol consumption. All depending on the vulnerability of the individual in getting a heart disease. Early diagnosis on the other hand makes treatment cheap which can reduce the complications as well as to reduce the cost of treatment. The revolution of the healthcare systems is very rapid. For a long time, scientists have improvised automated methods which have made forecasting of the risk easy and more accurate. Many researchers developed deep learning algorithms for ECG arrhythmia classification [2-9]. The computation efficiency and speed achieved through convolutional neural network for an individual patient [10].

In the field of medicine An electrocardiogram (ECG) is one of the most recognized and painless test that detects electrical activity in the heart using sensors that are placed on the skin over the chest [11]. Mainly there are three main parts to an ECG as in the Fig 1.1. The depolarization of the atria constitutes P wave and it is a small semi circular shape; Depolarization of the ventricles represents the QRS complex; and Repolarization of the ventricles constitutes the T wave. The rhythm of the heart in terms of beats per minute (bpm) can be easily calculated by counting the R peaks of the ECG wave during one minute of recording [12].

By taking a pulse into consideration heart rate can be calculated. If any issues related to heart damage, size of the heart chambers or abnormal heart beat then ECG is suggested. Heart rate calculation can be done in following ways as per the article[13]:

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1. Make use of distance across QRS complexes to calculate the heart rate.
   ECG is to assess the rhythm and heart rate, normally consists of small and large squares. For reference, usually large squares are used and in each large square 5 little squares are placed. The QRS complex is the tallest part on ECG. When one QRS complex transpires, it is an indication of one heartbeat has taken place. For example, let us take into 3.2 number of big squares separates adjacent QRS complexes. To determine heart rate the following calculation is used: $300/3.2=93.75$, which is normal as in table 1.1.

2. Second method.
   30 large squares on an ECG trace represents exactly 6 seconds, multiply by 10 i.e. $6 \times 10=60$ seconds will give the number of heartbeats occurred in one minute called beats per minute (BPM). For example, 7 beats counted in the period of 6 seconds then heart rate calculation is $7 \times 10 = 70$ BPM as shown in Fig. 1.2.

Atrial fibrillation is the common abnormal rhythm, replaces the normal heartbeat with an erratic pattern. Bradycardias is the Slow heart rhythms such as tachy-brady syndrome, atrioventricular (AV) heart block and bundle branch block. Tachycardia is a fast heart rhythm and includes:
   - ventricular tachycardia (VT)
   - atrial fibrillation (AF)
   - supraventricular tachycardia (SVT)
   - inappropriate sinus tachycardia
   - atrial flutter
   - ventricular fibrillation (VF).

Table 1.1 represents the normal resting heart rates at different ages according to the United States National Institutes of Health (NIH) [14].

Deep learning is a concept-oriented feature in the IT. It is instrumental in data prediction. The major obstacle is encountered when choosing the right prediction tool with the highest precision. Risk prediction has in numerous occasions been done manually. The process is however not so smooth due to several contributing risk factors that come with the disease. These factors include diabetes, hypertension and abnormal cholesterol levels. In most cases, researchers within the field of deep learning are focused on developing new data analysis methods. Not much has been done to determine the factors that affect the analysis and prediction of the results.

Deep learning methods have a wide application in the medical field. In this case, medical diagnosis is
Conducted through use-cases of deep learning networks. These include detection, segmentation, classification, prediction and other. Based on the results of the reviewed studies indicate that deep learning methods can be far superior in comparison to other high-performing algorithms. Therefore, it is safe to assume that deep learning is and will continue to diversify its uses [15].

Some of the sample public datasets features related to ECG signal consists of Age, Sex, Resting blood pressure (RBP), Cholesterol levels in the serum, chest-pain type, ST depression, Thal, Blood sugar, Maximum heartbeat rate achieved and risk levels of heart disease.

This paper is organized into three sections. First, an overview of Deep Learning and Python, next a description of research methodology, then review of deep learning in CVD diagnosis; finally, discussion on results, conclusions and future directions for development.

2. Deep learning and Python. Artificial Intelligence (AI) is a technique that emphasizes the development of intelligence machines, thinking and working like human cognitive behaviour. The model of an artificial neuron which takes an input, processes it, passed it through an activation function like the Sigmoid, return the activated output as in the Fig. 2.1.

Deep learning or deep neural network (DNN) is a subset of machine learning in Artificial intelligence that uses multiple layers for feature extraction from the input layers as represented in Fig. 2.2. The deep learning is so important because of its effectiveness on huge amount of data and ability to process large numbers of features when dealing with unstructured data. DNN achieved better results in some fields such as bioinformatics, computer vision, natural language processing (NLP), machine translation, speech and audio recognition.

The objective of a neural network is to get inputs, then to perform calculations on and to produce output to solve real world challenging problems like time series, classification of ECG signals etc. For image or object recognition Convolutional neural networks produces good results. The popular deep learning algorithms are:

- Convolutional neural networks (CNN)
- Recurrent neural networks (RNN)
- Auto-Encoders (AE)
Deep Boltzmann Machine (DBM)
- Long Short-Term Memory Networks (LSTMs)
- Deep Belief Networks (DBN)

The challenges for deep neural networks are computation time and overfitting. The hyper parameter such as Batching process can speed up computation process. Deep neural networks consider few parameters such as initial weights, learning rate and size to tilt overfitting some of the Regularization methods such as transfer learning, drop out, data augmentation and early stopping are applied.

Now a day’s data science widely uses a popular programming language called Python to construct deep learning algorithms. To develop deep learning models for prediction, classification, understanding, perception, creation Basic knowledge towards statistics, Linear algebra, calculus, Machine learning techniques and python libraries such as Matplotlib, Numpy, Pandas, Scipy, Neural network frameworks like Keras, Tensor Flow, Theono, Pytorch is required.

Theano is a python library introduced by Yoshua Bengio for building deep neural networks to train fast on machine. It builds the model with a highly optimized solution. The libraires Tensor flow and keras as backbone can be used with theono to expand the functionality.

Tensorflow is a Googles python library based on computational graphs applied for deep learning commercial grade applications. Using TensorFlow different types of deep neural networks like Autoencoders, Recursive Neural Tensor Network (RNTN), Recurrent neural network (RNN), Restricted Boltzmann Machine (RBM), Deep Boltzmann Machine (DBM), Multilayer perceptron (MLP) etc. can be built.

Keras is a portable python library with efficient numerical computation and high level neural network application programming interface (API) which allows us to define, train, developing and evaluating models of neural network with minimal lines of code.

Scikit-image is used for image processing and for statistical algorithms and data exploration stats models can be used. PyTorch is another open source library for machine learning mainly used for natural language processing and computer vision. It also includes a feature of deep neural networks built on a computational differentiation system.

Bioppy is a biosignal processing in python used for various biosignals such as Respiration, Electrocardiogram (ECG), Blood Volume Pulse (BVP), Electro Dermal Activity (EDA), Electroencephalography (EEG), Electromyography (EMG) for filtering and frequency analysis [16].

Deep Learning (DL) with python has the potential to change the future on medical imaging and diagnosis in achieving good results comparatively with the other traditional methods.

The general research methodology for deep learning in health care for disease detection and prediction includes the following Phases as represented in Fig 2.3:

Phase 1: Understanding dataset is the main step during the inception phase to avoid the problems in the development stage.

3.1. Review Process. The review process is done according to the PRISMA flow diagram and protocol [17]. The diagram divided into four sections as shown in Fig 3.1. Firstly, identification of articles in the related field, followed by eliminating the duplicate articles in the screening process. Then the eligible articles are considered for review. Finally, selected articles are considered for the study.

