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   Mohammed A. Younus Al-Sa'ati, Department of Administrative Affairs, The presidency of the University of Mosul, Mosul - Iraq
   Nashwan Mahmod Abdullah, Al-hadba University College, Mosul - Iraq

   A Beowulf Cluster is a type of apportioned parallel processing system, which consists of a collection of reticulated standalone computers working together as a single integrated computing resource generally having a single system image (SSI), that is the users generally view the clusters as a single system. The relatively low cost of two commodity components which are the fast CPUs designed primarily for the personal computer and networks designed to connect personal computers together (in local area network or LAN) makes full advantage of the use of Beowulf Cluster, in this paper we benefit from these components to build larger system. The model was implemented in this paper is using the message passing interface technique developed in C language and use Linux operating system and the goal is to build Beowulf cluster to solve large mathematic operation faster as an example for matrix multiplication and PI problem …etc. the same approach can be used in scientific applications that need supercomputing power or in various other areas like databases, multimedia, web services, etc. In addition the users can access any node of the cluster and use it independently as a local personal computer.

   Keywords— Parallel processing system; Network; Networking and Systems; Linux.

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   Shahzad Khan, Shahzad Shahid Peracha, Zain Ul Abideen Tariq
   Department of Communications System Engineering, School of Electrical Engineering and Computer Sciences (SEECS), National University of Sciences and Technology (NUST), Islamabad, Pakistan

   Abstract - The stream cipher A5/2 is used in GSM (Global System for mobile Communication) for authentication and data encryption. There have been numerous successful attacks that were launched on A5/2 hence breaking down its security. In this paper an evaluation of Cipher-text only attack is presented with an easy understanding of the equation solver; how the equations are generated and solved. Furthermore this paper also reviews that how hardware-only attacker can easily recover the initial states of A5/2 that is more than enough in decrypting all other frames without any pre-computation and storage of the information. It also tries to suggest corrections in the design, if any, based on the deeper analysis of the operations.

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   Syed Foysol Islam, Faculty of Engineering, University Of Development Alternative (UODA), Dhaka, Bangladesh
   Fahmi Ahmed, Faculty of Engineering, University Of Development Alternative (UODA), Dhaka, Bangladesh

   Abstract — In this Research paper, the simulations of call drop and handover failure in GSM network tele-traffic through OMNeT++ are presented. The results obtained in different scenarios are examined and analyzed which simulates a large business city in busy hour a number call attempts by the mobile phone users, with different characteristics of network coverage. This simulator is a discrete event simulator programmed in OMNeT++, focusing on the research of wireless or wired networks. It is also a flexible environment which allows its extension
to different aspects of GSM technology, such as the simulation of successful calls, call drops and handover failure probabilities etc.

Keywords - Graphical NED Editor; Integrated Development Environment; Mobile Station; Base Transceiver Station; Integrated Services Digital Networks; OMNET;

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Marzieh Izanlou, Dept. of Electrical and Computer Engineering, Science and Research Branch, Islamic Azad University, Tehran, Iran
Mohamadali Pourmina, Faculty of Electrical and Computer Engineering, Islamic Azad University, Tehran, Iran
Afrouz Haghbin, Faculty of Electrical and Computer Engineering, Islamic Azad University

Abstract — MIPv6 is a proper replacement for MIPv4 protocol which recommended by IETF. IPv6 lieu IPv4 has been chosen as convergence layer for next heterogeneous access networks. MIPv4 has limiting in protocol, but MIPv6 is created fundamental changes such as security enhancements, elimination of the Foreign Agent (FA) and route optimization. The MIPv6 characteristics defined by the IETF provides perspicuous host mobility within IPv6 networks. In MIPv6 MN is move between IP subnets without change in its original IPv6 address configuration. This means that MN ever is addressable in the internet via its Home Address (HoA). HoA is IPv6 address that is allocated to the MN in its home network. When away from the home network, MN can still detect by its HOA in the internet, Because packets routed to its HoA. Also In this way, mobility transparency of higher layer protocols like Transport layer or higher is achieved.

Keywords- MIPv6; routing; roaming capability

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Maan younus Abdullah, University of Mosul, Education College /computer science department, Mosul, Iraq

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Index Terms - DCTC, Voronoi, Tracking

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Dr. Rabiah Ahmad, Department of System & Computer Communication, Universiti Teknikal Malaysia Melaka (UTeM), Melaka, Malaysia
Prof. Shahrin Sahib, Department of System & Computer Communication, Universiti Teknikal Malaysia Melaka (UTeM), Melaka, Malaysia
M. P. Azuwa, Research Department, Cybersecurity Malaysia, Selangor, Malaysia

Abstract— Technical security metrics provide measurements in ensuring the effectiveness of technical security controls or technology devices/objects that are used in protecting the information systems. However, lack of understanding and method to develop the technical security metrics may lead to unachievable security control
objectives and incompetence of the implementation. This paper proposes a model of technical security metric to measure the effectiveness of network security management. The measurement is based on the effectiveness of security performance for (1) network security controls such as firewall, Intrusion Detection Prevention System (IDPS), switch, wireless access point, wireless controllers and network architecture; and (2) network services such as Hypertext Transfer Protocol Secure (HTTPS) and virtual private network (VPN). We use the Goal-Question-Metric (GQM) paradigm [1] which links the measurement goals to measurement questions and produce the metrics that can easily be interpreted in compliance with the requirements. The outcome of this research method is the introduction of network security management metric as an attribute to the Technical Security Metric (TSM) model. Apparently, the proposed TSM model may provide guidance for organizations in complying with effective measurement requirements of ISO/IEC 27001 Information Security Management System (ISMS) standard. The proposed model will provide a comprehensive measurement and guidance to support the use of ISO/IEC 27004 ISMS Measurement template.

Keywords- Security metrics; Technical security metrics model; Measurement; Goal-Question-Metric (GQM); Effective measurement; Network security management

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Anu Chaudhary, Department of information Technology, AKJ Institute of Technology and Management, GZB, India
K.K Gautam, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India
Nirbhay Aihlawat, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India

Abstract - The capability to provide network service even under a significant network system element disruption is the backbone for the survival of network technology in today’s world, keeping this view in mind, the present paper highlights cryptosystem and Cross-Layer Protocol. A global initial key distribution method based on public key certificate chain shall be presented. And also present a method for survivability strategy in mobile network.

Keywords: Survivability, Mobile Network, Key Distribution Cross-layer protocol

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Anu Chaudhary, Department of information Technology, AKJ Institute of Technology and Management, GZB, India
K.K Gautam, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India
Nirbhay Aihlawat, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India

Abstract:- The optimal and distributed provisioning of high through output in Mobile Ad Hoc Network (MANET) is a network consisting of a set of wireless mobile routers and Communication with each other. The Network Mobility(NEMO) for the traffic represents the moving behavior of directional antenna and mobile rooters. Use the Cross-layer protocol in ad hoc wireless network we can drastically improve the utilization through overlapping communication is the different direction for the traffic. This paper highlight the challenge to find out a route of effect the cross-layer protocol for traffic-management in Ad Hoc wireless network. In present paper we propose mobility for traffic management in Ad Hoc wireless network by use of theory of Cross-layer protocol.

Keywords:- Ad hoc Network, Cross-layer protocol, Directional Antenna, Mobile Router, Network Mobility

9. Paper 31051403: Performance of and Traffic management for a Mobile networks by using Cross-layer protocol (pp. 54-58)

Anu Chaudhary, Department of information Technology, AKJ Institute of Technology and Management, GZB, India
K.K Gautam, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India
Nirbhay Aihlawat, Department of Computer Science & Technology, Phonics Group of Institutions, Roorkee, India
Abstract:- Over the Recent years a considerable amount of effort has been devoted towards the traffic management and root is the important capability to provide best network technology in today’s world. Present paper we study the traffic management for mobile networks and we addresses current issue of the traffic management. Present the performance of Mobile Network by using Cross-layer protocol.

Index Terms:- Mobile Networks, call admission control, QoS (Quality of Service), route optimization


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Abstract — Anonymity has become a significant issue in security field by recent advances in information technology and internet. The main objective of anonymity is hiding and concealing entities’ privacy inside a system. Many methods and protocols have been proposed with different anonymity services to provide anonymity requirements in various fields until now. Each anonymity method or protocol is developed using particular approach. In this paper, first, accurate and perfect definitions of privacy and anonymity are presented then most important problems in anonymity field are investigated. Afterwards, the numbers of main anonymity protocols are described with necessary details. Finally, all findings are concluded and some more future perspectives are discussed.

Keywords-anonymity; privacy; online security
A Beowulf Cluster is a type of apportioned parallel processing system, which consists of a collection of reticulated standalone computers working together as a single integrated computing resource generally having a single system image (SSI), that is the users generally view the clusters as a single system.

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Keywords— parallel processing system ; Network; Networking and Systems; Linux.

I. INTRODUCTION (HEADING 1)

The increasing need of computing power, and the high cost of supercomputers and their low accessibility have all led us to the research in clusters that are providing services similar to supercomputers at a low cost. Clustering has come a long time ago since the beginnings of the 1960s with the advent of high capability of the microprocessors and high speed networks. This has gained further momentum with the development of standard tools for high performance distributed computing. Clusters give us the advantage of using low cost PCs over a network that provides us a cost effective form of parallel computing. This Concept has led research institutions in discussing the possibility of sharing computing resources and the ability to meet the needs is a major consideration towards developing new systems. Based on this knowledge and experience, HPC with Linux clusters are considered in order to build a parallel computing system that will act as core role in the next-generation systems for supercomputer. To achieve that we used "Beowulf Cluster". [1]

A Beowulf Cluster is a kind of supercomputers. More specifically, is an apportioned parallel computer built from commodity components. This approach takes advantage of the astounding performance now available in commodity personal computers. By many measures, including computational speed, size of main memory, available disk space and bandwidth, a single PC of today is more powerful than the supercomputers of the past. By harnessing the power of tens to thousands of such low cost but powerful processing elements, you can create a powerful supercomputer.

a computer cluster consists of a set of loosely connected computers that work together so that in many respects they can be viewed as a single system. Clusters are usually deployed to improve performance and/or availability over that single computer, while typically being much more cost-effective than single super computers of comparable speed or availability. A cluster is a group of linked devices (computer or embedded devices), working together closely so that they form a single node virtually. The components of a cluster are generally, but not always, connected to each other through wireless or wired (Ethernet) that allows data to move between the nodes. Nodes come in many types but are usually built from processors designed for the PC. If a node contains more than one processor, it is called an SMP (Symmetric Multiprocessor) node. [2]

The main purpose of a Beowulf cluster is to perform parallel computations for solving large mathematical operations very fast. This is accomplished by running applications across many nodes simultaneously. These applications may perform in parallel; that is, they may need to coordinate during execution. On the other hand, they may be performing an
embarrassingly parallel task, or a large group of serial tasks. The main key factor in application performance in all cases is local node performance.

The PXE (Pre-execution environment) is a protocol by which nodes can boot the system based on a network-provided configuration and boot image. The system is implemented as a combination of two common network services. First, a node will DHCP (Dynamic Host Configuration Protocol) for an address. The DHCP server will return an offer and lease with extra PXE data. This extra data contains an IP address of a local node performance.

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system on each compute node to switch over the system is generally difficult, as would maintaining multiple hard disk partitions with different operating systems or configurations. In such cases, building the Beowulf without the operating system on the local hard drive, if it even exists, can be a good solution. Diskless operation also has the added benefit of making it possible to maintain only one operating system image, rather than having to propagate changes across the system to all of the Beowulf nodes.

D. Secure shell

SSH allows for encrypted communication to and from the master node. The SSH daemon is the secure replacement of RLOGIN (remote login), RSH (Remote Shell) and Telnet (Telnet is a network protocol used on the Internet or local area networks to provide a bidirectional interactive text-oriented communications). The OpenSSH package is not installed automatically with Ubuntu, which means the SSH remote access clients like SSH (Secure shell) and SCP (Secure Copy) are not available to users immediately. The SSH service should be downloaded.

Once SSH has been downloaded the root user should be able to remotely access any of the nodes in the cluster. This ability can be tremendously useful when one needs to replicate configuration files across several nodes of the cluster or to restart a service without being at the console of the specific node.

E. Parallel Programming with MPI

Message Passing Interface (MPI) is an application program interface speciation that allows processes to communicate with one another by sending and receiving messages between them. It is typically used for parallel programs running on supercomputers, where the cost of accessing non-local memory is high [6]. The processes have separate address spaces they communicate by sending and receiving messages. Each process would be running on a separate node. MPI is also supporting shared memory programming model. This means that multiple processes can read or write to the same memory location.

Sometimes normal program can be used by all the processes, but with distinct parameters. In this case, no communication occurs among the separate tasks. When the strength of a parallel computer is needed to attack a large problem with a very complex structure, however, such communication is necessary [7].

F. Measuring MPI Performance

There are many tools have been developed for performance measuring like MPPTEST program and The SKaMPI test suite. To get best test results is always obtained of own application, but a number of tests are available that can give a general overview of the performance of MPI on a cluster.

G. Hardware Considerations

Building a cluster, access to computers on which to install the software is essential. Therefore, it makes sense to cover this early in the process.

For sure it is necessary to have at least two computer machines when building a cluster. It is not essential that these machines have the same levels of performance and specifications. The only main requirement is that they both share identical architecture. For instance, the cluster should only consist of all Intel machines or all Apple machines but not a mixture of the two. In theory it is possible to mix architectures when building a cluster by using Java, but that is outside the scope of this paper.

Strictly speaking, the only hardware requirements when building a cluster is two computers and some type of networking hardware to connect them with.

1. Clusters specification

To get full benefits of a cluster, the right hardware must be used. For optimal performance, all nodes except the master node should have same hardware specifications. This is because the fact that one node which takes longer to do its work can slow the entire cluster down as the rest of the nodes must wait for the slow node to catch up. This is not always the case, but it is a consideration that must be made. Having identical hardware specs also simplifies the setup process a great deal as it will allow each hard drive to be imaged from a master instead of configuring each node individually.

2. The Master Node

There are four main considerations when building the master node. They are Processor speed, Disk speed, Network speed, and RAM.

- Processor Speed

If the master node is participates in computation this will be critical. Many more tasks will be handled by master node than the slave nodes so a faster processor may be required to keep it from being behind others. Not forgetting that since the master node can be kept quite busy doling out work to the other nodes, a slowdown here can have a huge negative effect on the entire cluster as the slave nodes waste time waiting for their next instruction.

- Disk Speed

As we know, major work is done on the cluster, some time or another it need to be saved as files on a hard drive, disk speed for the master node is absolutely critical, made even more so due to the fact that most nodes make use of NFS (Network File System) which means that every node in the cluster will be competing for access of the master node's disk. A fast SCSI drive is recommended, but an IDE drive will work as well.
• Network Speed

This is critical as well. Time spent transmitting data is time wasted. The faster the network, the better the performance of the cluster. This can be mitigated by a good deal if the programmer expressly tries to minimize the ratio of time on the network to time on the processor but it never hurts to have more network speed. Fast Ethernet is recommended, Gigabit Ethernet is ideal but basically any network speed will work. While not part of the master node per se, it is strongly recommended that a switch be used instead of a hub when designing the cluster network.

• RAM

In the master node RAM is crucial for two reasons. First, the more RAM, the more processes can be run without ingressing the disk. Second, the Linux kernel can and will cache its disk writes to memory and keep them there until they must be written to disk. Both of these raise the speed of the master node which is critical to good overall cluster performance.

3. Slave Nodes

The slave nodes need to execute two tasks: Perform the computations assigned to them and then send that data back out over the network. So, their disk performance is not critical. In fact, it is normal to have nodes without hard drives in a cluster. These diskless nodes reduce the cost of building a cluster and eliminate some of the time required to set a cluster up. This document, however, assumes that the slave nodes will have no hard drives DRBL(Diskless Remote Boot in Linux).

The three most important hardware considerations for slave nodes are processor speed, network speed and RAM.

• Processor Speed

Nodes primary function is executing mathematical tasks, it makes sense that the fastest processor should be used. The more processing power the better. Multiple processors for the nodes can be desirable but add another degree of complexity for programming an applications for the clusters. Not only must the programmer take distributed processing into consideration, but SMP as well. As of the time of this writing, Intel Core I 5's offer a good price/performance ratio.

• Network Speed

This affects the slave nodes in exactly the same way that it does the master node. See that section above for more information.

• RAM

This affects the slave nodes in exactly the same way that it does the master node. See that section above for more information.

IV. DESIGN THE COMPUTER CLUSTER

Designing the Linux cluster model need a collection of Personal Computers (PCs) connected in one network together as a single resources in order to share their processors and other resources for computations and analysis that could be performed on any parallel machine. The cluster consists of a PC designated as the master while the other PCs on the network are the computational nodes as slaves.

The technology of cluster [8] that is being used is all active, that is, there is no primary or backup nodes. The cluster is designed from a set of heterogeneous mixture. The systems are networked together using a Fast Ethernet architecture of 100Mbps for data transfer and the cluster is designed in such a way that the nodes can access the master node and checks the status of the master through network commands issued from the node. Users should be able to log on to the master nodes through the client nodes.

The main steps of building the computer cluster is described in the following flowchart and this chapter will discuss each step alone:

Flowchart (1) main steps of building the computer cluster.

• Running MPI Program

In case of run a program with multiple inputs, a parallel client-server implementation might just run multiple
copies of the code serially with the server granting the different inputs to each client process. As each processor finishes its task, it is assigned a new input. Alternately, task parallelism can be implemented at a deeper level within the code.

V. TESTING PERFORMANCE AND RESULTS

The biggest challenge we had to do for the use of a Beowulf cluster was the conversion of an existing serial code to a parallel code based on the message passing philosophy. The main difficulty with the message passing philosophy is that we have to ensure that master node is distributing the workload equally between all the other nodes. Because all the nodes have to synchronize at each time step, each PC should finish its task in about the same period of time. If the load is uneven or if the load balancing is poor, the PCs are going to synchronize on the slowest node, leading to a worst situation.

Another hitch is the possibility of communication patterns that can deadlock. A typical example is if PC A is waiting to receive information from PC B, while B is also waiting to receive information from A.

• Calculating Value of Pi

A program to calculate the accurate value of mathematical constant PI (3.14) was evaluated for elapsed time and error in the calculated value. The benchmark value of PI was considered up to 25 decimal places whereas cluster computed the value up to 16 decimal places. Hence the error could be identified for 9 decimal places in accuracy as compared to the benchmark value using 25 decimal places. The error was observed to show very minor change which negligible and hence we focused mainly on the execution time of the program. Extensive use of machine file was made for submitting the processes.

