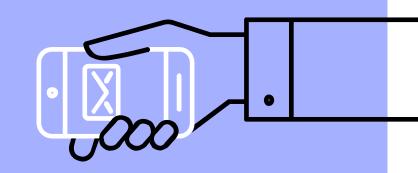
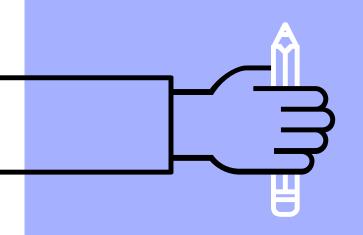


IMS + VoLTE Overview





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Aug 2018

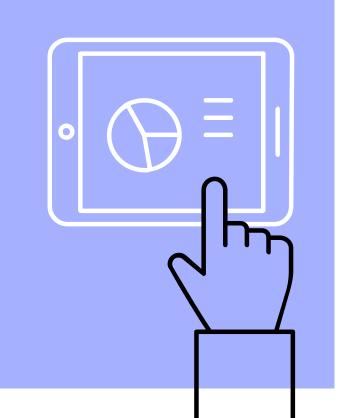
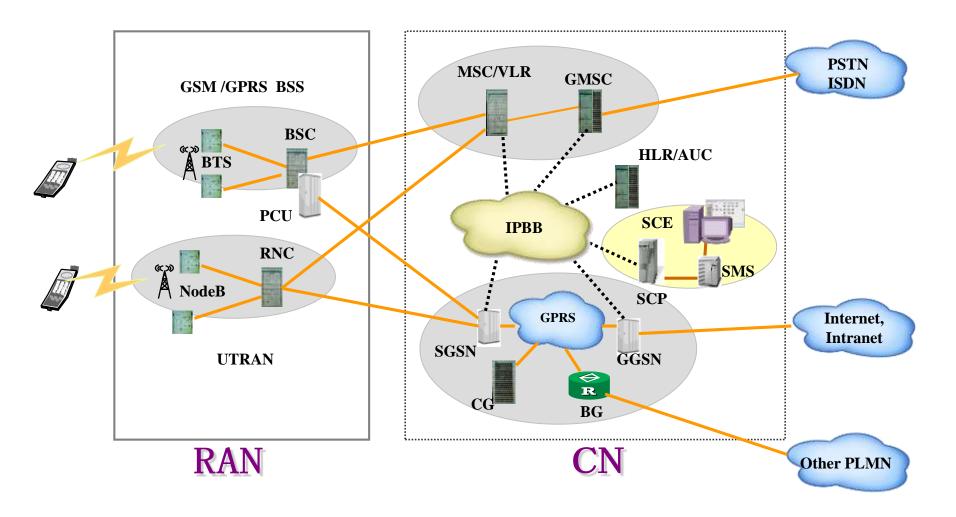


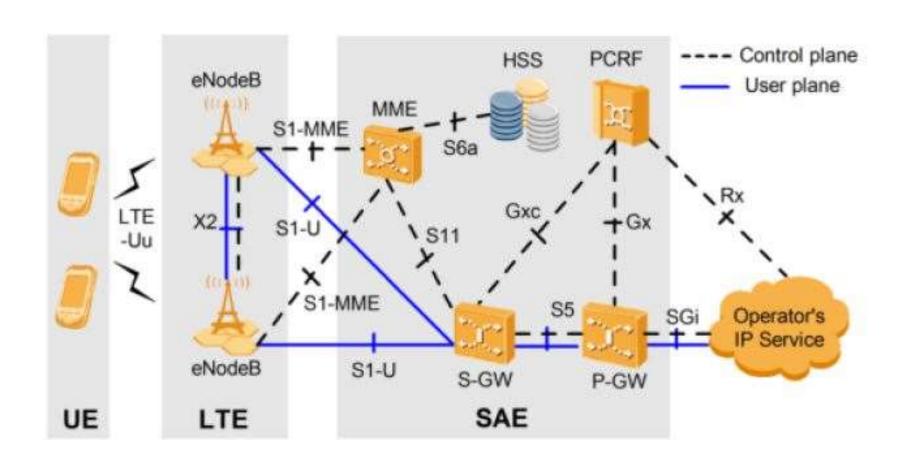
Table of Contents

- 2G/3G/4G Overview
- What is IMS/VoLTE/ViLTE?
- IMS / History & Motivation
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- IMS Reference Points & Protocols
- IMS Basic Signaling
- Volte Call Flow
- CSFB (Circuit Switch Fall Back)
- SRVCC (Single Radio Voice Call Continuity)
- Question & Answer
- Conclusion
- Exam

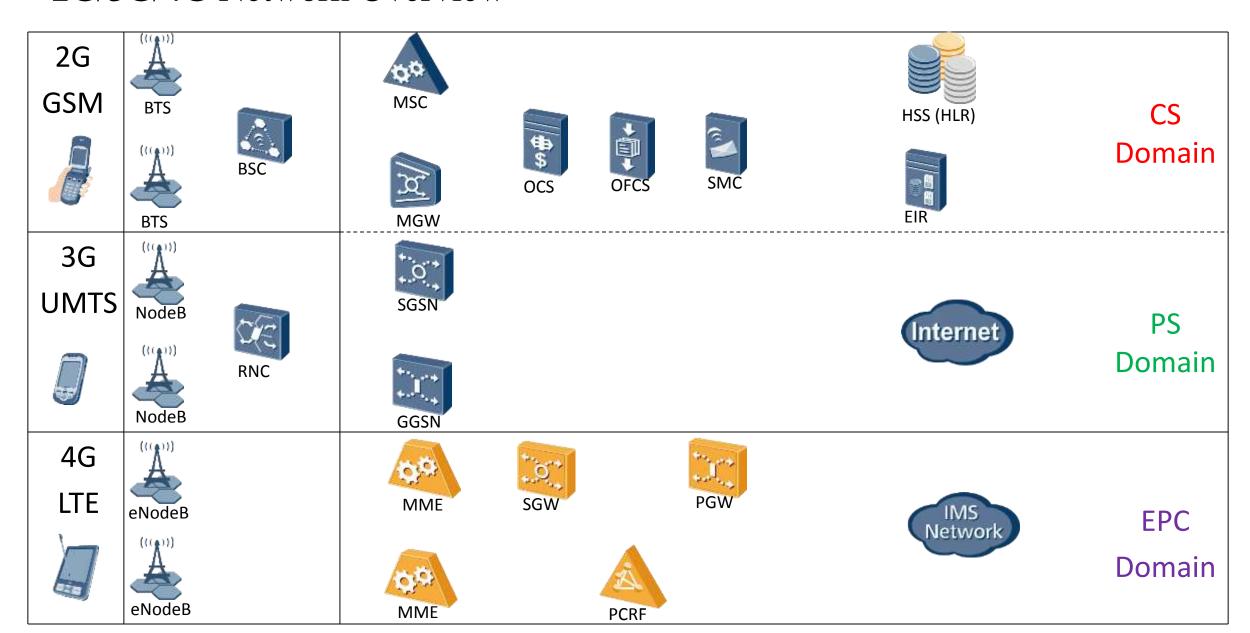
■ 2G/3G Overview



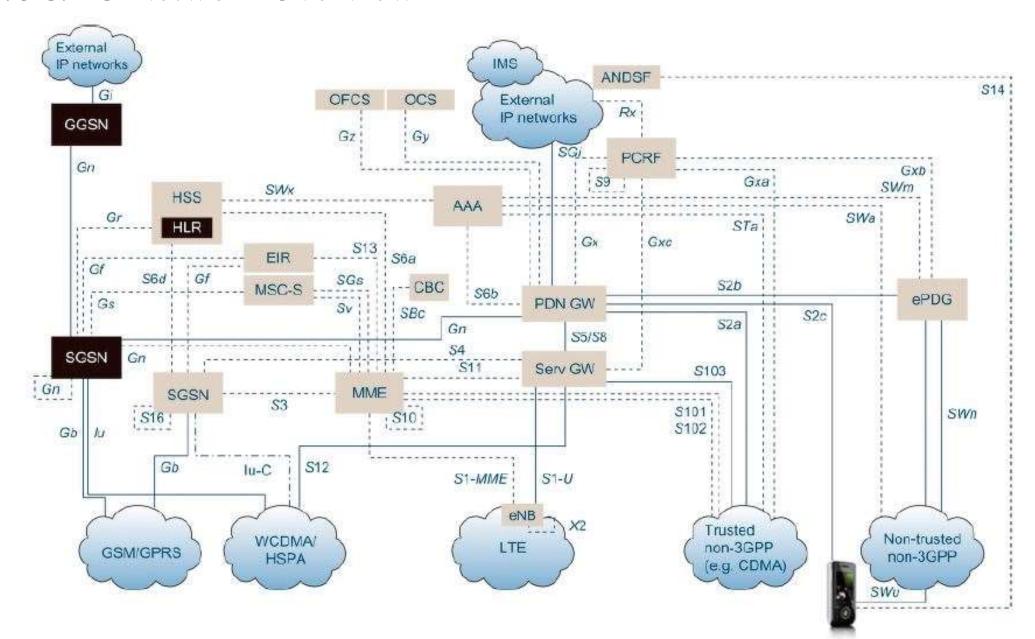
■ 4G/LTE Network Topology



■ 2G/3G/4G Network Overview



■ 2G/3G/4G Network Overview



■ What is IMS?