3.2. Literature sources and process of data collection. In order to identify the results, we used the keywords Deep learning and Cardiovascular disease, Deep learning and ECG. Mainly two literature sources were used namely, IEEE digital library, PubMed [18] and dblp computer science bibliography [19] from 2016 to till date. The articles from the mentioned databases are analyzed to get the information related to the deep learning methods for cardiovascular disease prediction. We undergone the three stages to collect the data from the sources.

Stage 1: Searching articles related to the keywords specified above.
Stage 2: Analysed the literature and excluded which are not related to research.
Stage 3: Finally at the result section presented the qualitative data such as type of data sets, deep learning methods.

Fig. 3.1. Review process flow diagram
4. Review Results. The performance analysis is done over several state-of-the-art models through the Type I measures such as positive measures like Accuracy, Sensitivity and Specificity. The main terms considered patient as positive sign for disease and healthy person as negative sign for disease [18].

Accuracy is to test its ability to correctly dive into the samples of healthy persons and patients. This can be stated mathematically as:

\[
\text{Accuracy} = \frac{TP + TN}{TP + TN + FP + FN}
\]

Sensitivity test is to determine the patient samples correctly through the proportion of true positive in patient samples. This can be stated mathematically as:

\[
\text{Sensitivity} = \frac{TP}{TP + FN}
\]

Specificity test is to determine the healthy samples correctly through the proportion of true negative in healthy samples. This can be stated mathematically as:

\[
\text{Specificity} = \frac{TN}{TN + FP}
\]

where \( FP \) stand for “false positive” which indicates the number of samples incorrectly identified as patient; \( FN \) for “false negative” which indicates the number of samples incorrectly identified as healthy. \( TP \) for “true positive” which indicates the number of samples correctly identified as patient; \( TN \) for “true negative” indicates the number of samples correctly identified as healthy.

5. Proposed methodology. The proposed model for heart disease prediction using ECG signals includes three major phases: Pre-processing, Feature Extraction, and Classification. In the first phase, both filtering process and heart beat detection takes place. After pre-processing, feature extraction will be carried out, from which the optimal features will be selected using a new Algorithm. Subsequently, the optimal features will be subjected to classification process, where the Optimized Neural Network (NN) will be used to predict the presence of heart disease. Further, to make the prediction model more accurate, the training of NN will be done using the proposed algorithm via selecting the optimal weight. Moreover, the proposed improved algorithm deals with the adaptive mutation process as the conventional algorithm concentrates only on the static mutation process [42]. The adaptiveness in proposed algorithm ensures the optimal tuning more appropriate. The overall architecture of the proposed scheme is demonstrated by Fig. 5.1.

The proposed heart disease prediction system using ECG signals will be implemented in Python, and the experimental investigation will be carried out. The performance analysis will be done by comparing the proposed model over several state-of-the-art models through the Type 1 measures and Type 2 measures. Here, Type I measures are positive measures like Accuracy, Sensitivity, Specificity, Precision, Negative Predictive Value (NPV), F1Score and Mathews correlation coefficient (MCC), and Type II measures are negative measures like False positive rate (FPR), False negative rate (FNR), and False Discovery Rate (FDR).

![Fig. 5.1. Overall Architecture of the Proposed Heart Disease Prediction Model.](image-url)
## Results of individual articles of deep learning and ECG based Cardiovascular disease prediction

<table>
<thead>
<tr>
<th>Ref</th>
<th>Method</th>
<th>Dataset/Datasource</th>
<th>Outcome</th>
</tr>
</thead>
<tbody>
<tr>
<td>[19]</td>
<td>1D-CNN</td>
<td>MIT-BIH, St.-Petersburg, PTB databases</td>
<td>Achieved accuracy of 97.70, sensitivity 99.71, and specificity 98.24 respectively.</td>
</tr>
<tr>
<td>[20]</td>
<td>LenET-5T CNN</td>
<td>Physionet Apnea ECG dataset, UCD dataset</td>
<td>Achieved accuracy of 97.1, sensitivity 100, and specificity 91.7, respectively.</td>
</tr>
<tr>
<td>[21]</td>
<td>DL-CCANet and TL-CCANet</td>
<td>MIT-BIH database, INCART database</td>
<td>DL-CCANet, achieved accuracies of 95.25 and 94.0, the TL-CCANet achieved the highest accuracy of 95.52.</td>
</tr>
<tr>
<td>[22]</td>
<td>DNN (Deep Convolutional neural network)</td>
<td>576581 ECGs from 449380 patients at Mayo clinic, Rochester Minnesota, from 1994 to 2017</td>
<td>Obtained accuracy of 90.</td>
</tr>
<tr>
<td>[23]</td>
<td>Deep Learning algorithm</td>
<td>Mayo clinic ECG laboratory in September 2018</td>
<td>Achieved accuracy of 86.5, sensitivity 82.5, and specificity 86.8, respectively.</td>
</tr>
<tr>
<td>[24]</td>
<td>Multilayer perceptron and CNN</td>
<td>physiobank.com, Kaggle.com</td>
<td>Achieved accuracy of 88.7 for MLP and 83.5 for CNN.</td>
</tr>
<tr>
<td>[25]</td>
<td>Linear support vector machine (SVM)</td>
<td>MIT-BIH arrhythmia database</td>
<td>Achieved accuracy of 97.77 and 97.08.</td>
</tr>
<tr>
<td>[26]</td>
<td>CNN</td>
<td>PhysioNet database</td>
<td>The performance on overall F1 score reached to 0.82 when algorithm applied.</td>
</tr>
<tr>
<td>[27]</td>
<td>CNN</td>
<td>PhysioNet database</td>
<td>The rhythm classification performance reached an overall F1 score of 0.82 when applying the algorithm to the hidden test set.</td>
</tr>
<tr>
<td>[28]</td>
<td>Deep convolutional neural network</td>
<td>MIT - BIH Arrhythmia database</td>
<td>Deep 1D-CNN achieved a recognition overall accuracy of 17 cardiac arrhythmia disorders (classes) at a level of 91.33 and classification time per single sample of 0.015.s</td>
</tr>
<tr>
<td>[29]</td>
<td>Convolutional neural network (CNN)</td>
<td>Publicly available database</td>
<td>Accuracy of 94.03 and 93.47 with and without noise removal respectively.</td>
</tr>
<tr>
<td>[30]</td>
<td>CNN-BLSTM network model</td>
<td>MIT - BIH Arrhythmia database</td>
<td>Achieved an accuracy of 96.59, a sensitivity of 99.93 and a specificity of 97.03</td>
</tr>
<tr>
<td>[31]</td>
<td>Deep neural network (DNN)</td>
<td>Physio net data and Novel dataset of 328 records from 328 unique patients</td>
<td>Rhythm classes were examined and had higher sensitivity of 94.1 than the averaged cardiologists of 78.4.</td>
</tr>
<tr>
<td>[33]</td>
<td>CNN and LSTM</td>
<td>MIT-BIH arrhythmia database</td>
<td>Accuracy of 99.26 and 94.20 were achieved under class-oriented scheme and subject-oriented scheme.</td>
</tr>
<tr>
<td>[34]</td>
<td>CNN</td>
<td>MIT-BIH arrhythmia database</td>
<td>As a result, classified the ECG beats into five categories with an average accuracy of 99.21.</td>
</tr>
<tr>
<td>[35]</td>
<td>Deep residual convolutional neural network (DR-CNN),</td>
<td>MIT-BIH arrhythmia database</td>
<td>Achieved the average detection accuracy of 99.0349, 99.4980 and 99.3347 for multiclass classification, ventricular ectopic beat (Veb) and supra-Veb (Sveb) recognition, respectively.</td>
</tr>
<tr>
<td>[36]</td>
<td>Deep multi-task learning approach</td>
<td>MIT-BIH arrhythmia database</td>
<td>Improved the accuracy of ECG data analysis by up to about 5.1</td>
</tr>
<tr>
<td>[37]</td>
<td>Extreme Learning Machine(ELM)</td>
<td>MIT-BIH arrhythmia database</td>
<td>This paper has a 1.7 increase, and the sensitivity of the N, S, V, and Q categories are 99.12, 75.2, 93.72, and 98.64, respectively.</td>
</tr>
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Table 4.2

