"Voice Call Establishment and Conterence Call over IP, a Hands-on SIP-Based Experiment"

# Part I. Establish and analyze a successful call between 2 SIP clients:



Client 2000 dialing 2010



Incoming call on 2010 from 2000

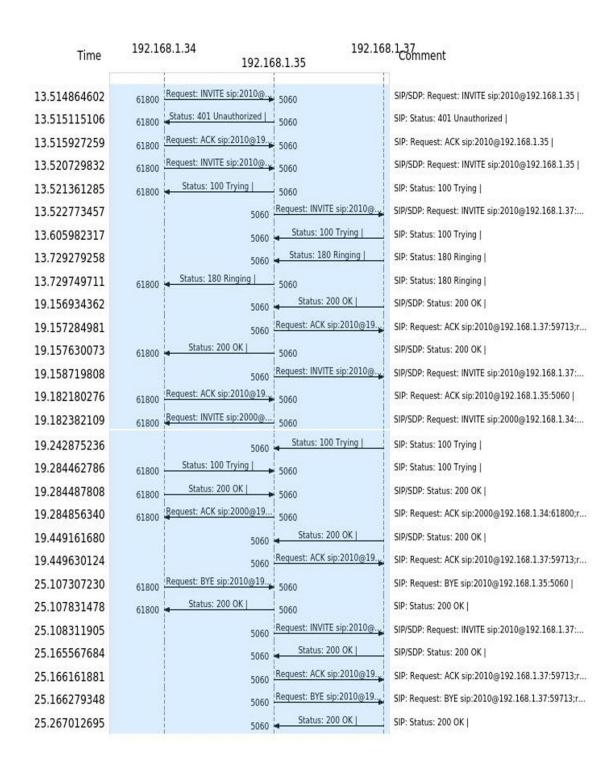


Call established between 2000 and 2010.

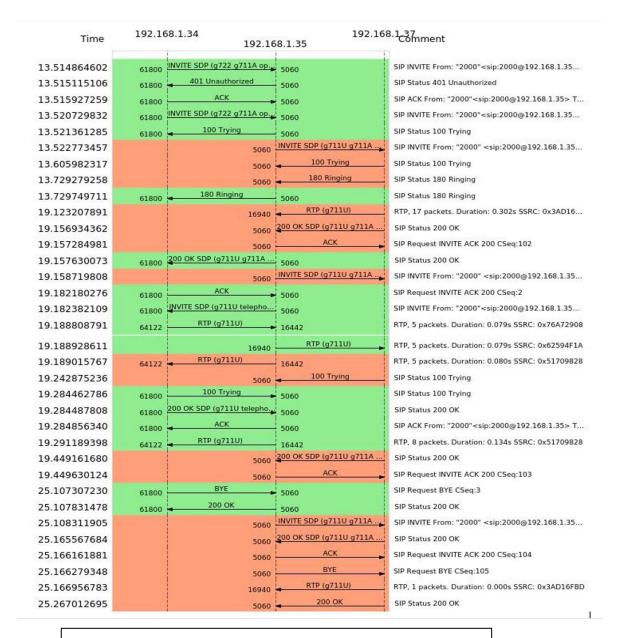
# <u>Part A)</u> Client 2000 and Client 2010 register with the PBX server whose destination IP address is 192.168.1.35 and is shown below:



## Process of call establishment and call termination is as follows below:



<u>Part B)</u> Logical Flow-chart of call establishment and call termination between Client 2000 and 2010 is as follows below:



Logical Flow-chart of Call Establishment and Call Termination

## Summary:

- Client 2000 and 2010 send Register packet to register with the proxy server with their IP address.
   The server responds back with an OK message.
- Once the registration process is complete, call establishment process starts with Client 2000 sending an Invite message to the proxy server.
- The proxy server then sends an invite to the callee client 2010 and send a Trying message to 2000. If the client 2010 is available, it sends a Trying message to 2000 and then an OK message followed by a Ringing Message.
- Client 2000 requests a connection by sending an ACK message to connect the call and start the media exchange. Once the media exchange is complete, any one of the clients sends a call termination message i.e. BYE which is acknowledged by all parties with an OK message.

```
▼ Session Initiation Protocol (100)
  Status-Line: SIP/2.0 100 Trying
        Status-Code: 100
        [Resent Packet: False]
        [Request Frame: 37]
        [Response Time (ms): 161]

    Message Header

     Via: SIP/2.0/UDP 192.168.1.35:5060; branch=z9hG4bK26db9cc8
     To: <sip:2010@192.168.1.37:64201;rinstance=2da60827ea55b9f7>
        SIP to address: sip:2010@192.168.1.37:64201;rinstance=2da60827ea55b9f7
     ▼ From: "2000" <sip:2000@192.168.1.35>;tag=as2d67d2d2
          SIP Display info: "2000"
        ▶ SIP from address: sip:2000@192.168.1.35
          SIP from tag: as2d67d2d2
        Call-ID: 6c01db684033133656ff026452232f18@192.168.1.35:5060
     CSeq: 102 INVITE
       Content-Length: 0
```