IMS is IP Multimedia Subsystem

- Base on IP bear network
- Use SIP protocol as core session control protocol
- Support access-independent and thus provide more competitive service packages

Multiple Access Modes

Mobile Network:

WiMax

LTE/ SAE

GSM/ WCDMA/

CDMA/TD-SCDMA

Fixed Network:

LAN, WLAN, xDSL

Multimedia Service

Session Service:

Voice and Video Call

Conference

Message

Non-Session Service:

IPTV、Media、Web

■ IMS Motivation / All IP Convergent

Legacy Core Network Arch.

- Vertical network, overlap in hardware, service and database
- Separated user and service

Mobile

Database

Charging

Service

Control

Bearer

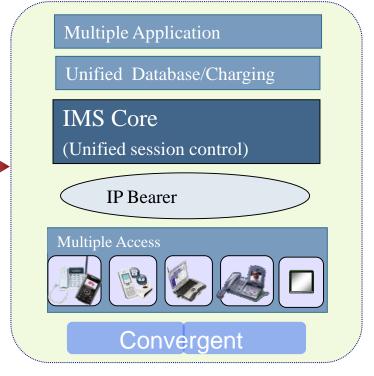
Network

Difficult to decrease OPEX/CAPEX

Fixed Network Database Charging Service Control Bearer Overlapping Cost

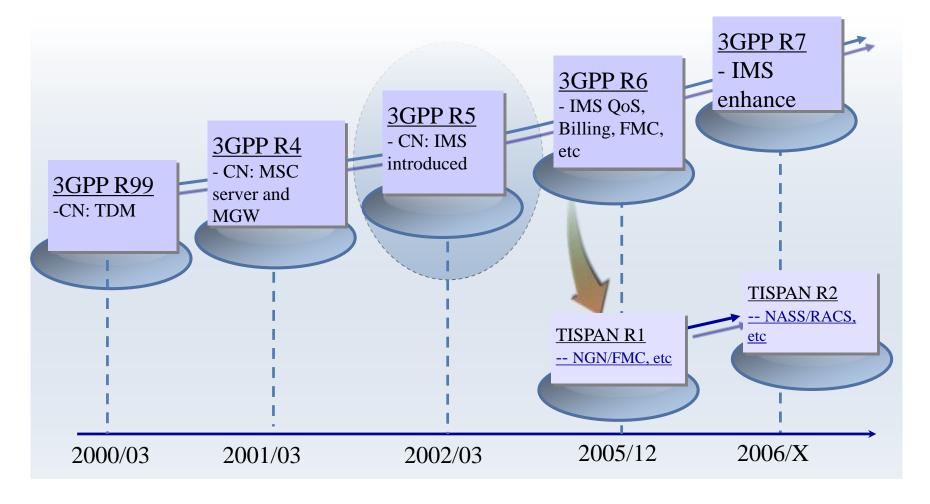
Converged Multi-service Arch.

- Horizontal network, converge fixed, mobile and data service network
- Unified database and service
- Access independent



History of IMS

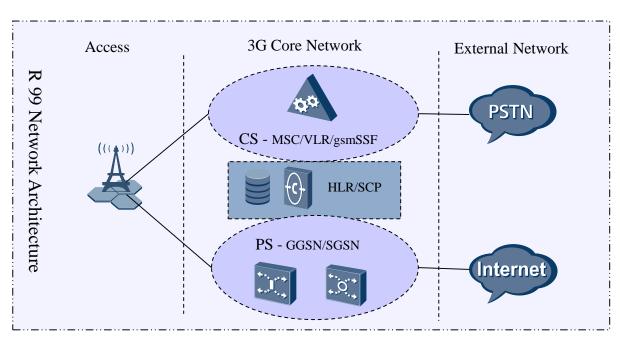
The IMS is introduced as part of 3GPP specifications at the R5 stage, act as the subsystem of PS domain.

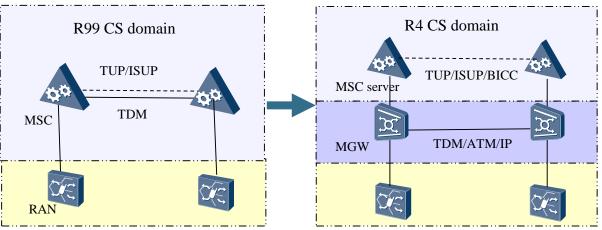


■ Network Architecture of 3GPP R99 and R4

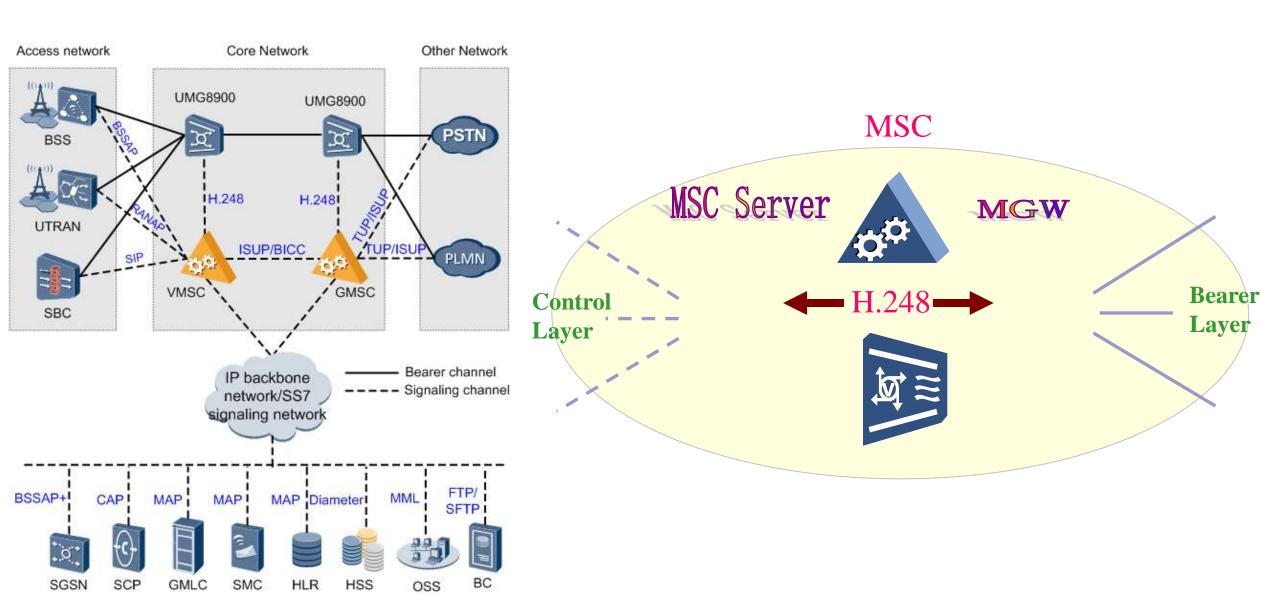
In R99, 3G Core Network is separated into CS and PS Domain.

In R4, MSC divided into MSC server and MGW, call control is separated with the media bearer function.