<table>
<thead>
<tr>
<th>Ref</th>
<th>Method</th>
<th>Dataset/Datasource</th>
<th>Outcome</th>
</tr>
</thead>
<tbody>
<tr>
<td>[38]</td>
<td>Transfer learning</td>
<td>Physio net/CinC Challenge 2017 dataset</td>
<td>The method achieved F1 F1 score of 0.89 and 0.86.</td>
</tr>
<tr>
<td>[39]</td>
<td>LSTM deep neural network</td>
<td>400,000 clean PPG (photoplethysmography) samples to learn typical PPG morphology and rhythm</td>
<td>Cardiac abnormality recognition approach recognises 60+ of ECG-detected PVCs in PPG signal, with a false positive rate of 23 - demonstrating the compelling power and value of this novel approach.</td>
</tr>
<tr>
<td>[40]</td>
<td>RNN</td>
<td>Physio net/CinC Challenge 2017 dataset</td>
<td>Achieved the maximum accuracy of 79</td>
</tr>
<tr>
<td>[41]</td>
<td>CNN</td>
<td>MIT-BIH arrhythmia database</td>
<td>Accuracy obtained 99.976</td>
</tr>
</tbody>
</table>

6. Discussion and Conclusion. Heart related illnesses and other cardiovascular diseases are a menace in the society and if they are not eradicated can cause drastic outcomes in the long run. Considering the mortality rate of the patients, this calls for the disease to be a major concern. Manually establishing the chances of getting the disease is very difficult and highly likely to be ineffective. However, deep learning techniques come in handy to solve the problem by evaluating the risk factors using the data presented in order to generate a forecast on the odds of as the output.

The objective of this paper was to study research papers in the field of deep learning methods on ECG based cardiovascular disease prediction. The available public datasets of ECG such as MIT-BIH, INCART, Physionet and Kaggle database. Despite the field of medicine being vast and generating a lot of data every single second, the infrastructure is set in place to store all the data and make it secure and readily available. From the findings from this research it is conclusive to say that Conventional Neural Network which is a method of deep learning has a high accuracy level on large datasets. Python is playing a major role in implementation of deep learning models. This may reduce the risk of late diagnosis by earlier prediction and improves the quality in health care. Deep learning methods in medical diagnosis will not replace practitioners, the development of DL models is to help doctors in early decision making.

7. Limitations and Future Research. Development of deep learning is still underdeveloped. There is still not enough research on the theories and algorithm for use in risk prediction. The accuracy and efficiency of the data mining methods used may be compromised especially when the wrong model is used or when handling large data capacities. There being enough research on complex data, the research may be altered by the gap in knowledge considering the several colliding explanations. It is expensive to build an expert heart disease prediction system.

Having the necessary deep learning tool models is very effective in forecasting. However, it does not really solve the problem at real time. Instead, the use of wearable sensors can most effectively solve the problem through giving feedback in real time. Due to the advancement of technology, the exposure to the internet and social media increases the bulk of data that is created every second. Data mining in the health industry is very essential. Data obtained could be structured, unstructured, complex or multileveled.

Smart phones have changed the health sector due to innovations such as medical apps. Online prescriptions, assessment, clinical decision support, treatment practice management, and self-care all are simplified. Smart watches that can keep track of various body activities such as pulse rate, temperature and continuous health management by using attached physiological sensors. Present deep learning tools rely on assumptions that variables are related, there is still a research gap regarding variables that are not related to each other in any particular manner. If a common model was introduced to handle all types of datasets this would be easy and more accurate. This will most efficiently make them more reliable and accurate in giving the right predictions.

The process of human learning is still a mystery and complex. It has not yet been implemented in machine learning. However, given the interest of researchers in the field of machine learning theories and algorithms, incorporation of multiple sensory modalities such as vision, sound and touch in self-supervised learning to predict one sensory experience from the others might become a reality.
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A HYBRID INTRUSION DETECTION SYSTEM FOR MOBILE ADHOC NETWORKS USING FBID PROTOCOL

D. RAJALAKSHMI * AND K. MEENA †

Abstract. A Security in a mobile ad hoc networks is more vulnerable and susceptible to the environment, because in this network no centralized environment for monitoring individual nodes activity during communication. The intruders are hacked the networks either locally and globally. Now a day’s mobile ad hoc network is an emerging area of research due to its unique characteristics. It’s more vulnerable to detect malicious activities, and error prone in nature due to their dynamic topology configuration. Based on their difficulties of intrusion detection system, in this paper proposed a novel approach for mobile ad hoc network is Fuzzy Based Intrusion Detection (FBID) protocol, to identify, analyze and detect a malicious node in different circumstances. This protocol it improves the efficiency of the system and does not degrade the system performance in real time. This FBID system is more efficient and the performance is compared with AODV, Fuzzy Cognitive Mapping with the following performance metrics: Throughput, Packet Delivery Ratio, Packets Dropped, Routing overhead, Propagation delay and shortest path for delivering packets from one node to another node. The System is robust. It produces the crisp output to the benefit of end users. It provides an integrated solution capable of detecting the majority of security attacks occurring in MANETs.

Key words: Security, Intrusion detection, AODV, MANET, Fuzzy, Cognitive Map

AMS subject classifications. 68M15

1. Introduction. A Mobile adhoc network is a complex wireless network, it consist of collection of mobile nodes, which forms a spontaneous network without the physical infrastructure, it allows individual, group of members and organizational members work together and communicate without the stable infrastructure [1]. Limitation of mobile adhoc networks are bandwidth and energy consumption.

A mobile adhoc network is shown in cf. Fig.1.1. It’s an infrastructure less network because the mobile nodes in the network dynamically change the paths with other nodes and transmit the data packets provisionally. In a MANET, nodes within the region or specified boundary means, it communicates with other nodes directly, otherwise it needs to rely on some other nodes to relay the messages from source to destination. The major security goals that need to be addressed in order to maintain a reliable and secure ad-hoc network environment. There are confidentiality, availability, non-repudiation, authentication and integrity. The security attacks in MANET can be roughly classified in two types: 1) Active Attacks and 2) Passive Attacks.

Hosts may misbehave or try to compromise security at all layers of the protocol stack. In Transport layer to provide secure end-to-end communication [2]. For that need to know keys to be used for secure communication, then it anonymity the communication. In Network layer, the misbehaving hosts may create the hazards: in terms of it disrupt the route discovery and maintenance. Due to that hazard, Delay, drop, corrupt and misroute the packets. It degrades the networking performance. In MAC layer, the misbehaving nodes may not cooperate to each other. Because disobey the protocol specifications for sellish gains.

Mobile Ad hoc networks are collections of mobile nodes that may enter and leave the network dynamically. No centralized controller and infrastructure. A major issue in Mobile ad-hoc network is security. This also aims of the work in MANET. To detection of malicious nodes forms a very essential one of the part an approach to security [3]. The main objective of this work is to detect the intrusions through Fuzzy logic that prevents the network from denying the active session or extract the confidential information that is being shared. The

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proposed work uses FBID to identify the malicious nodes of capture the intrusion over MANET that networks as well as provide the best solution to reduce the execution time over the network[4].