# SIP Trying packet

SIP Packet: Trying packet being sent from client 2000 requesting to connect a call with Client 2010 with IP address 192.168.1.37. The call ID of this packet is as seen in the image.

# Part C)

IP Address of Client 2000: 192.168.1.34

```
▶ Internet Protocol Version 4, Src: 192.168.1.34, Dst: 192.168.1.35
▶ User Datagram Protocol, Src Port: 55414, Dst Port: 5060
▼ Session Initiation Protocol (INVITE)
▶ Request-Line: INVITE sip:2010@192.168.1.35 SIP/2.0
▼ Message Header
▶ Via: SIP/2.0/UDP 192.168.1.34:55414; branch=z9hG4bK-524287-1---60 Max-Forwards: 70
▶ Contact: <sip:2000@192.168.1.34:55414; rinstance=90f6c38a272976db: ▶ To: <sip:2010@192.168.1.35>
▶ From: "2000"<sip:2000@192.168.1.35>; tag=d6f7fe25
Call-ID: 99142MDBhZGU5Mjc3Nzg4ZDc0Yjk4MDg2NzFlZjZiYTU0ZjM
▶ CSeq: 1 INVITE
```

## IP Address of Client 2010: 192.168.1.37

#### MAC address of PBX server: 08:00:27:09:01:5c

## Part II.1: Call on Hold:









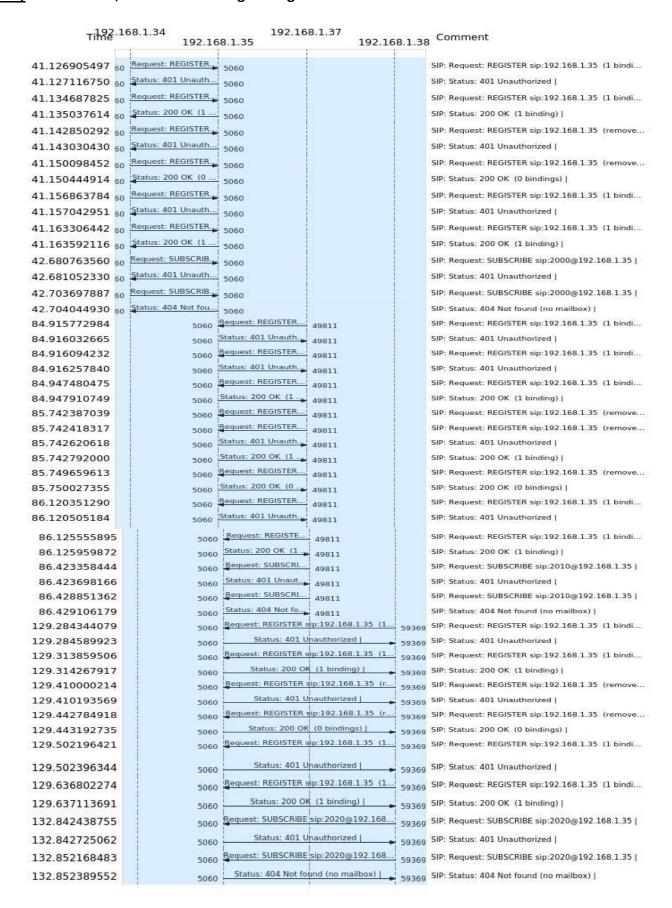
Call established between 2000 and 2010

Call established between 2000 and 2020 while 2010 is put on hold

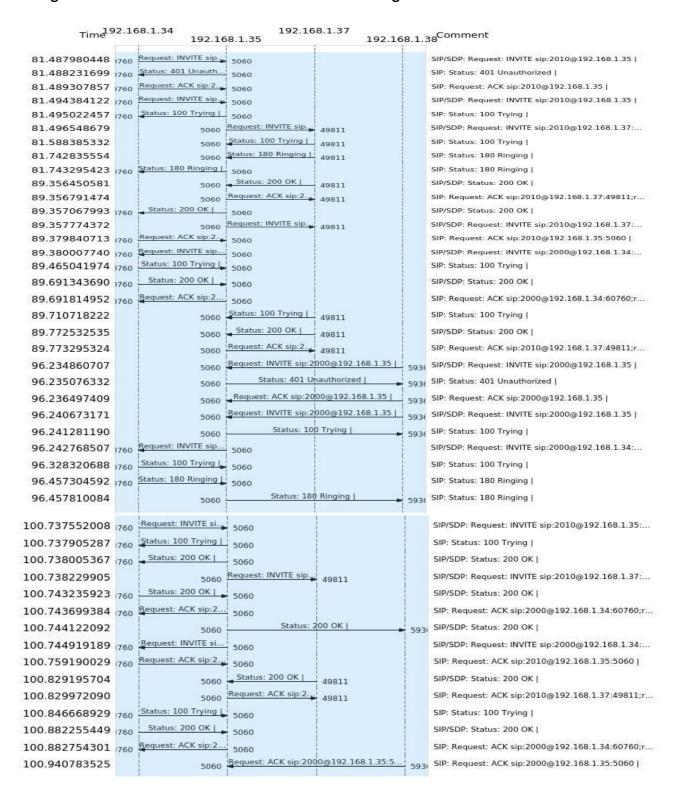
Call ended between 2000 and 2020 and taking 2010 off hold