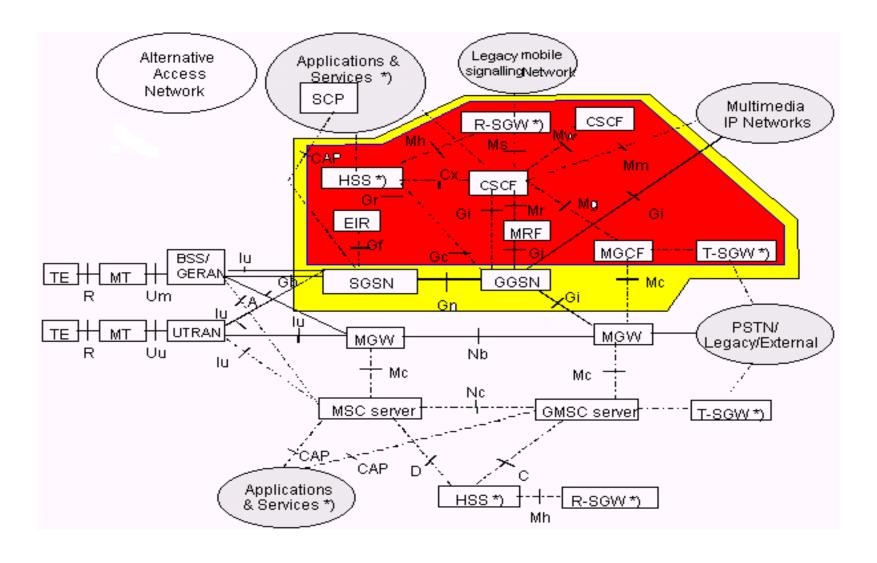




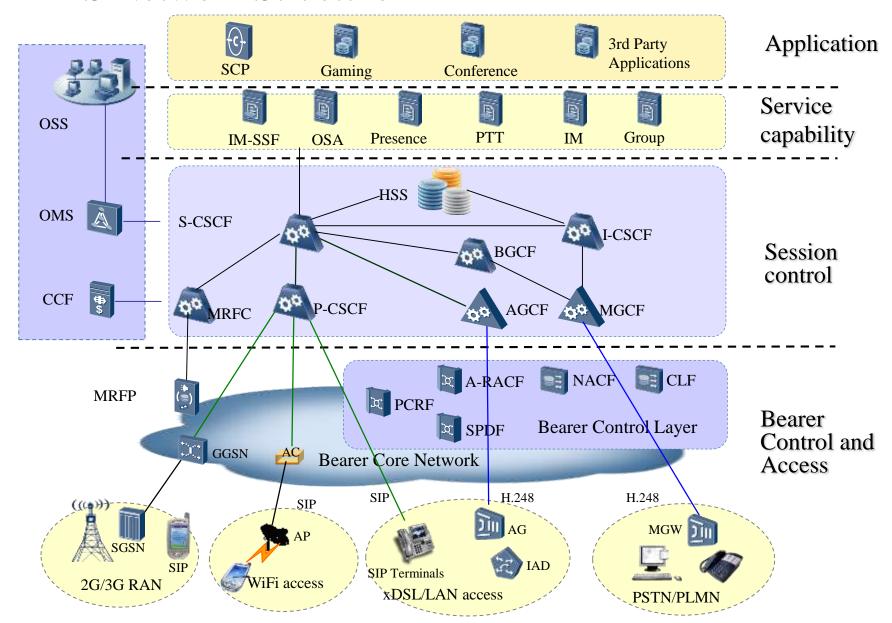
■ 3GPP R4 Network Architecture



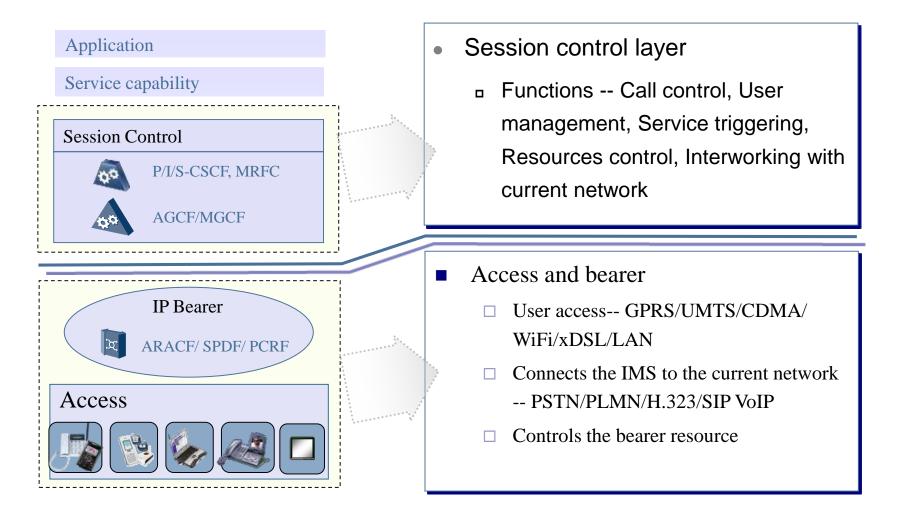
■ 3GPP R5 Network Architecture



■ IMS Network Structure



■ IMS Network Layers Architecture



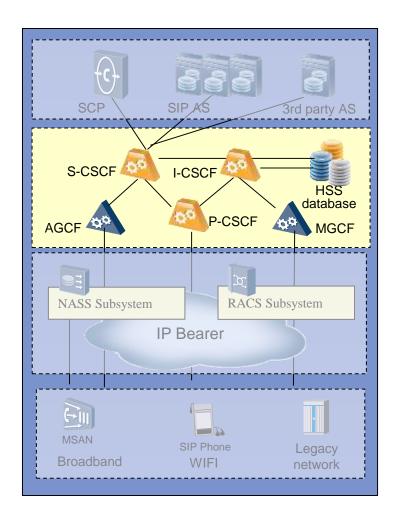


■ IMS Network Entities

The Network Elements of IMS in Session Control Layer are:

Function	NE		Function	NE	
Call control	P-CSCF	ØQ.	Network interworking	MGCF	00
	I-CSCF	Ø.		IM-MGW	
	s-cscf	Ø.		BGCF	Ø.
User management	HSS		Media	MRFC	Øø.
	SLF		resource	MRFP	(*)

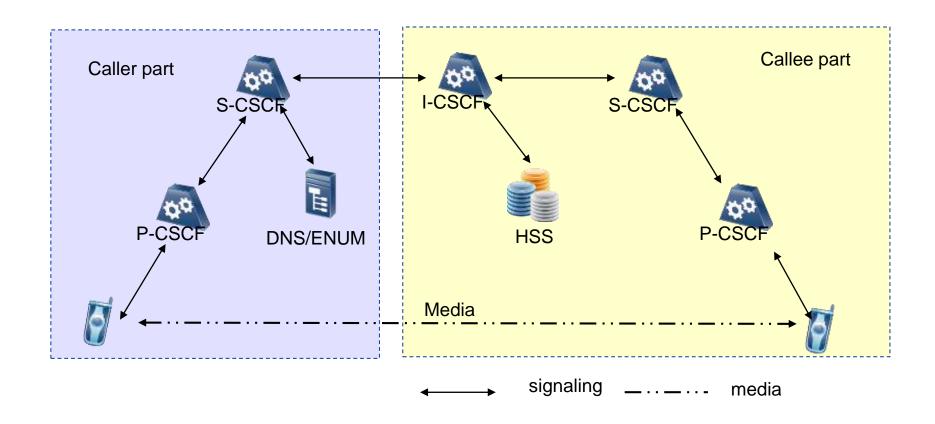
■ CSCF - Call Session Control Function



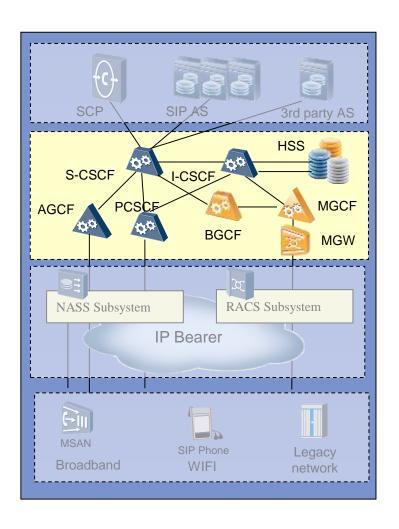
CSCF	Key Functions
P-CSCF	1: The 1st contact point to the IMS network in the
(Proxy)	visiting domain or home domain
	2: Access network control
	3: QoS control, NAT control and security control
I-CSCF	1: First entry to the IMS network of a carrier
(Interrogating)	2: S-CSCF assignment and session routing
	3: Topology hiding
S-CSCF	1: User registration authentication
(Serving)	2: Session route control (Normal, interworking,
	Emergency call)
	3: Service trigger

■ Simple Model for IMS Call Procedure

While two IMS users call each other, the P/I/S-CSCF can handle the whole signaling routing procedures.