The processes are allowed to move back and forth the master node depending upon the free resources. For example, say master node was allowed two processes. When the third process is to be scheduled, it will be scheduled on one of the slave node. In the meantime the fourth process is also queued and also one of the process on master node is terminated. Then there is no need to send the fourth process to the slave node, it will be executed on the master node itself.

Hence, processes were allocated dynamically depending on the free resources on a node, be it master or a compute node. Also note that the empty cells in table indicate that no output was obtained due to submission of processes beyond capacity of cluster.

<table>
<thead>
<tr>
<th>Number of Processes</th>
<th>Node = 1</th>
<th>Node = 3</th>
<th>Node = 5</th>
</tr>
</thead>
<tbody>
<tr>
<td>5</td>
<td>0.001481</td>
<td>0.001000</td>
<td>0.001211</td>
</tr>
<tr>
<td>10</td>
<td>0.001641</td>
<td>0.003463</td>
<td>0.00357</td>
</tr>
<tr>
<td>50</td>
<td>0.047269</td>
<td>0.108689</td>
<td>0.11757</td>
</tr>
<tr>
<td>100</td>
<td>0.115619</td>
<td>0.252974</td>
<td>0.2752</td>
</tr>
<tr>
<td>200</td>
<td>-</td>
<td>0.55463</td>
<td>0.584591</td>
</tr>
<tr>
<td>300</td>
<td>-</td>
<td>0.98335</td>
<td>0.993251</td>
</tr>
<tr>
<td>400</td>
<td>-</td>
<td>2.901754</td>
<td>2.925887</td>
</tr>
<tr>
<td>550</td>
<td>-</td>
<td>-</td>
<td>3.102552</td>
</tr>
</tbody>
</table>
in the input data, which includes the number of slaves to be spawned, \textit{numtasks}. Next, registering with nodes and receiving a \textit{taskid}, then distributes the input graph information to each of them. The server obtains the result from each of the slaves. Since each slave needs to work on a distinct subset of the set of matrix elements, they need to be assigned instance IDs in the range (0... \textit{numtasks}-1). The source code for serial and parallel is shown in Appendix A.

The matrix multiplication was run with forking of different numbers of tasks to demonstrate the speedup. Problem sizes were 100*100, 200*200, 300*300, 400*400, 500*500 and 600*600 in our experiments with one node attached to the master node. It is well known, the speedup can be defined as $Ts/Tp$, where $Ts$ is the execution time using serial program, and $Tp$ is the execution time using multiprocessor.

The execution times and corresponding speedups by using 20 processes with different problem sizes were listed in Figure 2. In, the corresponding speedup is increased for different problem sizes compared with the same problem size executed in serial.

![Figure 2](image1.png) Execution times and corresponding speedups by using 20 processes compared with serial execution

![Figure 3](image2.png) The speedup graph as $Ts/Tp$.  

I. CONCLUSION

Scalable computing clusters are rapidly becoming the standard platforms for high performance and large-scale computing. It is believed that message-passing programming is the most obvious way to help programmer to take advantage of clustering symmetric multiprocessors (SMP) parallelism. A high performance computing cluster is built using DRBL and OpenMPI. The usage of DRBL simplifies the building process and maintenance. A hard disk is installed on each client node only to provide the local swap space. This cluster with system less clients can be one with diskless clients if the memory capacity on each clients is large enough.

It can be observed from the execution of the MPI programs that number of nodes in a cluster must be in accordance with the target application. Also that a larger application needs more number of compute nodes else the problem cannot be solved due to shortage on resources. The time required for process migration and consolidation of the result on the master node increase with increase in number of nodes. Thus it can be noted that number of nodes must be increased with a care so that performance gain can be genuinely achieved.

The performance test based on the testing codes using MPI shows consistent results. It is demonstrated that such a high performance computing cluster can be constructed from scratch for potential applications on computational physics problems.

As the number of compute nodes in modern HPC clusters continues to grow, it is critical to design clusters with low power consumption and low failure rate. In particular, it is widely known that the internal disk drives of compute nodes (in the case of disk full clusters) are a major source of failures. In addition, these disk full HPC clusters tend to require more power and cooling requirements compared to diskless clusters.

The advantages of Beowulf Diskless Remote Boot computing cluster are evident for any organization that requires high computational power. This is, when we take into account the performance/price ratio, easy scalability and upgradeability and recycling properties of the hardware components. If this is true for any organization, we are convinced that it is imperative for an academic institution like our University. Therefore we make a proposal of deployment of such a device starting with a schematic installation to be eventually enlarged and improved.

REFERENCES


AUTHORS PROFILE

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Nashwan Mahmod Abdullah a lecturer in Al-hadba University College. The address of the thesis is digital image compression for bandwidth reduction using jpeg standard. I am interest in computer engineering. Computer networking and communications. the subjects are microprocessor programming. digital image processing. Computer applications.
Evaluation of Cryptanalytic Algorithm for A5/2 Stream Cipher

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Abstract—The stream cipher A5/2 is used in GSM (Global System for mobile Communication) for authentication and data encryption. There have been numerous successful attacks that were launched on A5/2 hence breaking down its security. In this paper an evaluation of Cipher-text only attack is presented with an easy understanding of the equation solver; how the equations are generated and solved. Furthermore this paper also reviews that how hardware-only attacker can easily recover the initial states of A5/2 that is more than enough in decrypting all other frames without any pre-computation and storage of the information. It also tries to suggest corrections in the design, if any, based on the deeper analysis of the operations.

Keywords- A/5, GSM, cryptanalysis, stream ciphers

I. INTRODUCTION

GSM (Global System for Mobile Communication) the digital cellular system that has covered the entire mobile communications in Europe and Asia. As in mobile communication, the main dealing is with real time applications such as data, voice and video so using stream cipher is the best option for achieving authentication and data encryption. GSM uses A5/1 Stream Cipher for the very same purpose. This algorithm is implemented in Europe and is stronger version. Its variant A5/2 is used in Asia though it is a weaker version. The design of both the ciphers were confidential but were revealed in 1999 by reverse engineering.

There are some core flaws in these ciphers that are exploited and hence the security of entire GSM is easily be compromised. In any case if mobile phone supports a weaker cipher the security can be compromised whatever algorithm for security is used by GSM.

The attacks on A5/2 have been mostly based on software implementation and the efficiency count is also based on software but in this paper we analyze hardware implementation of the attack with detailed illustrations and minor adjustments to the existing algorithm. With the help of cipher-text only attack we achieve our goal of breaking the security of A5/2 without any pre-computation and storage. Hence our primary focus is on providing a review over the cipher-text attack only based on hardware implementation.

II. BRIEF DESCRIPTION OF A5/2 STREAM CIPHER

A5/2 is a stream cipher in which sender and receiver must be synchronized as this cipher is synchronous stream cipher requiring key stream, plain text and producing cipher text by XORing the plaintext with the key stream. A5/2 requires 64 bit key that we denote by K= (k₀, k₁, k₂, k₃, ……….. k₆₃) the key must be belonging to GF(2⁶₄) it also required the 22 bit frame number that acts as an Initialization Vector (IV). The Initialization Vector must be defined under the GF (2²²) we identify the IV = (IV₀, IV₁, ….. IV₂₁). There is no privacy in the frame number as it is publicly known. In A5/2 there are four Linear Feedback Shift Registers (LFSRs). The length of each LFSR is relatively prime to each other. We recognize them R₁, R₂, R₃ and R₄ and the length of each register is 19, 22, 23, 17 bits respectively. In order to retain the desired properties of the LFSR we choose the primitive polynomial with maximum period and large linear complexity. R₁, R₂ and R₃ are the registers used for producing the key stream and R₄ is used to control the remaining 3 registers with the help of clocking signals. The internal structure of A5/2 is described in Table 1.

A. Initialization Phase

Initially the LFSRs are filled up with the 64 bit values of the key K, but before this all the registers are filled up with the 0. The key bits are inserted one bit at a time to all the registers in parallel. The first bit of the key is XORed with the ith position of register; each register is filled up with the 64 bit key. After every cycle the registers are clocked unconditionally. The similar step is followed for the Initialization Vector (IV) and its 22 bit frame number is inserted in the registers.
This process is summarized in following four steps:

**Step 1:** Initially all registers are filled with 0
\[ R1 = 0; R2 = 0; R3 = 0; R4 = 0 \]

**Step 2:** 64 bits of Key (K) from 0 to 63 bits are inserted
\[ R1, R2, R3, R4 \text{ are regularly clocked} \]
\[ R1[0] = R1[0] \oplus K_i \]
\[ R2[0] = R2[0] \oplus K_i \]
\[ R3[0] = R3[0] \oplus K_i \]
\[ R4[0] = R4[0] \oplus K_i \]

**Step 3:** 22 bits of Initialization Vector (IV) from 0 to 21 bits are inserted
\[ R1, R2, R3, R4 \text{ are regularly clocked} \]
\[ R1[0] = R1[0] \oplus IV_i \]
\[ R2[0] = R2[0] \oplus IV_i \]
\[ R3[0] = R3[0] \oplus IV_i \]
\[ R4[0] = R4[0] \oplus IV_i \]

**Step 4:** R1 [15], R2 [16], R3 [18] and R4 [10] are assigned 1

**B. The Key Generation Phase**

After the initialization phase the register R4 is clocked 99 times and the output is discarded. After this phase the registers R1, R2 and R3 are clocked irregularly based on the majority bits of Register R4. The clocking is determined by the bits R4[3], R4[7], and R4[10] in each clock cycle. The majority of the three bits are computed, and the registers R1, R2 and R3 are then clocked based on the majority function. R1 is clocked if R4[10] agrees with the majority. R2 is clocked if R4[3] agrees with majority and R3 is clocked if R4[7] agrees with the majority bit. In this way the registers are clocked irregularly and in each cycle at least two of the three registers are clocked.

**C. Output Stream Bit Generation**

In each register the majority of two bits and the complement of a third bit is calculated.
\[ R1; \text{majority (bit 12, complement of bit14, bit15)} \]
\[ R2; \text{majority (bit 9, bit13, complement of bit16)} \]
\[ R3; \text{majority (complement of bit 13, bit16, bit18)} \]

The result of each majority bit and the right most bit of each register is XORed giving out the output bit. In this fashion 228 bits are generated the first 114 bits are used to encrypt the link from network to the subscriber and the remaining 114 bits are used to encrypt the link from subscriber to the network.

**III. Cryptanalysis Of A5/2 Cipher**

The cryptanalysis of the A5/2 stream cipher is presented in detail in [1] along with its hardware implementation. We present the an overview of each the individual blocks of hardware implementation of the proposed algorithm with some minor adjustments that we feel have been ignored. The cryptanalytic attack on the A5/2 stream cipher exploits some properties of the cipher blocks to deduce the initial secret states of the LFSRs. Before we look into the cryptanalytic architecture we elaborate the process of encryption tracking up to bit level the impact on the equations that are generated by the proposed architecture.

<table>
<thead>
<tr>
<th>LFSR</th>
<th>Length of LFSR</th>
<th>Primitive Polynomial</th>
<th>Clocking bit</th>
<th>Tapped bits</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>19</td>
<td>(x^{18} + x^{17} + x^{16} + x^{13} + 1)</td>
<td>8</td>
<td>13, 16, 17, 18</td>
</tr>
<tr>
<td>2</td>
<td>22</td>
<td>(x^{21} + x^{20} + 1)</td>
<td>10</td>
<td>20, 21</td>
</tr>
<tr>
<td>3</td>
<td>23</td>
<td>(x^{22} + x^{21} + x^{20} + x^{7} + 1)</td>
<td>10</td>
<td>7, 20, 21, 22</td>
</tr>
<tr>
<td>4</td>
<td>17</td>
<td>(x^{16} + x^{11} + 1)</td>
<td></td>
<td>16, 11</td>
</tr>
</tbody>
</table>
A. Brief Overview of Encryption Process

We have a brief overview of the encryption process as discussed by Bogdanov, Eisenbarth, and Rupp in their paper titled “A Hardware-Assisted Realtime Attack on A5/2 Without Precomputations” [1], to understand the impact of bit level processes on the cryptanalysis. The data blocks ID_0, ID_1 and ID_2 as shown in Fig 2 are passed through error correction coding block. In this case the error correction is provided by convolutional coding that adds redundancy on the data blocks and transforms 267 bit blocks into 456 bit blocks i.e CD_0, CD_1 and CD_2 respectively. That is further passed on to another block that performs reordering and interleaving on the coded blocks to evade the effect of burst errors that spreads out the errors in multiple blocks that appear as isolated errors easily corrected. The result of the reordering and interleaving block is the sixteen plaintext blocks. Thus we have data of three 456 bit blocks CD_0, CD_1 and CD_2 spread over sixteen 114 bit blocks P_0, P_1 and so on upto P_15. The A5/2 key stream generator takes in the initial 64 bit Key ‘K’ and 22 bit initialization vector IV (IV_0, IV_1, ..., IV_21) and generates sixteen 114 bit key stream blocks S_0, S_1 and so on upto S_15. The stream blocks are XORed with the plaintext blocks to give cipher text blocks C_0, C_1 and so on upto C_15. This completes the encryption process as illustrated in Fig2.

We briefly narrow down to the reordering and interleaving block to see what really happens inside. As illustrated in the detailed diagram in Fig3, the three 456 bit blocks under consideration CD_0, CD_1 and CD_2 are reordered and interleaved resulting in the chunks of eight 57 bit blocks. The details of interleaving are narrowed down in the following section. The data of the three 456 bit blocks is spread over sixteen 114 bit plaintext blocks. As shown in Fig 3, the data of the block CD_0 is spread over eight blocks P_0, P_1, ..., P_7. Similarly the data of block CD_1 is covered by eight blocks P_8, P_9, ..., P_15 and that of block CD_2 is covered by eight blocks P_16, P_17, ..., P_31. The bits in the first four 57 bit chunks of CD_0 are placed at the even positions of the plaintext blocks P_0, P_2, P_4 and P_6 while the odd positions are covered by the bits of last four chunks of the preceding block. Similarly the bit in the last four 57 bit chunks of CD_0 are placed at the odd positions of the plaintext blocks P_8, P_10, P_12 and P_14 while the even positions are covered by the bits of first four chunks of the CD_1 block. The similar procedure follows for CD_1 and CD_2. We notice a specific property of the 456 bit encoded blocks CD_0, CD_1 and CD_2. This property can be summarized in following equation:

\[ \text{cdi,2j} \oplus \text{cdi,2j+1} \oplus \text{cdi,2j+2} \oplus \text{cdi,2j+3} \oplus \text{cdi,2j+6} \oplus \text{cdi,2j+8} \oplus \text{cdi,2j+9} = 0 \quad \text{where} \quad 0 \leq j \leq 184 \quad (1) \]

We exploit this specific property for cryptanalysis to find the initial secret states of the LFSRs of the A5/2 stream cipher. We notice that for any 456 bit block CD_i we get 185 different equations satisfying above condition. Since the data of each of the CD_i block is spread over eight plaintext blocks, we select the span of eight plaintext blocks to get the required bits.
satisfying above condition. These bits are carried forward onto the ciphertext blocks thus our focus shifts to eight consecutive stream bit blocks and respective ciphertext blocks satisfying above equation. This phenomenon is illustrated in Fig 3. The equation that we get becomes:

\[ c_{f}(i, 2j) \oplus c_{f}(i, 2j+1) \oplus c_{f}(i, 2j+2) \oplus c_{f}(i, 2j+3) \oplus c_{f}(i, 2j+6) \]
\[ \oplus c_{f}(i, 2j+8) \oplus c_{f}(i, 2j+9) \oplus s_{f}(i, 2j) \oplus s_{f}(i, 2j+1) \oplus s_{f}(i, 2j+2) \oplus s_{f}(i, 2j+3) \oplus s_{f}(i, 2j+6) \]
\[ \oplus s_{f}(i, 2j+8) \oplus s_{f}(i, 2j+9) = p_{f}(i, 2j) \oplus p_{f}(i, 2j+1) \oplus p_{f}(i, 2j+2) \oplus p_{f}(i, 2j+3) \oplus p_{f}(i, 2j+6) \]
\[ \oplus p_{f}(i, 2j+8) \oplus p_{f}(i, 2j+9) \]
\[ = c_{d,i,1} \oplus c_{d,i,2} \oplus c_{d,i,3} \oplus c_{d,i,4} \oplus c_{d,i,5} \oplus c_{d,i,8} \oplus c_{d,i,10} \oplus c_{d,i,11} = 0 \]

where \( j = 1 \) (3)

After reordering and interleaving we observe from Fig 4, that this equation depends on five plaintext blocks and consequently five ciphertext and stream bit blocks. For the equation (3) the blocks are \( P_0, P_2, P_3, P_4, P_5 \). We refer to Fig 4 in the following section to see how this information helps us in understanding the cryptanalysis architecture. From this point we move on to the brief overview of the cryptanalysis process.

B. Overview of Cryptanalytic Process

As discussed in the previous section, we exploit the property of the encoded blocks that reflects in different form when going through the interleaving and reordering process. We keep track of the bits...
required to satisfy one of the equations given as equation (3). The overview of the cryptanalytic architecture is given in Fig 5. The Process of cryptanalysis begins by identifying the main unknown values that are to be sorted out to break the cipher. During the process what we are available with are sixteen cipher text blocks from air. Then we know that the stream blocks used to encrypt the plaintext blocks are unknowns. To determine the stream bits we go back into the process of encryption and identify the secret initial states of the LFSR’s being the main candidate of unknowns to be found out. We denote the initial secret states of the LFSR, as $\alpha_0, \alpha_1, \alpha_2, \alpha_3, \alpha_4$, and so on till $\alpha_{77}$. This gives us total of 78 initial secret states that we are required to find out during the process.
During the encryption process and generation of stream bits using majority function of the three LFSRs R1, R2 and R3 we observe that we get different combinations of the terms \(a_6, a_j, a_2, a_s\) to \(a_{0j}\) that contribute to each of the stream bit \(s_{h,k}\). We have following combinations possible:

\[
18c_2 + 21c_2 + 22c_2 + 61 = (18)(17)/2 + (21)(20)/2 + (22)(21)/2 + 61 = 594 + 61 = 655 \text{ (plus a constant)}
\]

The above expression shows that there are 61 values comprising of single variable \(a_6, a_j, a_2, a_s\) to \(a_{0j}\) while 594 values are quadratic combinations of these single variables. We take these quadratic combinations as new variables, summing total variable up to 655. The different combinations of these variables give the stream bits as output. This assumption is used in [1] to propose the cryptanalytic architecture. The said architecture as shown in Fig 5 comprises of five main blocks:

1) Ciphertext Module (CM)
2) Equation Generators (EG)
3) Linear System of Equation (LSE) Solver
4) Key Tester (KT)
5) Control Logic Unit (CLU)

We briefly touch each of these blocks and present an overview of their internal working.