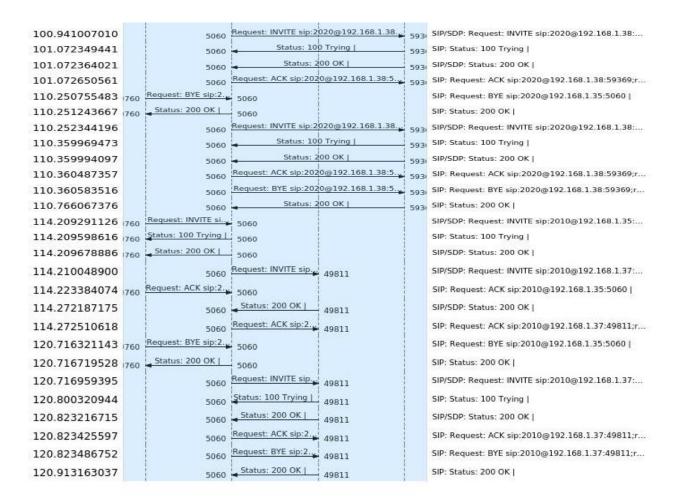
Call resumed between 2000 and 2010

### Part A) Devices 2000, 2010 and 2020 registering with the PBX server is as shown below:

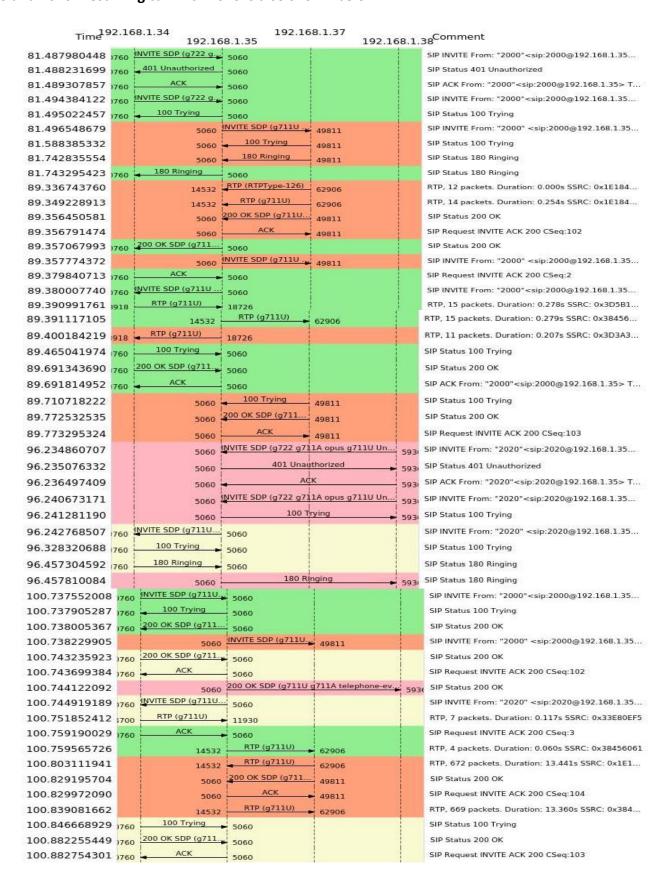


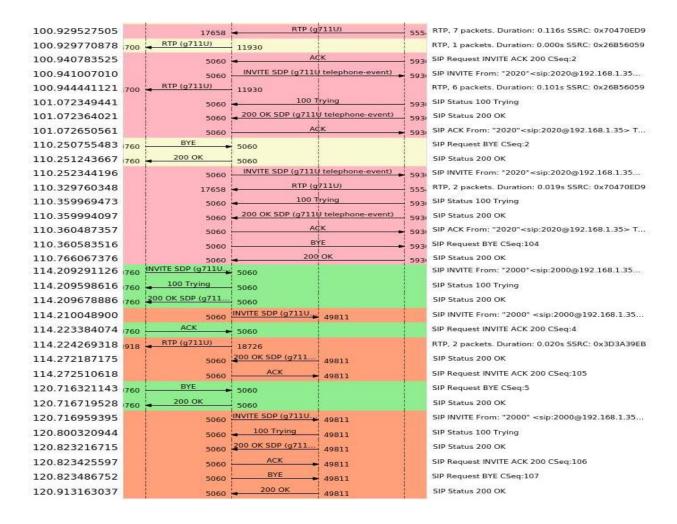
# Process of call establishment between 2000 and 2010, and the process of putting 2010 on hold while establishing a call between 2000 and 2020. Process of resuming the call with 2010 is also shown below:





# <u>Part B)</u> Logical Flow-chart of call establishment and putting 2010 on hold while establishing call between 2000 and 2020. Resuming call with 2010 is also shown below:





#### Summary:

- Client 2000, 2010 and 2020 send Register packet to register with the proxy server with their IP address. The server responds with OK message.
- Once the registration process is complete, call establishment process starts with Client 2000 sending an Invite message to the proxy server.
- The proxy server then sends an invite to the callee client 2010 and send a Trying message to 2000. If the client 2010 is available, it sends a Trying message to 2000 and then an OK message followed by a Ringing Message.
- Client 2000 requests a connection by sending an ACK message to connect the call and start the media exchange.
- Now client 2020 wants to call 2000 and sends an Invite packet to proxy server. The same process of call
  establishment follows and once client 2000 picks the call, media exchange starts between 2000 and