2. Related Work. Security is an essential requirement of mobile adhoc networks. These networks are more vulnerable and threats are increased due to lack of security and very hard to implement the centralized security control for authentication purpose [5]. The main attributes of security requirement of mobile adhoc networks are [6]:

**Confidentiality:** It has a set of rules; it limits the unauthorized users accessing the network.

**Data Integrity:** It gives the information as trustworthy and accurate data.

**Availability:** It’s reliable access to the authorized users.

**Denial of Service:** It is attacked by malicious nodes or selfish nodes.

The characteristics of Mobile Adhoc Networks are:

**Dynamic Topologies:** Randomly the network topology has changed, and inside the node also moving freely with different speed at changeable times.

**Energy - Constrained Operation:** In wireless networks the individual node rely on batteries, entire energy is drained due to continuous monitoring or active in all times till the energy has exhausted.

**Limited Bandwidth:** In wireless communication the signals or data’s are dropped due to noise, interference, fading, multiple accessing technique and weather conditions. If the data’s are dropped then automatically the throughput has reduced and it leads to the less bandwidth consumption.

**Security Threats:** Security is lacking in wireless networks. Due to their infrastructure less network any one can hack the system in the form of passive attacks and active attacks.

The challenges of Mobile Adhoc Networks are [7]:

**Scalability:** To measure the performance of network, the scalability is the main issue. How many packets or data to be delivered to the particular destination without data loss, it can be measured by scalability. Now a day the overall performance of the network is based on throughput, it is low in wireless environment. For this improvement, crucial research work is progress on.

**Quality of Service:** Need to improve the quality of service robustness, algorithms, protocols and policies to be addressed in a very effective manner in wireless region. Quality of service can be measured by the following factors: Delay, Jitter, Bandwidth and Throughput [8].

**Client–Server Model Shift:** In wireless domain, Client - Server concept is not applicable in real time, because there is no stable state for server, connections, IP addresses and authorization mechanisms. Here the individual node may act as a client or server, that is, peer-to-peer communication. For this purpose the traditional client-server model to be implemented in a better way for wireless communication.

**Security:** In Mobile adhoc networks the suspicious or malicious node they can enter the network, and compromise the network. The individual nodes are dynamically moving at irregular time intervals; in that case the malicious nodes are attacking the network and observing the nature of data [9].
Interoperation with the Internet: Now a day without internet we cannot do anything. In mobile adhoc networks the configuration and set up is differ from one network to another. The interface is main issue of connecting the different type of networks. It can be avoided by assigning foreign agent to the mobile IP.

Energy Conservation: Energy conservative networks are popular in adhoc networks. For better performance the entire battery is fully utilized for the active networks. Still the energy is constrained in wireless environment. It will be improved to implement a better routing algorithm for transmission and reception.

Node Cooperation: In wireless environment, the node cooperation is more important, because the individual nodes are independent; the malicious nodes are acting as a dependent node to all other nodes. These malicious nodes are charged in a very high manner and it consumes the entire battery power.

3. Problem Definition. In this paper, our objective is to solve the weakness of watchdog methods; Ambiguous collisions, Receiver Collisions, Limited Transmission Power, Partial Dropping, False Misbehavior Report and Collision [10-11]. The Proposed work will be anomaly based intrusion detection system that is lack of monitoring capability and entering the malicious nodes inside the network. These issues are solved by fuzzy based intrusion detection protocol. The ultimate goal of the paper is finding the malicious node in effective way when compared to all other existing methods, measured by the following scenarios:

1. Ambiguous Collisions: In two nodes are communicated Node 1 and Node 2, it transmits a packet from node 1 to node 2 and vice versa, collision occurs in node 1 and node 2 it forward the packet. Node 1’s collided by Node 2 transmissions, so neighbourhood nodes are not able to do the communication. Node 1 continuously monitors the same node, in this case malicious node accessed or hacking the network throughput.

2. Receiver Collisions: Two nodes are communicated say Node 1 and Node 2. The senders send a packet to Node 1 and monitor the action of node 1 and send the packet to node 2. Sender does not give any assurance to deliver the packet successfully at node 2. In this case collision occurs in node 2 means, again node 1 it resends the packet to node 2, due to number of times sending a same frame leads the malicious nodes it access the packet.

3. Limited Transmission Power: A misbehaving node consumes the transmission power, such that the signal is high in previous node or sender and too weak in the destination node. It leads to malicious node will enter and hack the network bandwidth.

4. Partial Dropping: In Watchdog mechanism, the packets are transmitted from one hop to another to reach the destination, but watchdog doesn’t aware about where the packet reside in it and which hop to transfer the packet to the desired destination in the network. This lack of information leads to misbehaviour nodes are entered in the network; this also cannot be detected by watchdog. If it suspects the node become misbehavior, it forced to forward the packets to threshold bandwidth, and drops the packet.

5. False Misbehavior Report: The malicious nodes divert or falsely define the trusted node becomes misbehavior node. It diverts the monitoring controls to the trusted node. In that duration the malicious node access the network and grasp the information and leave the network.

6. Collusion: Multiple node collusion is well planned activity. For example two nodes are communicated say 1 and 2. It collides as due to malicious node. Here node 1 forwards the packet to node 2, but it not responds to the initiator, so node 2 it discards the packet. The consecutive untrusted nodes communicated in a single routing path. The Malicious node limits or spoils the communication.

4. Methodology. Due to the revolution of science and technology, it’s more difficult to take decision and its leads to the issue of unclear or not expected results are generated, and it is very hard to analysed [12]. Fuzzy concept has the capability to take decisions in a correct manner through a formal mathematics and logic; it generates the qualitative data or predicted data. The fuzzy concepts are capable of handling humanistic type of problems.

The word fuzzy refers two things: true or false [13]. Any action or event, the current state is changed suddenly or continuously, it cannot be represented as either true or false. For this purpose, the fuzzy concept plays a vital role in emerging applications. In Boolean concepts the output is represented by 0 or 1[14 -15]. Fuzzy system values are defined in the range from 0 to 1 or yes/ no. But in fuzzy 1.0 represents "Absolute or
Fig. 4.1. *Fuzzy Logic Architecture*

The fuzzy set represents the one form of uncertainty. Suppose in regular activity of network, who are uncertain about the non-malicious or innocence of the trusted node. The uncertainty in this situation is very hard to find the malicious nodes and innocent nodes. In order to represent this type of uncertainty, assign a value to each possible crisp set. This value defines the amount of confirmation or certainty of the nodes in that network. The uncertainty is also called as a fuzzy measure. Fuzzy measures solve the problem after considering all available data’s, and then it takes the better relevant decision for the given input. It’s shown in cf. Fig. 4.1.

There are four components of fuzzy logic architecture:

1. **Rules**: It contains set of rules and regulations that govern to take the better relevant decision making system. Due to the better relevant decision making system, it reduces the fuzzy rule.

2. **Inference**: It matches the rules according to the fuzzy input set, if it not matched with the existing rule, then the new rule will be implemented based on the fuzzy input. Afterwards the new rules are combined to take the better relevant decision.

3. **Fuzzifier**: Before entering to the inference process, we need to convert the crisp input to fuzzy input set values. This conversion process is performed in fuzzifier.

4. **Defuzzifier**: It is used to convert the output of inference process, that is fuzzy output set into crisp output to the benefit of end users.

The benefits of Fuzzy logic systems are:

- Implementation of fuzzy logic system is easy and understandable.
- It provides a very efficient solution to the complex problems in the emerging trends.
- It deals with uncertainty in engineering.
- The system is robust.
- Easily modified to improve or enhance the system performance.

The limitations of Fuzzy logic systems are:

- Ambiguity: There is no systematic methods to solve the problems in real time issue.
- Lack of mathematical description: Proof of techniques is more complex and difficult to obtain for all possible scenarios.
- Verification and Validation is more complex.
- Don’t have the capability of machine learning.

