MSIP: A Protocol for Efficient Handoffs of Real-time Multimedia Sessions in Mobile Wireless Scenarios

Dissertation

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for the degree of

Master of Technology

by

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under the guidance of

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2003

Dedicated to

my **Family**, **Thoops**,

my **Friends**

and

All the **Orphans** in this world

Dissertation Approval Sheet

This is to certify that the dissertation titled

MSIP: A Protocol for Efficient Handoffs of Real-time Multimedia Sessions in Mobile Wireless Scenarios

By

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is approved for the degree of **Master of Technology**.

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Abstract

The support for IP mobility has become very important with the increasing growth of mobile applications. One of the key challenges for the deployment of such wireless Internet infrastructure is to efficiently manage user mobility. Mobility is handled by Mobile IP (at the network layer) and Session Initiation Protocol (at the application layer). The main aim of these schemes is to provide seamless connectivity for ongoing communications, that is, to keep the handoff latency delay as less as possible. The delay constraint becomes even more crucial for real-time applications. The Mobile IP scheme has drawbacks like triangular routing, longer delays and the need for tunneling management. SIP solves many of the problems in Mobile IP, but it involves the time consuming process of obtaining a new IP address from the DHCP server.

In this report, we propose a hybrid scheme, MSIP, that is the integration of Mobile IP and SIP. MSIP reduces the handoff delay for real time applications. It avoids the need for tunneling the packets through out the handoff period and the communication need not be hung up till a new IP is given by the DHCP server. Thus MSIP gives better performance than Mobile IP ans SIP, for real-time applications.

Acknowledgments

I express my sincere gratitude towards my guide **Prof. Sridhar Iyer** for his constant help, encouragement and inspiration through out the project work. Without his invaluable guidance, this work would never have been a successful one. I would also like to thank the members of the *Mobile Computing Research Group* at KReSIT, namely Ashish Shinde, Srinath Perur, Raghu, Vijay Rajsinghani, Nitin Salodkar, Vijaya Kumar, Sayed Imranali, Kiran Diwakar, Kalyan Rao, and Kalpesh Patel for their valuable suggestions and helpful discussions. I would also like to thank all my friends at KReSIT, for their support and encouragement.

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IIT Bombay
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Contents

List of Figures

Abbreviations

MN	Mobile Node
CN	Correspondent Node
MIP	Mobile IP
SIP	Session Initiation Protocol
MA	Mobility Agent
HMA	Home Mobility Agent
FMA	Foreign Mobility Agent
DHCP	Dynamic Host Configuration Protocol
BS	Base Station
MH	Mobile Host
CoA	Care of Address
DAD	Duplicate Address Detection

Chapter 1

Introduction

1.1 Introduction to Wireless Networks

In recent times, the wireless networks have become very popular. With the ever increasing use of mobile devices, the demand for host mobility and support for various mobile applications is also growing considerably. Wireless LANs are being deployed on airports, conferences, etc. People have started using portable laptops to access Internet and other resources using wireless networks while moving. For any mobile node, if it wants to take full advantage of this mobility provided, the mobile node should be able to move to any point in the network and still retain all its ongoing communications.

1.2 Challenges of Mobility in Wireless Networks

Multihop wireless networks are faced by challenges not present in the wired networks. Mobility of stations in wireless networks gives rise to issues like route changes, link failures, and need for change of IP addresses. These reasons require changes at various layers of protocol stack.

1.2.1 IP and Routing

The stations in wireless network do not remain in the same subnet due to mobility; hence either their IP addresses need to be changed and/or the packets should be forwarded to them. These requirements have lead to development of mobile IP[1] where the addresses are assigned to mobile hosts dynamically and the packets are appropriately forwarded to them. Session Initiation Protocol[6] handles mobility at the application layer. In

SIP, the node, when moves to a foreign domain, gets a new IP address from the DHCP server, and resumes its communication at the new IP address.

1.2.2 Delay Issues

The cell handoff delay is the period of time between the moment at which the mobile node detects the subnet change, and the time at which it receives the first packet of its ongoing communication in the new subnet. Existing mobility protocols have mainly been designed for network, and application layers, and the majority of studies refer to the inherent mobility support provided by the wireless network. The primary design goal of any scheme, that handles mobility, is to keep the handoff delay as less as possible. If the applications are real-time then this constraint on delay becomes even more crucial, as the real time applications are highly delay-sensitive. Schemes like Mobile IP and SIP exist that handle mobility at network and application layers respectively, but there are several issues that are yet to be resolved in these schemes. The problem with the Mobile IP scheme is the very famous triangular routing problem, and also that all the packets need to be tunneled through the home agent, to the foreign agent and then to the mobile node, leading to much longer delays[1][5] . The problem with SIP is that after the the mobile node moves to a new subnet, it should contact a DHCP server to get a new IP[6] , and then inform the other communicating party about its new IP. The process of getting a new IP from the DHCP server is quite time consuming.

1.3 Related Work

Several schemes have been proposed to handle mobility in IP networks and voice applications. One such scheme is Mobile Voice Over IP, an application-level protocol [8]to support terminal mobility in real-time applications such as voice over IP, on a wireless local area network. Its implementation is based on the ITU-T H.323 protocol boundaries in the network. Mobile Voice over IP (MVOIP), provides a mechanism to maintain a VOIP call even as the underlying network addresses of the hosts engaged in the call need to change.

1.4 Thesis Objective and Scope

This work aims at improving the performance for real-time applications, in terms of delay, by using *MSIP*, a combination of network and application layer mobility management models. MSIP reduces the signaling load and provides fast handoff for ongoing conversations. The main focus of this work is to present the design and architecture of MSIP. The implementation of MSIP in Network Simulator NS-2 is presented, and simulation results are discussed. The scope of this work is limited to IPV4 networks, not to IPV6. This is due to inclusion of *R*oute Optimization in IPV6, which avoids the longer delays due to triangular routing.

1.5 Thesis Outline

The rest of the thesis is outlined in this section. In Second chapter, Mobile IP scheme is discussed in detail. The working of Mobile IP, and problems of using it for real-time applications are also discussed there. Third chapter discusses the operation of Session Initiation Protocol, and describes the drawbacks in SIP. In chapter four, the motivation for MSIP to improve the handoff delay is explained. Fifth chapter explains the design and working of MSIP in detail. Sixth chapter describes the MSIP implementation in NS. The simulation set up and results are discussed in chapter seven. Finally, thesis ends with the conclusion and the scope for the future work in chapter eight.