1) Ciphertext Module (CM)

This module contains buffers to store the ciphertext blocks an the Initialization Vectors (IV). For ciphertext block it has 24 memory locations to store the blocks in the groups of eight that are further required for processing by the Equation Generator (EG) block. The Initialization Vectors (IV) are stored in 16 memory modules. The first bit of each of the 24 ciphertext
memory modules is connected to the Equation Generator (EG) module that is rotated forward or backward to access other bits. Similarly the same bit are provided to Key Tester (KT) module.

2) Equation Generator (EG)

The Equation Generator Module consists of three EG sub modules named as EG₀, EG₁ and EG₂. The three sub modules are meant to generate equations for CD₀, CD₁ and CD₂. Each of the EG sub module operates on eight ciphertext blocks and generates 185 equations.

![Fig 5: Overview of Cryptanalytic architecture][1]

- Fig 5: Overview of Cryptanalytic architecture [1]

satisfying the condition given in equation (1). The total of 555 equations generated from the three EG sub modules are then passed on to the LSE Solver for further processing. The three sub modules of EG operate on the eight consecutive ciphertext blocks to generate equations from the data spread over these blocks. The sub module EG₀ operates on blocks C₀ to C₇. The sub module EG₁ operates on the blocks C₈ to C₁₅ and consequently EG₂ operates on blocks C₂₄ to C₂₅. Fig3 clearly shows how the data of CD₀, CD₁ and CD₂ are spread over the plaintext blocks and then carried onward to the ciphertext blocks. Each of the sub modules EGᵢ contains eight Stream Generator SG blocks and a stream combiner (SC) block as shown in Fig 6. Each of the Stream Generators SGᵢ is meant to generate the 655 variable coefficients (along with a constant) for the 114 stream bits associated with each of the eight ciphertext bits. Each ciphertext bit is XORed with the constant value generate from the SGᵢ and then stored in the buffer. The Stream Combiner (SC) has a job of taking appropriate values from the output of each of the SGᵢ and combine them and pass it on to the LSE block. As shown in Fig 4 we observe that we need the current value and some of the old values to form an equation.

a) Suggested Improvement & Analysis

Fig 6 shows the buffers at the output of the SGᵢ but we show with the help of Fig 4 that there needs to be some correction as there are only two buffers to store the current and the previous coefficient values, while we need to have three buffers because we get the right previous value before two stream bits due to the effect of interleaving. We show this correction in Fig 8.

Focusing on Stream Generator (SG) we have architecture as shown in Fig 7 for all three LFSRs. Instead of single row vector LFSRs was have multiple row LFSRs with each row representing 61 unknown variables and a constant value while the locations of these LFSRs represents the coefficients of those variables as their dependencies at those positions of LFSRs. Fig 7 shows the vector LFSR for R₁ only and their respective dependencies. These are explained in detail in [1]. The addition of the extra buffer would increase hardware as 24 memory locations of 656 bits are added in EG module and 24 more are added in KT module. It does not increase the power consumption because only two resisters at a time are clocked to generate the equations. But there will be power overhead for there are two registers being updated for each output of SGᵢ in present design as shown in Fig 6 while after correction three registers are updated as shown in Fig 8.

3) Linear System of Equation (LSE) Solver

The Linear System of Equation (LSE) Solver module buffers in 555 equations in its buffers while each equation is represented by 655 variable coefficients while 655 unknowns are to be found. The process used for this purpose is Guass Jordon Elimination that is discussed in [1].
The equations are arranged in \( m \times n \) matrix where \( m \) is the number of rows and that equals to 555 representing the total number of equations while \( n \) represents the total number of variable inclosing the constant value that is equal to 656. Using this process we are able to get a candidate value of initial state variables \( \alpha_0, \alpha_1, \alpha_2, \alpha_3, \ldots, \alpha_{60} \). This candidate is then passed on to the Key Tester (KT) for verification if the calculated candidate for the initial secret state is correct or not.

4) **Key Tester (KT)**

The key tester module takes the candidate values from the LSE Solver and verifies its correctness. The Key Tester (KT) module contains the same A5/2 encryption architecture. It sets the initial state candidate values in the LFSRs and calculates the output stream bits that are XORed with right ciphertext bits from Ciphertext Module (CM) and then passed on to stream combiner that take in right values, generates equation satisfying condition in equation (q) to test if its values are equal to zero. If any of the values comes out to be one then the candidate fails and new candidate value is generated by LSE Solver and then tested again at Key Tester (KT) module. These values are tested for all possible values of R4 LFSR. Similarly the equations in Equation Generator (EG) module are generated for all possible values for R4 LFSR.

5) **Control Logic Unit (CLU)**

The Control Logic Unit (CLU) is used to control the operations of all the four modules by clocking then and giving out control signals. This is also discussed in detail in [1].
We need three buffers instead of two altogether for each equation to store the current values and the last two values.

**IV. CONCLUSION**

We presented the overview of the cryptanalytic attack on the A5/2 cipher and the proposed hardware architecture to carry out the cryptanalysis and suggested some minor corrections that contribute to the improvement in the design based on the bit level analysis of the data.

**REFERENCES**


Simulation And Modeling Of Handover Failure And Call Drop In GSM Network For Different Scenarios

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Abstract—In this Research paper, the simulations of call drop and handover failure in GSM network tele-traffic through OMNeT++ are presented. The results obtained in different scenarios are examined and analyzed which simulates a large business city in busy hour a number call attempts by the mobile phone users, with different characteristics of network coverage. This simulator is a discrete event simulator programmed in OMNeT++, focusing on the research of wireless or wired networks. It is also a flexible environment which allows its extension to different aspects of GSM technology, such as the simulation of successful calls, call drops and handover failure probabilities etc.

Keywords—Graphical NED Editor; Integrated Development Environment; Mobile Station; Base Transceiver Station; Integrated Services Digital Networks; OMNET;

I. INTRODUCTION

Building a simulation of a telephone system GSM cellular mobile OMNET using the simulator to measure the parameters Recommended minimum quality voice service. The main objective of this research paper is to create the different environments of simulation and measurement on the system of GSM cellular phone. Another objective is to design and construct a simulation of a cellular mobile telephone system GSM by using the simulator OMNET. In this research paper the specific objective is to analyze the operation under minimum parameters on the voice channel in the implementation of a simulation of a cellular telephone system GSM technology through OMNET and analyzing the results of building simulation.

II. SIMULATION TECHNIQUE FOR ANALYSIS

OMNET simulator generates an output of the simulation, which is given into data files, output vector files, output scalar files, and possibly the users own output files. The output vector file allows observing the behavior of each MS in simulation time. This is to analyze the behavior of the minimum standards of quality that must be provided to the GSM Cellular System. OMNET simulator generates an output vector file which allows observing the behavior of each MS in simulation time. This is to analyze the behavior of the minimum standards of quality that must be provided to the GSM Cellular System.

OMNET contains graphical publishers Scalars and Plove which generates result in graphical form, therefore facilitating the analysis of the simulation [1].

III. RESULT ANALYSIS IN DIFFERENT SCENARIOS

A. TABLE-1: General Characteristics of Scenario 1

<table>
<thead>
<tr>
<th>Properties</th>
<th>Characteristics Data</th>
</tr>
</thead>
<tbody>
<tr>
<td>Simulation area:</td>
<td>1 km(^2) (1000 m \times 1000 m)</td>
</tr>
<tr>
<td>Number of MS:</td>
<td>50 mobile stations</td>
</tr>
<tr>
<td>Speed of MS:</td>
<td>(minimal 0.0 m/s) (maximum 7.1 m/s) (average 1.7 m/s)</td>
</tr>
<tr>
<td>Power measurements:</td>
<td>one measure per second</td>
</tr>
<tr>
<td>MSISDN:</td>
<td>6009000xx, where xx is the MS number (from 0 to 49)</td>
</tr>
<tr>
<td>Number of MSC:</td>
<td>1</td>
</tr>
<tr>
<td>Number of BTS:</td>
<td>1</td>
</tr>
<tr>
<td>Transmission power of BTS:</td>
<td>4 dBm</td>
</tr>
<tr>
<td>Position of the BTS:</td>
<td>The simulation area has permanent coverage of the BTS.</td>
</tr>
<tr>
<td>Number of traffic channels of the BTS:</td>
<td>14</td>
</tr>
<tr>
<td>Calls processing in MS:</td>
<td>Exponential function with inter-arrival time of 10 min</td>
</tr>
<tr>
<td>Probability of intra - MSC calls:</td>
<td>0% (every MS calls always to another MS out of the simulation that is assigned to a factitious MSC not present in the simulation).</td>
</tr>
</tbody>
</table>
• Scenario 1:
  There is a single BTS situated in the center, which manages the total area. The transmitted power attenuation of 4 dBm corresponds to a circumference of radius 731 m. This scenario includes 50 MSs moving inside the zone of study, 30 of which have linear trajectories and 20 have random ones. This scenario is quite simple, but contains enough GSM parameters to obtain several conclusions. The configuration file also collects information about the initial position and speed of each MS. Remember that not all MSs have the same speed. The service time distribution is exponential with average service time established in 2 min (120 s). The single BTS is also positioned on the center. Over this point, (500; 500) of the simulation area, the MS has an attenuation which fits with the maximal in the graphic. Due to the BTS having 14 traffic channels to serve the communication demand, the number of busy traffic channels is always less than or equal to 14. When the BTS has the 14 traffic channels assigned, congestion in calls is produced. During congestion, any new call attempts will be rejected. This scenario considers that all calls are out-MSC, i.e. every MS in the simulation area is calling to a fictitious MS depending on another MSC not present in the simulation area [2].

By applying this scenario we simulate the Omnet & get the following Graphs. The different analytical graphs of the same MSs according to the scenario are given below:

**FIGURES OF SCENARIO 1**

**FIGURE-1**: Analysis of the results Received Power & Simulation Time at MS (5) in scenario 1. Here the examples of MS in Scenario-1, shows the received power of each MS from one BTS. This figure shows the graph of MS (5) from BTS (0). In this scenario, there is 50 MS & 1 BTS.

**FIGURE-2**: Analysis of the results Received Power & Simulation Time at MS (10) in scenario 1. Figure-2 shows the graph of MS (10) from BTS (0). Here MS in Scenario-1 shows the received power of each MS from one BTS. In this scenario, there is 50 MS 30 linear & 20 random. Total power is 4 dB. The trajectories of every MS in the area during a busy hour simulation show the BTS coverage. Fig.1 shows MS with 5 identifier and with a linear (path type 0) trajectory. Fig.2 shows MS with 10 identifier and with a random (path type 1) trajectory.

**FIGURE-3**: Analysis of the total channel of BTS (0) in scenario 1. The total Traffic Channel (TCH) is a logical channel that allows the transmission of speech or data. In most second generation systems, the traffic channel can be either full or half rate.

B. **TABLE-2: General Characteristics of Scenario 2**
### Properties | Characteristics Data
---|---
Simulation area: | 4 km² (2000 m × 2000 m)
Number of MS: | 50 mobile stations
  - 30 MS with linear trajectories
  - 20 MS with random trajectories: change direction after lifetime (random between 0 to 200 s)
Speed of MS: | (minimal 0.0 m/s) (maximum 7.1 m/s) (average 1.7 m/s)
Power measurements: | One measure per second
MSISDN: | 6009000xx, where xx is the MS number (from 0 to 49)
Number of MSC: | 1
Number of BTS: | 3
Transmission power of BTS: | 7 dBm
Position of the BTS: | there is a small area without coverage of the BTSs.
Number of traffic channels of the BTS: | 7
Calls processing in MS: | Exponential function with inter-arrival time of 10 min
  - Exponential service time distribution (duration of calls) with average service time of 3 min (180 s)
Probability of intra-MSC calls: | 33% (probability that a call generated in the current MSC has as destination another MS located in the same MSC).

#### Scenario 2:
In this scenario, there is three BTSs situated approximately equal distances, which manages the total area of MSC. The transmitted power attenuation of 7 dBm in Table-2 summarizes the main characteristics of Scenario-2. Table-2 is an extract of the configuration file 'omnetpp.ini'. As in Scenario 1, it is a busy hour. The area simulation of Scenario-2 is larger than the first scenario. In this scenario we assign 2 kilometers long to both sides of the square area. Some parameters are similar in both scenarios, as number of MSs. This scenario also includes 50 MSs moving inside the zone of study, 30 of which have linear trajectories and 20 have random ones [3].

In this case, the simulation is on a network with a single MSC; however, the program can simulate calls between MSs within the simulation and other MSs which depends on a MSC out of the simulation area. Scenario-1 worked only with calls from a MS to a fictitious MS connecting to another MSC, but Scenario 2 considers calls between MSs present in the area and so is depending on the current MSC. This scenario assigns the probability of 33% of intra-MSC calls and also indicates average rates of service time & call generating distributions. The average service time call is now 3 minutes and the calls generating process of each MS is an exponential distribution of average rate of 1/600 calls per second, that means call attempts every 10 min. We consider three BTSs in the simulation. We have defined these three BTSs to be approximately of equal range. The most significant BTS characteristics are 7 traffic channels and a transmission power of 7 dB m. We observe that in this case there are MSs moving in zones without coverage. Any call attempt on those positions will fail. By applying this scenario we simulate the Omnet & get the following Graph.

**FIGURES OF SCENARIO 2**

**FIGURE-4:** Analysis of the results Received Power & Simulation Time at MS (5) in scenario 2.

Here are the examples of MS in Scenario-2 that show the received power of each MS from the three BTS. In this scenario number of MS is 50. This figure shows the graph of MS (5) from BTS (0).

**FIGURE-5:** Analysis of the results Received Power & Simulation Time at MS (5) in scenario 2.

In figure-5, the examples of MS in Scenario-2 that show the received power of each MS from the three BTS. This figure shows the graph of MS (5) from BTS (0), BTS (1) & BTS (2). In this scenario, number of MS is 50.
In figure-6, the example of MS in Scenario-2 shows the received power of each MS from the three BTS. Here in this scenario, number of MS is 50. This figure shows the graph of MS (10) from BTS (0).

Here are the examples of MS in Scenario-2 that show the received power of each MS from the three BTS. This figure shows the graph of MS (10) from BTS (0), BTS (1) & BTS (2). Here in this scenario, number of MS is 50, 30 linear & 20 random. Total power is 7 dB. The trajectories of every MS in the area during a busy hour simulation show the BTS coverage. Fig.04 & 05 shows have MS with 5 identifier with a linear trajectory. Here path type is 0. Fig.06 & 07 shows MS with 10 identifier with a random trajectory, here path type is 1.

A type of time code designed to match the real time of clocks. Two frames of time code are given. On the other side, Call drop is the common term for a wireless mobile phone call that is terminated unexpectedly as a result of technical reasons, including presence in a dead zone [4].

Offered Load is the total traffic load, including load that results from retries, submitted to a telecommunications system, group of servers, or the network over a circuit in the sector of telecommunication [5].

In frame relay, the data rate, as measured in bits per second (bps) offers the network for delivery. The aggregate offered load can be less than the access rate supported by the access link and the port speed of the frame relay network device but can never exceed that is less [6].
In communication networks, such as Ethernet or packet radio, throughput or network throughput is the average rate of successful message delivery over a communication channel. This data may be delivered over a physical or logical link, or pass through a certain network. Throughput is sometimes normalized and measured in percentage, but normalization may cause confusion regarding what the percentage is related to. Channel utilization and packet drop rate in percentage are less ambiguous terms. The channel utilization, also known as bandwidth utilization efficiency, in percentage is the achieved throughput related to the net bit rate in bit/s of a digital communication channel. For example, if the throughput is 70 Mbit/s in a 100 Mbit/s Ethernet connection, the channel utilization is 70%. In a point-to-point or point-to-multipoint communication link, where only one terminal is transmitting, the maximum throughput is often equivalent to or very near to the physical data rate (the channel capacity), since the channel utilization can be almost 100% in such a network, except for a small inter-frame gap [3]. The throughput is usually measured in bits per second (bit/s or bps), and sometimes in data packets per second or data packets per time slot. The system throughput or aggregate throughput is the sum of the data rates that are delivered to all terminals in a network.

Total Traffic Channel (TCH) is a logical channel that allows the transmission of speech or data. In most second generation systems, the traffic channel can be either full or half rate [7].