  2020. Whilst this call is being established, client 2010 is put on hold by sending an Update message.
  After media exchange is complete between 2000 and 2020, call is terminated between them.
- Client 2000 then sends an Update message to client 2010 to resume the call. Once the media exchange is complete, any one of the clients sends a call termination message i.e. BYE which is acknowledged by all parties with an OK message.

```
▼ Session Initiation Protocol (180)

▼ Status-Line: SIP/2.0 180 Ringing

Status-Code: 180

[Resent Packet: False]
[Response Time (ms): 161]

▼ Message Header

▶ Via: SIP/2.0/UDP 192.168.1.35:5060; branch=z9hG4bK26db9cc8

▼ Contact: <sip:2010@192.168.1.37:64201; rinstance=2da60827ea55b9f7>

▶ Contact URI: sip:2010@192.168.1.37:64201; rinstance=2da60827ea55b9f7

▼ To: "2010"<sip:2010@192.168.1.37:64201; rinstance=2da60827ea55b9f7>; tag=21126f78

SIP Display info: "2010"

▶ SIP to address: sip:2010@192.168.1.37:64201; rinstance=2da60827ea55b9f7

SIP to tag: 21126f78

▼ From: "2000" <sip:2000@192.168.1.35>; tag=as2d67d2d2

SIP Display info: "2000"

▶ SIP from address: sip:2000@192.168.1.35

SIP from tag: as2d67d2d2

Call-ID: 6c01db684033133656ff026452232f18@192.168.1.35:5060

▶ CSeq: 102 INVITE

User-Agent: X-Lite release 5.6.1 stamp 99142

Allow-Events: talk, hold

Content-Length: 0
```

# **SIP Ringing packet**

We can see a Ringing packet sent from client 2000 to Client 2010 via the server with IP address 192.168.1.37 to respond to a call setup request. The call ID of the packet is as seen in the image.