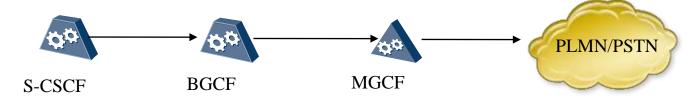
■ IMS Interworking Nodes



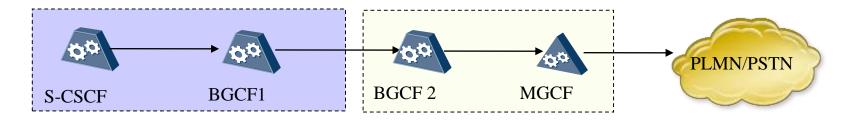
NE	Key Functions
MGCF	Media Gateway Control Function
	1:Controls the IMS-MGW for establish/modify/delete media channels
	2: Selects the I-CSCF for incoming calls from PSTN/CS.
	3: Performs protocol conversion between ISUP and SIP.
IM-MGW	IM-MGW IMS-Media Gateway Function
	1: Terminate bearer channels from a switched circuit network and
	media streams from a packet network
BGCF	Breakout Gateway Control Function
	1: Select a proper MGCF for interworking with the PSTN/CS domain

Connection to Legacy Networks

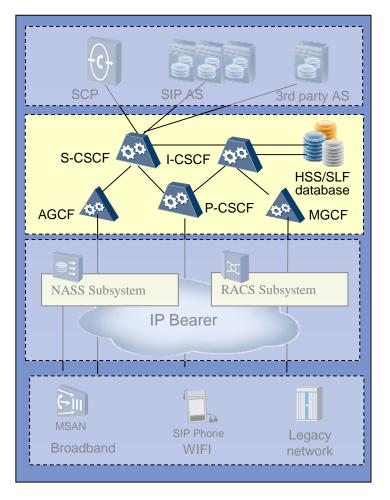
While an IMS user call a PSTN/CS user



While an IMS user call a PSTN/CS user which belongs to other operator

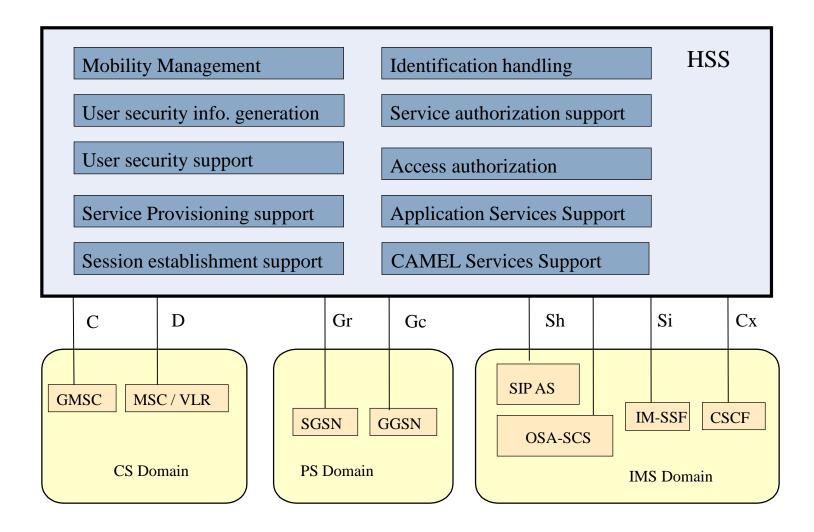


■ Database Function – HSS and SLF

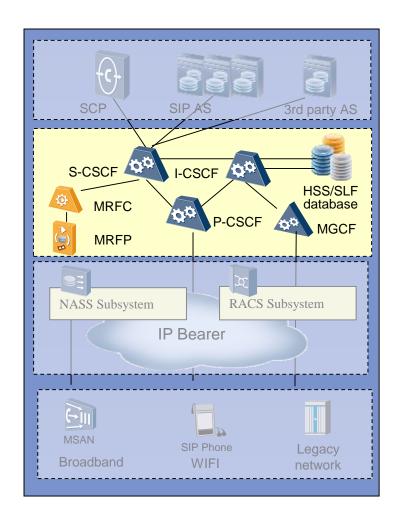


NE	Key Functions
HSS	1: User Identification, Numbering and addressing information;
Home Subscriber Server	2: User Security information: Network access control information for authentication and authorization;3: User Location information at inter-system level;4: User profile information (iFC, etc).
SLF	When an operator has more than one HSS,
Subscription Locator Function	SLF is used to select the related HSS, and usually SLF is combined with the HSS

■ Database Function – HSS and SLF

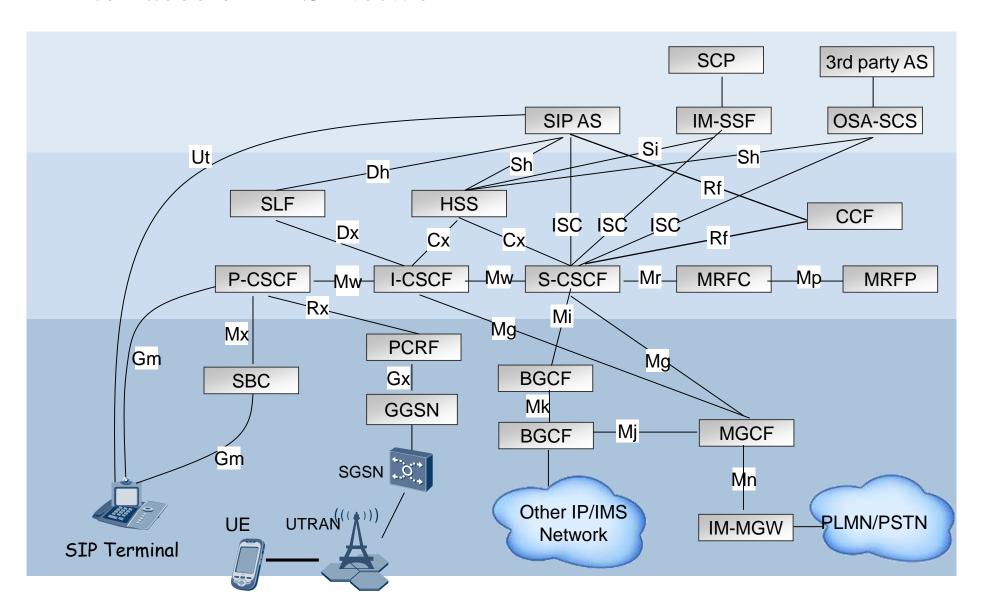


■ IMS Multimedia Resource Function



NE	Key Functions
MRFC	Multimedia Resource Function Controller
	1: Controls the media stream resources in the MRFP.
	2:Interprets information coming from an AS and S-CSCF (e.g session identifier) and control MRFP accordingly.
MRFP	Multimedia Resource Function Processor
	1: Provides resources to be controlled by the MRFC.
	2: Mixes incoming media streams (e.g. for multiple parties).
	3: Processes media streams (e.g. audio transcoding, media analysis).