C. TABLE 03: General Characteristics of Scenario 3

<table>
<thead>
<tr>
<th>Properties</th>
<th>Characteristics Data</th>
</tr>
</thead>
<tbody>
<tr>
<td>Simulation area:</td>
<td>9 km² (3000 m x 3000 m)</td>
</tr>
<tr>
<td>Number of MS:</td>
<td>85 mobile stations</td>
</tr>
<tr>
<td></td>
<td>50 MS with linear trajectories</td>
</tr>
<tr>
<td></td>
<td>35 MS with random trajectories: change direction after lifetime (random between 0 to 200 s)</td>
</tr>
<tr>
<td>Speed of MS:</td>
<td>(minimal 0.0 m/s) (maximum 7.1 m/s) (average 1.7 m/s)</td>
</tr>
<tr>
<td>Power measurements:</td>
<td>one measure per second</td>
</tr>
<tr>
<td>MSISDN:</td>
<td>60090000xx, where xx is the MS number (from 0 to 84)</td>
</tr>
<tr>
<td>Number of MSC:</td>
<td>1</td>
</tr>
<tr>
<td>Number of BTS:</td>
<td>7</td>
</tr>
<tr>
<td>Transmission power of BTS:</td>
<td>7 dBm</td>
</tr>
<tr>
<td>Position of the BTS:</td>
<td>there is a small area without coverage of the BTSs.</td>
</tr>
<tr>
<td>Number of traffic channels of the BTS:</td>
<td>7</td>
</tr>
<tr>
<td>Calls processing in MS:</td>
<td>Exponential function with inter-arrival time of 10 min</td>
</tr>
<tr>
<td></td>
<td>Exponential service time distribution (duration of calls) with average service time of 3 min (180 s)</td>
</tr>
<tr>
<td>Probability of intra-MSC calls:</td>
<td>5% (probability that a call generated in the current MSC has as destination another MS located in the same MSC)</td>
</tr>
</tbody>
</table>

- Scenario 3:
  In this scenario, there are seven BTSs situated at several distances that manage the total area of MSC. The transmitted power attenuation of 7 dB m. in Table-3 summarizes the main characteristics of Scenario-3. Table-3 is an extract of the configuration file 'omnetpp.ini'. As in Scenario 1, it is a busy hour. The area simulation of Scenario 3 is larger than the first and second scenarios. In this scenario we assign 3 kilometers length to both sides of the square area. Some parameters are similar in above scenarios.
  This scenario includes 85 MSs moving inside the zone of study, 50 of which have linear trajectories and 35 have random ones. In this case, the simulation is on a network with a single MSC; however, the program can simulate calls between MSs within the simulation and other MSs that depends on a MSC out of the simulation area. Scenario 1 worked only
with calls from a MS to a fictitious MS connecting to another MSC, but Scenario 2 considers calls between MSs present in the area, and so is depending of the current MSC. We assign the probability of 5% of intra-MSC calls in this scenario. In this table we indicate average rates of service time and call generating distributions. The average service time call is also 3 minutes in this scenario and the calls generating process of each MS is an exponential distribution of average rate of 1/600 calls per second, that means call attempts every 10 min. Here we consider seven BTSs in the simulation. We have defined these seven BTSs to be equivalent. The most significant BTS characteristics are 7 traffic channels and a transmission power of 7 dB m. We observe that in this case there are MSs moving in zones without coverage. Any call attempt on those positions will fail. By applying this scenario we simulate the Omnet & get the following Graph.

**FIGURES OF SCENARIO 3**

**FIGURE-12:** Analysis of the results Received Power & Simulation Time at MS (5) in scenario 3.

Here the examples of MS in Scenario-3 that shows the received power of each MS from the seven BTS. This figure shows the graph of MS (5) from BTS (0).

**FIGURE-13:** Analysis of the results Received Power & Simulation Time at MS (5) in scenario-3.

In figure-13, the examples of MS in Scenario-3 that show the received power of each MS from the seven BTS. This figure shows the graph of MS (5) from BTS (0) to BTS (6). Here in this scenario, number of MS is 85.

**FIGURE-14:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 3.

Here the examples of MS in Scenario-3 that shows the received power of each MS from the seven BTS. In this scenario, number of MS is 85. This figure shows the graph of MS (10) from BTS (0).

**FIGURE-15:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 3.
Here are the examples of MS in Scenario-3 that show the received power of each MS from the seven BTS. This figure shows the graph of MS (10) from BTS (0) to BTS (6). In this scenario, number of MS is 85, 50 linear & 35 random. Total power is 9 dB. The trajectories of every MS in this area during a busy hour simulation show the BTS coverage. Fig.12 & 13 shows MS of 5 identifier with a linear trajectory. Here path type is 0. Fig.14 & 15 shows MS with 10 identifier consisting of a random trajectory, here path type is 1.

D. TABLE 04: General characteristics of Scenario 4

<table>
<thead>
<tr>
<th>Properties</th>
<th>Characteristics Data</th>
</tr>
</thead>
<tbody>
<tr>
<td>Simulation area:</td>
<td>4 km² (2000 m x 2000 m)</td>
</tr>
<tr>
<td>Number of MS:</td>
<td>100 mobile stations</td>
</tr>
<tr>
<td></td>
<td>60 MS with linear trajectories</td>
</tr>
<tr>
<td></td>
<td>40 MS with random trajectories: change direction after lifetime (random between 0 to 200 s)</td>
</tr>
<tr>
<td>Speed of MS:</td>
<td>(minimal 0.0 m/s) (maximum 7.1 m/s) (average 1.7 m/s)</td>
</tr>
<tr>
<td>Power measuremens:</td>
<td>one measure per second</td>
</tr>
<tr>
<td>MSISDN:</td>
<td>6009000xx, where xx is the MS number (from 0 to 99)</td>
</tr>
<tr>
<td>Number of MSC:</td>
<td>1</td>
</tr>
<tr>
<td>Number of BTS:</td>
<td>20</td>
</tr>
<tr>
<td>Transmission power of BTS:</td>
<td>7 dBm</td>
</tr>
<tr>
<td>Position of the BTS:</td>
<td>there is a small area without coverage of the BTSs.</td>
</tr>
<tr>
<td>Number of traffic channels of the BTS:</td>
<td>7</td>
</tr>
<tr>
<td>Calls processing in MS:</td>
<td>Exponential function with inter-arrival time of 10 min</td>
</tr>
<tr>
<td></td>
<td>Exponential service time distribution (duration of calls) with average service time of 3 min (180 s)</td>
</tr>
<tr>
<td>Probability of intra-MSC calls:</td>
<td>33% (probability that a call generated in the current MSC has as destination another MS located in the same MSC)</td>
</tr>
</tbody>
</table>

- Scenario 4:
  In this scenario there is twenty BTSs situated at several distances, which manages the total area of MSC. The transmitted power attenuation of 7 dB m. in Table 4 summarizes the main characteristics of Scenario-4. Table-3 is an extract of the configuration file 'omnetpp.ini'. As in Scenario-1, it is a busy hour. The area simulation of Scenario-4 is larger than the first scenario and small from second scenario, here we assign 2 kilometers length to both sides of the square area. Some parameters of this scenario are also similar in above scenarios. This scenario includes 100 MSs moving inside the zone of study, 60 of which have linear trajectories and 40 have random ones. In this case, the simulation is on a network with a single MSC; however, the program can simulate calls between MSs within the simulation and other MSs which depends on a MSC out of the simulation area. Scenario 1 worked only with calls from a MS to a fictitious MS connecting to another MSC, but Scenario 4 considers calls between MSs present in the area, and so is depending on the current MSC. Here we assign the probability of 33% of intra-MSC calls. Table 4 indicates average rates of service time and call generating distributions. The average service time call is also 3 minutes and the calls generating process of each MS is an exponential distribution of average rate of 1/600 calls per second that means call attempts every 10 min. Here we consider twenty BTSs in the simulation. We have defined these twenty BTSs to be approximately equal. The most significant BTS characteristics are 7 traffic channels and a transmission power of 7 dB m. We observe that in this case there are MSs moving in zones without coverage. Any call attempt on those positions will fail. By applying this scenario we simulate the Omnet & get the following Graph.

FIGURES OF SCENARIO 4

FIGURE-16: Analysis of the results Received Power & Simulation Time at MS (5) in scenario 4.
Here are the examples of MS in Scenario-4 that show the received power of each MS from the twenty BTS. Here, the scenario number of MS is 100. This figure shows the graph of MS (5) from BTS (0).

![Figure 17](image17.png)

**FIGURE-17:** Analysis of the results Received Power & Simulation Time at MS (5) in scenario 4.

In figure-18, the examples of MS in Scenario-4 that show the received power of each MS from the twenty BTS. This figure shows the graph of MS (5) from BTS (0) to BTS (19). Here, the scenario number of MS is 100.

![Figure 18](image18.png)

**FIGURE-18:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 4.

Here the examples of MS in Scenario-4 that shows the received power of each MS from the twenty BTS. This figure shows the graph of MS (10) from BTS (0) to BTS (19). In this scenario, number of MS is 100, 60 linear & 40 random. Total power is 12 dB. The trajectories of every MS in this area during a busy hour simulation show the BTS coverage. Fig.16 & 17 shows MS with 5 identifier consisting of a linear trajectory, here path type is 0. Fig.18 & 19 shows MS with 10 identifier consisting of a random trajectory, here path type is 1.

![Figure 19](image19.png)

**FIGURE-19:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 4.

Here in figure-18, the examples of MS in Scenario-4 that show the received power of each MS from the twenty BTS. Here in this scenario, number of MS is 100. This figure shows the graph of MS (10) from BTS (0).

![Figure 20](image20.png)

**FIGURE-20:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 4.

Here in figure-18, the examples of MS in Scenario-4 that show the received power of each MS from the twenty BTS. Here in this scenario, number of MS is 100. This figure shows the graph of MS (10) from BTS (0).

![Figure 21](image21.png)

**FIGURE-21:** Analysis of the results Received Power & Simulation Time at MS (10) in scenario 4.

Here in figure-18, the examples of MS in Scenario-4 that show the received power of each MS from the twenty BTS. Here in this scenario, number of MS is 100. This figure shows the graph of MS (10) from BTS (0).

**IV. RESULT ANALYSIS OF THE GRAPHS**

In the above graphs, linear trajectories are characterized by 'path Type = 0'; random paths are categorized by 'path Type = 1'. The program assigns a unique MSISDN to every MS, which follows the format 6009000xx, where xx is the identifier of each MS in the simulation. The configuration files collect information about the initial position and speed of each MS. All MSs have different speed [8]. Here, MSs following linear trajectories generate regular graphics of power attenuation that present symmetries, whereas MSs with random movements have no regular representations of power attenuation. The number of traffic channels is assigned during a busy hour simulation. Due to the BTSs having several traffic channels to serve the communication demand, the number of busy traffic channels is always less than or equal to the number of channels we use here. When the BTS assigned 7 traffic channels, congestion in calls is produced. During congestion, any new call attempts will be rejected. The calls generation process is a Poisson's process with average generating call rate of 6 calls per hour, which is an average inter-arrival time of 10 min or 600 s. At the beginning of a call, the MS assigned a traffic channel in the BTS which reaches the MS with the largest power. After a time, it is possible that another BTS covers the MS with a larger power [9].
The network releases a handover when the difference in power exceeds a given threshold. GSMSIM has configured this threshold to be 9 dB. Another point of these simulations are the zones without coverage. When the received power from a BTS is less or equal than −102 dB m, GSMSIM considers that the MS is out of coverage. In scenario-1, there is no area in out of coverage and the other three scenarios have few areas in out of coverage. Finally we assume that scenario-4 is more efficient than the other scenarios because in this scenario there is more channels with respect to the MSs comparing the others scenarios. By analyzing the figures we assume that, when BTS & Channel increase with respect to area, the call drop will decrease proportionately. In this case, hand over will increase [10].

By analyzing the graph of scenario-3, we calculate the blocked calls, successful calls, failed handover & successful handover.

These figures represent the percentage of successful and failed handovers and percentage of successful and blocked calls for each BTS during the simulation.

Considering the BSC of the scenario-2, the blocked calls during the busy hour simulation reaches the 16% (40 blocked calls and 210 successful calls). On the other word, the BSC of the scenario-3, the blocked calls during the busy hour simulation reaches the 6.37% (32 blocked calls and 470 successful calls). Possible solutions to decrease these high percentages may be:

- Check the traffic channel assignation algorithm for call attempts. If the BTS with more influence (power transmission) over a MS has any idle traffic channel, then examine the second BTS with more influence at scenario-2 [15].

Finally, the right part of the Figure-5(A) shows the number of handovers for every BTS and BSC. The BSC begins 31 handover during the simulation, 10 of which fail. It would be necessary to evaluate reasons for the very high percentage of failed handover operations (32%) [11]. The lower part of Figure-5(B) also shows the number of handovers for each BTS and BSC. The BSC begins 62 handover during the simulation. 12 are failed in those handover. It would be necessary to evaluate reasons for the very high percentage of failed handover operations (19%) [11].

V. CONCLUSION

The criteria used in each scenario, are defined to provide a service of highest quality to users. This leads to multiple tests and changes in input parameters of each simulation. These parameters make complex designs because the network must have Multiple factors to provide quality service. The size to optimize coverage, the power to radiate the BTS for greater capacity and to ensure quality throughout the service must be also taken into account [12][13].

Traffic demand due to the number of users in the cell determines the number of timeslots to be configured to ensure the minimum number of missed calls (missed calls). Throughout, the different simulations showed that the number of calls lost depends on the value of timeslots defined by each TRX at the station base. It is also noted that the power is a factor involving the number of dropped calls, i.e. less power than the number of dropped calls (Broken calls) [14]. Observing the percentage of calls of different scenarios compared to the minimum quality standards that must provide the GSM system, one can conclude that these depend on the network design itself, such as the number of timeslots, the power, the size of the cell, the number of users [15]. The greater the number of network users, the higher the number of timeslots. The greater the number of timeslots, the lower the number of missed calls. The greater their power, greater the coverage and lower the number of dropped calls. After a large number of simulations, one can observe the Cellular operators by the large amount of demand and use limited radio spectrum, opting to provide more coverage than quality [16].

VI. FUTURE SCOPE
The project can be extended to study the behavior of the Handover and the development of module, the interface that enables communication between the BTS and BSC. This project can be exploring by the great discrete event simulation tool OMNET further for the study and design of different types of networks.

VII. RECOMMENDATIONS

OMNET does not require licensing since it is Open Source. This project can be used as a guide for using the simulation tool OMNET to serve as a resource in the study of different types of networks in the area Telecommunication Systems Engineering.

VIII. REFERENCES


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Route optimization and roaming capability based MIPv6 protocol in internet network

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Abstract— MIPv6 is a proper replacement for MIPv4 protocol which recommended by IETF. IPv6 lieu IPv4 has been chosen as convergence layer for next heterogeneous access networks. MIPv4 has limiting in protocol, but MIPv6 is created fundamental changes such as security enhancements, elimination of the Foreign Agent (FA) and route optimization. The MIPv6 characteristics defined by the IETF provides perspicuous host mobility within IPv6 networks. In MIPv6 MN is move between IP subnets without change in its original IPv6 address configuration. This means that MN ever is addressable in the internet via its Home Address (HoA). HoA is IPv6 address that is allocated to the MN in its home network. When away from the home network, MN can still detect by its HoA in the internet. Because packets routed to its HoA. Also In this way, mobility transparency of higher layer protocols like Transport layer or higher is achieved.

Keywords- MIPv6; routing; roaming capability

I. INTRODUCTION

Mobile IPv6 (MIPv6) is a commonly accepted standard to address global mobility of Mobile Nodes (MNs) [1]. This is one of the main protocols to manage mobile node (MN) movements; refer to IETF documentation. This allows the MN to acquire and register a new IPv6 address in each visited network. Terminology used in Mobile IPv6 as follows [2]:

A node that can change its situation from one network to another, while still being reachable pre its home address, this called Mobile Node (MN). Corresponding Node (CN) is a mobile node or a fixed node that communicates or corresponds with the MN by exchanging packets with MN. The individual network that manages the MN is Home Network (HN). Foreign Network (FN) is other network that the MN is attached lieu of its HN. Home address (HoA) is an irreversible IP address assigned to MN within its home network. Home Agent (HA) is a router on a MN's home network with which the MN has registered its current CoA [3], [4]. While the MN is away from home, the HA arrests packets on the home network destine to the MN's address, encapsulates them, and tunnels them to the MN's registers CoA. Access Router gives connectivity to the mobile node at its other point of attachment to the Internet. Binding is the association of the home address of a Mobile Node (MN) with a care-of-address for that MN, along with the remaining lifetime of that association. Binding Update which including the Home Address (HOA) and the CoA [5], [6]. Care-of-Address (CoA) is An IP address associated with a MN while visiting a foreign network; the subnet prefix of this IP address is a foreign subnet prefix. Among the multiple care-of-addresses that a MN may have at a time, the one registered with the MN's Home Agent is called its primary CoA.

The paper is organized as follows: Section 2 and 3 describe the scenario 1 and scenario 2 of the proposed scheme, respectively. And, both of sections discuss simulation results. Finally, Section 4 concludes this paper.

II. SCENARIO 1

(EFFECTS MIPv6-ROUTE-OPTIMIZATION-ENABLED)

The objective is to demonstrate the effects of Mobile IPv6 (MIPv6) mechanisms while two mobile nodes communicate with each other. Cases for route optimization enable and disable are evaluated.

The IPv6 network is composed by four WLAN access points connected through an IP cloud. The core of the network, represented by the IP cloud, has a constant latency of 0.1 seconds. This makes easier to note the effects of the different MIPv6 mechanisms over the application delay.

Table I and II summarize the simulation parameters and the network parameters, respectively.

A. Simulation scenario 1

MN_A and MN_B communicate to each other by running a very light video application as a source of constant UDP traffic. Initially the mobiles are placed at their corresponding home networks. Then MN_A is served by home agent HA_A and MN_B is served by home agent HA_B. Both mobiles use MIPv6 to roam among the various access points in the network. The movement performed by the nodes can be described as follows:

- MN_A: (1A)- MN_A moves in a counterclockwise trajectory roaming through all four access points in the network.
- MN_B: (1B)- MN_B moves, first in a clockwise trajectory roaming through all four access points in
communicate with correspondent nodes. There are two possible mechanisms used by MIPv6: Route optimization and Tunnel/reverse tunnel (route optimization disabled).

Notice that when route optimization is enabled the application delay is reduced compared to the case when route optimization is disabled. Below you will find a more detailed explanation for this effect.

Fig. 2 represents video conferencing traffic and video conferencing packet delay, respectively.

2) Mobile IPv6 measurements

Two mobiles communicating with each other. This means that at some point in the simulation the mobile nodes will be acting as both a mobile node and/or a correspondent node. This cause interesting MIPv6 effects that can be observed at the "Mobile IPv6 Traffic" statistic panel:

- When both mobiles are away from their corresponding home networks a double MIPv6 overhead will occur, either:

  Two MIPv6 tunnels will be needed for the mobiles to communicate (mipv6_route_optimization_enabled). In this case the application response time delay will be mainly produced by the three times the data packet must pass through the IP cloud (Internet). Given the latency configured for the IP cloud (0.1 sec), the total application delay will be approximately 0.3 seconds.

  Two IPv6 extension headers (routing extension header and destination extension header) will be used (at the same time) to transport the data traffic when using route optimization mechanism (mipv6_route_optimization_disabled). In this case the application response time delay will be mainly produced by only one time the data packet must pass through the IP cloud (approximately 0.1 sec). This is when both mobiles are away from home but located in different networks. Now, when both mobiles are located at the same access point, the data packets will just go through the access point, reducing the application response time even more.

- When only one mobile is away from its home network, it will act as a mobile while the other one will perform correspondent node operations. In this case:

  One MIPv6 tunnel will be needed to communicate (mipv6_route_optimization_enabled).

