

Introduction of VolTE

No: KT_1512_10 To: RJIL NPE

From: KT RF Part



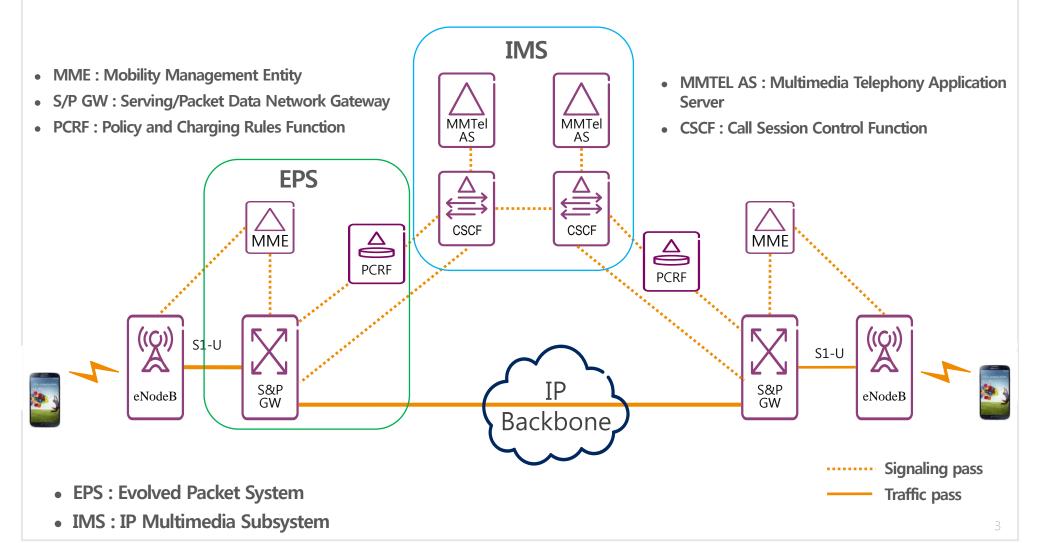


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2	Call Flow
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01 VolTE - Overview

Network Structure for VoLTE



01 VolTE - Overview

Voice service on LTE network

- ☐ SVLTE (Simultaneous Voice and LTE)
 - ✓ Method: UE supports both LTE (data) and legacy networks (circuit service) simultaneously.
 - ✓ Pros : No interworking needed between LTE and legacy, Faster than CSFB, (Voice + LTE data) is possible
 - ✓ Cons: High battery consumption (Using both modems simultaneously), Limited supplier for mobile
 - ✓ Operator : CDMA2000 operators in Korea : LGU+ at first LTE launching

☐ CSFB (Circuit Switch Fall-Back)

- ✓ Method: UE falls back to a legacy network for voice service
- ✓ Pros : Low battery consumption, Simple UE structure (cost down)
- ✓ Cons: Interworking needed between LTE and legacy, Delay for switching system, (Voice + LTE data) is impossible (3G data only)
- ✓ Operator : Most of WCDMA operators, it's a 3GPP standard. (in Korea : KT/SKT)

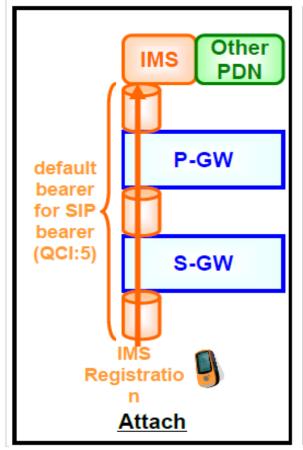
☐ VolTE (Voice Over LTE)

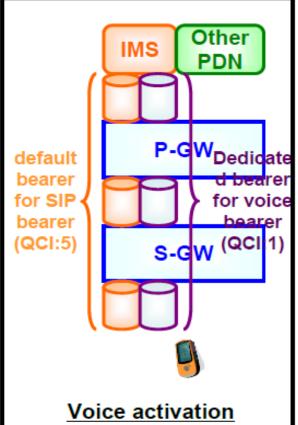
- ✓ Method : Supports Voice call on LTE network
- ✓ Pros : Low battery consumption, Simple UE structure (cost down), Faster than CSFB, (Voice + LTE data) is possible
- ✓ Cons : Requires IMS system, Complex core network, Requires National Wide network coverage
- ✓ Operator : Most of LTE operators are preparing
- > SRVCC (Single Radio Voice Call Continuity): Technology for transferring VoLTE call to the legacy (3G/2G) CS network while the call is in progress to . It is specified 3GPP REL.10