■ Interfaces of IMS Network



■ IMS Interface and Protocols

NE A	NE B	Interface	Application Layer Name	Transport Layer Protocol
UE	P-CSCF	Gm	SIP	IPSec / UDP
PGW	P-CSCF	SGi	IP	IP
P-CSCF	I-CSCF	Mw	SIP	UDP
S-CSCF	MGCF	Mj	SIP	UDP
S-CSCF	ATS	ISC	SIP	UDP
MGCF	I-CSCF	Mg	SIP	UDP
SRVCC IWF	ATS	12	SIP	UDP
P-CSCF	PCRF	Rx	Diameter	SCTP
I-CSCF	HSS	Сх	Diameter	SCTP
S-CSCF	HSS	Сх	Diameter	SCTP

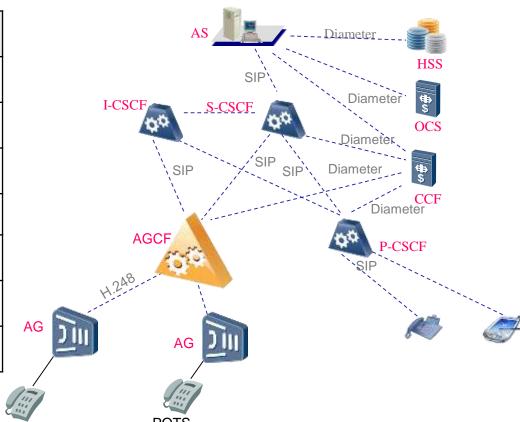
SIP & H.248 / Reference Points

Reference point (SIP)	Endpoints
Gm	UE, PCSCF
Mw	P-CSCF, I-CSCF ,S-CSCF
ISC	I-CSCF, S-CSCF,AS
Mg	S/I-CSCF,MGCF
Mi	S-CSCF, BGCF
Mj	BGCF,MGCF
Mk	BGCF,BGCF
Mx	SBC,P-CSCF
Mr	S-CSCF,MRFC

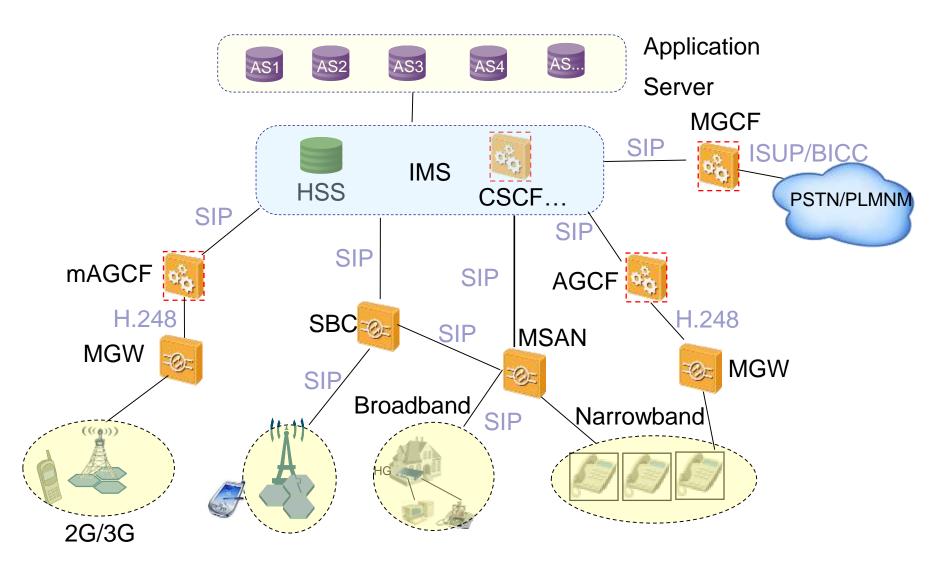
Reference point (H.248)	Endpoints
Mn	MGCF, IM-MGW
Мр	MRFC, MRFP

■ Diameter / Reference Points

Reference point	Endpoints
Сх	I-CSCF, S-CSCF, HSS
Dx	I-CSCF, S-CSCF, SLF
Rx	P-CSCF,PCRF
Gx	GGSN, PCRF
Rf	CCF,S-CSCF,AS,MRFC
Sh	HSS, AS
Dh	SLF, AS

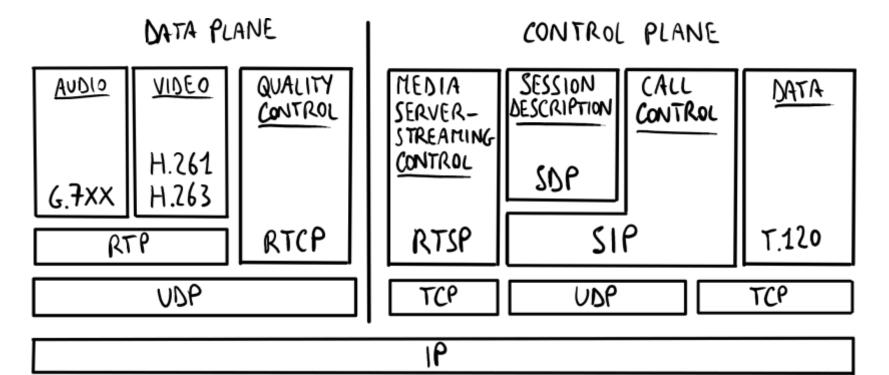


■ IMS Evolving Target Network



■ SIP Protocol

- The SIP was originated in Mbone (Multicast Backbone) experiment in 1996.
- IETF Standard defined by RFC 3261. (July, 2002)
- The Session Initiation Protocol (SIP) is an application layer protocol (Signaling) for creating, modifying and terminating sessions with one or more participants.
- Can be used for Voice, Video, Instant Messaging, Gaming, etc.
- SIP is Text based like HTTP, Request / Reply Protocol and mostly used on UDP.



SIP Protocol

- Use URIs for addressing Single Communications Identity
 - mailto:dbaron@MIT.edu for email
 - xmpp:dbaron@MIT.EDU for instant messaging
 - □ sip:dbaron@MIT.EDU for voice and video
- - Host name
 - Port(optional parameter) _
 - URI parameter
 - Header and Body(optional parameter)

Sip:28970808@huawei.com:5061;user=phone?Subject=foo

Username replaced by numbers for telephone applications.

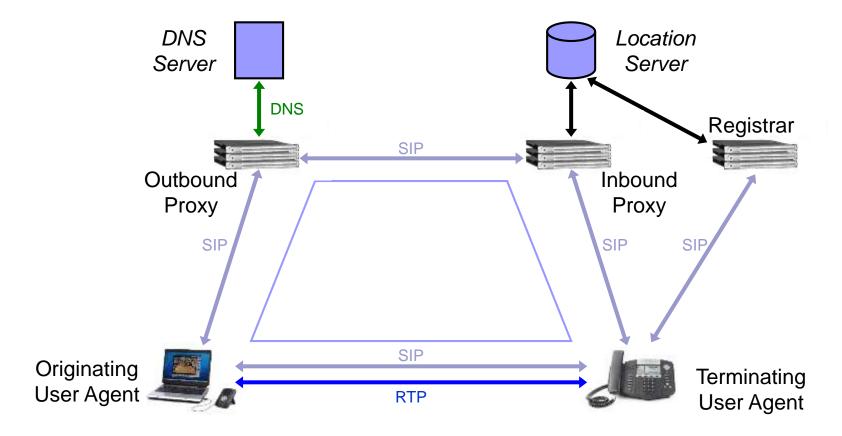
SIP Components

User Agents

- ☐ Cleints Make Request
- ☐ Servers Accept Requests

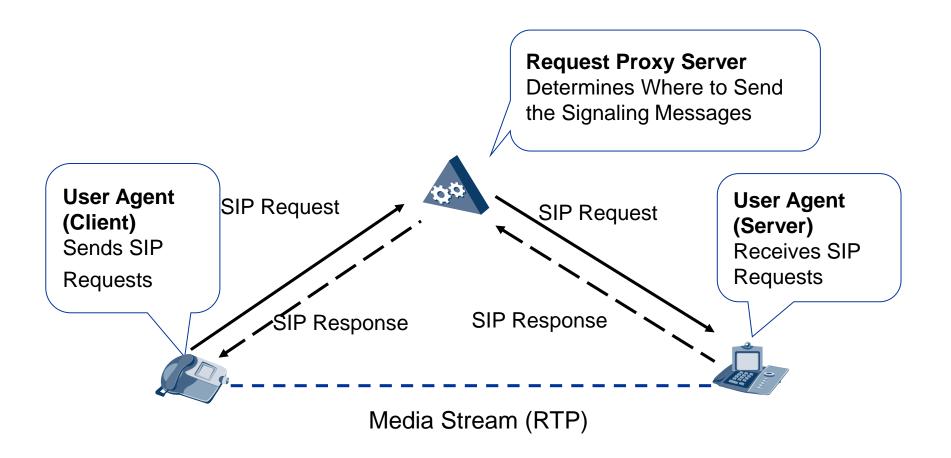
Server Types

- Redirect Server
- Proxy Server
- Registrar Server
- Proxy Server
- Location Server