  Just one MIPv6 extension header (at a given time) will be used to transport the data traffic when using route optimization mechanism (mipv6_route_optimization_disabled).

  Fig. 3 represents the result of MIPv6 traffic.

  Fig. 4 shows the packet delay variation in cases enable and disable routing optimized. Variance among end to end delays for video packets received by this node. End to end delay for a video packet is measured from the time it is created to the time it is received. The packet delay variation is 2.5 in optimized enable routing that is less than optimized disable routing.

3) Visited access points

Under this statistic panel it is possible to observe all access points that were visited by both mobiles. Each bar in the graph represents an access point visited by the mobiles, and the bar width represents the time the mobile used the access point.
until it move to a different one. The colors of the bars have been set so each one identifies one of the four access points according to the color of the annotation circle placed at the access point's position. Fig. 5 shows the visited access points.

III. SCENARIO 2 (ROAMING CAPABILITY IN MIPv6)

This scenario utilizes 802.11b WLAN interface with roaming capability to simulate hand-offs between mobile IP agents who are also WLAN access points. Here are some configuration specifics you have to follow to use WLAN roaming capability.

- Wireless Lan Parameters. BSS Identifier should be explicitly set on all the WLAN nodes in the network.
- There should be only one access point for a BSS network (Wireless LAN Parameters. Access Point Functionality)
- 802.11b uses the following physical attribute values: Wireless Lan Parameters. Data Rate (bps) -- 11Mbps Wireless Lan Parameters. Physical Characteristics -- Direct Sequence
- IP auto addressing scheme will assign IP addresses based on the BSS ID, i.e. all the WLAN nodes sharing the same BSS ID will be assigned an IP addresses from the same IP network.

The Mobile IP NET is a mobile subnet containing a mobile router and a client node. The mobile router node uses the mobile IP home agent service from the HA WLAN router. The MR in mobile subnet is manually configured with common BSS ID and IP network address as that of the HA WLAN router. All the foreign agents are also WLAN routers with different BSS Identifiers.

Table III and IV summarize the simulation parameters and the network parameters, respectively.

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<table>
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<th>Parameters</th>
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<td>Flow type of ip-traffic-flow</td>
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A. Simulation Scenario 2

The Mipv6 network, which simulated in this scenario, inclusive a mobile subnet that has a mobile router and client node. This scenario represented in fig. 6. In figure 6 the mobile subnet first stands in home domain. When simulation starts the packet which exchanges between RPG-Server and RPG-Client is serving via HA. The mobile subnet gently moves, when approaching the first neighbor domain, MR sends BU massage to HA and HA responses with BA massage. If update is allow, starts transmittal with new domain. Hereinafter the packet transmitted by server outset goes to HA and then transmission to FA and achieve to client in mobile subnet.

Figure 6. Simulated scenario 2: (a): Network, (b): Subnet

B. Simulation Result For Scenario 2

The RPG-Server node sends traffic to the RPG-Client. As the mobile subnet moves along the trajectory, it changes the access point and changes the mobile agent as well. The packets will be tunneled to different foreign agents as the mobile node changes its access points. Fig. 7 shows the tunneled traffic sent for HA and the tunneled traffic received for FAs.

As mobile subnet is move at trajectory RPG-Server send traffic to RPG-Client. Traffic sent is Total number of all RPG packet bits sent per second by this node to other RPG nodes in the network and traffic received is Total number of RPG packet bits received per second by this node from all other RPG packet sources in the network. Fig. 8 shows the Traffic sent by RPG-Server and traffic received by RPG-Client. Fig. 9 shows throughput at various domains, which is namely total data traffic in bits/sec successfully received and forwarded to the higher layer by the WLAN MAC. This statistic does not include the data frames that are 1) unicasts addressed to another MAC, 2) duplicates of previously received frames, and 3) incomplete, meaning that not all the fragments of the frame were received within a
certain time, so that the received fragments had to be discarded without fully reassembling the higher layer packet.

Fig. 10 represents the media access delay that is namely the total of queuing and contention delays of the data, management, delayed Block-ACK and Block-ACK Request frames transmitted by the WLAN MAC. For each frame, this delay is calculated as the duration from the time when it is inserted into the transmission queue, which is arrival time for higher layer data packets and creation time for all other frames types, until the time when the frame is sent to the physical layer for the first time. Hence, it also includes the period for the successful RTS/CTS exchange, if this exchange is used prior to the transmission of that frame. Similarly, it may also include multiple number of back off periods, if the MAC is 802.11e-capable and the initial transmission of the frame is delayed due to one or more internal collisions.

IV. CONCLUSION

In scenario 1 has been perusing the effect of mipv6 route optimization on delay, sent and received traffic, tunneled traffic in MIPv6, delay and traffic for video conferencing. We perceive that with proper routing, situation of all of the traffics improve to a considerable extent. Even in optimal routing the incision in communication, was much less. In the other hand, delay and Video packet delay dispersion slake to a considerable extent.

The problem considered in Scenario 2, is the mobile roaming in the MIPv6 network. It was observed that, whenever the mobile enters to a new AP, traffic just has been tunneling for that AP. E.g. sending traffic of HA is totally of receiving traffic by FAs. Also media access delay in HA is more than FAs.

References

Figure 5. (a) Active access point for MN-A

Figure 5. (b) Active access point for MN-B

Figure 7. (a): stacked, The tunnelled traffic sent for HA and the tunnelled traffic received for FAs

Figure 7. (b): overlaid, The tunnelled traffic sent for HA and the tunnelled traffic received for FAs

Figure 8. (a): stacked, Traffic sent by RPG-Server and traffic received by RPG-Client

Figure 8. (b): overlaid, Traffic sent by RPG-Server and traffic received by RPG-Client

Figure 9. (a): stacked, WLAN throughput at various domain

Figure 9. (b): overlaid, WLAN throughput at various domain

Figure 10. (a): stacked, The media access delay

Figure 10. (b): overlaid, The media access delay
Utilization DCTC and Voronoi of Tracking in Wireless Sensor Networks
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Abstract—Wireless sensor networks (WSN) may consist of several to thousands of sensors that share the need to organize for network data collection sink routing. This paper addressed the problems of tracking moving of wireless sensor network objects. The traditional tracking method, called Dynamic Convoy Tree-Based Collaboration (DCTC) presented. In additional describe a method, called Distributed computation of Voronoi cells in sensor networks. Proposed solutions of WSNs challenges using converge cast traffic, covering networks configuration and efficient routing routines. We also present the intermediate routing sensor nodes expend an excessive amount of their energy resources. thus can achieve superior tracking accuracy with faster tracking convergence speed and reducing the network lifetime.

Index Terms - DCTC, Voronoi, Tracking

I. INTRODUCTION

Wireless sensor networks in any applications are used to gather intelligence about field conditions. Monitoring the activity or assess conditions and influences. A major requirement for any applications sensor networks is to reliably aggregate and disseminate information within a time frame that allows the command control to take necessary tactical decisions. This calls for communication systems that can provide high data rates with high reliability while using minimum bandwidth and power. In other words, the underlying communication network must be robust, reliable, and scalable. The choice of network architecture (topology) has strong influence on the effectiveness of the tactical applications in wireless any application sensor networks. Network architecture affects network characteristics such as latency, robustness, and capacity [1,2,3].

Almost all of the common applications require knowledge of the position of the car to work properly. Even when no applications directly take benefit of the car park, can the underlying data dissemination protocols take Feature a lot of this information [4]. To example, geocasting and guidance geographical dependence on the car and are desirable for many of the scenarios and be more suitable for applications Voronoi region is less than 6 proof. By the Euler formula [5,6]

The complexity of data routing and processing (data fusion) also depends on the topology. In this paper will be fined how to route packets efficiently in networks is an interesting and challenging research topic.

The rest of the paper is organized as follows. In the next section, we briefly outline the principle of voronoi methods. In Section III, we describe the proposed algorithm. Simulation results are presented and discussed in Section IV. Finally, we conclude in Section V.

II. Principals of VORONOI Methods

Static point Voronoi tessellations are well known in the literature, and algorithms have been used for many years (see [7] for a summary). Less well known are dynamic algorithms, that allow point creation, deletion and movement, and also Voronoi tessellations of more complex objects - typically line segments as well as points. Algorithms for generating the simple point Voronoi tessellation have improved significantly in theoretical efficiency in recent years. Where the whole structure may be constructed at once, randomized incremental algorithms such as [8,9] can create these diagrams in expected time O(n log n), which is optimal. However, as a major motivation for this work concerned the maintaining of a map when one or more objects are moving, an alternative technique was developed that maintained the Voronoi spatial relationships while map objects were being inserted, deleted, or displaced. This is achieved by determining when the Voronoi cell of a moving point gain or loses a neighboring cell, moving the point to that location, and locally updating the topological structure accordingly. For the case of all points moving simultaneously, [10] give a rather complex theoretical efficiency based on Davenport-Schinzel sequences, but in the case of one point being inserted at a time by splitting it from the nearest pre-existing point and then moving it to its destination (see below) the expected time efficiency should again approximate O(n log n)The WSN of active sensors suffers from serious inter-sensor in- terference and imposes new design and implementation challenges. Show that the adaptive sensor scheduling scheme can achieve superior tracking accuracy with faster tracking convergence speed.

The concept of the Voronoi diagram [7], a well-known construct from computational geometry, is used to find a maximal breach path in a sensing field. In two dimensions, the Voronoi diagram of a set of discrete points (also called sites) divides the plane into a set of convex polygons, such that all points inside a polygon are closest to
only one point. In Figure 1a, 10 randomly placed nodes divide the bounded rectangular region into 10 convex polygons, referred to as Voronoi polygons. Any two nodes si and sj are called Voronoi neighbors of each other if their polygons share a common edge. The edges of a Voronoi polygon for node si are the perpendicular bisectors of the lines connecting si and its Voronoi neighbors. Since by construction, the line segments in a Voronoi diagram maximize the distance from the closest sites, the maximal breach path must lie along the Voronoi edges. If it does not, then any other path that deviates from the Voronoi edges would

![Voronoi diagram](image)

Figure 1. (a) Voronoi diagram of 10 randomly placed nodes; (b) Voronoi polygon for node S, constructed by drawing perpendicular bisectors of the lines connecting S and its neighbors.

Issues in wireless sensing networks are network architecture design and data routing. Hence, several researchers have addressed the issue of communication jamming in a wireless sensor network and its effect on the performance of the network. Xu et al. [11,12] discuss radio interference attacks on wireless sensor networks. They study the feasibility and effectiveness of jamming attacks on wireless networks and examine the critical issue of detecting the presence of jamming attacks. They also propose four different jamming attack models that can be used by an adversary to disable the operation of a wireless network, and evaluate their effectiveness in terms of how each method affects the ability of a wireless node to send and receive packets.

### III. Target Tracking Using Sensor Networks

Target tracking has been a classical problem since the early years of electrical systems. Sittler, in 1964, gave a formal description of the multiple-target tracking (MTT) problem [11]. The goal of the MTT problem is to find the moving path for each target in the field. Target tracking using a sensor network was initially investigated 2004 [13]. With the advances in the fabrication technologies that integrate the sensing and the wireless communication technologies, tiny sensor motes can be densely deployed in the desired field to form a large-scale wireless sensor network. Challenges and difficulties, however, also exist in a target tracking sensor network:

1) Tracking needs collaborative communication and computation among multiple sensors.
2) Each sensor node has very limited processing power.
3) Each node also has tight budget on energy source. Thus, for data processing and tracking should consider the impact of power saving mode in each node.

#### A) Different Approaches of Target Tracking

The method will need to handle a large number of moving objects at once. While our method uses a hierarchy to connect the sensors:

1) The leaves are sensors
2) The querying point as the root
3) The other nodes are communication nodes

The main idea of STUN is showed in the example figure 2 showed that the message-pruning hierarchy. Consider the detection messages from sensors that detected the arrival of the car. Sensors A's message will update the detected sets of all its ancestors. The message from sensors B and D do not update the detected sets of their parents and thus will be pruned at X. The main advantage of STUN Message pruning and routing routing while the disadvantage Building the tree (the structures of the tree).

![Message pruning hierarchy](image)

Figure 2 is a message-pruning hierarchy.

#### B) Tracking Tree Management Dynamic Convoy Tree-Based Collaboration (DCTC)

A dynamic convoy tree-based collaboration (DCTC) framework for tracking a mobile target is proposed in [14]. Heuristics are used to predict the object’s moving direction. A dynamic tree is then created by adding or pruning the sensors near the moving target. The root of the tree can dynamically refine the readings gathered from various tree nodes.

Since the coverage area of individual sensor nodes usually overlaps, the work in [15] attempts to periodically search the smallest subset of nodes that covers the monitoring area. This group of nodes is referred to as the area-dominating set. A distributed spanning tree, induced by the initial interest flood over the area-dominating set, is used to aggregate the reply messages from various event sources. DCTC relies on a tree structure called “convoy tree”. The tree is dynamically configured to add some nodes and prune some nodes as the target moves that DCTC–main idea. This paper studies the Efficient of detecting and tracking a mobile target, and monitoring a particular region surrounding the target in sensor networks. Figure 3 showed that the sensor nodes surrounding an adversary tank detect and track the tank and its surrounding area which may include enemy surrounding area. DCTC relies on a tree structure called convoy tree, which includes sensor nodes around the moving target, and the tree...
is dynamically configured to add some nodes and prune some nodes as the target moves. Figure 1 illustrates how to use the convoy tree to track a mobile target. As the target first enters the detection region, sensor nodes that can detect the target collaborate with each other to select a root and construct an initial convoy tree. Relying on the convoy tree, the roots collect information from the sensor nodes and refine this information to obtain more complete and accurate information about the target using some classification algorithms [16, 17].

The region around it in an energy efficient way, and the network should forward this information to the sinks in a fast and energy efficient way. The data report can be saved locally waiting for other node’s query, or can be forwarded to single or multiple data centers (the sinks), which can be a static command center or moving soldiers. As the sensor nodes surrounding the moving target should promptly provide robust and reliable status information about the mobile target and design goals. Such as is tracking an important target (e.g., an important person) in a parade. As design goals, moving target should promptly provide robust. The data report can be saved locally waiting for other node’s query, or can be forwarded to single or multiple data centers (the sinks), which can be efficient way, and the network should forward well as its surrounding area, and one of them (i.e., the root) generates a data report.

This is information to the sinks in a fast and energy efficient way. DCTC is a framework to detect and track the mobile target and monitor its soldiers. These nodes collaborate among themselves to aggregate data about the tank.

IV. Simulation and Results Discussion

Scenario: 200 → 1000 sensor nodes are thrown randomly in area of 640m x 540m. Each node has 2J (2*10^6 µJ) of energy with sensing radius = 30m and communication radius = 60m. Intruder objects are supposed moving specific paths. No data aggregation is allowed. The Utilized tools and module descriptions as a tools of OMNET++, C#, and Matlab.

The module description under OMNET++: Layer 0 module: Represented for physical layer. It consists of gates (in/out) and be responsible for making connection between the node and its neighbors. Its behaviors include forward messages from higher layer to its neighbors and vice versa. MAC module: Represented for pre-processing packet layers. It consists of gates (in/out) and queues (incoming queue and outgoing queue). When the queue is full, it deletes some latest messages to make sure that there is enough room in the queue for new messages. It helps to evaluate performance of the node. (Note: In current simulation, this module is temporary eliminated to speech up the performance) Application module: Represented for application layer consisting of gates (in/out). Note that, at anytime, after sending a message, the module automatically sends a decrease_energy message to energy module (through the coordinator) to let the module decrease the energy by one unit. Coordinator module: an interface to connect all modules together. It categories incoming messages to delivery them to the right module. For example, when receiving a decrease_energy message, it will forward the message to energy module. Sensor module: represented for sensor board in a sensor node. if sensor_switch parameter is “on” (set to 1), the module consumes energy, so, after an interval (timer), the module send decrease_energy message to the energy module (through the coordinator). When the timer ticks, the waiting timer decreases. The waiting timer is set by sensor_refresh messages from application module, if the waiting timer is zero, the module will turn “off” (sensor_switch parameter is set to 0). Radio module: represented for the radio board in a sensor node. if radio_switch parameter is “on” (set to 1), the module consumes energy, so, after an interval (timer), the module send decrease_energy message to the energy module (through the coordinator). Energy module: represented for battery in a sensor node. at the beginning, each sensor node is set to a specific energy level (energy parameter). if the module receives a decrease_energy message, it decreases the energy level by one.
Figure 4: DCTC and Voronoi Routing

We found that the DCTC had a very small increase in routing overhead than using voronoi as shown in Figure 1, because of delayed receipt of route reply by the source nodes.

1. Conclusions

However, each sensor mote has limited capabilities in terms of power, sensing, and processing abilities. Therefore, comprehensive and accurate data can be obtained only through the collaboration of sensor nodes in the network as a single node does not have the capability to provide this information. We have discussed the importance of coverage and connectivity, Which are two fundamental factors for ensuring efficient resource management in wireless sensor networks, and surveyed various methods and protocols, which are optimally cover a sensing field while maintaining global network connectivity at the same time.

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Effective Measurement Requirements for Network Security Management

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Abstract—Technical security metrics provide measurements in ensuring the effectiveness of technical security controls or technology devices/objects that are used in protecting the information systems. However, lack of understanding and method to develop the technical security metrics may lead to unachievable security control objectives and incompetence of the implementation. This paper proposes a model of technical security metric to measure the effectiveness of network security management. The measurement is based on the effectiveness of security performance for (1) network security controls such as firewall, Intrusion Detection Prevention System (IDPS), switch, wireless access point, wireless controllers and network architecture; and (2) network services such as Hypertext Transfer Protocol Secure (HTTPS) and virtual private network (VPN). We use the Goal-Question-Metric (GQM) paradigm [1] which links the measurement goals to measurement questions and produce the metrics that can easily be interpreted in compliance with the requirements. The outcome of this research method is the introduction of network security management metric as an attribute to the Technical Security Metric (TSM) model. Apparently, the proposed TSM model may provide guidance for organizations in complying with effective measurement requirements of ISO/IEC 27001 Information Security Management System (ISMS) standard. The proposed model will provide a comprehensive measurement and guidance to support the use of ISO/IEC 27004 ISMS Measurement template.