Chapter 2

Mobile IP

2.1 Overview

Mobile IP is a transparent solution that handles mobility at the network layer. This scheme was designed to solve the mobility problem by allowing the mobile node to use two IP addresses: a fixed home address and a care-of address that changes at each new point of attachment. This solution is transparent because, when two nodes communicate with each other and one of them moves to a new subnet, then the other node is completely unaware of this mobility, and it continues its communication as if nothing has happened. In Mobile IP, it is assumed that every mobile node (MN) has its home network, and a statically allocated IP address on its home network. In Mobile IP, the support for mobility is provided by adding IP tunneling to IP routing. The Mobile IP architecture mainly consists of

1. Mobile Nodes (MN)

2. Correspondent Nodes (CN) : The CN participates in communication with the MN. The CN can be either fixed or a mobile node.

3. Home Agents (HA) : The default router on the home network, HA stores the current locations of all the mobile nodes in its network. It intercepts the packets from CN and tunnels them to the current location of the mobile node.

4. foreign Agents (FA) : The default router on the foreign network, FA receives the tunneled packets from HA and forwards them to the MN in its network.

2.1.1 Main Functions in Mobile IP

Mobile IP consists of three main functions[13],

1. **M**obile agent discovery: When a mobile node is away from home, it wants to find agents so it does not lose access to the Internet. There are two ways of finding agents. The first is by selecting an agent from among those periodically advertised, and the second is by sending out a periodic solicitation until it receives a response from a mobility agent. The mobile node thus gets its care-of-address which may be dynamically assigned or associated with its foreign agent.

2. **R**egistration with home agent: The mobile node registers its care-of-address with its home agent in order to obtain service. The registration process can be performed directly from the mobile node, or relayed by the foreign agent to the home agent, depending on whether the care-of-address was dynamically assigned or associated with its foreign agent. Note that simultaneous registrations with multiple care-of-addresses is possible.

3. **P**acket delivery using IP tunneling: This is the period after the registration process and before the service time expiration, provided that the mobile node stays in the service area. During this period, the mobile node gets forwarded packets from its foreign agent which were originally sent from the mobile node's home agent. Tunneling is the method used to forward the message from home agent to foreign agent and finally to mobile node.

2.2 How Mobile IP Works

The Home agents and the foreign agents will broadcast agent advertisements in their subnets. The mobile node detects that it has moved to a new subnet, when it hears the advertisement from an agent which is not its HA. Then it sends a registration request to its HA through FA with a care-of-address(CoA), which is generally the address of the foreign agent itself. The HA then replies either accepting or denying that request. The CN will be completely unaware of this mobility and will transmit the packets to the mobile nodeś home address. The HA will intercept these packets and performs the IP-in-IP encapsulation and tunnels them to the CoA of the mobile node (FA). The FA then decapsulates the packets and forwards them to the mobile node.

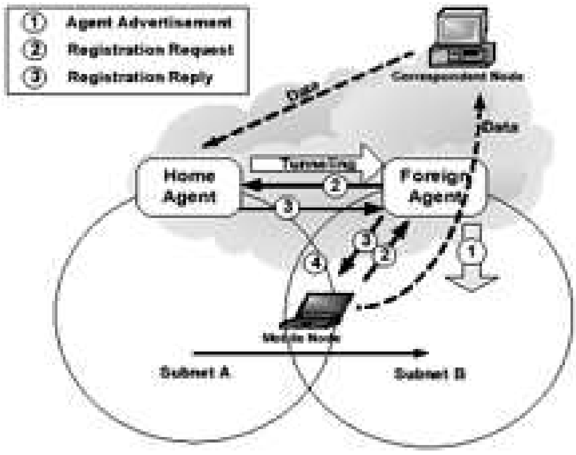

Figure 2.1: Working of Mobile IP

2.3 Problems with Mobile IP

The famous problem with Mobile IP is *triangular routing*. All the packets from CN to MN, should take the longer path from CN to HA, from HA to FA, and then from FA to MN, whereas the packets from MN to CN, go directly to CN. One solution was proposed to solve this problem, *Route Optimization*.

Route Optimization : Route optimization solves the triangular routing problem by using binding updates to inform the correspondent host about the current IP address. However, route optimization has several drawbacks[6] :

1. Route optimization requires changes in the IP stack of the correspondent host, since it must be able to encapsulate IP packets, and store care-of addresses of the foreign agent or mobile host.

2. Only the home agent may send binding updates to correspondent hosts. This means that there will be an extra delay before the correspondent host finds out where to send the packets, during which the old foreign agent must forward the packets to the correct location.

3. The mobile host needs to rely on the old foreign agent forwarding packets to its new foreign agent until the correspondent host has got the binding update. There is no requirement saying that the foreign agent must do so.

Because of the requirements that are put on the correspondent hosts, it cannot be expected that route optimization will be widely employed in a near future. Moreover, the home and foreign agent can become bottlenecks since they must handle the tunnels for a possibly large number of mobile hosts.

2.4 Summary

Mobile IP provides transparent mobility which is needed to keep TCP connections alive when the user is moving. This is because, Mobile IP does not require the change of IP address with the change in point of attachment of the node, thus Mobile IP is the right choice for such applications. This scheme is not suitable for real-time applications because of long tunneling delays.

Chapter 3

Session Initiation Protocol : SIP

3.1 Overview

The Session Initiation Protocol (SIP) is the application-layer solution to handle mobility. This protocol is used for establishing and tearing down multimedia sessions[6]. SIP is a control-signaling protocol and has been standardized by IETF for the invitation to multicast conferences and Internet telephone calls.

There are certain functions or characteristics that a signaling protocol should be capable of performing. They are

1. User Location function
2. Session or call establishment function
3. Session negotiation function
4. Call Participant Management function
5. Feature Invocation

In SIP, every user is addressed using an email-like address "user@host", where user is a user name or phone number and host is a domain name or numerical address. The SIP architecture consists of Registrar, User Agents, Proxy Server, Redirect Server.