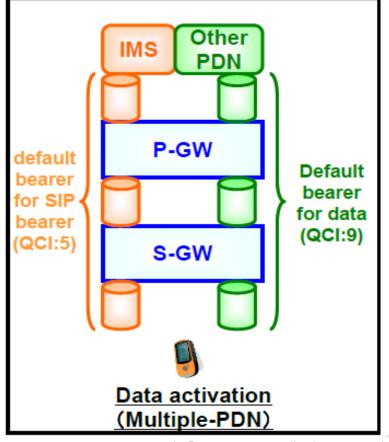
01 VolTE – Overview

QOS based bearers

- The SIP Bearer is constructed as a default bearer on IMS APN in attach procedure.
- The Voice bearer is constructed as a dedicated bearer on IMS specific APN in voice activation phase.
- Bearer for data is constructed as default bearer on other APN in attach procedure.







01 VolTE - Overview

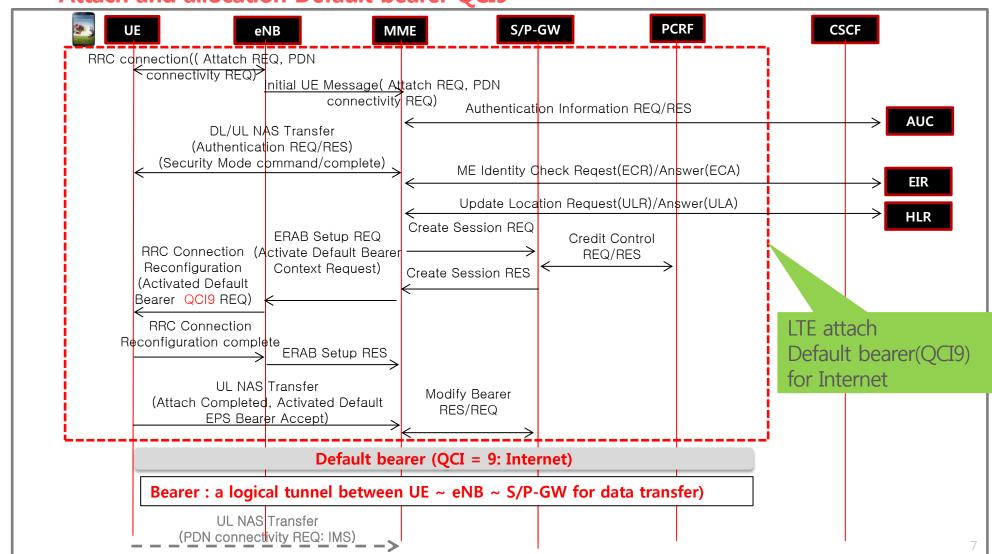
QCI (QoS Class Identifier)

	QCI	Resource type	Priority	Packet Delay Budget(ms)	Packet error Loss Rate	services	remark
	1	GBR	2	100	10-2	Conversation Voice	Volte RTP
	2	GBR	4	150	10-3	Conversation Video (live streaming)	PSVT
•	3	GBR	3	50	10 ⁻³	Real-time Gaming	
	4	GBR	5	300	10-6	Non-conversation Video (buffered streaming)	
	5	Non-GBR	1	100	10 ⁻⁶	IMS signaling	rjil,kt ims
	6	Non-GBR	6	300	10-6	Video (buffering streaming)	KT default bearer for data
	7	Non-GBR	7	100	10-3	Voice, Video (live streaming), interacting gaming	
	8	Non-GBR	8	300	10-6	TCP-based (WWW, email, FTP) ;privileged subscriber	
	9	Non-GBR	9	300	10 ⁻⁶	TCP-based (WWW, email, FTP) ;non- privileged subscriber	RJIL default bearer for data