Keywords—Security metrics; Technical security metrics model; Measurement; Goal-Question-Metric (GQM); Effective measurement; Network security management

I. INTRODUCTION (HEADING 1)

Network security is defined as the security of devices, security of management activities related to the devices, applications/services, and end-users, in addition to security of the information being transferred across the communication links [2]. How much protection is required in ensuring the use of information and associated networks to conduct the business are well managed? How to identify and analyze network security controls to mitigate the network security risks? These questions have derived to implement and maintain secure and functional network is absolutely critical to the success of any organization’s business operations [2][3]. Thus, it is important to measure network security effectiveness in handling the risks from the current threats, vulnerabilities and attacks.

According to [4], the practical challenges and issues are what to measure and what information to report in facilitates the senior management for any decision making. Obviously, the reported information is often based on what is easier to measure instead of what is actually meaningful strategically [5], [6], [7]. Does network security management is among the “easier” information to measure?

Some organizations may be reported the measures from out of context perspective, without a baseline for comparison, or present simple measurements that do not show any kind of correlation, which greatly (or even completely) limits the value of the reported information [5][8].

A. Requirements From ISO/IEC 27001 ISMS Standard

ISO/IEC 27001:2005 Information Security Management System (ISMS) [9] is intended to bring formal specification of information security under explicit management control. It is a mandated specific requirement, where organizations can therefore be formally audited and certified compliant with the standard.

The standard provides some confidence level of information protection among business organizations. With the existence of ISO/IEC 27001 ISMS certification, these organizations can increase their protection of information by having independent assessment conducted by the accredited certification body. The certificate has proven the potential marketing to the most business organizations, where a total of 7536 organizations have already been certified worldwide [10]. Obviously, there are other 27000 series that support this standard, including ISO/IEC 27002 Code of practice for information security management [11], ISO/IEC 27003 ISMS implementation guidance [12], ISO/IEC 27004 Information security management – Measurement [13] and ISO/IEC 27005 Information security risk management [14].

There are 133 security controls in Annex-A of ISO/IEC 27001 ISMS standard. ISO/IEC 27002 [11] provides the best practice guidance in initiating, implementing or maintaining the security control in the ISMS. This standard regards that “not all of the controls and guidance in this code of practice may be
applicable and additional controls and guidelines not included in this standard may be required.”

Information security measurement is a mandatory requirement in this standard where a few clauses are stated in [9]:

- “4.2.2(d) Define how to measure the effectiveness of the selected controls or groups of controls and specify how these measurements are to be used to assess control effectiveness to produce comparable and reproducible results;
- 4.2.3(c) Measure the effectiveness of controls to verify that security requirements have been met;
- 4.3.1(g) documented procedures needed by the organization to ensure the effective planning, operation and control of its information security processes and describe how to measure the effectiveness of controls;
- 7.2(f) results from effectiveness measurements; and
- 7.3(e) Improvement to how the effectiveness of controls is being measured.”

Moreover, the new revision of ISO/IEC 27001:2013 [15] standard has also highlighted the importance of effective measurement in their mandatory requirement clauses 9 - Performance evaluation.

B. Summary

The standard highlighted that the organization must evaluate the information security performance and the effectiveness of the ISMS. The evaluation of the effectiveness should include but not limited to: (i) monitor and measure information security processes and controls; (ii) methods to use when monitor and analyze measurement for valid or significant result; (iii) time and personnel to perform the monitoring and measurement; (iv) determine time, duration and personnel to analyze the measurement results.

Thus, in ensuring the ISMS effectiveness, the information security measure can facilitate the management to make decision by the collection, analysis, evaluation and reporting of relevant performance-related measurements.

The importance of information security measurement is well defined and highlighted in both standards. Most of the research papers focused on information security metrics for general IT systems. However, lack of research on technical security metrics [16][17][18][19]. Thus, our research is focusing on the development of technical security measurement that will be incorporated in the technical security metric model.

II. RELATED WORK

In understanding the requirements, the security metric, measure and effective measurement must be defined. “Whatever the driver for implementing ISO 27001, it should no longer be just about identifying the controls to be implemented (based on the risk), but also about how each control will be measured. After all, if you can’t measure it, how do you know it’s working effectively?” [20].

In our previous study [21], we defined information security metrics is a measurement standard for information security controls that can be quantified and reviewed to meet the security objectives. It facilitates the relevant actions for improvement, provide decision making and guide compliance to security standards. Information security measurement is a process of measuring or assessing the effectiveness of information security controls that can be described by the relevant measurement methods to quantify the data and the measurement results are comparable and reproducible.

Apparently, we also mapped the definitions of security metric, security measure and effective measurement from the previous studies [2][6][20][22][23][24][25][16][26][17][27][28][29][30][18][31][32][33][19][34] (refer to Table 1).

From Table I, we grouped the eight (8) components of security metrics and supported by the components in security measures. The definitions of security metric and security measures are quite similar through the analysis of the descriptions. To ease the understanding, the metric is also sometimes called a “measure” [27]. However, in the development of TSMM, we intend to develop a security metric that can consist of a few security measures.

We also derived the eight (8) criteria of the effective security metric (ESM) that are supported by the following statement:

a) Meet security objectives - ESM should gauge how well organization is meeting its security objectives. It should also have a clearly defined set of variables which are acceptable, unacceptable and excellent range of values that can be easily identified by the audience to which the measure is communicated.

b) Quantifiable values – ESM should be a quantitatively measurable that derived from precise and reliable numeric values and expressed by using understood and unambiguous units of measure.

c) Simple measurement – ESM should be easily recognize and comprehended by the audience for which they are intended. The measurement method should be produced by a process or procedure to collect data, determine the data source, scale or score, analysis, and reporting of relevant data. The right and competent personnel should be identified to conduct the measurement and able to analyze and produce the accurate report.
<table>
<thead>
<tr>
<th>Security Metric</th>
<th>Security Measure</th>
<th>Effective Measurement</th>
</tr>
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<tbody>
<tr>
<td>(1) Security Objectives</td>
<td>• Clearly defined acceptable value</td>
<td>• Meet security objectives and requirements</td>
</tr>
<tr>
<td></td>
<td>• Performance goals and objectives (efficiency, effectiveness)</td>
<td>• Clearly defined</td>
</tr>
<tr>
<td>(2) Quantifiable, computed value</td>
<td>• Quantifiable information</td>
<td>• The value is objective and quantifiable</td>
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<tr>
<td></td>
<td>• Scope of measurement (Process, performance, outcomes, quality, trends,</td>
<td>• Determine the Key-Performance-Indicator (KPI)</td>
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<td></td>
<td>conformance to standards and probabilities)</td>
<td></td>
</tr>
<tr>
<td>(3) Method of Measurement</td>
<td>• Easily identified</td>
<td>• Simple measurement</td>
</tr>
<tr>
<td></td>
<td>• Quantitative indications by some attributes of a control or process</td>
<td>• Low cost and easy access</td>
</tr>
<tr>
<td></td>
<td>• Process of data collection, data from security assessment process</td>
<td>• Capability to measure accurately</td>
</tr>
<tr>
<td></td>
<td>• Repeatable</td>
<td></td>
</tr>
<tr>
<td>(4) Analysis of Data</td>
<td>• Apply formulas for analysis</td>
<td>• Consistent value</td>
</tr>
<tr>
<td></td>
<td>• Track changes</td>
<td>• Accurate time and data</td>
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<td></td>
<td>• Quantifiable information for comparison</td>
<td>• Comparable and reproducible results</td>
</tr>
<tr>
<td></td>
<td>• Comparable to a scale/benchmark/Predicted baseline</td>
<td>• Security controls are implemented correctly, operating as intended, and meeting the</td>
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<tr>
<td></td>
<td>• Repeatable</td>
<td>desired outcome</td>
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<td></td>
<td>• Quantifiable</td>
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<td></td>
<td>• Meaningful result (score, rating, rank, or assessment result)</td>
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<td></td>
<td>• Process of data collection, data from security assessment process</td>
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<td>• Security controls are implemented correctly, operating as intended, and</td>
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<td>meeting the desired outcome</td>
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<td></td>
<td>• Monitor the accomplishment</td>
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<td></td>
<td>• Increase confidence level</td>
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<td></td>
<td>• Security improvement</td>
<td></td>
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<tr>
<td>(6) Reporting relevant data</td>
<td>• Communicated/Reported</td>
<td>• Present to targeted audience/Stakeholder</td>
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<tr>
<td></td>
<td>• Intended audience</td>
<td></td>
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<tr>
<td>(7) Decision making</td>
<td>• Facilitate decision making</td>
<td>• Facilitate corrective action</td>
</tr>
<tr>
<td></td>
<td>• Facilitate decision making</td>
<td></td>
</tr>
<tr>
<td>(8) Requirement to Standard, regulatory,</td>
<td>• Align with business goals and regulations</td>
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<tr>
<td>financial and organizational reasons</td>
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</table>

**d)** Comparable result – ESM should produce a baseline for comparison purposes, repeatable or consistently reproducible, so that different people at different times can make the same measurement. Apparently, this supports the adequacy of in-place security controls, policies, and procedures; security controls are implemented correctly, operating as intended, and meeting the desired outcome.

**e)** Corrective action - ESM should provide the appropriate timeliness and frequency of measurement for the change of measurement target so that the latency of measures does not defeat their purpose. ESM should be collected and reported in a consistent manner. ESM should provide the management to decide the new investment in additional information security resources, identify and evaluate non-productive security controls, and prioritize security controls for continuous monitoring.

**f)** Targeted audience/Stakeholder – ESM should be easily identified by the audience/stakeholder to whom the measure is communicated. For example, provide the relevant measures that produce the significant result for the management to make decision.

**g)** Security Improvement – ESM should provide some indicators that could be a sign of relevant security characteristics that prescribes the meaning of obtained security values and achieves to some level of improvement.

**h)** Align with business goals - ESM should provide a benefit to the business it supports.

The development of our TSMM is based on the above criteria and to focus on security performance for the relevant controls (see Fig.1).
III. RESEARCH METHOD FOR DEVELOPMENT OF TECHNICAL SECURITY METRIC MODEL (TSMM)

The GQM approach was originally developed by Basili et.al [1] in evaluation and measurement of software products and development processes. Ever since developed, this approach was used consistently focus on the software measurement and processes [35]. There were also a few research studies on business processes [36][37][38] and security metrics [26][39][40][41][42][43][44]. However, there is no research study conducted for measuring the network security management using the same approach.

To achieve the objective of developing the TSMM, we propose a research method based on a combination of approaches. The outcome of this research method is the introduction of network security management metrics as attributes to the TSMM.

The first approach is to define the technical security metric (TSM). We set our goal to meet the requirements from ISO/IEC 27001 ISMS standard. The paradigm of Goal-Question-Metric (GQM) [1] is used and described further which to align with standard requirement (Fig.2).

We combine the developed Goal-Question-Metric (GQM) paradigm and data of literature review (Fig.3) as a first step. This approach is used for developing the initial TSM in a top-down manner, from general objective to the relevant metrics or outputs and combines the inputs from the literature review. The application results in GQM models, leading to the initial TSMM. However, this initial development work remains subjective and potentially incomplete.

In the second approach (Step 2), we use the GQM method consists of four phases [45]: planning, definition, data collection, interpretation (see Fig.4). The explanation of these phases is based on the compliancy to the requirement controls of ISO/IEC 27001 standard [9] for A.10.6 Network security management (NSM); A.10.6.1 Network controls; and A.10.6.2 Secure network services.

Our implementation adopts the processes and activities by [41] and [46].

- **The Planning phase:** The NSM-team is established and the compliance requirement is clearly delivered. The desired improvement areas such as performance, security and monitor are identified. The team selects and characterizes the products or controls to be studied. The result of this phase is a project plan that outlines the characterization of the products or controls, the schedule of measuring, the organizational structure, and necessary awareness and training for people involved in measurements.

- **The Definition phase:** The measurement goals are defined. This phase is also to identify and analyze the perception and understanding of effective measurement requirement from ISO/IEC 27001 standard [9]. We will create a new template to gather all related information based on some other templates from ISO/IEC 27004 [13] and NIST SP800-55 [27]. For the purpose of this, the interviews may be conducted with people (management and technical) involved in the process or product under study. Based on the goals, relevant questions are developed to identify the specific quality attributes and to re-define the goals.
precisely. For each question a hypothesis with an expected answer should be defined. Next, the metrics are defined for each question and checked on consistency and completeness. Results of this phase are an analysis of compliance plan and a measurement plan.

- The **Data Collection phase** – the team is required to prepare the data collection within their knowledge and availability. The data may be extracted manually or electronically and may involve automated data collection tools. Results of this phase are to develop the data support system consisting of spreadsheets, statistical tools, database applications and presentation tools.

- The **Interpretation phase** - the collected data is processed and analyzed according to the metrics defined. The measurements result should be able to answer the questions, and with the answers it can be evaluated if the initial goals are attained. Moreover, the measurement result should provide some values that describing the performance measurement of the security controls.

![Data Collection phase](image1)

**Figure 3.** Data from literature review

![Four phases of GQM-method](image2)

**Figure 4.** The four phases of GQM-method [45]

The second approach is used as a validation/improvement of the first step. It is based on a literature review of security metric standards and guidelines and measurement methods for network security controls. This approach is a bottom-up, being an analysis of the literature to identify the metrics currently used. A comparative analysis is developed between the metrics and those defined through GQM. This comparison is summarized in an analysis table.

As shown in Fig.5, we map the GQM-method with ISO/IEC 27004 template for an information security measurement construct and show the synchronization link (relevant colored-box). We refer to this standard as a reference and example to form a GQM-Measurement plan.

Once the literature is completely surveyed, the development of GQM-Measurement plan should be ready. The relevant people should be interviewed to validate the initial TSMM. Finally, the TSMM is accordingly revised.
A. GQM-Measurement Plan

We develop a GQM-Measurement plan consists of goals, questions, and metrics in a hierarchical structure (see Fig. 6) based on [1][45].

In developing the goals, the security objectives of A.10.6, A.10.6.1 and A.10.6.2 of ISO/IEC 27001 requirement controls [9] are referred. At this stage, the understanding of the security control requirements is very important. The understanding can be obtained through the interview with the relevant people and checking available process or product descriptions [46]. If goals are still unclear, a reference to ISO/IEC 27002 [11], FDIS ISO/IEC 27033 [2] and NIST SP800-55 [25] can also assist.

The proposed questions shall refine the goals make them operational enough so that it would not create difficulties to reveal the relationship to the collected data and ease the interpretation of the answers towards the goals [46]. The questions are also derived from the literature reviews.

The questions are stated in a quantitative way where data can be collected by measurements. We provide the expected answers to the questions and formulated as hypotheses. Through hypotheses, we can learn the effect from measurements and compare the knowledge before and after measurements.

According to [1][41][46], we can define several metrics for each question. It is also possible that one metric may be used to answer different questions under the same goal. We choose metrics with quantitative level making it possible to assign numbers to a quality attribute. Metrics are defined to answer the relevant questions and should be able to support or reject the stated hypotheses (if any).

A simple Goal-Measurement plan is developed for the purpose of this discussion (as full development of plan is currently in progress). The example of GQM-Measurement plan as stated in Table II.

| Goal | G1 | A.10.6.1 Network controls - Networks shall be adequately managed and controlled, in order to be protected from threats, and to maintain security for the systems and applications using the network, including information in transit.

| Question | Q1 | What are the risk levels for network controls and security controls that protect your information? |
| Metric | M1.1 | Risk Assessment = Asset Value x Threat x Vulnerability |
| Metric | M2.1 | Frequency of audit logging review |
| Metric | M2.2 | Security Incidents report (IDS/IPS/user report) - Comparison of number of total incidents with the threshold. |
| Question | Q3 | How often the security assessment and/or penetration testing are conducted within a year? |
| Metric | M3.1 | Frequency of assessment conducted |
| Metric | M3.2 | Success or failure rate for corrective action |
| Metric | M3.3 | Conducted by trained/experience staff |
| Question | Q4 | How often your organization conduct the security updates for network controls? |

Figure 5. Synchronization between GQM-Method and ISO/IEC 27004 Measurement Template

Figure 6. The GQM Paradigm by Basili et.al [1]
The objective of this paper is to identify and to define a set of metrics for the TSMM with a systematic and scientific approach to comply with ISO/IEC 27001 standard. We use the GQM approach on the TSMM and review with regards to the literature. The result of this paper is the enrichment of the TSMM with suited network security management metrics.

Although the initial developed TSMM are validated through literature analysis, their testing in a real case would provide a concrete instantiation and validation of their relevance. The GQM-Measurement plan is currently being attended.

As part of the next step of our future work, the metrics will be integrated into the initial TSMM and a case study is to be conducted using our GQM-Measurement plan. This will validate the final TSMM.

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A Survivability Strategy in Mobile Network by Key Distribution and Cross-layer protocol

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Abstract:
The capability to provide network service even under a significant network system element disruption is the backbone for the survival of network technology in today’s world, keeping this view in mind, the present paper highlights cryptosystem and Cross-Layer Protocol. A global initial key distribution method based on public key certificate chain shall be presented. And also present a method for survivability strategy in mobile network.

Keywords:
Survivability, Mobile Network, Key Distribution Cross-layer protocol

Introduction:
Network survivability is considered to cope with increasing demand for reliable network system. Network survivability is an essential aspect of reliable communication service. Survivability consists not only of robustness against failure occurring due to natural faults. In mobile networks infrastructure element such as base station (BS), and base station Controller (BSC), wired links, and mobile switch centre (MSC), are employed to provide and maintain essential services, hence the operation interruption of a network component affects overall or partial network services. wireless access network have unique characteristics to support mobile users which can result in different survivability and security aspect [1]. There for wireless survivability strategies must be designed to improve the service available rate of the network component system [1-2]. Due to the mobility if node, the network topology is highly dynamic and all traffic suffers from frequent path breaks. The survivability of routing protocols of such networks must be able to perform efficiently and effectively. In this paper we propose a solution on traditional survivability strategy in mobile network. Survivability is a critical requirement for reliable services in any network. This paper highlights the challenge of providing Survivability. Over the years, cross-layer designs ,which let two or more protocols from non-adjacent layers function in concert, have become very popular. since these tend to sacrifice generality for performance improvements. The two modularity, which provides flexibility in protocol update and specialization, which uses the specificities of a network to improve performance. Cross-layer designs may be best understood by explain in their opposite-layered scheme. The latter prevent communication between non-
Survivability:

Traditional security research is primarily focused on the detection and prevention of intrusion and attacks rather than on continued correct operation while under attack. Fault tolerance is usually concerned with redundancy that is required to detect and correct up to a given number of naturally occurring faults. Nature is not malicious, and conventional failure model make significant assumptions, in particular, assuming faults to be independent and random. The presence of intelligent adversarial attacks can protocol vulnerability often become more important considerations in the presence of an adversary.