Gateways

■ Simplified SIP Network Architecture



Logical Entities of SIP

Logical SIP entities



User Agents (UA)

- User Agent Client (UAC):Initiates SIP requests
- User Agent Server (UAS):Returns SIP response
- Both UAC and UAS can terminate a call



Network Servers

- Register: Maintains the location of SIP users, sip clients needs to update the location using register request
- Proxy: decides next hop and forwards request
- Redirect: Accepts SIP request and translates to new address

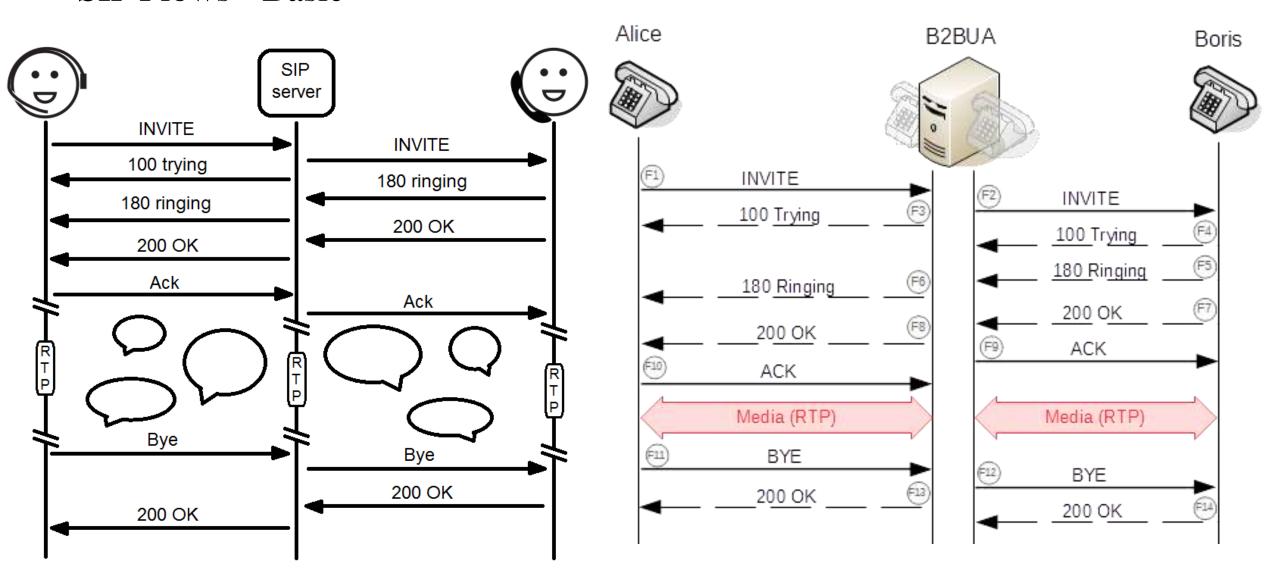
SIP Methods

Method	Description
INVITE	Requests a session
ACK	Final Response to the INVITE
OPTIONS	Ask for server capabilities
CANCEL	Cancels a pending request
BYE	Terminate a session
REGISTER	Send user's address to server

SIP Responses

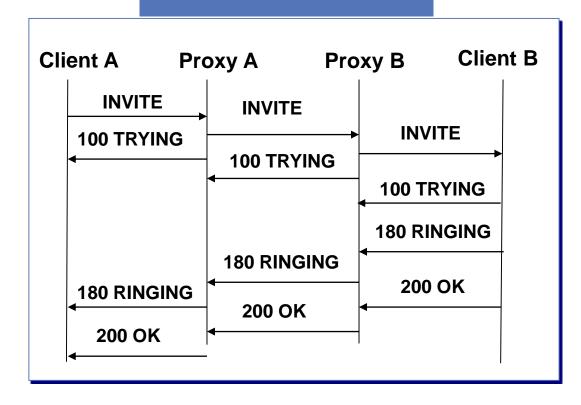
Response	Туре	Sample
1XX	Provisional	100 Trying
2XX	Successful	200 OK
3XX	Redirection	302 Moved Temporarily
4XX	Client Error	404 Not Found
5XX	Server Error	504 Server Timed-Out
6XX	Global Failure	603 Decline

■ SIP Flows - Basic

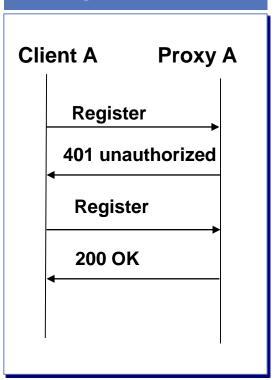


■ SIP Flows - Basic

Session setup



Registration



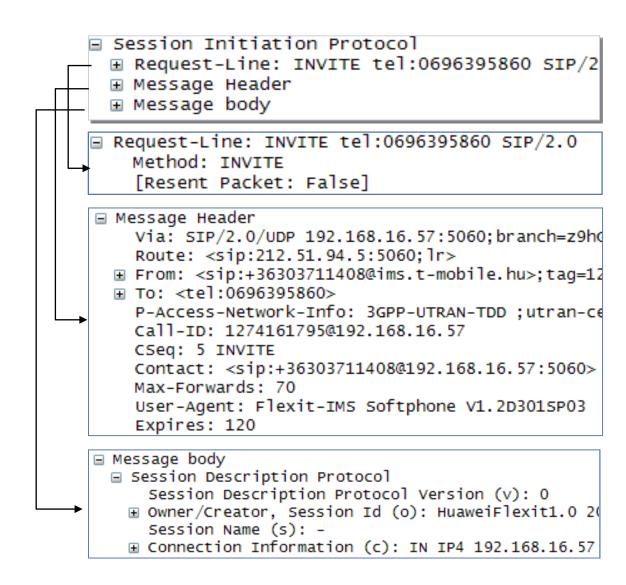
■ SIP Message Structure

SIP is a Text based protocol and comprised by 3 parts:

Request-Line

☐ Header

□ Body



■ SIP Request / Response Example

```
Method
Request
          INVITE Sip: Gonzalo. Camarillowericsson.com SIP/2.0
 Line
          View SIP/2.0/UDP [5555::aaa:bbb:ccc:dd];branch=z9hG4bknashds
          Max-Forwards: 70
                                                                     Request-URI
          Route: <sip;pcscfl.visitedl.net;lr>, <sip;scscfl.homel.net;lr>
          From: <sip:user1_public1@home1.net>;tag=171828
          To: <sip:Gonzalo.Camarillo@ericsson.com>
                                                              Header Field
          Call-ID: cb03a0s09a2sdfqlkj490333
          Cseq: 127 INVITE
          Contact > < sip: [5555: | aaa;bbb;ccc;ddd] >
          Content-Type application/sdp
                                                            Header Field Name
          Content-Length: 248
                                                          leader Field Value
          o -- 2987933615 2987933615 IN IP6 5555;;aaa;bbb;ccc;ddd
          c=IN IP6 5555::aaa:bbb:ccc:ddd
          t-907165275 O
          m=audio 3458 RTP/AVP 97 96 0 15
                                                                     Status
          a-rtpmap:97 AMR
                                                                                (SIP/2.0)(200)(OK)
                                                                                                               Reason phrase
                                                                      Line
          a=fmtp:97 mode-set=0,2,5,7; maxframes=2
                                                                                Via: SIP/2.0/UDP [5555::aaa:bbb:ccc:ddd];branch=z9hG4bKnashds7
          a=rtpmap:96 G726-32/8000
                                                                                 Record-Route: <sip:scscf1.home1.net;lr>, <sip:pcscf1.<u>visited1.net:lr></u>
                                                                                 From: <sip:user1 public1@home1.net>;tag=171828
                                                                                                                                          Status codes in SIP
                                                                                 To: <sip:Gonzalo.Camarillo@ericsson.com>;tag=314159
                                                                                                                                          1xx - Provisional responses
                                                                                 Call-ID: cb03a0s09a2sdfglkj490333
                                                                                                                                          2xx - Success
                                                                                 CSeq: 127 INVITE
                                                                                 Contact: <sip:[5555::eee:fff:aaa:bbb]>
                                                                                                                                          3xx - Redirection
                                                                                 Content-Type: application/sdp
                                                                                                                                          4xx - Client Error
                                                                                 Content-Length: 220
                                                                                                                                          5xx - Server Error
                                                                                                                                          6xx - Global Failures
                                                                                     2987933615 2987933615 IN IP6 5555::eee:fff:aaa:bbb
                                                                                 c=IN IP6 5555::eee:fff:aaa:bbb
                                                                           ody
                                                                                 t=907165275 0
                                                                                 m=audio 3458 RTP/AVP 97 0
                                                                                 a=rtpmap:97 AMR
                                                                                 a=fmtp:97 mode-set=0,2,5,7; maxframes=2
```