5. **Performance Evaluation.** Fuzzy Based Intrusion Detection System can be formed using two disjoint classes of fuzzy cognitive mapping. It has lot of advantages, cost-effective, perceptive and time consumption is very less. Fuzzy Based Intrusion Detection System is better than the normal fuzzy cognitive mapping, because here two disjoint zones are used to produce the better result in real time scenarios.

   1. Region Zone
   2. Field Zone

   There are no intermediate relations are exist between Region and field zones. The numbers of elements are not mandatorily equal between these region and field zones. Region zone values are taken from the real vector

---

**extreme truth**, 0.85 represents very truth, 0.35 represents sometimes truth, 0.0 represents absolute false. The range of fuzzy system is represented by the truth value. Fuzzy logic handles the reasoning capability with the fuzzy concepts. Fuzzy logic is not a logic, it’s a fuzzy, logic used to describe the fuzziness [16-18].
zone of range represented by \( p \) and Field zone values are taken from the real vector zone of range represented by \( q \). Set of zones are represented \( F \), it has the dimension from \( F_1, \ldots, F_m \) of the field zone, where \( F \), range lies from \( a_1 \) to \( a_m \), where \( a_x = 1 \) or \( 0 \) for \( y = 1, 2, \ldots, s \). If \( a_x = 1 \) it means the zone \( F_i \) is in ON State and if \( a_x = 0 \) means OFF state. \( R \) denotes the zone \( R_1, R_2, \ldots, R_n \) of the range zone, where \( R \) range lies from \( a_1 \) to \( a_n \), where \( a_x = 1 \) or \( 0 \) for \( x = 1, 2, \ldots, n \). If \( a_x = 1 \) it means the zone \( R_i \) is in ON State and \( a_x = 0 \) means OFF State.

Fuzzy Based Intrusion Detection System is a fixed graph; it represents the value of Region to Field zone with rule, conditions and policies, as zones and causalities as edges. It denotes the fundamental association between Zone \( R \) and \( F \). When zones of the fuzzy based intrusion detection system are fuzzy sets it is also called as fuzzy zones. The weights zero, plus or minus one are called as simple fuzzy based intrusion detection system. The system architecture for intrusion detection is shown in Fig.5.1.

Let \( R_1, R_2, \ldots, R_n \) be the zones of the region zone \( R \) of an FBID and \( F_1, F_2, \ldots, F_m \) be the zones of the Field zone \( F \) of an FBID. Let matrix \( Z \) be defined as \( Z = (z_{xy}) \) where \( z_{xy} \) is the weight of the fixed edge \( R_x F_y \), \( Z \) is called as the cognitive relational matrix of FBID.

Consider the relationship between the research work and scholar. Suppose the Region zone as perception is based on the research work say \( R_1, R_2, \ldots, R_5 \) and the field zone define the perception is based on the Scholar say \( F_1, F_2, \ldots, F_5 \). Let define the zones \( R_1, R_2, \ldots, R_5 \) and \( F_1, F_2, \ldots, F_5 \) as follows. For Region Zone,
- \( R_1 \): Research Work is Good.
- \( R_2 \): Research Work is Poor.
- \( R_3 \): Research Work is Average.
- \( R_4 \): Research Work is Different Variety.
- \( R_5 \): Research Work is not useful.

For Field Zone,
- \( F_1 \): Good Scholar
- \( F_2 \): Bad Scholar
- \( F_3 \): Average Scholar

![Fig. 5.1. Intrusion Detection System](attachment:image.png)
The cognitive relational fixed graph of the research work - scholar model is plotted in Fig 5.2. The cognitive relational matrix \( Z \) derived from the above graph is:

\[
Z = \begin{bmatrix}
1 & 0 & 0 \\
0 & 1 & 0 \\
0 & 0 & 1 \\
1 & 0 & 0 \\
0 & 1 & 0 \\
3 & 7 & 7 & 7 & 7 & 5
\end{bmatrix}
\]

(5.1)

If \( C = (1, 0, 0, 0, 0) \) is passed on in the cognitive relational matrix \( Z \), the instantaneous vector, \( CZ = (1, 0, 0) \) implies that the scholar is a good scholar. \( DZ^T = (1, 0, 0, 1, 0) \) implies that the research work is good and he/she did a research work is different variety. \( DZ^T = C_1, C_1Z = (2, 0, 0) \) after threshold, \( C_1Z = (1, 0, 0) \) implies that the scholar is good, so on.

The Fuzzy cognitive relational membership is:

\[
\mu_x(R_i) = \begin{cases} 
1 & \text{if Row sum of maximum value} \\
0 & \text{if Row sum of minimum value} \\
R_i - \text{Row sum of minimum value} & \text{if Row sum of minimum value is} \\
\text{Row sum of maximum value} - \text{Row sum of minimum value} & \text{less than or equal to } R_i \text{ and} \\
& \text{less than or equal to } \text{Row sum of maximum value}
\end{cases}
\]

\[
\mu(x) = \begin{cases} 
\frac{(x - p)}{(q - p)} & \text{if } p \leq x \leq q \\
\frac{(r - x)}{(r - q)} & \text{if } q \leq x \leq r
\end{cases}
\]

(5.2)

\( \mu \) is a fuzzy subgroup of \( CZ \). Then \( \mu(x^{-1}) = \mu(x) \) and \( \mu(x) \leq \mu(e) \) for all \( x \in CZ \), where \( e \) is the identity element of CG.

\[
\mu(x) = (\mu((x^{-1})^{-1}) \geq \mu(x^{-1}) \geq \mu(x))
\]

Hence for \( x \in CZ \),

\[
\mu(e) = \mu(xx^{-1}) \geq \min(\mu(x), \mu(x^{-1})) = \mu(x)
\]
A Hybrid Intrusion Detection System for Mobile Adhoc Networks using FBID Protocol

Table 5.1
Analysis of AODV, FCM and FBID Protocol

<table>
<thead>
<tr>
<th>Performance Metrics</th>
<th>AODV</th>
<th>FCM</th>
<th>FBID</th>
</tr>
</thead>
<tbody>
<tr>
<td>Throughput</td>
<td>Moderate</td>
<td>High</td>
<td>Very High</td>
</tr>
<tr>
<td></td>
<td>60 to 75%</td>
<td>70 to 80%</td>
<td>80 to 90%</td>
</tr>
<tr>
<td>Packet Delivery Ratio</td>
<td>Moderate</td>
<td>High</td>
<td>Very High</td>
</tr>
<tr>
<td>Packets Dropped</td>
<td>Low</td>
<td>Low</td>
<td>Very Low</td>
</tr>
<tr>
<td>Routing Overhead</td>
<td>Less at moderate congestion</td>
<td>Low at reasonable congestion</td>
<td>Very low at less congestion</td>
</tr>
<tr>
<td>Propagation delay</td>
<td>Less at moderate congestion</td>
<td>Low at moderate congestion</td>
<td>Very low at less congestion</td>
</tr>
<tr>
<td>Shortest Path</td>
<td>Moderate</td>
<td>High</td>
<td>Very High</td>
</tr>
</tbody>
</table>

\[
\mu(y) = \text{Max} \{\text{Min} \, \mu_p k_1(\text{input}(i)), \{\mu_p k_2(\text{input}(j))\}\}
\]

\[
Z(x, y) = \begin{cases} 
\mu_p(x) & \text{if } \mu_q(y) = 1, \\
\mu_q(y) & \text{if } \mu_p(x) = 1, \\
0 & \text{if } \mu_p(x) < 1, \mu_q(y) < 1 
\end{cases}
\]  
(5.3)

\[
Z(x, y) = \begin{cases} 
1 & \text{if } \mu_p(x) \leq \mu_q(y) \\
\mu_q(y) & \text{if } \mu_p(x) > \mu_q(y) 
\end{cases}
\]  
(5.4)

To maintain the consistency using fuzzy logics, the probability is 1.