The session is active when the first INVITE Request is issued and the call is established. The session terminates when a BYE Request is issued. It also supports mobility using proxy or redirecting servers to reach the Callee. SIP is also independent of transport protocol. SIP is a client-server protocol where the client generates the SIP Request messages while the server receives the messages and send responses. There are three

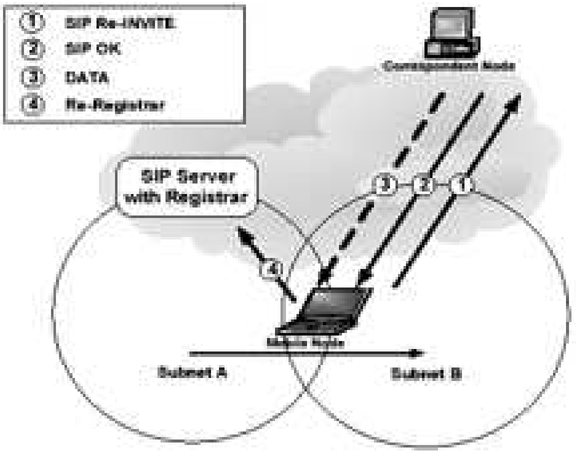

Figure 3.1: Working of SIP

kinds of servers in use with SIP. They are namely the Proxy servers, Redirect Servers and the User Agent Servers (UAS). When a Proxy Server receives a Request message, it acts like a hop server or a router and forwards the Request message to the next Proxy Server or the User Agent Server at the receiving end-system. As for a Redirect Server, when it receives a Request message, it returns an IP address of the Callee's User Agent Sever so that the Caller, which is a client, would have to resend his/ her request to the UAS.

SIP Methods

1. *Register* Every user will register itself with the Registrar. In turn Registrar binds a permanent address to the current location. These registrations are not permanent, these will expire after certain period of time thus requiring periodic registrations.

2. *Invite* This method is generally invoked at the beginning of a communication. Session description messages will be exchanged during this period. This function could also be invoked when the state of the session has changed or has to be changed.

3. *Options* Returns users capabilities to the requester.

4. *ACK* This is only used during the session initiation process, at the end of three way handshake process.

5. *Cancel* Terminates an incomplete session.

6. *BYE* Closes an open session.

The working of SIP in presence of proxy and redirect servers is explained in the figure 3.2.

3.2 How SIP Works

3.2.1 SIP with Proxy Server

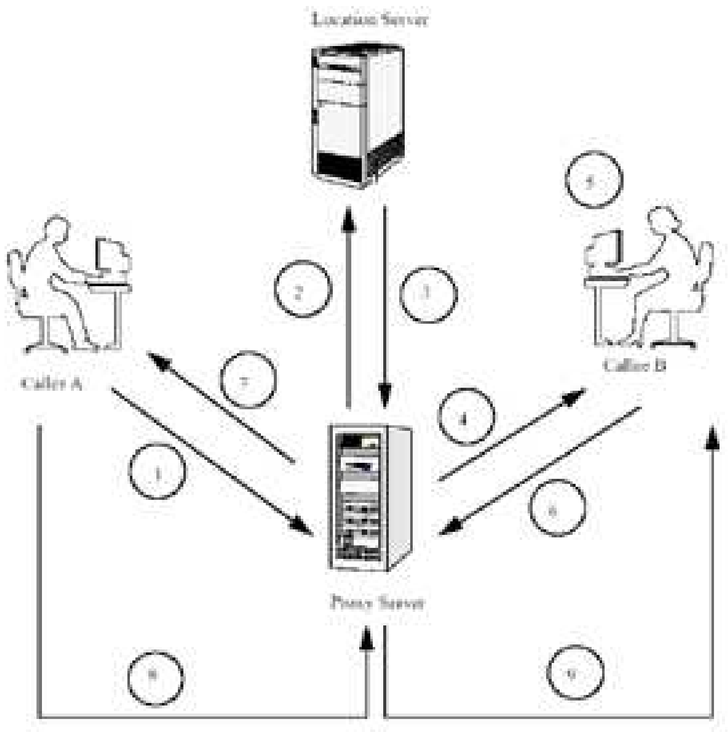

Figure 3.2: SIP in presence of a Proxy Server

The above figure shows the working of SIP in the presence of a proxy server. The steps are as follows:

1. SIP Client at Caller A site tries to translate the domain abc.com to an IP address where the server can be found. Once this is complete, Caller A issues the INVITE request.

2. Caller A, also known as the User Agent Client (UAC), creates a session with an INVITE request for Callee B, sip:B@abc.com. This request is forwarded to the

SIP Proxy Server of B. This proxy server at B s site looks for Callee B at the abc.com domain in its Location Server, usually a DNS. The Location Server or DNS then returns the reply that Callee B is known locally.

3. The Proxy Server then re forwards the request message to Callee B, the host location which is also the User Agent Server (UAS).

4. Callee B accepts the request and a response is returned through the Proxy Server, and sends 200 message, which means OK.

5. The Proxy Server then forwards the message to Caller A to indicate that the request is a success.

6. Caller A or UAC, returns an ACK message to the Proxy Server for the UAS of Callee B to mean that Caller A acknowledges the last response from Callee B.

7. Finally, this forwarded message reaches Callee B or the UAS. Once this ACK is received, the call is ready to begin. The call will only end when a BYE request is issued, which is then followed by the 200 OK message and finally the ACK message back for 200 OK.

3.2.2 SIP with Redirect Server

The above figure 3.3 shows the working of SIP in the presence of a Redirect server. The steps followed are :

1. Caller A, creates a session with an INVITE request for Callee B, sip:B@abc.com. This request is forwarded to the SIP Redirect Server of B.

2. This redirect server at B s site looks for Callee B at the abc.com domain in its Location Server, usually a DNS.

3. The Location Server or DNS then returns the reply that Callee B is logged in as BDomain.abc.com. The redirect server knows this through a static configuration, database entry or dynamic binding set up by Callee B with a SIP REGISTER message.