There are a number of definitions of survivability. The one we use here is from the Software Engineering Institute, which emphasizes timeliness, survivability under attack and failure, and that detection of attack is a vital capability. Survivability is the capability of a system to fulfill its mission in a timely manner. Even in the presence of attacks or failures. Survivability goes beyond security and fault tolerance of focus on delivery of essential service even when system is entered or experience failures. And rapid recovery of full service when conditions improve. Unlike traditional security measures that require central control and administrative, survivability address highly distributed unbonded network environment that lack central control and unified security policies.

The Three Rs: Resistance, Recognition, and Recovery

The focus of survivability is on delivery of essential services and preservation of essential assets. Essential service and asserts are those system capabilities that are critical to fulfilling mission objectives. Survivability depends on three key capabilities: resistance, recognition, and recovery. Resistance is the capability to detect attacks as they occur and to evaluate the extent of damage and compromise. Recovery, a hallmark of survivability is the capability to maintain essential service and asserts during attacks, limit the extent of damage, and restore full service following attack.

We further extend this definition to require that survivability be able to quickly incorporate lesson learned from failure, evolve, and adapt to emerging threats. We call this survivability feature refinement.

We can classify survivable mobile wireless networking requirement into four categories based on [3]: (i) resistance requirement; (ii) recognition requirement; (iii) recovery requirements; and (iv) refinement requirement. We can also describe a requirement definition process [4]. This includes the definition of system and survivability requirement, legitimate and intruder usage requirement, development requirement, operation requirement, and evolution requirement. Essential service must be identified and specified for the penetration, exploration, and exploitation phases of the attack.
The approach has guided this work and is recommended for more detailed requirement analyses for future mobile wireless network. Ultimately, there are two distinct aspects of survivability that apply at all networking layers.

**Information access requirement:**

Does the user have access to the information or service required to complete the task in the presence of failure or attack? For e.g. it is possible to replicate service or information and provide them locally when the network gets partitioned? End – to-end communication should not be mandated in these cases.

**End-to-end communication requirement:**

On the other hand there are interactive application , inter-personal communication such as voice calls, or dynamically generated information such as current sensor data, which require true end – to – end connectivity . Do existing session survive? Can the user create new session to reach the intended communication end-point even in the presence of failures and attacks? This requires that the communication end-point themselves survive and that the communication end-points themselves survive and that the adversary must not be able to permanently partition the network. Furthermore, the adversary must not be able to permanently disable access to required services such as authentication, naming, resource discovery, or routing.

**Mobile Network Survivability:**

Existing work on survivability in the context of cellular telephone networks concentrates primarily on infrastructure survivability (for e.g. see the outage index metrics and does not consider adversarial attacks[5-6]. However, they offer some insight on quantifying survivability and the role of network management tools. Networks are vulnerable during upgrades, especially those involving software [7]. Furthermore, rapid evolution leads to learning -cure problems as well as – over – concentration leads or service into single points of failure. This problem is exacerbated by deficits in network management tools to operate and maintain increasingly complex system. Architectural improvement applicable to mobile include the use of redundant networks.

**Base Station:**

In more environment, a cell that is geographical region unit is geographical region unite is covered by the radio frequency of a base station. Each call is controlled by a BS which has a fixed connection to a BSC (or RNC). In mobile network infrastructure element such as base station controller (BSC), wired links and mobile switch centre (MSC) are employed to provide and maintain essential service,. Hence the operation interruption of a network component affects overall or partial network services.

A radiation antenna is classified as omnidirectional and directional with an omnidirectional antenna, a single frequency spreads out in all directions of 360 coverage. A cell is directional antenna with each different set channel.

**System State of Base Station:**

The BS system, including antenna parts , cannot provide partial or whole service function for coverage cell when single or more fatal failures occur in the BS system , in this paper, we consider that system failures are caused by key distribution method. For example by interrupt sequence mishandling, overall system operation falls into failure state because of unanticipated handled interruption to a component of the system.

**Key distribution frame work:**

In mobile computing environment , when a mobile host moves to the visited domain it needs to be authenticated by the current domain before gaining the service of the provider in the domain. If the mobile host requires the current visited domain to provide service it will need a shared key with current domain authentication server. An effective method is using the hybrid authentication server. An effective method is using the hybrid authentication method including key cryptographic (PKC) system. PKC can verify the identity of the owner of a public key certificate can verify the identity of the owner of a public key and avoid the attack of impersonation. It is impossible to take single public key certificate (SPKC) authority to disseminate the PKC in the interrupt. so a sable Hierarchical Public key distribution (HPKD) framework is presented. According to the scale of the mobile users, the number of the layer of the framework can be decided. Figure 1 is an example of framework. The top level node S A1 is the root of the survivability certificate.
The distributing process of public key certificate (PKC) is described as follows:

1. SCA1 Sends PK1, SC2 to SCA2;
2. SCA2 Sends PK1||PK2, SC2||SC3 to SCA3;
3. SCA3 Sends PK1||PK2||PK3, SC2||SC3||SC4 to SCA4;
   ...
(n-1) SCA_{n-1} sends PK^1||PK^2||...||PK^{n-1}, SC2||SC3||SC4||...|SCn
(n) SCA_n Sends PK1||PK2||...||PK^n, SC2||SC3||SC4|......

After the distributing process is performance the AS and user gain a certificate chain respectively.

A network is connected if these are a path between every pair of nodes. And a network is biconnected if the toss of any one link leaves the network connected. By key distributed framework it is clear that if user is linked by SCA1 then it is important that every node connected by each other.

**Cross-Layer Protocol:**

It is repeatedly argued that layered architectures are not suitable for mobile networks, to illustrate this point, researchers usually present what they call across-layer design proposal. Cross-layer design refers to protocol design done by actively exploiting the dependence between protocols gain. This is unlike layering, where the protocols at the different layers are designs independently. Protocols can be designed by respecting the rules of the reference architecture.

In a layer architecture, this would means designing protocols such that a higher-layer protocol only makes use of the services at the service is being provide.

**Cross-layer module:**

A key concept at the physical cross-layer is the capacity of survivability region, which characterizes a tradeoff between achievable capacities at different links for survivability. Consider a network with A, B and C_L as the link gain, power, noise, respectively. Denote D_i as the interference coefficient from link k_1 to the link k_2. Assume that each node has a power budget X_{max}. Thus the power control with a physical Cross-layer module interference model may be formulated as

\[
\max \sum \gamma n
\]

\( \gamma \) is dual variable. \( \gamma \) play a key role in coordinating the survivability networking layer demand and layer supply.

n-capacity region at the physical cross-layer

\[ n = \log(1+\text{SINR}) \] for every \( k_2 \)  
\[ \text{SINR} = \frac{AB}{\sum AB + C_L} \] for every \( L \)  

Because of interference, the power control problem is a non convex optimization problem that is present physical cross-layer module. The capacity approximitly.

**Conclusion:**

In this paper, we have proposed a scheme for mobile service use of BS system and key distribution and Cross-layer protocol. The key distribution takes full in to account and the certificate chain is transferred in clear text, impeders can observe the home SCA controls and cross-layer protocol for many mobile host. When the mobile host gets to visited domain, it may get a survivability scheme.

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**Fig Frame work of Key Distributionure-1**


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Effect of Cross Layer optimization of Traffic Management in Ad HOC Mobile Network

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Abstract:- The optimal and distributed provisioning of high through output in Mobile Ad Hoc Network (MANET) is a network consisting of a set of wireless mobile routers and Communication with each other. The Network Mobility(NEMO) for the traffic represents the moving behavior of directional antenna and mobile rooters. Use the Cross-layer protocol in ad hoc wireless network we can drastically improve the utilization through overlapping communication is the different direction for the traffic. This paper highlight the chalenge to find out a route of effect the cross-layer protocol for traffic-management in Ad Hoc wireless network. In present paper we propose mobility for traffic management in Ad Hoc wireless network by use of theory of Cross-layer protocol.

Keywords:- Ad hoc Network, Cross-layer protocol, Directional Antenna, Mobile Router, Network Mobility

Introduction:- AD HOC NETWORKS are multiple wireless networks consisting of a large number of radio equipped nodes that may be as simple as autonomous sensors. These type of network are useful in any situation where temporary network connectivity is needed such as in disaster relief. A mobile ad hoc network (MANET), is a network comprising wireless mobile rooters (MRs) that communication with each other without centralized control. The dynamic of wireless ad hoc networks as a consequence of mobility and disconnection of mobile host. The MRs that are within each other’s radio range can communication directly. Each mobile rooter’s acts as host in MANET environment. Mobile rooters are free to join or leave the network at any point of time.

Here we are working towards implementing wireless ad hoc community network (WACNEC) that was small, low cost directional antenna, known as ESTAR (Electronically Steerable passive Array Radiator) antenna, with each user terming [1,2].Due to unreliability of wireless links, it has been of interest to study the impact of physical-layer techniques on the design ,including medium access control(MAC),packet scheduling, power control, routing, transport protocol, and ultimately the QoS at the application level in the wireless networks.

Mobility system define rooter, movement patterns in ad hoc networks. Since MANETs are currently not deployed on are large scale and due to inherent randomness of mobility modes, research in evaluating the performance of routing protocols on various system of mobility. In this paper the performance of MANET for the effect of cross-layer protocol [3]. In[3],a cross-layer design approach is employed to improve the performance of combined cooperative schemes. The cross-layer information is minted
in a separated data structure and is shared among layers.

Whatever may be routing scheme, they all rely on using Omni-directional antenna. The use of directional antenna to find out a route and use it in database has not been explored properly. Here we proposed Cross-layer protocol where each node keeps certain neighborhood information dynamically through the Maintenance of an Angle – SINR Table.

For experimental purposes we have considered A framework to evaluate the impact of different mobility models on the performance of MANET routing cross-layer protocol is provided in [4] various protocol independent matrices are provided to capture interesting mobility characteristics restrictions.

Mobility models:- The mobility model is designed to described the movement pattern of Mobility models used in the simulation study of MANET.: traces base model and synthetic base model [4]. The traces base model obtains deterministic data from the real system. This mobility model is still in its early stage of research, therefore it is not recommended to be used. The synthetic based model is the imaginative model that used statics. The movement of each MN to its destination has a pattern that can be described by a statistical model that expresses the movement behavior in the real environment.

The Framework:-

Angle – SINR Table :-

In order to make the directional routing effective, a node should know how to set its transmission direction effectively to transmit a packet to its neighbors. So each node periodically collects its neighborhood information and forms an Angle- SINR Table (AST). \( \text{Sinu}^2 \ m(t) \) (Signal – to – Interference and Noise Ratio) is a number associated with each link \( l^n \ n, m \), and is a measurable indicator of the strength of radio connection from node \( n \) to node \( m \) at an angle \( u \) with respect to \( n \) and as perceived by \( m \) at any point of time \( t \). AST of node \( n \) specifies the strength of radio connection of its neighbors with respect to \( n \) at a particular direction . Angle - SINR Table for node \( n \) time \( t \) is shown below (Table I) where we assume that nodes I, j and k are the neighbors of \( n \).

**TABLE I. ANGLE – SINR TABLE (AST) FOR NODE n**

<table>
<thead>
<tr>
<th>Azimuth Angle (degree)</th>
<th>I</th>
<th>j</th>
<th>K</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>( \text{Sinu}^2 n_i(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
</tr>
<tr>
<td>30</td>
<td>( \text{Sinu}^2 n_i(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
</tr>
<tr>
<td>60</td>
<td>( \text{Sinu}^2 n_i(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
</tr>
<tr>
<td>...</td>
<td>...</td>
<td>...</td>
<td>...</td>
</tr>
<tr>
<td>330</td>
<td>( \text{Sinu}^2 n_i(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
</tr>
<tr>
<td>360</td>
<td>( \text{Sinu}^2 n_i(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
<td>( \text{Sinu}^2 n_j(t) )</td>
</tr>
</tbody>
</table>

In order to form AST, each node periodically sends a directional request in the form of a directional broadcast, sequentially in all direction . in this work it has been done 30 degree interval, covering the entire 360 degree space sequentially. A node is \( i \) in the neighborhood of \( n \) will wait until it receives all the request packets generated by \( n \) in all direction in that occasion. In others word, node I accumulates the entire column of the AST of \( n \) for node I, I accumulates the entire column of the AST of \( n \) for rooters i. Here, rooters i, after receiving the first request from \( n \), has to wait a
pre-specified amount of time to make sure that the directional broadcasts by n in all direction are over. Rooters I sends this information from all the neighbors of n, the Angle-SINR Table of n would be complete.

Omni transmission both transmission range and spatial reuse can be substantially enhance by having nodes concentrate transmitted energy only towards their destination direction, thereby achieving higher signal to noise ratio.

When there is a need to utilize only the directional characteristic, the demands are more since this is possible only when the rooters which wants to transmit and the rooters which wants to receive are synchronized with their respective related modes (i.e.) one rooters is in the transmit rooters and other is in the transmit mode and other is in the receive mode and are pointing towards each other as shown in figure 1.

If \( x_1 = 60^\circ, x_2 = 120^\circ, x_3 = 180^\circ, x_4 = 240^\circ, x_5 = 300^\circ, x_6 = 360^\circ \) then

\[
\max \sum \psi n
\]

\( \psi \) is dual variable. \( \psi \) play a key role in coordinating the survivability networking layer demand and layer supply.

This is the most important task of any Cross-layer protocol for Mobile Network is to identify the set of non-interfering transmissions is an area and to coordinate the activities of the various senders. As we discussed above, the notion of non-interfering transmissions depends on the antenna orientations of the senders. Thus, an indoor directional Cross-layer protocol must not only identify the set of possible concurrent transmissions but also determine their orientations. Directional antenna has the potential to provide the necessary interference reduction by spatially confining transmission.

**CROSS-LAYER PROTOCOL:**

It is repeatedly argued that layered architectures are not suitable for mobile networks. To illustrate this point, researchers usually present what they call across-layer design proposal. Cross-layer design refers to protocol design done by actively exploiting the dependence between protocols gain. This is unlike layering, where the protocols at the different layers are designed independently. Protocols can be designed by respecting the rules of the reference architecture.

In a layer architecture, this would mean designing protocols such that a higher-layer protocol only makes use of the services at the service is being provide.

**Physical cross-layer module:**

A key concept at the physical cross-layer is the capacity of survivability region, which characterizes a tradeoff between achievable capacities at different links for survivability.

Consider a network with A, B, and C, as the link gain, power, noise, respectively. Denote \( D_{l_{ij}} \) as the interference coefficient from link \( k_1 \) to the link \( k_2 \). Assume that each node has a power budget \( X_{\text{max}} \). Thus the power control with a physical Cross-layer module interference model may be formulated as

\[
\max \sum \psi n
\]

\( \psi \) is dual variable. \( \psi \) play a key role in coordinating the survivability networking layer demand and layer supply.

n-capacity region at the physical cross-layer

\[ n = \log (1 + \text{SINR}) \text{ for every } k_2 \quad E \]

\[ \text{SINR} = \frac{AB}{\sum AB + C_L} \text{ for every } L \quad E \]

Because of interference, the power control problem is a non convex optimization probe that is present physical cross-layer module[9] capacity approximatly.
CONCLUSION: In this paper we use the Cross-layer protocol and directional antenna to find out a route optimization and we also present the effect of the Cross-layer protocol for traffic management in Ad Hoc wireless network. And we propose a system of traffic management in Ad Hoc wireless network by use of theory of directional antenna.

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Performance of and Traffic management for a Mobile networks by using Cross-layer protocol

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Abstract:-
Over the Recent years a considerable amount of effort has been devoted towards the traffic management and root is the important capability to provide best network technology in today’s world. Present paper we study the traffic management for mobile networks and we addresses current issue of the traffic management. Present the performance of Mobile Network by using Cross-layer protocol.

Index Terms:-
Mobile Networks, call admission control, QoS (Quality of Service), routing optimization

1. Introduction:-
Over the recent years a considerable amount of effort has been devoted towards the performance, evaluation for the traffic management of wireless mobile networks of wireless mobile networks (WMN). A considerable amount of research efforts has been used to characterize user and calling behavior and their performance impact on wireless mobile networks. At present the mobility in most mobile in most mobile networks is confined to the end users only.

With the development of mobile compfor the devolopement of traffic management the call admission schemes are ganarely adopt[1] mobile user authentication is absolutely nessesary as mobile user want to request service provided by the provide survival in the visited traffic management a networks could as simple as of forum held in a city ,state ,country ,whole world between the people . where people communicating with each other[2] .the system may need to block inas a mobile user ression the CAC schemes are generally adopted by setting thresholds for hand off calls and new call differently given the traffic condition and it is the maximum number of users that can be supported. The system may need to block incoming users if all of the entire band width has been used up to provide the highest QoS to existing users. However if these existing users can be degraded to a lower but acceptable QoS level, it is possible to reduce the blocking probability without degrading the QoS of existing users. A graceful degradation mechanism is proposed in [3]. Thus a system could free some bandwidth allocation for new users. In this paper we address current issues in traffic management for cellular mobile networks. In traffic management coming user that can be supported .the system may need to block incoming user and congestion control, courcoubetis and series device new procedures and tools for the analysis of network traffic measurement.
2. MODEL DESCRIPTION:-

We consider uplink communication in a wireless mobile networks. As an accepted call does not always send data frames. Then for best traffic we consider the activity factor \( l \) as the probability that a call is active. We represent QoS requirement of traffic by required transmission rate. The required transmission rate can be obtained by setting the target level.

Often these intra – and inter –traffic interferences of call can be large so that the target bit error rate of traffic interferences(BERIT) can not be achieved temporarily, which is called outage. The outage probability needs to approach zero as close as possible and can be different for each class. Here we assume for traffic management the allowed outage probability is the same.