> PSVT : Packet Switched Video Telephony service

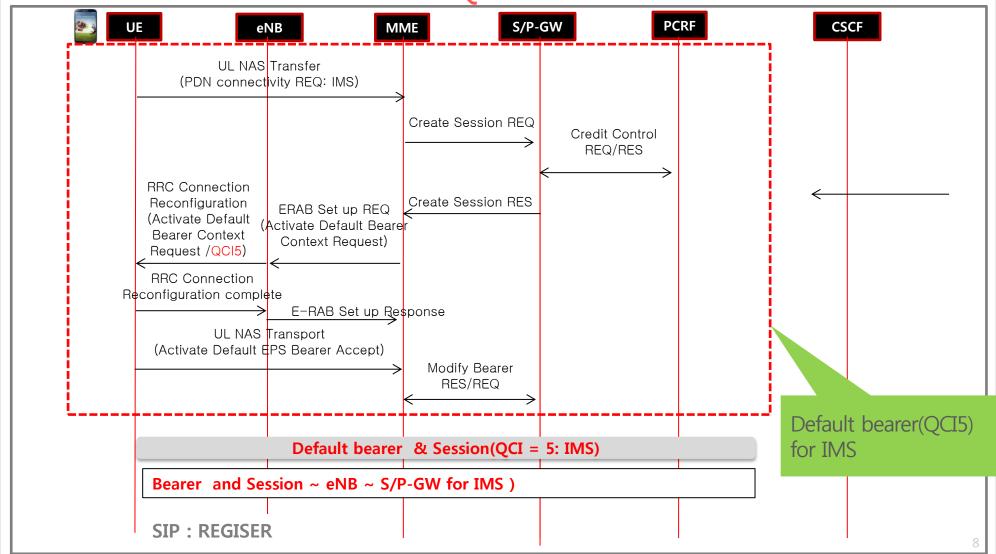
01 VolTE – Call Flow

Attach and allocation Default bearer QCI9



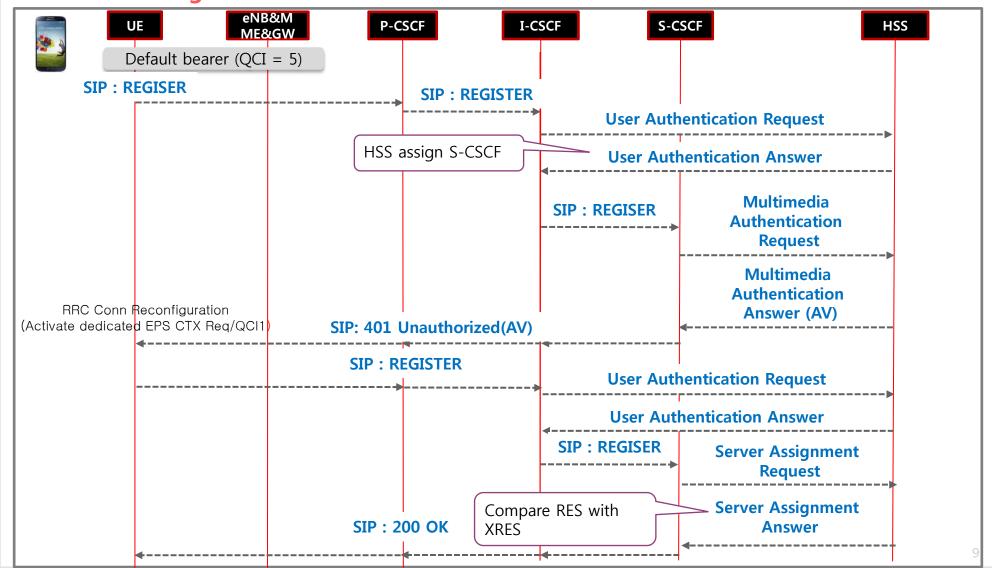
01 Volte - Call Flow

Attach and allocation Default bearer QCI5



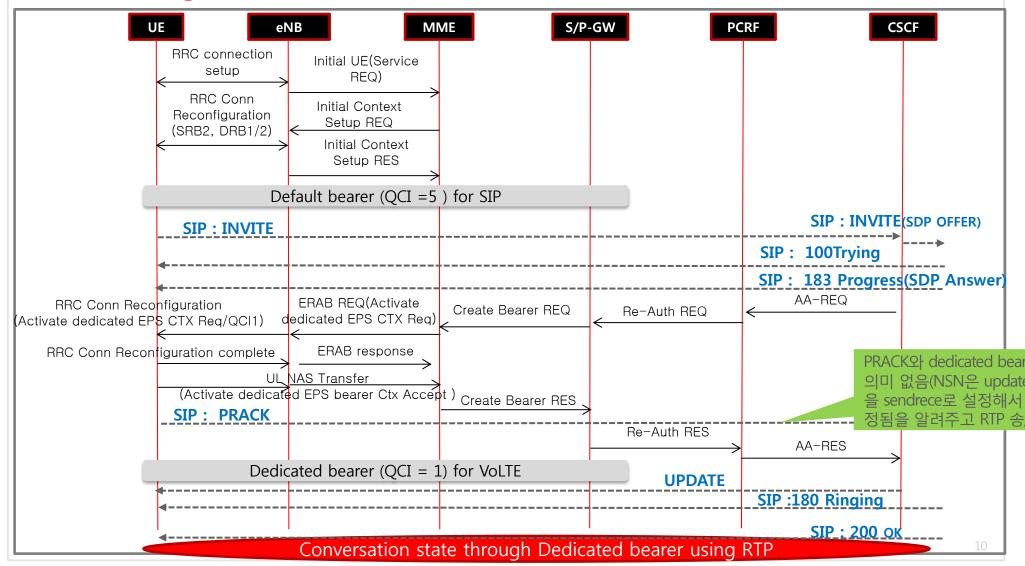
02 VolTE - Call Flow

Attach to register IMS

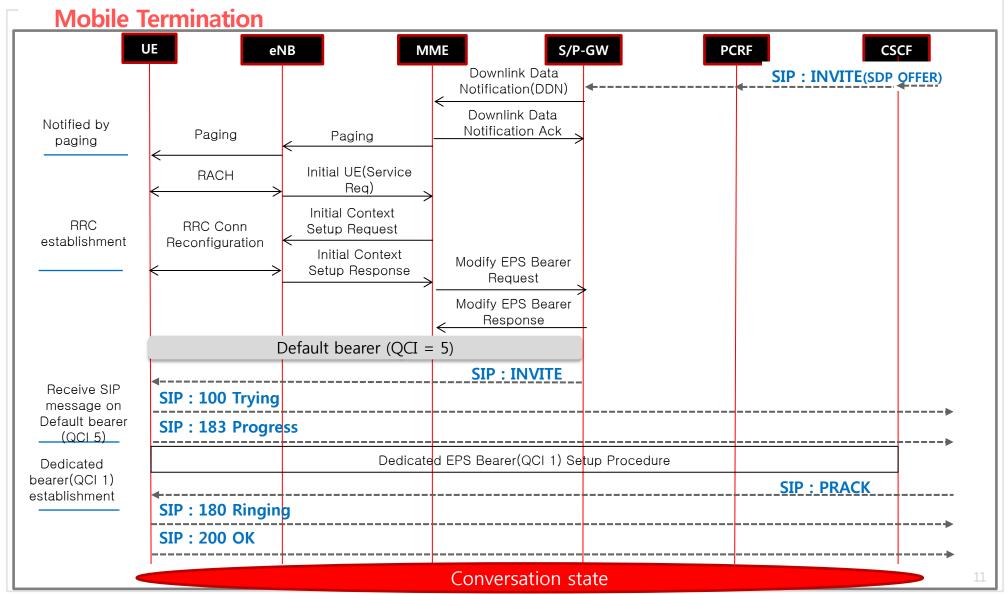


02 Volte – Call Flow

Mobile Origination



02 Volte - Call Flow



03 Protocol - SIP Method (Request / Respond)

SIP Method	Description	Definition
INVITE	Indicate that a client being invited to participate in a call session	RFC 3261
ACK	Confirms that the client has received a final response to and INVITE request	RFC 3261
BYE	Terminates a call; can be sent by either the caller or the called party	RFC 3261
CANCEL	Cancels any pending request	RFC 3261
OPTION	Queries the capabilities of servers	RFC 3261
REGISTER	Registers the address listed in the To header field with a SIP server	RFC 3261
PRACK	Provisional acknowledgement	RFC 3262
SUBSCRIBE	Subscribes an event to the Server	RFC 3265