# Part C)

IP Address of Client 2000: 192.168.1.34

```
▶ Internet Protocol Version 4, Src: 192.168.1.34, Dst: 192.168.1.35
▶ User Datagram Protocol, Src Port: 55414, Dst Port: 5060
▼ Session Initiation Protocol (INVITE)
▶ Request-Line: INVITE sip:2010@192.168.1.35 SIP/2.0
▼ Message Header
▶ Via: SIP/2.0/UDP 192.168.1.34:55414; branch=z9hG4bK-524287-1---60 Max-Forwards: 70
▶ Contact: <sip:2000@192.168.1.34:55414; rinstance=90f6c38a272976db:
▶ To: <sip:2010@192.168.1.35>
▶ From: "2000"<sip:2000@192.168.1.35>; tag=d6f7fe25
Call-ID: 99142MDBhZGU5Mjc3Nzg4ZDc0Yjk4MDg2NzFlZjZiYTU0ZjM
▶ CSeq: 1 INVITE
```

#### IP Address of Client 2010: 192.168.1.37

#### IP Address of Client 2020: 192.168.1.38

#### MAC address of PBX server: 08:00:27:09:01:5c

```
Ethernet II, Src: IntelCor_a2:f4:a4 (c0:b6:f9:a2:f4:a4), Dst: PcsCo
    Destination: PcsCompu 09:01:5c (08:00:27:09:01:5c)
    Source: IntelCor_a2:f4:a4 (c0:b6:f9:a2:f4:a4)
    Type: IPv4 (0x0800)
    Internet Protocol Version 4, Src: 192.168.1.34, Dst: 192.168.1.35
    User Datagram Protocol, Src Port: 55414, Dst Port: 5060
    Session Initiation Protocol (INVITE)
    Request-Line: INVITE sip:2010@192.168.1.35 SIP/2.0
    Message Header
    Via: SIP/2.0/UDP 192.168.1.34:55414; branch=z9hG4bK-524287-1--Max-Forwards: 70
    Contact: <sip:2000@192.168.1.34:55414; rinstance=90f6c38a27297
    To: <sip:2010@192.168.1.35>; tag=c65aad2a
    Call-ID: 991420GNjNTE5NjlmZDMZYTFiNjQzOTY2MDBjN2ZiMDUwYjY
```

# Part II.2: Call Conferencing



Call established between 2000 and 2010



Call established between 2000 and 2020 while 2010 is put on hold



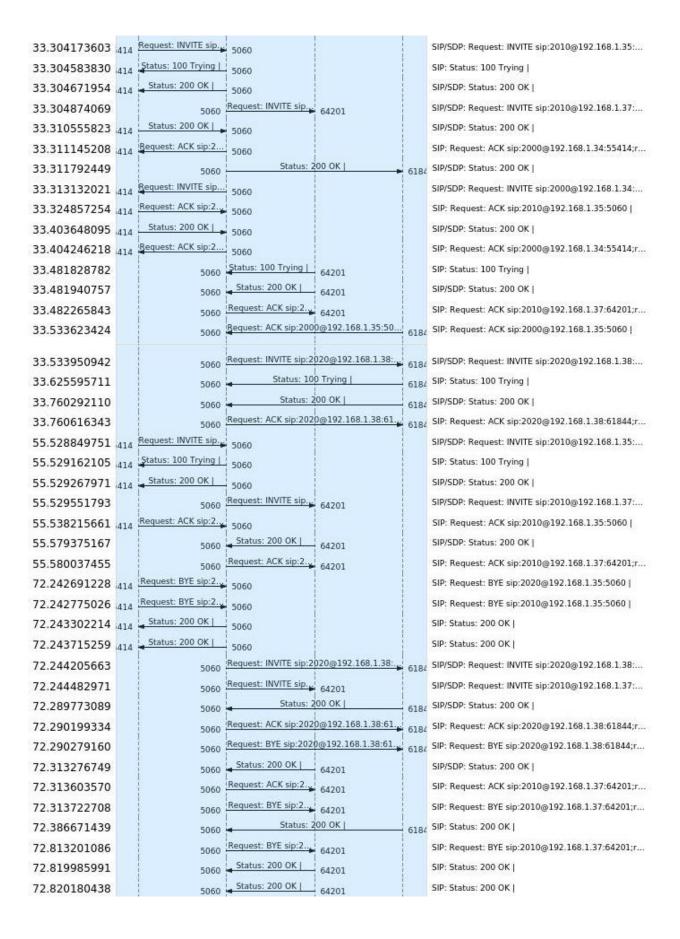
Call conferenced between 2000, 2010 and 2020.