Degree of truth (T) + level of indeterminacy (I) + Degree of false (F) = 1

Incomplete information on a variable the proposition is:

Degree of truth (T) + level of indeterminacy (I) + Degree of false (F) < 1.

Contradictory sources of information on a variable, the proposition is:

Degree of truth (T) + level of indeterminacy (I) + Degree of false (F) > 1.

In FBID directly give the results of one type of network into another type of networks. FBID divide the number of zones into two zones, that is region and field, and relational represents are sent from one network to another network. It gives better predicted results based on the previous data. So FBID provide more benefits when compared to existing systems. It’s shown in Table 5.1.

6. Experiments and Results. In this simulation evaluating the performance of the FBID protocol using IEEE802.11 standards. Our results are generated based on the FBID protocol. In our simulations, we are concentrate on the Throughput, Packet delivery ratio and energy consumption through the following factors: Ambiguous Collision, Limited Transmission power, False Misbehavior Report, Partial dropping, Receiver collision and collusion all are shown in the Fig. 6.1.

7. Conclusion. Mobile adhoc network is an autonomous collection of nodes. Nodes are changed the position randomly or dynamically throughout the communication. This infrastructure less environment the hackers are easily entered and access the network. These issues are solved by the following methodologies and protocols ACK, S-ACK, AACK, EAACK, AODV, DSR and DSDV etc. But these methodologies still they are suffer to improve the system performance. The proposed fuzzy based intrusion detection protocol, it improves the performance of watchdog limitations. It can be measured by number of factors; Throughput, Packet delivery ratio, propagation delay, routing overhead and finding the shortest path. This on demand based fuzzy based intrusion detection protocol; it improves the performance and doesn’t degrade the networking functionalities.
Fig. 6.1. (a) Ambiguous Collision; (b) Limited Transmission Power; (c) False Misbehavior Report; (d) Partial Dropping; (e) Receiver Collision; (f) Collusion

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A Hybrid Intrusion Detection System for Mobile Adhoc Networks using FBID Protocol


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MONTE CARLO SIMULATIONS OF COUPLED TRANSIENT SEEPAGE FLOW AND COMPRESSION STRESS IN LEVEES

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Abstract. The purpose of this research is to compare the results from two different computer programs of flow analyses of two levees at Port Arthur, Texas where rising water of a flood from Hurricane Ike occurred on the levees. The first program (Program 1) is a two-dimensional (2-D) transient finite element program that couples the conservation of mass flow equation with accompanying hydraulic boundary conditions with the conservation of force equations with accompanying x and y displacement and force boundary conditions, thus yielding total head, x displacement, and y displacement as unknowns at each finite element node. The second program (Program 2) is a 2-D transient finite element program that considers only the conservation of mass flow equation with its accompanying hydraulic boundary conditions, yielding only total head as the unknown at each finite element node. Compressive stresses can be computed at the centroid of each finite element when using the coupled program. Programs 1 and 2 were parallelized for high performance computing to consider thousands of realizations of the material properties. Since a single realization requires as much as one hour of computer time for certain levees, the large realization computation is made possible by utilizing HPC. This Monte Carlo type analysis was used to compute the probability of unsatisfactory performance for under seepage, through seepage, and uplift for the two levees. Respective hydrographs from the flood resulting from Hurricane Ike were applied to each levee. When comparing the computations from the two programs, the most significant result was the two programs yielded significantly different values in the computed results in the two clay levees considered in this research.

Key words: coupled transient seepage/stress analysis, finite element method, high performance computing

AMS subject classifications. 65Y05, 35J66, 76S05

1. Introduction. Practising engineers commonly perform transient finite element seepage analyses that account only for seepage due to hydraulic boundary conditions but do not account for forces applied to the levee with resulting stresses in the levee due to rising water [1]. In clay levees, excess pore pressures beyond those from just hydraulic boundary conditions are developed from stresses in the levee soil as the river rises. Seepage design information can therefore be incorrectly estimated without considering these excess pore pressures. A detailed geotechnical explanation of the proper use and limitations of transient seepage in levees is given in [2, 3]. The trend in academic research as in [2, 3] and in commercial software documentation such as in [4] is that a fully coupled transient seepage/stress analysis is needed for clay levees. Other studies describing the importance of coupling in modelling seepage flow are [5, 6, 7, 8].

However, only one or just a few values of each soil parameter are often used in current research, although there is significant variation in these values. Unique features of this research are that 6000 realizations of the soil properties are considered, and specific failure modes were investigated. Thus, a two-dimensional (2-D) finite element transient seepage program (SEEP2D-TRAN-HPC) that considers only Darcian seepage flow and a 2-D coupled transient finite element seepage/structural plane strain program (SEEP2D-COUPLED-HPC)
were written for parallel, high performance computing (HPC) to analyse the failure modes of under seepage, through seepage, and uplift using 6000 realisations of material properties described in [9]. This allows a Monte Carlo type analysis to investigate the fraction of the realisations that produced unsatisfactory performance using criteria for the three failure modes. This is equivalent to the probability of unsatisfactory performance (PUP) for each failure mode.

When comparing independent simulations in a previous study, we found that 6000 realisations of the material properties were sufficient to yield estimates of the probability of unsatisfactory performance within one percentage point. This result was consistent over 30 levees representing a wide variety of geometries and soil types. Based on these results, we chose to use 6000 realisations of the material properties in this study.

2. Purpose. The purpose of this study was to compare the difference in results from SEEP2D-TRAN-HPC and SEEP2D-COUPLED-HPC for under seepage, through seepage, and uplift using as a measure the PUP for each failure mode using two levees from Port Arthur, Texas as test levees. This paper reports the results of the study, gives the equations used, and describes the computational challenges encountered in developing a successful parallel coupled transient seepage/compressive stress program.

3. Levees. Two levees in Port Arthur, Texas are considered in this paper. Figure 3.1 shows the first levee where the respective soil types as defined by the Unified Soil Classification System (USCS) and are as follows: CL - low plasticity clay, CH - high plasticity clay, SM - silty sand, and SP - poorly graded sand. The figure shows the water level at the elevation of the crown (10.5 ft (3.2 m)), but the transient solution applies a hydrograph from the flood resulting from Hurricane Ike. The land side water level is held at the ground surface elevation of 0 ft. Figure 3.2 shows the hydrograph of the flood on the gulf side of the levee where simulation time = 0 days when the water level from the gulf reaches 0 ft (0 m). The hydrograph was developed using the Coastal Storm Modelling System (CSTORM) [10], the coupled Advanced Circulation (ADCIRC) programs [11, 12], and the Steady-State Spectral Wave (STWAVE) model [13, 14].

Figures 3.3 and 3.4 show the second levee and its hydrograph. As before, the usable part of the hydrograph starts at the elevation of the ground surface on the land side of the levee (7.5 ft (2.29 m)). However, on the gulf side, the water level exceeds the elevation of the crown of 16.5 ft (5.03 m) which results in over-topping. For seepage inside the levee, the hydrograph used in the computations is capped at 16.5 ft (5.03 m).