NOTIFY	Notifies the subscriber of a new Event	RFC 3265
PUBLISH	Publishes an event to the Server	RFC 3903
INFO	Sends mid-session information that does not modify the session state	RFC 6086
REFER	Asks recipient to issue a SIP request (call transfer)	RFC 3515
MESSAGE	Transports instant messages using SIP	RFC 3428
UPDATE	Modifies the State of a session without changing the state of the dialog	RFC 3911

03 Protocol - RTP

☐ RTP (Real-time Transport Protocol)

- ✓ RTP was developed by the Audio-Video Transport Working Group of the (IETF) in 1996.
- ✓ RTP is designed for end-to-end, real-time, transfer of streaming media such as telephony, videoconference, TV and Push-to-talk services.
- ✓ RTP can be used with several protocols such as SIP, Jingle, RTSP, H.225, H.245 and etc. (These protocols transfer signals and RTP transfer Traffic.)

☐ RTP payload types for VoLTE

Media type	Subtype	Rate(Hz)	IETF	3GPP	Notes
Audio	AMR	8000	RFC4867	26.071	Mandatory
Audio	AMR-WB	16000	RFC4867,3267	26.171	Mandatory
Audio	EVRCB	8000	RFC4788	C.S0055	for CDMA
Audio	EVRCWB	8000	RFC5188	C.S0055	for CDMA