4. Caller A or UAC sends an ACK message to the Redirect Server that it received its response message.

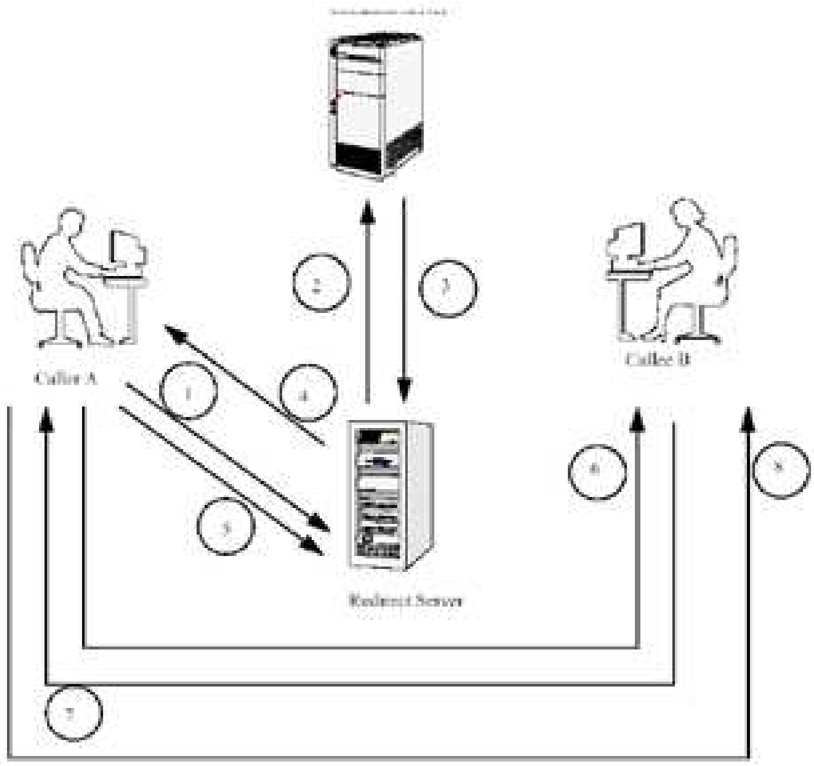

Figure 3.3: SIP in presence of a Redirect Server

5. Caller A now resends the INVITE request for Callee B, sip:B@BDomain.abc.com.

6. This time, Callee B or UAS, returns an ACK message to the UAC of Caller A to mean that Callee B acknowledges the last response from Caller A and the request is a success.

7. Finally, Caller A or UAC sends an ACK message to Callee B that it receives its 200 response message for request success.

The SIP Redirect server is preferred to the Proxy server, because the redirect server just relays the current location of the user, whereas the proxy server participates in the entire signaling session and thus increasing the load on the server. If the load on the server, that is the number of users it had to handle is less, then proxy server and redirect server may perform equally well. If the load is very high, then the redirect server gives a better performance than that of a proxy server.

3.3 Problems with SIP

1. SIP is not preferred for TCP applications.

2. The process of getting a new IP from the DHCP server involves a significant amount of delay (typically 1-2 sec). The messages involved in this process is explained in the figure 3.4.

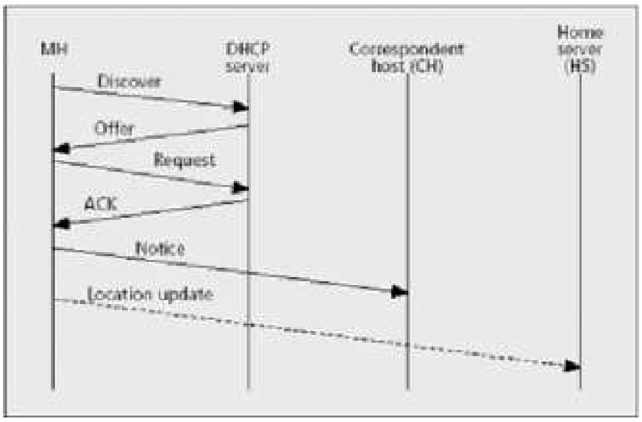

Figure 3.4: Delay involved in DHCP

3. The communication resumes more quickly in Mobile IP than SIP. In Mobile IP, this communication starts as soon as the mobile node hears the Agent advertisement in the new subnet, which is not the case with SIP.

3.4 Summary

Session Initiation Protocol solves many of the problems of Mobile IP, like triangular routing, longer tunneling delays. The performance of SIP also depends on the DHCP implementation. The delay involved in obtaining IP address from DHCP server is the main bottle neck in the working of SIP. So, in DHCP implementation if the duplicate address resolution *DAD* is done in the background after assigning the IP address to the node, the performance improves a lot.

Chapter 4

Motivation for MSIP

4.1 Motivation

The real-time applications are highly delay-sensitive. In order to support mobility for such applications, the main challenge would be to provide seamless connectivity for the ongoing communications, with as lesser delay as possible. The motivation for MSIP is to reduce the long delay problems associated with the existing schemes, Mobile IP and SIP and improve th performance for real-time applications. These problems are

1. In Mobile IP all the packets for the entire session, the period in which the mobile node (MN) is in foreign network, should travel through the longer path that from correspondent node (CN) to home agent (HA) then to foreign agent (FA) and then to MN, thus resulting in longer delays, not desirable for real time applications.

2. In SIP, the process of obtaining a new IP address from the DHCP server, when a handoff occurs, involves lot of delay.

In spite of the above mentioned problems associated with Mobile IP and SIP, these schemes have certain advantages over each other. In Mobile IP, after the node moves to the new subnet, the communication resumes more quickly. This is because, the agents will broadcast the advertisements periodically with available CoA's. As soon as the mobile node listens this advertisement it sends a registration to its home agent and restarts the communication. In SIP, the packets need not be tunneled through the longer path after the handoff, but can be sent directly to the node's new location. The main aim of MSIP is to reduce the handoff delay for real time applications by integrating these good features from both Mobile IP and SIP. In this new scheme we introduce a agent called **Mobility Agent** which is an integration of SIP Redirect Server and Mobile IP Home Agent.

4.2 Problem Statement

The real-time applications are highly delay sensitive. The handoff latency should be as less as possible for such applications. Existing schemes Mobile IP and SIP have delay problems in them, explained in the above paragraph. The main aim of this work is to reduce the handoff disruption time and give better performance than Mobile IP and SIP, for real-time applications.

Chapter 5

MSIP Proposed Solution

5.1 Overview

In this report, we propose MSIP, an integrated scheme, combination of Mobile IP and SIP, that results in a very less handoff latency time. This combination of network and application layer mobility management models reduces the signaling load and provides fast handoff for ongoing conversations.

5.2 How MSIP Works

If the Correspondent Node(CN) wants to communicate with the Mobile node(MN), they start the communication using the SIP call establishment mechanism. If the MN moves to a new subnet, with the session being still active, then, the MN switches to Mobile IP scheme and receives the packets in the new subnet with tunneling being done by its *Home Mobility Agent*. The tunneling is required only until the MN receives a new IP address from the DHCP server in the new subnet. As soon as the MN gets the new IP, it sends the SIP REINVITE message, with the same Call-ID, to CN informing it about its new IP. From then on, tunneling is stopped, and the packets will be sent to the current location of the MN directly.