SDP Example

v=0

o=HuaweiSoftX3000 868 868 IN IP4 10.216.9.200

s=Sip Call

c=IN IP4 10.216.6.108

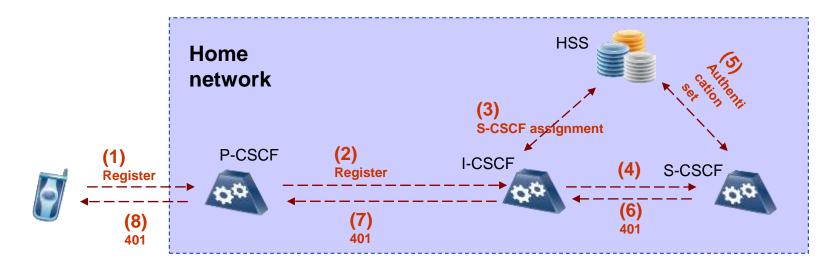
t=0 0

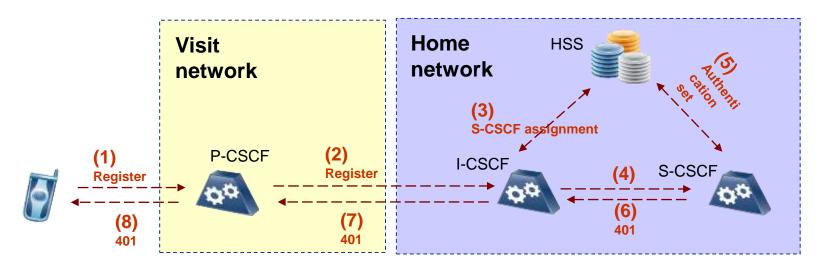
m=audio 17368 RTP/AVP 8

a=rtpmap: 8 PCMA/8000

SDP Parameter	Parameter Name	Remarks
V	Version number	v=0
0	Origin containing name	o= <user name=""> <session id=""> <version> <network type=""> <address></address></network></version></session></user>
S	Subject	
С	Connection	Connection IP address(10.216.6.108)
t	Time	t= <start time=""> <stop time=""></stop></start>
m	Media	Media format (audio); Port number(17368)
а	Attribute	Media encoding (PCM A Law); Sample rate (8000Hz)