# Part A) Devices 2000, 2010 and 2020 registering with the PBX server is as shown below:

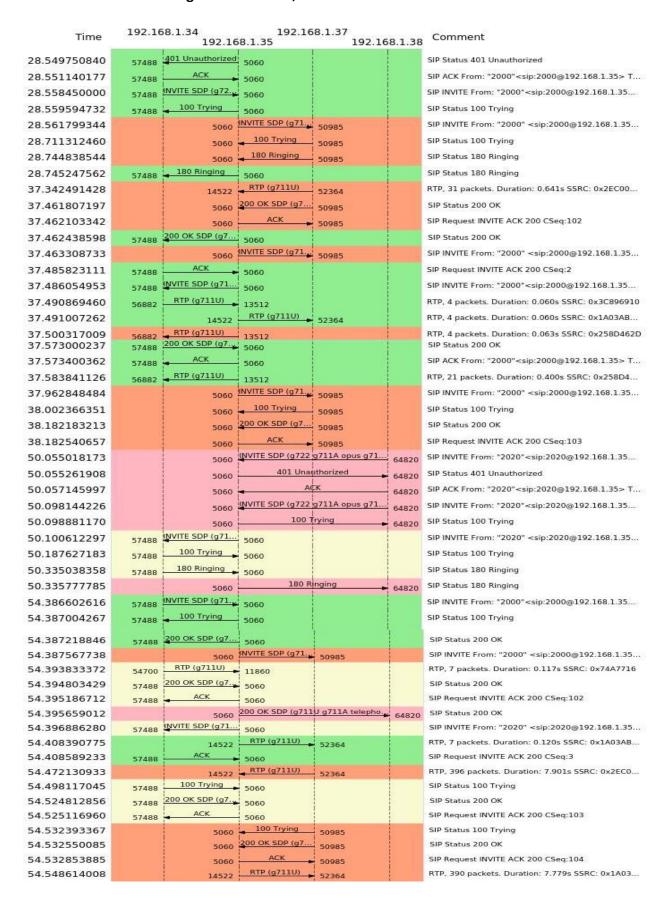


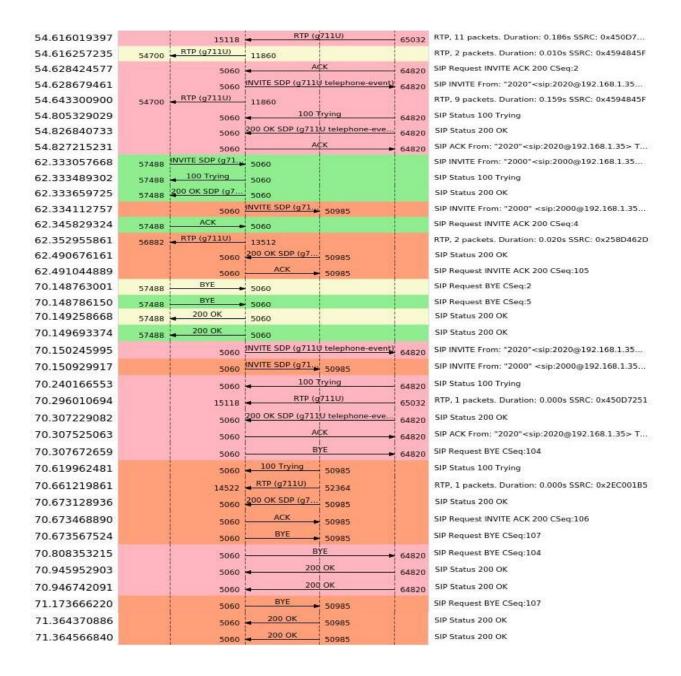
Process of call establishment between 2000 and 2010, and the process of putting 2010 on hold while establishing a call between 2000 and 2020. Process of call conferencing between 2000, 2010 and 2020 is shown below:





# <u>Part B)</u> Logical Flow-chart of call establishment and putting 2010 on hold while establishing call between 2000 and 2020. Call conferencing between 2000, 2010 and 2020 is also shown below:





### **Summary:**

- Client 2000, 2010 and 2020 send Register packet to register with the proxy server with their IP address. The server responds with OK message.
- Once the registration process is complete, call establishment process starts with Client 2000 sending an Invite message to the proxy server.
- The proxy server then sends an invite to the callee client 2010 and send a Trying message to 2000. If the client 2010 is available, it sends a Trying message to 2000 and then an OK message followed by a Ringing Message.
- Client 2000 requests a connection by sending an ACK message to connect the call and start the media exchange.
- Now client 2020 wants to call 2000 and sends an Invite packet to proxy server. The same process of call
  establishment follows and once client 2000 picks the call, media exchange starts between 2000 and
  2020.

- In order to add client 2010 to the ongoing call, proxy server merges the call by sending an INVITE message to 2010. Media exchange takes place between 2000, 2010 and 2020.
- Once the media exchange is complete, any one of the clients sends a call termination message i.e. BYE which is acknowledged by all parties with an OK message and the call is terminated.

```
Session Initiation Protocol (INVITE)
Request-Line: INVITE sip:2010@192.168.1.35 SIP/2.0

    Message Header

  Via: SIP/2.0/UDP 192.168.1.34:55414; branch=z9hG4bK-524287-1---60d9a9058b8
    Max-Forwards: 70
  Contact: <sip:2000@192.168.1.34:55414;rinstance=90f6c38a272976db>
  To: <sip:2010@192.168.1.35>
  From: "2000"<sip:2000@192.168.1.35>;tag=d6f7fe25
     Call-ID: 99142MDBhZGU5Mjc3Nzg4ZDc0Yjk4MDg2NzFlZjZiYTU0ZjM
  CSeq: 1 INVITE
     Allow: OPTIONS, SUBSCRIBE, NOTIFY, INVITE, ACK, CANCEL, BYE, REFER, INFO,
     Content-Type: application/sdp
     Supported: replaces
     User-Agent: X-Lite release 5.6.1 stamp 99142
     Content-Length: 334
Message Body
```

# SIP Invite packet

➤ We can see an Invite packet sent from client 2000 to server with IP address 192.168.1.35 requesting to connect a call with Client 2010. The call ID is as seen in the image. There are two parts to a SIP packet: message header and message body.

#### Part C)

IP Address of Client 2000: 192.168.1.34

```
▶ Internet Protocol Version 4, Src: 192.168.1.34, Dst: 192.168.1.35
▶ User Datagram Protocol, Src Port: 55414, Dst Port: 5060
▼ Session Initiation Protocol (INVITE)
▶ Request-Line: INVITE sip:2010@192.168.1.35 SIP/2.0
▼ Message Header
▶ Via: SIP/2.0/UDP 192.168.1.34:55414;branch=z9hG4bK-524287-1---60 Max-Forwards: 70
▶ Contact: <sip:2000@192.168.1.34:55414;rinstance=90f6c38a272976db: ► To: <sip:2010@192.168.1.35>
➤ From: "2000"<sip:2000@192.168.1.35>;tag=d6f7fe25
Call-ID: 99142MDBhZGU5Mjc3Nzg4ZDc0Yjk4MDg2NzFlZjZiYTU0ZjM
▶ CSeq: 1 INVITE
```

```
Internet Protocol Version 4, Src: 192.168.1.37, Dst: 192.168.1.35

User Datagram Protocol, Src Port: 64201, Dst Port: 5060

Session Initiation Protocol (INVITE)

Request-Line: INVITE sip:2000@192.168.1.35 SIP/2.0

Message Header

Via: SIP/2.0/UDP 192.168.1.37:64201; branch=z9hG4bK-524287-1---11
    Max-Forwards: 70

Contact: <sip:2010@192.168.1.37:64201; rinstance=2da60827ea55b9f7

To: <sip:2000@192.168.1.35>

From: "2010"<sip:2010@192.168.1.35>; tag=c296202c
    Call-ID: 99142MmM50DVkZDE5ZjFkOWRiYzFjNjA4YTMxMzRkN2I1NWU
```

#### IP Address of Client 2020: 192.168.1.38

#### MAC address of PBX server: 08:00:27:09:01:5c