4. Equations for transient seepage in the saturated zone. Transient flow where only conservation of mass and hydraulic boundary conditions from the rising water on the levee are considered is governed by

\[
\frac{\partial}{\partial x} \left( k_{xs} \frac{\partial \phi}{\partial x} \right) + \frac{\partial}{\partial y} \left( k_{ys} \frac{\partial \phi}{\partial y} \right) = m_v \frac{\partial \phi}{\partial t} \tag{4.1}
\]

where \( k_{xs} \) is the saturated hydraulic conductivity in the x direction, \( k_{ys} \) is the saturated hydraulic conductivity in the y direction, \( \phi \) is total head, \( m_v \) is the volumetric compressibility of the soil, \( x \) is the x coordinate, \( y \) is the y coordinate, and \( t \) is time. As discussed earlier, a coupled formulation where both hydraulic boundary conditions and the structural loads applied to the levee by the water, causing stresses, strains, and displacements of the soil, is preferred, especially for clay levees. There are many levels of sophistication available for this coupled analysis.
as described in [4]. The initial version of SEEP2D-COUPLER-HPC uses a simple plane strain constitutive model for the soil and an uncoupled unsaturated flow assumption. Later versions can easily have more options.
Conservation of forces inside a finite element is given by

\[
\{\sigma\} + \gamma_w (\phi - y) \begin{pmatrix} 1 \\ 0 \end{pmatrix} = [C] \{\epsilon\}
\]

where

\[
\{\sigma\} = \begin{pmatrix} \sigma_{xx} \\ \sigma_{yy} \end{pmatrix}
\]

and

\[
\{\epsilon\} = \begin{pmatrix} \epsilon_{xx} \\ \epsilon_{yy} \\ \gamma_{xy} \end{pmatrix} = \begin{pmatrix} \frac{\partial u}{\partial x} \\ \frac{\partial v}{\partial y} \\ \frac{\partial u}{\partial x} + \frac{\partial v}{\partial y} \end{pmatrix}
\]

where \(\sigma_{xx}\) is the normal stress in the x direction, \(\sigma_{yy}\) is the normal stress in the y direction, \(\gamma_{xy}\) is the shearing stress, \(\epsilon_{xx}\) is the normal strain in the x direction, \(\epsilon_{yy}\) is the normal strain in the y direction, \(\gamma_{xy}\) is the shearing strain, \(u\) is the displacement in the x direction, \(v\) is the displacement in the y direction, \(\gamma_w\) is the density of water, and \([C]\) is the soil plane strain constitutive matrix given by

\[
[C] = \frac{E}{(1 + \nu)(1 - 2\nu)} \begin{pmatrix} 1 - \nu & \nu & 0 \\ \nu & 1 - \nu & 0 \\ 0 & 0 & \frac{1 - 2\nu}{2} \end{pmatrix}
\]

where \(E\) is Young’s Modulus and \(\nu\) is Poisson’s Ratio.

The coupled governing equation for conservation of flow in the saturated zone is

\[
\frac{\partial}{\partial x} \left( k_{xs} \frac{\partial \phi}{\partial x} \right) + \frac{\partial}{\partial y} \left( k_{ys} \frac{\partial \phi}{\partial y} \right) = \frac{\partial}{\partial t} \left( \frac{\partial u}{\partial x} + \frac{\partial v}{\partial y} \right)
\]

The coupled governing equation for conservation of flow in the saturated zone is

\[
\frac{\partial}{\partial x} \left( k_{xs} \frac{\partial \phi}{\partial x} \right) + \frac{\partial}{\partial y} \left( k_{ys} \frac{\partial \phi}{\partial y} \right) = \frac{\partial}{\partial t} \left( \frac{\partial u}{\partial x} + \frac{\partial v}{\partial y} \right)
\]

5. Equations for transient seepage in the unsaturated zone. The pore pressures from SEEP2D-TRAN-HPC, SEEP2D-COUPLED-HPC, and a steady-state version of these programs (SEEP2D-SS-HPC) are input into slope stability programs similar to [15] where negative pore pressures are not considered. To keep equivalent sophistication for purposes of comparison, negative pore pressures (\(\phi < y\)) are not considered here. Thus conservation of force in a finite element in the unsaturated zone is given by (see Eq. 4.2 for comparison)

\[
\{\sigma\} = [C] \{\epsilon\}
\]

Also, for a finite element in the unsaturated zone, the uncoupled version of conservation of flow is used. Starting with the equation for the suction head, \(\psi\),

\[
\psi = y - \phi
\]

conservation of flow is modelled by Richards equation [16],

\[
\frac{\partial}{\partial x} \left( k_r k_{xs} \frac{\partial \phi}{\partial x} \right) + \frac{\partial}{\partial y} \left( k_r k_{ys} \frac{\partial \phi}{\partial y} \right) = \frac{\partial}{\partial t} \left( \frac{\partial u}{\partial x} + \frac{\partial v}{\partial y} \right) = \frac{d\theta}{d\psi} \frac{\partial \phi}{\partial t}
\]

where \(k_r\) is the relative hydraulic conductivity (\(0 < k_r \leq 1\)), and \(\theta\) is the moisture content. Also, \(k_r\) and \(\theta\) are functions of \(\psi\) resulting in Eq. 5.3 being a difficult non-linear equation to solve.
6. Moisture content and relative hydraulic conductivity. The equations for moisture content and relative hydraulic conductivity as given in [17] are used in this work.

6.1. Moisture content. Moisture content is modeled by

\[ \theta(\psi) = B(\psi) \frac{\theta_s}{\left\{ \ln \left[ e + \left( \frac{\psi}{\beta} \right)^n \right] \right\}^m} \]  

(6.1)

\[ B(\psi) = 1 - \frac{\ln \left( 1 + \frac{\psi}{C_r} \right)}{\ln \left( 1 + \frac{1000000}{C_r} \right)} \]  

(6.2)

where \( \theta_s \) is the saturated moisture content; and \( \beta, C_r, m, \) and \( n \) are soil parameters.

6.2. Relative hydraulic conductivity. Using the change of variables,

\[ \psi = e^z \]  

(6.3)

\[ k_r(\psi) = \frac{\int_{\ln \psi}^{\ln 1000000} \theta(e^z) - \theta(\psi)}{\int_{\ln \varepsilon}^{\ln 1000000} \theta(e^z)} \theta'(e^z) \, dz } \]  

(6.4)

\[ \theta'(\psi) = -B(\psi) \frac{mn\theta_s \left( \frac{\psi}{\beta} \right)^{n-1}}{\beta \left\{ \ln \left[ e + \left( \frac{\psi}{\beta} \right)^n \right] \right\}^m \left[ e + \left( \frac{\psi}{\beta} \right)^n \right]} \]  

(6.5)

where \( \varepsilon \) is a small positive number. Eq. 6.4 is numerically integrated to obtain a value for \( k_r \).

7. Unsatisfactory performance criteria. Unsatisfactory performance for the three failure modes of under seepage, through seepage, and uplift is determined by provided criteria. There are two unsatisfactory performance criteria for each failure mode, and they are limit state (LS) and engineering practice (EP). The LS criteria are typically used for forensic analysis and extreme events, while the EP criteria are used in the design stage.

7.1. Under seepage. A positive value of vertical exit gradient, \( i_v \), at the toe is used for testing under seepage. This is computed by selecting two node points with the same x value at the toe of the levee as shown in Fig. 7.1 and using

\[ i_v = \frac{\phi_1 - \phi_2}{y_2 - y_1} \]  

(7.1)

where \( \phi_1 \) is the total head of the first selected node and beneath the ground surface, \( \phi_2 \) is the total head of the second selected node and on the ground surface, \( y_1 \) is the y coordinate of the first selected node, and \( y_2 \) is the y coordinate of the second selected node.

The critical vertical exit gradient, \( i_{cv} \), at the toe is computed from

\[ i_{cv} = \frac{\gamma_s}{\gamma_w} - 1 \]  

(7.2)

where \( \gamma_s \) is the density of the saturated soil. The limit state criterion for unsatisfactory performance for under seepage is

\[ i_v \geq i_{cv} \]  

(7.3)

The engineering practice criterion for unsatisfactory performance for under seepage is

\[ i_v \geq 0.5 \]  

(7.4)
7.2. Through seepage. The exit gradient, $i_p$, near the toe and perpendicular to the land side slope of the levee at the toe is used for testing through seepage. From Fig. 7.2, this research uses

$$i_p = \frac{\phi_A - \phi_P}{d}$$  \hspace{1cm} (7.5)

The three node points, 1, 2, and 3, are determined from the nodes of the finite element mesh in the embankment at the toe.