The detailed working of our new scheme is explained in figure 5.1

If Correspondent Node (CN) wants to communicate with Mobile Node (MN), then

1. CN sends an INVITE message to the Mobility Agent (MA) in MN's home domain.

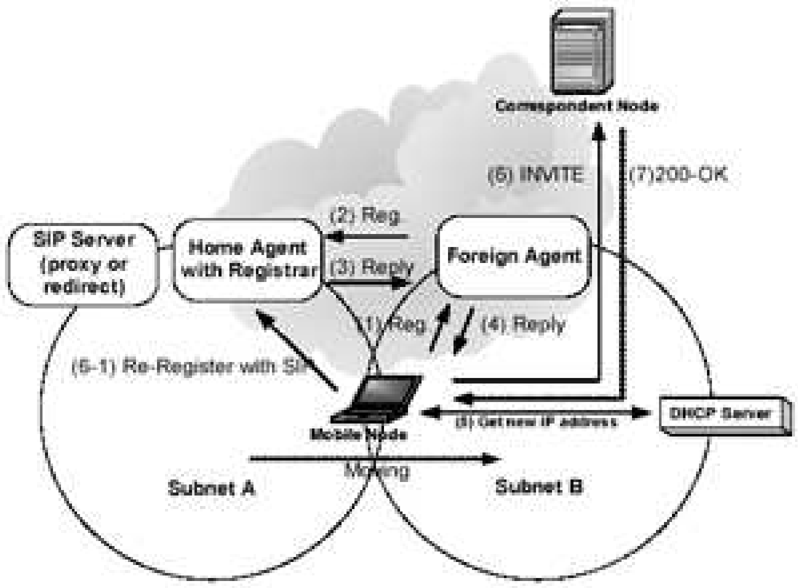

Figure 5.1: Working of MSIP

2. The Mobility Agent performs a DNS lookup for the current location of the MN, and gives this information to the CN.

3. CN sends the INVITE message directly to the current location of MN, and the call is established.

Meanwhile, if the MN moves to a new subnet during the ongoing communication with CN, resulting in a handoff, then,

4. MN sends a Registration request to its Home Mobility Agent (HMA) through Foreign Mobility Agent (FMA).

5. MN gets the reply from its HMA accepting its registration.

6. Soon after this, the transmission resumes, as in normal Mobile IP scheme.

7. The packets will be tunneled to the MN through FMA by the HMA.

8. Mean while, the MN contacts DHCP server to get a new IP address.

9. After it gets a new IP, it sends the SIP REINVITE message to the Correspondent Node (CN), with its new location.

10. From then on, the packets will be sent directly to the MNś new current location without tunneling.

11. The MN sends a SIP RE-REGISTER message to its HMA, so that any new connections in future can be correctly redirected to the MN s current location.

The messages involved in the handoff process are given in figure 5.2 .

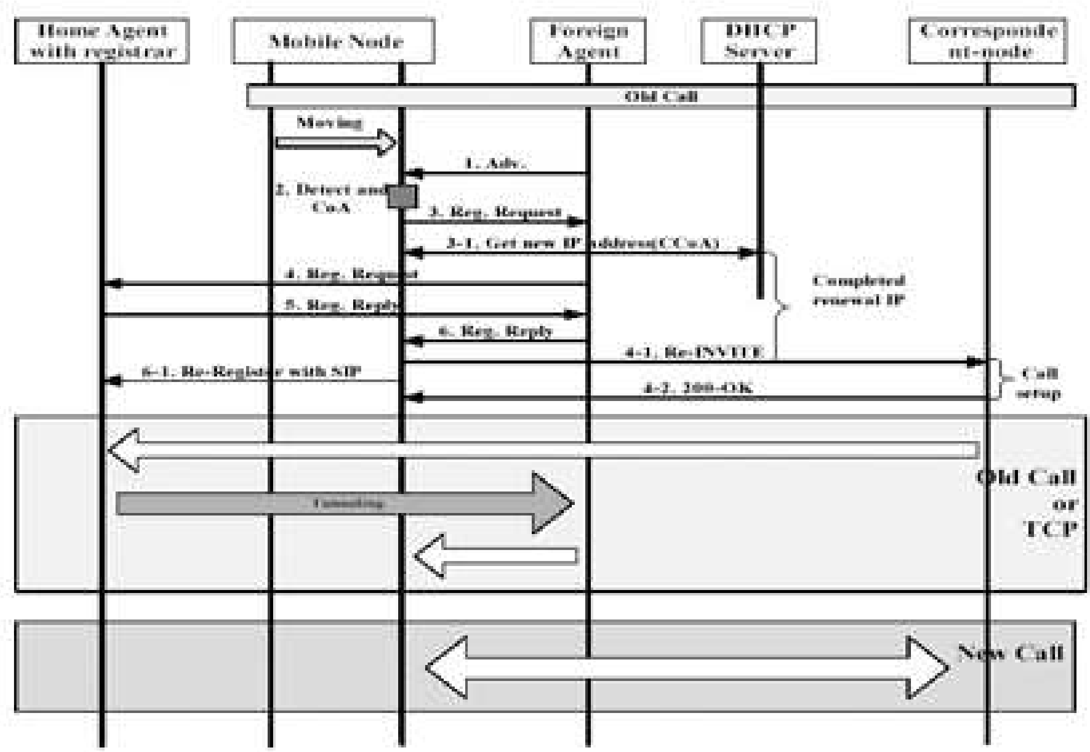

Figure 5.2: Message flow during a Handoff

5.3 Advantages

Our scheme, MSIP, overcomes the above mentioned problems associated with Mobile IP and SIP. The communication resumes as soon as the node hears the agent advertisement in the foreign network as it happens in Mobile IP, without waiting until the node obtains the new IP address from the DHCP server asin SIP scheme. Also, the packets need not be tunneled through the longer path, for the entire session as in Mobile IP.

5.4 Summary

The gain in performance when MSIP is used, also depends on the DHCP implementation. MSIP gives better performance for real-time applications. But still, for long-lived TCP connections like Telnet, FTP, Mobile IP is preferred as it doesn't involve the need for change in IP addresses, with change in point of attachment.