3. OUTAGE PROBABILITY FOR TRAFFIC:-

In a mobile network a traffic management the supports a single class of calls, the outage probability is given by [2].

\[
P_{\text{out}} = P \{ N^a + M^a > (3/2) G(x^1 - (Yb/N_0)^{+1}) \} \quad (1)
\]

When \( N^a, M^a, G, X, Y^b, \) and \( N_0 \) represent the number of active calls in the current call, similarly in a network that support L-Class of calls, we obtain

\[
P_{\text{out}} = P \{ \sum_{i=1}^{L} (Ybi / Ybj) Ci (N_i^a + M_i^a) \geq A_j \}
\]

\[
= P \{ \sum_{i=1}^{L} \theta_i (N_i^a + M_i^a) \geq \eta j \} \quad (2)
\]

When i, j represent traffic call classes (TCC), \( C_i \) is the number of orthogonal codes needs for a TCC, ‘i’. By the Gaussian random variable from the limit theorem and we can write control the outage probability of a TCC’ ‘j’. As

\[
P_{\text{out}} = Q (\eta j - \lambda / \partial \lambda ) \quad (3)
\]

Where \( Q (\xi) = 1/ \sqrt{2 \pi} \int_{-\infty}^{\xi} e^{-x^2/2} dx \)

And represent the total traffic receive single power (TRSP) i.e. \( \sum_{i=1}^{L} \theta_i (N_i^a + M_i^a) \)

Therefore \( \bar{\lambda} = (1+f_1) \sum_{i=1}^{L} \theta_i \bar{N}^a_i \quad (4) \)

And \( \partial^2 \bar{\lambda} = \sum_{i=1}^{L} \theta_i^2 (\partial^2_i + f_2 \bar{N}^a_i) \)

Where \( \bar{N}^a_i \) and \( \partial^2 \bar{\lambda} \) indicate the mean and variance of \( N^a \).

According to the assumption of TCC equal outage probability for each class we can \( \eta I = \eta j \) for all I and j there for TCC received single power meets the following relation.

\[
\theta I / \theta j = CiXi (3G - 2CjXj) / CjXj(3G+ 2CiXj) \quad (6)
\]

This indicates that the power allocation refers the target of TCC outage probability.

Call Admission Control (CAC):

System Model:-

The Communication system under consideration can be defined as

\[
r[k] = \sum_{i=0}^{L} h[i] & [k-i] + z[k] \quad (7)
\]

Where \( r[k] \) received call sequence
unknown channel for traffic with memory \( L' \), \( z[k] \) is an independent and identically distributed Gaussian notice sequence. Then traffic management symbol sequence \( s[k] \) is drawn from \( M \)-ary alphabet, \( A \) with equal probability, the vector version of (1) can be written as

\[
\begin{bmatrix}
1 \\
2 \\
3 \\
4 \\
5 \\
6 \\
7 \\
8 \\
9 \\
10
\end{bmatrix}
\begin{bmatrix}
S[K - L] ... S[K] \\
\vdots \\
\vdots \\
\vdots \\
\vdots \\
S[1 - L] ... S[1] \\
S[-L] ... S[0]
\end{bmatrix}
\begin{bmatrix}
h[1] \\
h[1] \\
h[0] \\
h[0]
\end{bmatrix}
+ 
\begin{bmatrix}
r[1] \\
r[0] \\
z[1] \\
z[0]
\end{bmatrix}
\]

Where \( S_k \) is toeplitz data matrix.

**Call Admission Control for Traffic Management:**

For call Admission Control for traffic Management [CACTM] the outage probability is very small defined as

\[
\frac{\partial Q(\eta)}{\partial \eta} < 0
\]

we can show that

\[
\frac{\partial P_{\text{out}}}{\partial N_i} = \alpha_i \frac{\partial P_{\text{out}}}{\partial N_i} > 0
\]

where \( \alpha_i \) is the active factor for (CACTM) a class I call. It is clear that the average rate for mobile network [ARRMN] and outage probability increase with the number of users. Call admission control is a mechanism used in networks to administer quality of service (QoS). Whereas the CAC problem in time division multiple access (TDMA) based cellular networks is simply resalable to the number of physical channels available in the network, it is strongly related to the physical layer performance in WCDMA networks since the multi-access interference in them is a function of the number of users and is a limiting factor in ensuring QoS. The CAC mechanism will thus rely on the “Soft Capacity” of the W-CDMA networks as determined by the level of multi-access interference, often characterized by the signal to interference ratio. In such systems the CAC design leads to a significant interaction between the physical and medium access control layers.

Any given networks have a finite resource that is the number of node, links and buffers and the bandwidth are finite. Thus there are maximum numbers of packets that can be in a network at any given time. Although there is consideration related to the economics of network that favors operating at or close to full capacity there are other considerations related to QoS that provide impacts to operating at less then full capacity.

The higher the packet traffic in a network or part of a network, the greater the average delay per packet due to the limited resources. i.e. if there are more packets the QoS is lowered. Thus in order to maintain QoS the number if calls is to be limited. Rejection if calls create a perception in customer mind regarding provides inability. End to end all problem faced by the network is one od the measuring and forecasting QoS, maximizing call blocking probability and maximizing throughput while maintaining QoS.

The typical parameters that must be managed are latency, filter, bandwidth and packet loss rare [1], [2]. Packet loss is mainly due to buffer overflow packet corruption can occur, this is erroneous reception of nits due to
physical layer impairments. A highly loaded network affords less loss of overloaded of overcrowded buffers. Light loading can also reduce end to end delay and in wireless network based on W-CDMA protocols, lower packet corruption caused by interference. QoS proves coming refers to network capability to different classes of traffic to implement QoS proves coming, a desired QoS is negotiated between the customer and the network on each call and the network QoS parameters are set accordingly.

Physical layer issue are an essential components of QoS management in wireless network, especially with mobile with platforms as varying channels condition and number of users directly affect reliability of communication. Thus QoS schemes must potentially integrate functions at the physical and medium access control (MAC) layers.

CAC has emerged as one key component of such schemes [3].

**Numerical Result:**

We now compare the performance of the consider two CACs through numerical analysis. The system bandwidth is 2.50(MHz) and each code can carry information bits at the rate of 19.2(kbps) so that the processing gain is 256. Two types of calls are considered to manifest the effect of traffic parameters on performance. Class 1 and 2 calls are voice traffic and we set their transmission rates after channel coding at 19.29(kbps). They have different Mobile Network Average Revenue Rate (MNARR) for the traffic management requirement of less than $10^{-4}$ and $10^{-6}$, respectively, and their activity factors are set at 1.0. The coefficient for intercall interference modeling are chosen as $f_1 = 0.114$ and $f_2 = 0.44[12]$.

**CONCLUSION:**

In this paper, we consider Call Admission Control for Traffic Management [CACTM] in Mobile Networks. Through the mathematical analysis and also present outage probability and a system model’s for CAC and we also present an example for Call Admission Control for Traffic Management [CACTM].

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A Fast Survey Focused on Methods for Classifying Anonymity Requirements

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Abstract—Anonymity has become a significant issue in security field by recent advances in information technology and internet. The main objective of anonymity is hiding and concealing entities’ privacy inside a system. Many methods and protocols have been proposed with different anonymity services to provide anonymity requirements in various fields until now. Each anonymity method or protocol is developed using particular approach. In this paper, first, accurate and perfect definitions of privacy and anonymity are presented then most important problems in anonymity field are investigated. Afterwards, the numbers of main anonymity protocols are described with necessary details. Finally, all findings are concluded and some more future perspectives are discussed.

Keywords-anonymity; privacy; online security

I. INTRODUCTION

Utilization of computer networks has been raised in recent years especially internet has become the most famous computer network in all over the world. While we are sending email or talking to our family members through internet, a lot of data or information packets are sent through internet. These packets consist of information of sender and receiver and etc. Since the packets are transmitted by several hops, everybody can monitor them and access to various information such as who started the contact and with whom and some other useful information. Although it is possible to conceal packet contents from a viewer by cryptography, the information of IP header is still accessible for a viewer. For this reason, in two past decades, some improvements have been emerged about anonymity and preserving privacy in formal and public communication field. So far, several systems have been designed and such systems are using by military groups, journalist and public sections. These systems are used to hide identities in virtual internet world. Today, there are various applications which need some methods to provide anonymity for performing their particular tasks. Some examples of these applications could be electronic voting, electronic commerce and etc.[1] Anonymity can be a branch of preserving privacy but preserving privacy is a concept wider than keeping anonymity of entities. Anonymous communication give a possibility to have contacts without disclosing their identities and it does not contain all aspects of privacy. Indeed, anonymity try to conceal operation agents’ information while preserving privacy also hides whatever they perform[1].

Ignoring anonymity aspects causes to jeopardize people privacy. Hence, anonymity is one the most important issues in information security and preserving people privacy. So many applications need anonymity practically. In [3, 4, 5] these applications were categorized as follows:

- Searching information anonymously
- Maintain communication patterns to prevent traffic analysis
- Providing freedom of speech in fanatic environments
- Electronic voting
- Anonymous using of location based services
- Electronic payments
- Electronic cash
- Anonymous web browsing
- Anonymous e-mail
- Anonymous publishing

Anonymity attributes and also the level of anonymity are different in various applications. Therefore, analyzing of anonymity requirements which are used for determining accurate anonymity features in a service are very important and they must be done with high accuracy. For instance, applying a complete level of anonymity is not mostly a best
choice because it causes some problems in many systems. There is no ability to follow and pursue entities in a complete anonymous system while the capability of imputing operations and attacks to people or entities in the system gives a possibility to hamper people’s wrongdoings [2]. Consequently, anonymity must be applied with respect to the organization completely or under some particular conditions.

II. PRESERVING PRIVACY, ANONYMITY

Information is a lifeline in the most institutes, developed organizations and scientific communities. In the institutes and organizations in which information is really important, a quick and proper way is necessary to access to information. Organization and institutes should create informatics infrastructure and try to organize their information. One of success keys in institute, organizations and scientific communities in information age is speeding to generate and offer worthwhile information. After organizing information, it is necessary to provide regular and correct use of this information for others. Along with moving to developed organization based on information technology, it is essential to plan some other methods for preserving information.

Information security points to preserve information and minimize information revealing risks in unauthorized parts. Information security is a set of tools for preventing thefts, attacks, crimes, espionage and sabotages and etc. it is a science for studying various approaches to preserve data in computers and communication systems against access to unauthorized changes. Preserving privacy could be a subcategory of information security. Privacy means such a person can separate his/her information and disclose them on others view by his/her choice. Everyone has some private information which wants to keep them from others.

III. ANONYMITY ISSUES

Today, providing anonymity approaches are considered specially preserving entities’ privacy in electronic commerce and electronic voting and etc. As it is mentioned before, content of messages could be protected by cryptography methods but message rout, source and destination of message, sending time, message length and some kind of information would still remain. Sometimes, valuable information can be accessible only by observing people communication pattern. Access to entities’ information in a communication would be a violation of their privacy and anonymity can prevent revealing of this kind of information. Accordingly, anonymity can be a branch of information security [1]. Nowadays, there are a lot of applications that need anonymity and each application requires special anonymity attributes. For example imagine an electronic payment system that users can search their items and select and buy them. Most customers do not like to show their identity and their private information like interests and preferences. Thus, besides concealing users’ identity connection between users’ different operations must be hided. However, customers’ anonymity should be applied in a limited way to preserve authority of trades correctly. It means that in electronic payment system, anonymity must be applied a different way. When an entity makes a wrongdoing, it would be possible to remove its anonymity and expose its real identity. As a matter of fact, the ability of imputing responsibility of operation to people gives a possibility to hinder crime activities in system by ordaining some rules and politics [2].

On the contrary, suppose an online medical consultation such that gives consultation to patients by hiding patients’ identities. Since patients’ backgrounds have very serious role in correct consultation, hiding information can damages system operation. Consequently, unlike electronic payment covering users’ background might destroy accuracy of disease detection in a medical system.

Several protocols were proposed to provide anonymity in applications until now. It is necessary to have an organized method for developing software security because existence of this kind of method gives a capability to users for analyzing and describing application requirements. Therefore, it can reduce complexity of software analyzing and designing. Furthermore, it can save cost and time because it can recognize and move system problems in initial phase of software development.

VI. ANONYMITY PROTOCOLS

Anonymous communication means the communication layer must not reveal potentially identifiable information such as the user’s IP address or location. This can be met by so-called anonymity protocols such as mix networks [6], onion-routing systems [7].

A. Mix-Net Protocol

The Mix-Net protocol is the base for some other anonymity protocols, Web Mixes [8], ISDN-Mixes [9], and Stop-and-Go-Mixes [10], to name a few. This protocol uses some nodes, called Mix, between sender and receiver. Mixes act as mediators for sending messages and provide the anonymity of the sender against the receiver. Moreover, Mixes are used for hiding a connection against attacks. Figure1 shows this protocol.
In the first step of executing the protocol, the sender defines a sequence of mixes. This can be accidentally or contractually. In this case it supposed the defined sequence is static. Then the sender encodes its message (M) by using the general key (Ki) of mixes. The sender adds the receivers address (AR) to the encoded message of last mix. The form of message is shown below:

$$K_n(R_n, K_{n-1}(R_{n-1}, \ldots , K_2(R_2, K_1(R_1, K_a(R_0, M), AR)))\ldots$$

In message form, there is a random number (Ri) besides of each encoded part. Therefore, before encoding the sender adds a set of random bits besides each part and this prevents the data from dictionary attack.

The important point in mix-net protocol is that even if one mix stay safe against traffic analyzing attack, the connection between sender and receiver will stay safe, because there is no relationship between the input and output of each mix [12, 13, 14].

### B. Onion Routing Protocol

In the Onion-Routing protocol, the sender and receiver can identify each other. The basic goal of this protocol is to make an anonymous connection from others’ viewpoint, and to prevent network traffic analysis. This protocol uses a group of computers named Onion Routers. When a user has a request for sending a message, first the user considers a sequence of Onion Routers, then, makes a data for each router and uses layer encoding with general key encoding to preserve every router’s data from other routers [15].

Each node peels a layer of onion, and this means the node decodes the information with its own private key that is related to itself and sends the result to next routers. After finishing this process of peeling a rout of onion routers is created between sender and receiver that can have an anonymous connection. According to this explanation the onion routing protocol creates a two-sided real-time connection between sender and receiver.

### V. PREVIOUS WORKS ON CLASSIFICATION

These days, anonymity and preserving privacy are becoming very important issues in the digital world [5]. As a matter of fact, the requirements and the level of anonymity for different applications are different, and in many of the applications anonymity should be applied in a controlled and conditional manner. The concept of conditional privacy preservation has been widely studied in vehicular communications especially in VANETs [16]. The works in [17-20] are number of proposed methods to achieve conditional privacy.

Naessens et al. in [21, 22] introduced a methodology for designing controlled anonymity systems. This methodology defines four categories of requirements: Anonymity requirements, controlled requirements, applicability requirements and trust requirements. In their methodology, anonymity requirements come in a graph like “Unlinkable(X, Y)” which is called Linkability graph. In this system X and Y can be any kinds of operations or features. This graph shows for doing any operations what features needed to be accessed and what privileges will be required. Moreover, the proposed methodology uses Petri-Nets to represent the sequence of operations in a system. For each operation, it defines what kind of privileges will be required, when the operation will be finished, and what kind of privileges we will gain. The most important issue regarding this methodology is that it is not possible to consider all anonymity requirements from all aspects and put them into Unlinkable forms. Moreover, in this methodology there is no approach for detecting entities that might try to break the system rules.

Kavaki et al. in [23] proposed a methodology named Pris for considering privacy requirements in software development process. It is a Goal-Oriented methodology and defines the requirements as goals. The conceptual model that is used in Pris comes from Enterprise Knowledge Development framework that is shown in Figure 2. In this methodology, to reach the goals, they are divided into sub-goals until it is possible to reach each goal with a process. There exist several issues about this methodology; it divides all requirements (goals) into two categories: organizational goals and privacy goals which are too general. There are many applications with anonymity requirements, and these requirements are very different in each application; hence, sometimes considering the requirements in the form of such goals is not possible.
TABLE I: DIFFERENT KINDS OF ANONYMITY [10]

<table>
<thead>
<tr>
<th>Traceability</th>
<th>Linkability</th>
<th>Identifiability</th>
<th>Anonymity</th>
</tr>
</thead>
<tbody>
<tr>
<td>✓</td>
<td>✓</td>
<td>✓</td>
<td>0 (no Anonymity)</td>
</tr>
<tr>
<td>x</td>
<td>✓</td>
<td>✓</td>
<td>1</td>
</tr>
<tr>
<td>✓</td>
<td>x</td>
<td>✓</td>
<td>2</td>
</tr>
<tr>
<td>x</td>
<td>x</td>
<td>✓</td>
<td>3</td>
</tr>
<tr>
<td>✓</td>
<td>✓</td>
<td>x</td>
<td>4</td>
</tr>
<tr>
<td>x</td>
<td>✓</td>
<td>x</td>
<td>5</td>
</tr>
<tr>
<td>✓</td>
<td>x</td>
<td>x</td>
<td>6</td>
</tr>
<tr>
<td>x</td>
<td>x</td>
<td>x</td>
<td>7</td>
</tr>
</tbody>
</table>

As well as, Gürses et al in [24] proposed a methodology named CREE for Confidentiality Requirements Elicitation and Engineering which is applied to a real world project in the health care area. However, this work as stated by the authors is a primary effort and is also limited to the confidentiality and do not cover the anonymity concerns.

De Win et al. in [5] proposed a categorization for anonymity. In this categorization, they explained three features of anonymity such as traceability, linkability, and identifiability. They also proposed some combination of these features for any application that needs to be anonymous. For example if an entity is not traceable, linkable and identifiable, this entity is not anonymous, but if this entity is untraceable, unlinkable, and unidentifiable, it has the complete level of anonymity. In this approach the different combinations of these features make different levels of anonymity. Although this categorization is better than other works in this area but, this categorization is not complete enough, because they just consider some features of entities which mostly are related to the messages of entities or the connection between those entities. However, in this categorization they do not consider the features of entity itself which is selected to be anonymous.

VI. CONCLUSION

We live in electronic society and thus many of us read online news, manage online back account, buy online and chat with friends every day. Since we spend a lot of our daily time on the internet, anonymity treats are rising extremely. Storage memories are inexpensive; hence, the information of our activities can be saved and marinated with very low cost. Fortunately, a lot of efforts have been performed to preserve users’ privacy and to anonymize users’ communications in cyberspace up to now. The numbers of these existence anonymity protocols and methods with different approaches were studied in this paper. An accurate Knowledge of anonymity requirements in the system could be helpful to develop more secure and utilisable software and to have more safe online communication in the future.

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