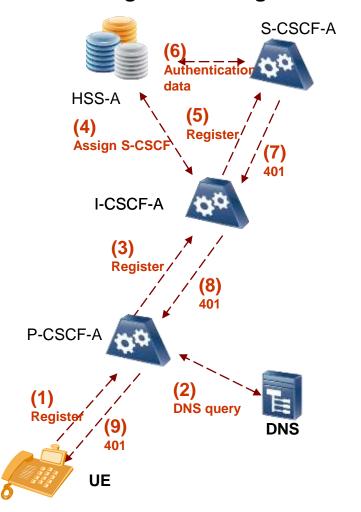
■ IMS User Registration Procedure

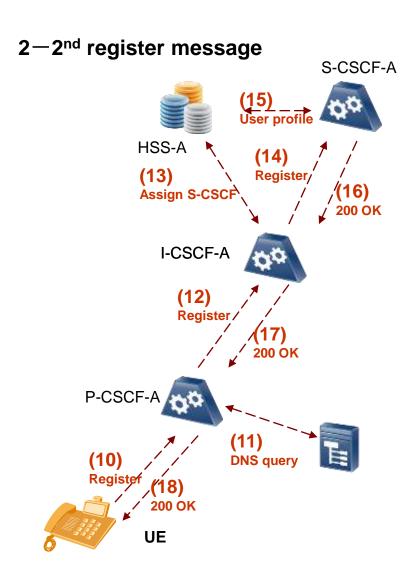




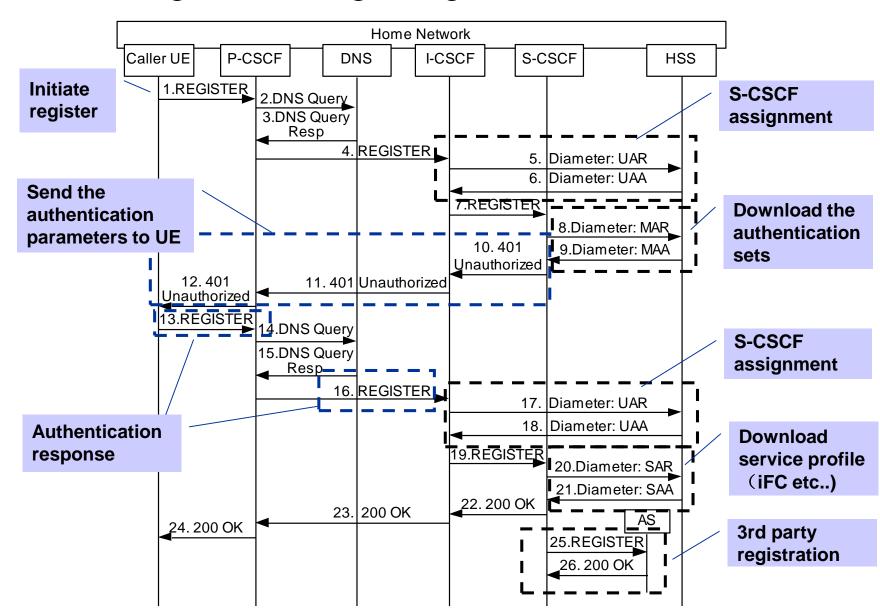
■ IMS User Registration Procedure

1 — Initial register message



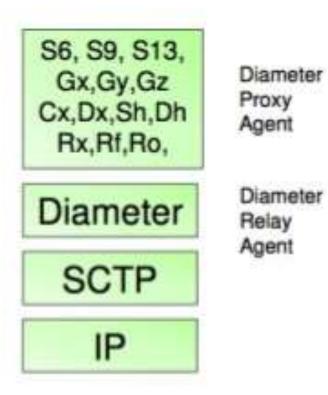


■ IMS Registration Signaling Flow

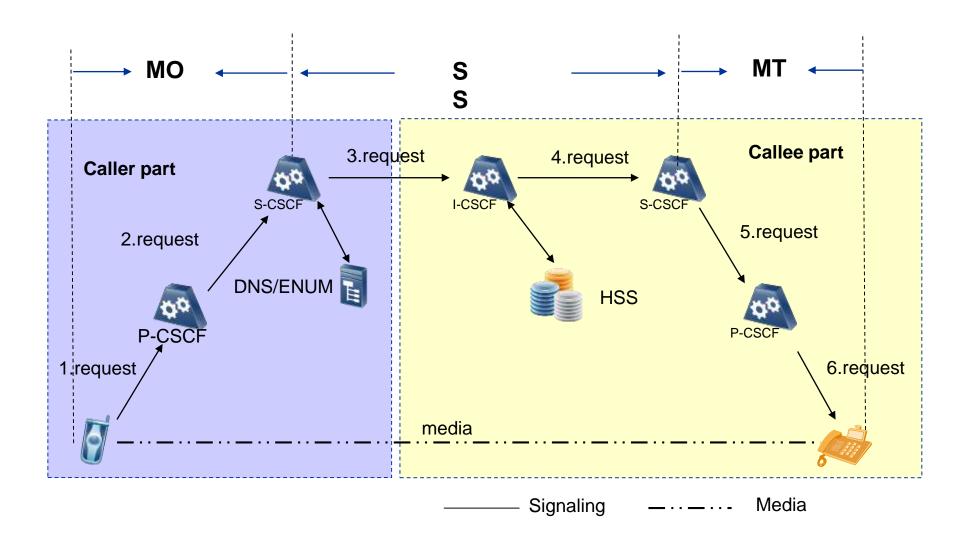


Related Diameter Messages

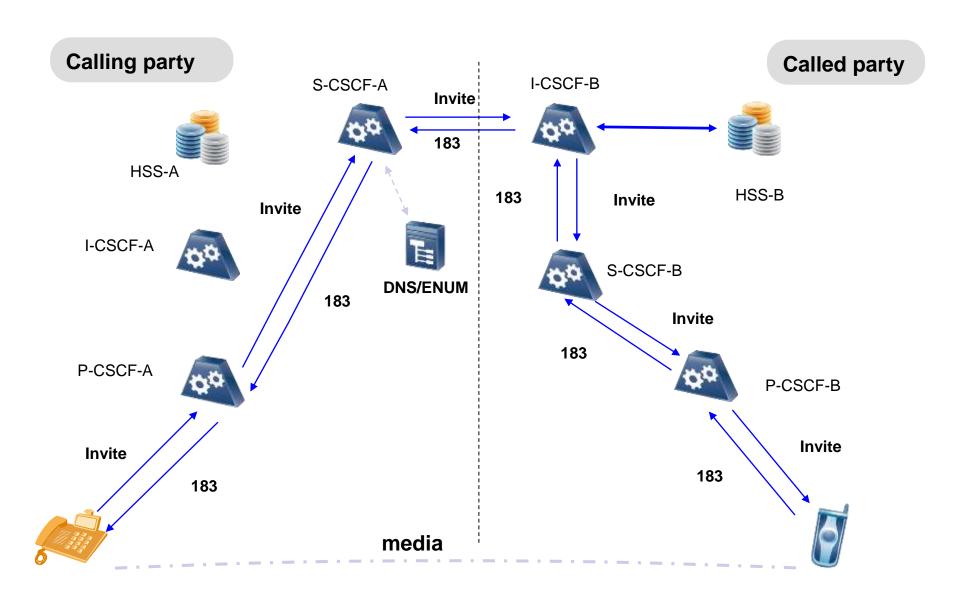
Message	Explanation	
UAR	User Authorization Request	
UAA	User Authorization Answer	
MAR	Multimedia Authentication Request	
MAA	Multimedia Authentication Answer	
SAR	Server Assignment Request	
SAA	Server Assignment Answer	



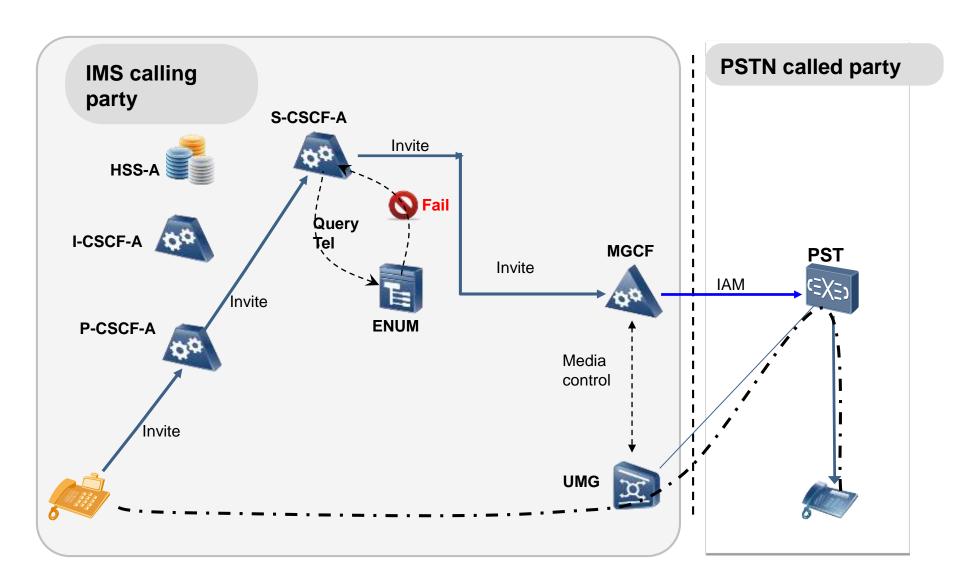
Simple IMS Call Flow



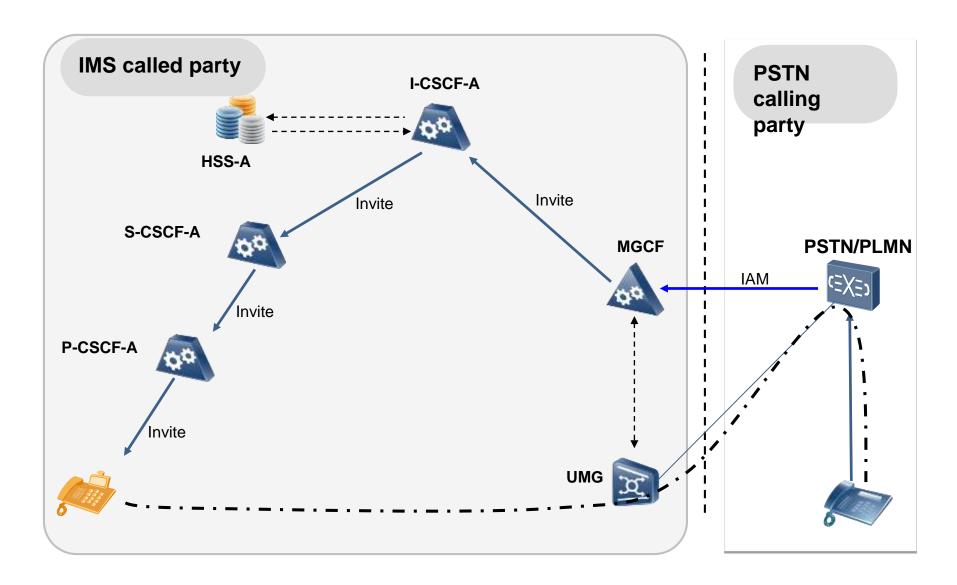
■ Basic Session Establishment between 2 IMS Users



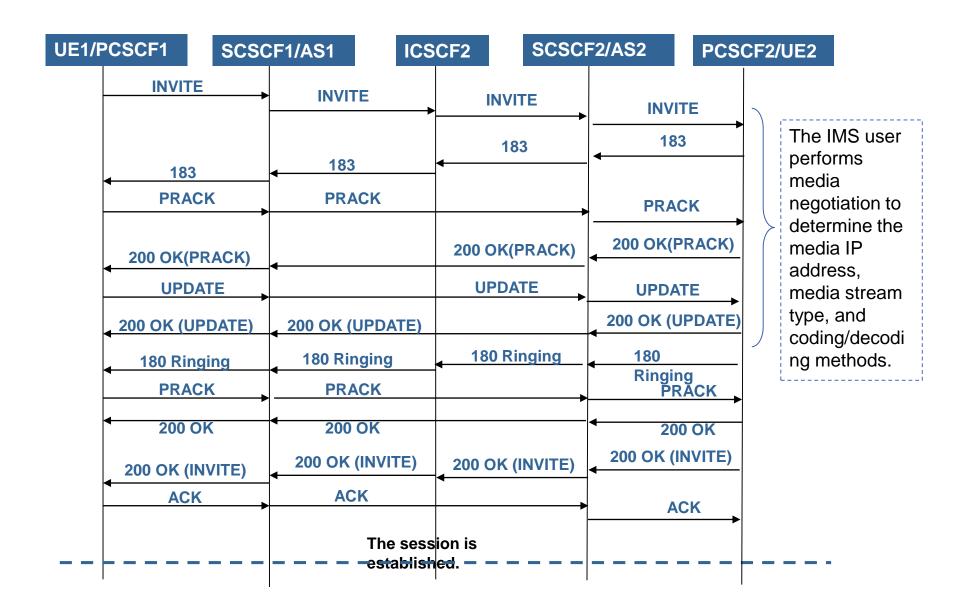
■ IMS User Call PSTN User



■ PSTN User Call IMS User



■ IMS Session Establishment Signaling Flow



Thanks For Your Attention!



Hamid Reza Bolhasani Bolhasani@gmail.com