Chapter 6

MSIP Implementation in NS

6.1 Node Architecture in NS

6.1.1 Mobile IP

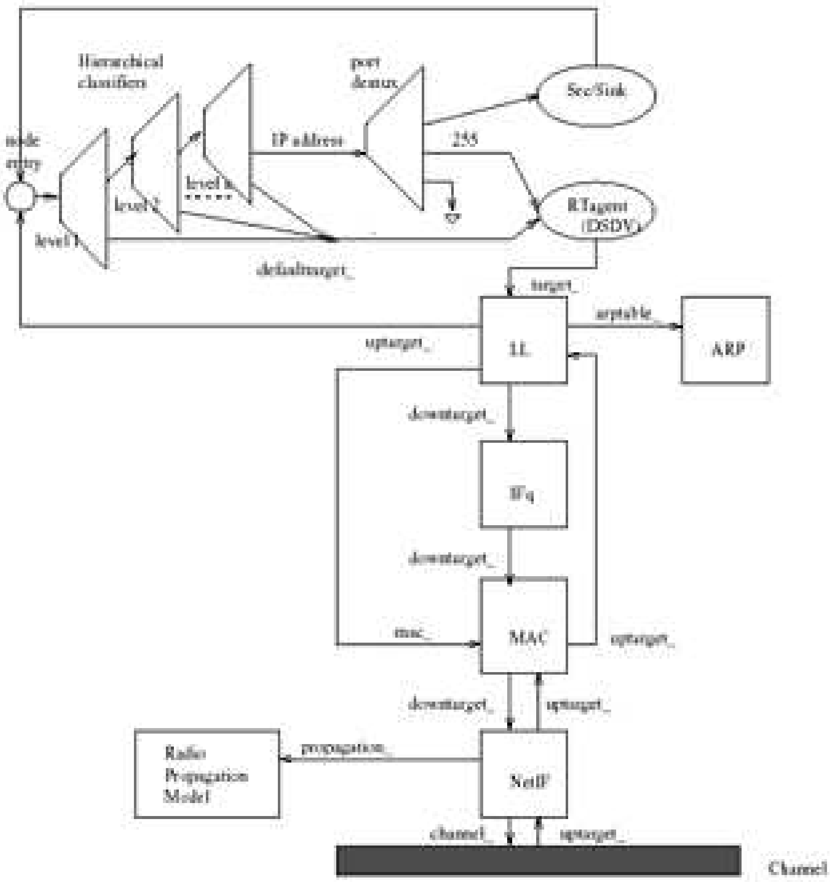

Figure 6.1: Node Architecture in Mobile IP

The HA and FA nodes are defined as MobileNode/MIPBS having a registering agent (regagent_) that sends beacon out to the mobile nodes, sets up encapsulator and decapsulator, as required and replies to solicitations from MHs. The MH nodes are defined as MobileNode/MIPMH which too have a regagent_ that receives and responds to beacons and sends out solicitations to HA or FAs. The MobileNode/MIPMH node is very similar to this except for the fact that it doesn't have any encapsulator or decapsulator.

The MobileNode/MIPBS node routinely broadcasts beacon or advertisement messages out to MHs. A solicitation from a mobile node generates an ad that is send directly to the requesting MH. The address of the base-station sending out beacon is heard by MH and is used as the COA (care-of-address) of the MH. Thus as the MH moves from its native to foreign domains, its COA changes. Upon receiving reg_request (as reply to ads) from a mobile host the base-station checks to see if it is the HA for the MH. If not, it sets up its decapsulator and forwards the reg_request towards the HA of the MH. In case the base-station is the HA for the requesting MH but the COA doesn't match its own, it sets up an encapsulator and sends reg-request-reply back to the COA (address of the FA) who has forwarded the reg_request to it. so now all packets destined to the MH reaching the HA would be tunneled through the encapsulator which encapsulates the IP packet header with a IPinIP header, now destined to the COA instead of MH. The FA s decapsulator receives this packet, removes the encapsulation and sends it to the MH. If the COA matches that of the HA, it just removes the encapsulator it might have set up (when its mobile host was roaming into foreign networks) and sends the reply directly back to the MH, as the MH have now returned to its native domain.

6.1.2 Session Initiation Protocol

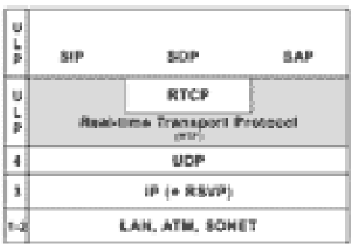

Figure 6.2: SIP in Protocol Stack

The SIP handles the mobility at the application layer. The SIP patch has been downloaded from the National Institute of Standards Technology website. The patch is applied on the NS2.1b9a version. The downloaded SIP patch was for wired scenario, and mobility was not supported. In order to support mobility, DHCP agent had to be included in NS. DHCP Agent would be responsible for allotting new IP addresses, when requested for, to the mobile nodes roaming in the foreign domains. The pseudo code for DHCP implementation is given in figure 6.3.

6.2 MSIP Node Architecture

The main idea here, is to use MSIP for only real-time applications. At the link layer, we can use Mobility Policy Table, which decides whether to use Mobile IP or SIP based upon the type of application. The Mobility policy table specifies how the packets should be sent and received for each traffic flow matching certain characteristics. If RTP header is present in the packet, MSIP scheme will be used or else Mobile IP normal scheme will be used.

The implementation of *MSIP* involved changes in *MIPMHAgent* class and *SIPAgent* class.

The pseudo code for MSIP implementation in NS is shown in figure **??**. In our experiments, we are testing for sip traffic, thus MSIP scheme will be used. When the mobile node moves into the foreign network, the node hears an advertisement from FA, so it initiates the handoff process, by sending a DHCP_REQUEST to the DHCP server. The server then allocates a new IP to the mobile node by sending DHCP_REPLY. The node updates its routing table about the changed IP, and propagates this information in the network by initiating startUp() in the DSDVAgent class. Then the SIPAgent is then informed about this IP change. It updates the data structure *ua_state_* . The SIPAgent sends a invite message with new IP as its contact address, to the CN. The CN then transmits the packets directly to the new IP address of the MN, and thus tunneling would be stopped.

```
MIPMHAgent::recv(Packet* p, Handler *h)
{
        hdr_mip *miph = hdr_mip::access(p);
        hdr_ip* iph = hdr_ip::access(p);

        if (miph->type ==MIPT_ADS) {
                if(iph->saddr() != ha_) {
                        miph->type = DHCP_REQUEST;
                        iph->daddr() = DHCP_Server;
                        send(p, 0);
                }
        }
        if(miph->type == DHCP_REPLY) {
                Change myNode_Addr = new_IP;
                update_My_RouteTable();
                // Propagate the change of IP in the network
                DSDVAgent->startUp();
        }
}
DHCPAgent::recv(Packet *p, Handler *h)
{
        hdr_mip *miph = hdr_mip::access(p);
        hdr_ip* iph = hdr_ip::access(p);

        if(pkt->type ==         DHCP_REQUEST) {
                //Allocate a new IP address
                new_IP = allocate_NewIP();
                miph->type = DHCP_REPLY;
                send(p,0);
        }
}
```