The limit state criterion for unsatisfactory performance of through seepage at the toe is given by

$$i_p \geq i_{cv} \cos \theta$$  \hspace{1cm} (7.6)

where $\theta$ is the angle between the land side slope of the levee and the ground surface. The engineering practice criterion for unsatisfactory performance of through seepage at the toe is given by

$$i_p \geq 0.5 \cos \theta$$  \hspace{1cm} (7.7)

7.3. Uplift. Uplift is computed at the toe typically beneath the confining layer of the levee (see Fig. 7.3) to check the factor of safety with respect to heave or uplift. The factor of safety for uplift is

$$f_{up} = \frac{\gamma_s (y_B - y_A)}{\gamma_w (\phi_A - y_A)}$$  \hspace{1cm} (7.8)

which represents a ratio of weight divided by the uplifting force. The limit state criterion for unsatisfactory performance of uplift is given by

$$f_{up} \leq 1$$  \hspace{1cm} (7.9)

and the engineering practice criterion for unsatisfactory performance of uplift is

$$f_{up} \leq 1.5$$  \hspace{1cm} (7.10)
8. Initial and boundary conditions. The initial conditions are that the water level is at the ground surface, and only the weight of each finite element of the levee is the applied force. Also, the bottom of the levee is fixed for no x or y displacement and considered impervious to water flow. Boundary conditions for flow are as follows: (1) Total head is applied to the gulf side of the levee as dictated by the provided hydrograph. (2) The elevation of the ground surface on the land side of the levee is the total head applied on this boundary. (3) The land side slope of the levee is an exit face where the phreatic surface (pore pressure = 0) that intersects this downstream slope must be determined by an iterative solution. (4) For forces, the water from the hydrograph is applied as water pressure on the levee.

9. Computational details and challenges. The governing equations for coupled flow and displacement with three degrees of freedom per node ($u$, $v$, and $\phi$) were implemented into the finite element program, SEEP2D-COUPLED-HPC. SEEP2D-TRAN-HPC with only $\phi$ as a degree of freedom was also written. For the coupled solution, only the weight of the levee is first applied with the water at the ground surface to obtain initial displacements of the nodes. Then for a given small time increment, $\Delta t$, the water level is advanced a small amount using the hydrograph and a new increment of displacement and total head for each node is computed. The non-linear nature of the equations requires iteration at each time step to achieve a solution.

The finite element meshes of the two levees were generated on the PC using SEEP2D [18, 19] in the Groundwater Modeling System [20]. Once generated, these mesh files were uploaded to a large HPC system. SEEP2D-TRAN-HPC and SEEP2D-COUPLED-HPC are written in Fortran and use the Message Passing Interface (MPI) [21] for parallelization. The HPC system used is a Cray XC40/50 at the ERDC DSRC named “Onyx” [22]. Table 9.1 shows the current configuration of Onyx. Six thousand MPI processes were run on 6000 cores with only collecting the results of the success or failure of each realisation needing communication.

Each non-linear iteration at each time step creates a system of simultaneous, linear equations to be solved. For SEEP2D-TRAN-HPC, this system of equations is symmetric and positive-definite. Thus, the conjugate gradient method [23] works very well. However, for SEEP2D-COUPLED-HPC, this system of simultaneous, linear equations is nonsymmetric and lacks the positive-definite quality. An iterative solver (bi-CG stabilized [23]) was first tried without success. A banded direct solver was then implemented which solved the equations but took significantly more computer time.

The system of non-linear equations for SEEP2D-TRAN-HPC is notoriously difficult to solve. This difficulty is even more acute when running SEEP2D-COUPLED-HPC. Many researchers use Newton’s method [24], while...
Table 10.1

Probability of unsatisfactory performance for the first levee for under seepage, through seepage, and uplift for hydraulic boundary conditions only.

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others use the Picard iteration method [25]. Still others add line search techniques to accelerate the convergence to a solution. An advantage of transient solutions is that decreasing $\Delta t$ makes the solution easier to converge. This work chose a simple combination of the Picard iteration, a small $\Delta t$, and a modified convergence criterion to obtain successful convergence for both levees tested.

The convergence criteria for a given non-linear iteration often used is

$$\max_{j=1}^{N} |\phi_{j}^{k-1} - \phi_{j}^{k}| \leq \varepsilon$$

(9.1)

where $N$ is the number of node points of the finite element mesh, $j$ is the node point number, $\phi_{j}^{k-1}$ is the total head at node, $j$, for iteration $k-1$, $\phi_{j}^{k}$ is the total head at node, $j$, for iteration $k$, and $\varepsilon$ is a small number. The problem with Eq. 9.1 is that the iterations oscillate between different worst nodes and convergence is never achieved. This work successfully used

$$\frac{\sum_{j=1}^{N} |\phi_{j}^{k-1} - \phi_{j}^{k}|}{N} \leq \varepsilon$$

(9.2)

which smooths out the oscillation weakness in Eq. 9.1.

10. Results. Each core with its realisation of the material properties is run and returns either a 0 for satisfactory performance or a 1 for unsatisfactory performance for each failure mode. These 0’s and 1’s are then collected and summed for each respective failure mode and divided by 6000. Table 10.1 gives results for the first levee from SEEP2D-TRAN-HPC (only hydraulic boundary conditions considered), Table 10.2 gives results for the first levee from SEEP2D-COUPLED-HPC (coupled seepage/compressive stress), Table 10.3 gives results for the second levee from SEEP2D-TRAN-HPC, and Table 10.4 gives results for the second levee from SEEP2D-COUPLED-HPC.

11. Analysis. The following observations are made:

- When stresses in the levee were not considered, there is no appreciable change in the calculations of flow in the levee (Tables 10.1 and 10.3).
- When accounting for force and hydraulic boundary conditions in a coupled analysis, drastic increases of the PUP for under seepage and uplift occurred (Tables 10.2 and 10.4).
- The very rapid increase in the water level between time = 0 and time = 0.1 days in the second levee caused a rapid increase in pore pressures that then dissipated slowly as the water level stayed relatively constant. This is predicted behaviour in that excess pore pressures build up and slowly decrease over time. It also explains why PUP for under seepage and uplift gradually decreased although the water level of the flood remained almost constant.
- Through seepage did not encounter any PUP because of the slowness of water flow in the embankment.
Monte Carlo Simulations of Coupled Transient Seepage Flow and Compressive Stress in Levees

Table 10.2
Probability of unsatisfactory performance for the first levee for under seepage, through seepage, and uplift for coupled seepage/compressive stress.

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Table 10.3
Probability of unsatisfactory performance for the second levee for under seepage, through seepage, and uplift for hydraulic boundary conditions only.

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Table 10.4
Probability of unsatisfactory performance for the second levee for under seepage, through seepage, and uplift for coupled seepage/compressive stress.

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12. Conclusions. The challenge of obtaining a stable solution of coupled flow/compressive stress in an HPC environment performing 6000 realisations of material properties was successfully met, despite the need to solve a non-linear system of equations at each time step. The results of the research showed that a coupled analysis is essential for considering failure modes of under seepage, through seepage, and uplift in clay levees. Further, if the resulting pore pressures at the nodes of the finite element mesh for clay levees are to be input into slope stability programs, the coupled analysis is also essential for a more accurate representation of the real-world phenomenon of slope failure. Further research is needed if a faster iterative solver instead of the slow direct solver is desired for the coupled program. The computer runs for the coupled option took as long as one hour, indicating that this type of analysis requires HPC resources to perform.
REFERENCES


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