Figure 6.3: DHCP implementation in NS

```
MIPMHAgent::recv(Packet* p, Handler *h)
{
        hdr_mip *miph = hdr_mip::access(p);
        hdr_ip* iph = hdr_ip::access(p);

        if (miph->type ==MIPT_ADS) {
                if(iph->saddr() != ha_) {
                        miph->type = DHCP_REQUEST;
                        iph->daddr() = DHCP_Server;
                        send(p, 0);
                        informSIPAgent(new_IP);
                }
        }
        if(miph->type == DHCP_REPLY) {
                Change myNode_Addr = new_IP;
                update_My_RouteTable();
                // Propogate the change of IP in the network
                DSDVAgent->startUp();
        }
}

SIPAgent::informSIPAgent(new_IP) {

        // Change the Agent\'s node address in the
        //       data structure maintained by SIP Agent
        ua_state_.caller_location.node_ = miph->newDHCPAddress;
        ((SIPAgent*) node_->SIP_AGENT)->addr()=miph->newDHCPAddress;
        node_->address_ = miph->newDHCPAddress;
        addr()=miph->newDHCPAddress;
        sendInviteToCN(newIP);
}
```


Figure 6.4: MSIP in NS

Chapter 7

Simulation and Results

7.1 Simulation SetUp

This chapter explains the simulation of MSIP. The simulations are done using the public domain simulator NS-2. The SIP patch[9] has been applied on the NS2.19ba version. The main aim of the experiments is to measure the handoff latency time and compare with the existing schemes. The following are the assumptions that were made during the simulation.

1. The DHCP delay, time taken for obtaining the a new IP address from the DHCP server, is taken as 1.76 sec. This value is taken after practically measuring the DHCP delay on the subnet of our department, KReSIT, IIT Bombay.

2. The DHCP server is present on the base station node itself.

3. The authentication delays are not considered in our simulations.

4. The transmission range of the nodes is 250m.

5. Mobile IPV4 is used.

7.1.1 Aim of the experiments

The main aim of the simulation is to measure the handoff disruption time. To compare the performance of MSIP for real time applications with Mobile IP and SIP, in terms of

1. The time taken for the first packet to reach the destination after the handoff,

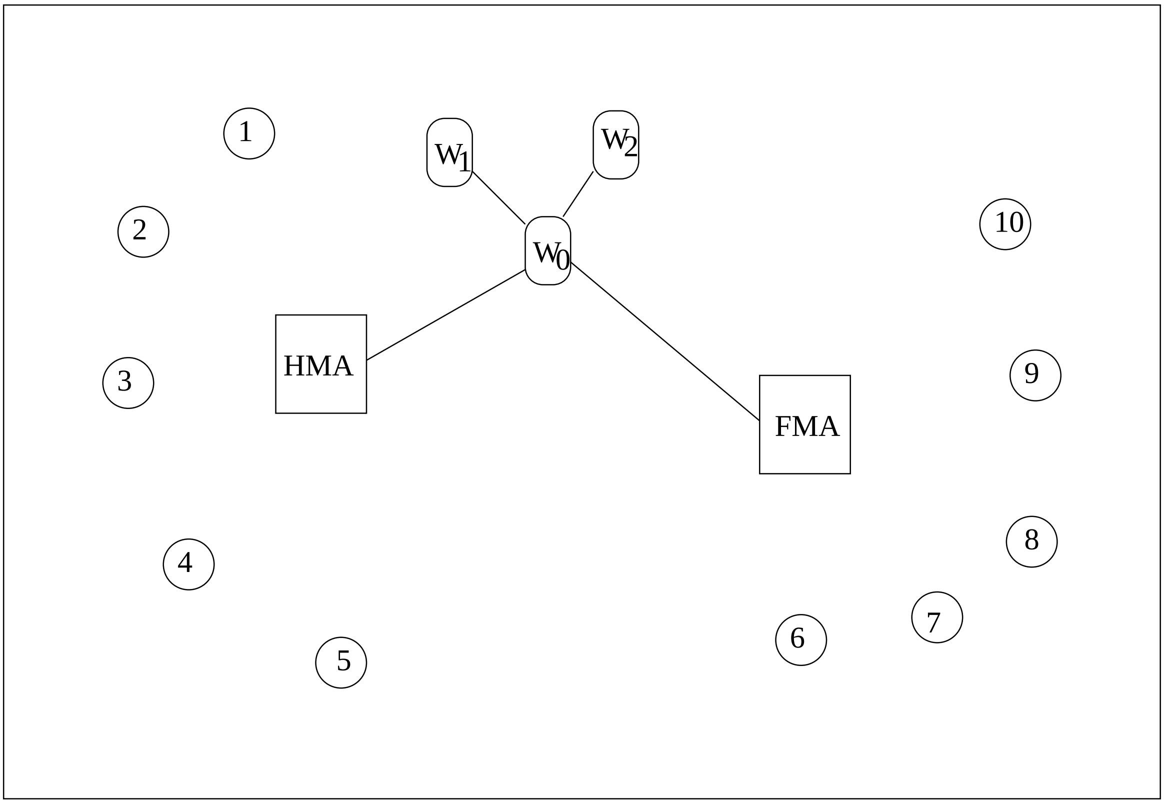

Figure 7.1: Simple Topology

2. The overall throughput, that is the number of packets transmitted in a given amount of time.

7.2 Scenario and Results

In the scenario taken for the simulation, all the nodes are placed in an area of size, 670m x 670m There are 3 domains, the home domain, the foreign domain and the wired domain. There is 5 wireless nodes each, in the home and the foreign domain. The nodes are numbered **1**, **2**, **3**, **4**, **5**, **6**, **7**, **8**, **9**, and **10** The hierarchical addressing is switched on. There are 3 wired nodes as well in this topology namely **W0**, **W1** and **W2**, to resemble the real life situation where the network is a combination of both wired and wireless nodes. The topology is shown in the figure 7.1 .

In this scenario, the Home Mobility agent is on the node HMA which acts as both Mobile IPś home agent and SIPś redirect server. One sip agent is on node **1** in the home domain, and other sip agent is on the node **6**. Here RTP traffic is set up on the SIP connection, with a packet size of 500 bytes. The mobile node **1** moves to the foreign

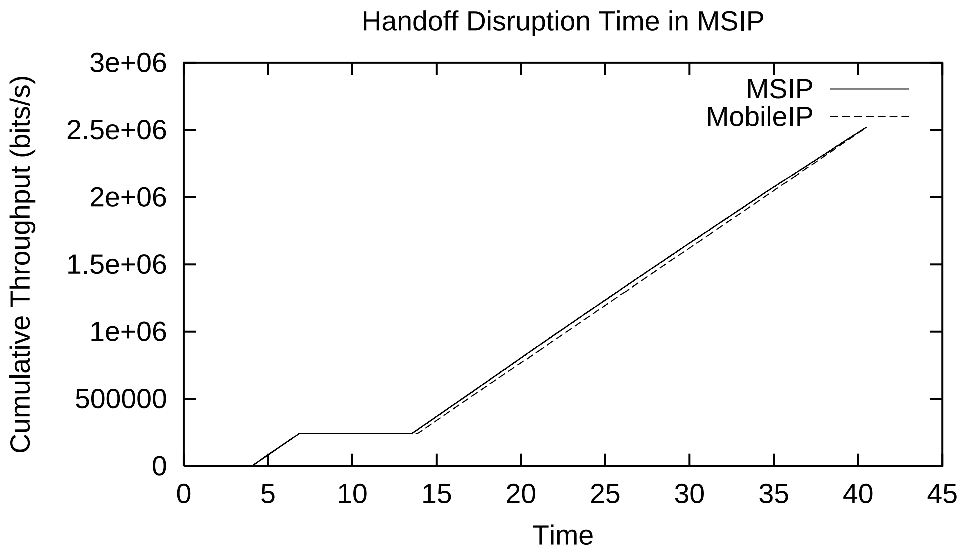

Figure 7.2: Delay in MIP Vs MSIP

domain in the middle of their communication, where **FMA** is the *mobility agent*.

The figure 7.2 describes the comparison of throughput achieved, with Mobile IP and MSIP. From the above graph we could observe that the handoff disruption time is same in both the schemes Mobile IP and MSIP. The disruption period is 4.5 sec. The total number of bytes received in MSIP is also greater than that of Mobile IP within the same duration of time. Actually this gain in number of bytes received depends upon the distance between the HA and MN. If this distance is more then with MSIP, we can receive more data with same amount of handoff disruption delay.

7.3 Analysis on Results

The performance of Mobile IP and MSIP depends heavily on the distance between the Home Agent (HN) and the Mobile Node, and also on the distance between the Correspondent Node (CN) and the Mobile Node. All simulations are performed using the network topology shown in Figure. In this simulations because we are interested in packet loss during call in progress and handoff latency, the CN acts as a RTP traffic source, producing fixed length packets (500 bytes). The MN acts as a sink. The results observed are given in the graph 7.3.

From the graph 7.3, its clear that the disruption during handoff with SIP-mobility becomes shorter than that of Mobile IP approach as the distance between the MN

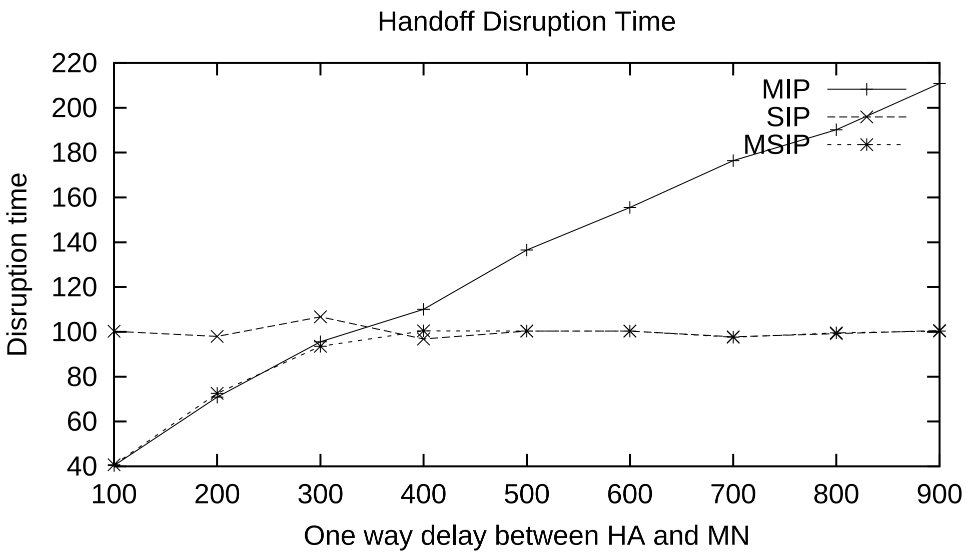

Figure 7.3: Disruption Time vs. Delay

and its home network increases, since the SIP-mobility handoff mainly depends on the distance between CN and MN. However, the disruption time in MIP increases directly proportional to the distance between HA and MN, since the MN must register its IP address with HA whenever it moves a new subnet. Finally, the integrated mobility management approach MSIP shows the most effective results since it benefits both from MIP, when the MN to home network delays are small, and SIP when updating the CN is faster than updating the home network agents.

Chapter 8

Future Work and Conclusions

MSIP performs better than Mobile IP and SIP. The better performance here means a lesser handoff delay. In our hybrid scheme it takes lesser time, for the first packet (after the handoff), of the ongoing communication, to reach the MN. There are some issues yet to be resolved in this scheme, like authentication. Experiments need to be done, to measure these authentication delays, required for exchange of extra control messages (Extra, because, in our scheme, control messages of both Mobile IP and SIP are involved).

The delay involved in obtaining a new IP address from a DHCP server mainly depends on its implementation. The duration of the hand-off period would be significantly reduced if the mobile host could immediately begin using the new IP address while verifying in the background that it is not in use by any other host. In addition, the accuracy and speed of determining when handoff is necessary could be increased

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