Text	T140	1000	RFC4628		Mandatory
Video	H263	90000	RFC4628	26.235	Mandatory
Video	H264,MP4V-ES	90000	RFC6184,3016		options

☐ RTCP (Real-time Control Protocol)

- ✓ The RTP specification describes two sub-protocols; RTP and RTCP (Real-time Transport Control Protocol).
- ✓ RTCP is used to monitor transmission statistics and Quality of Service (QoS) and aids synchronization of multiple streams.

03 Protocol - SIP Response

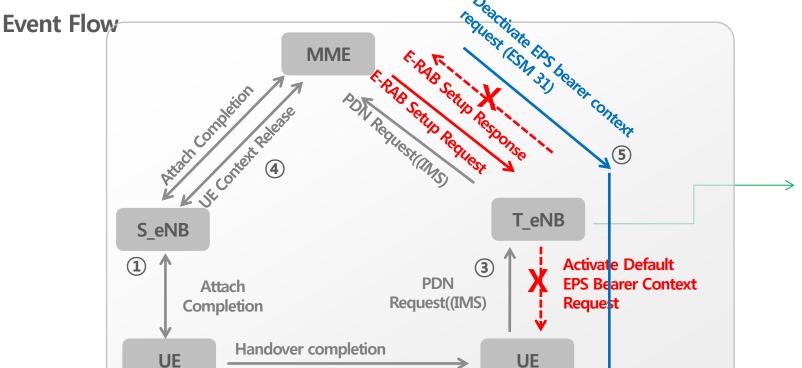
SIP Response	Codes	Description
Information/Provisional	1XX	Request received and being processed
Successful	2XX	The action was successfully received, understood, and accepted
Redirection	3XX	Further action needs to be taken (typically by the sender) to complete the request
Client Failure	4XX	The request contains bad syntax or cannot be fulfilled at the server
Server Failure	5XX	The server failed to fulfill unapparent valid request
Global Failure	6XX	The request cannot be fulfilled at the server

Refer to Appendix. SIP codes.

04 Event Cases – Attach

Attach Failure due to no response for E-RAB Setup from eNB

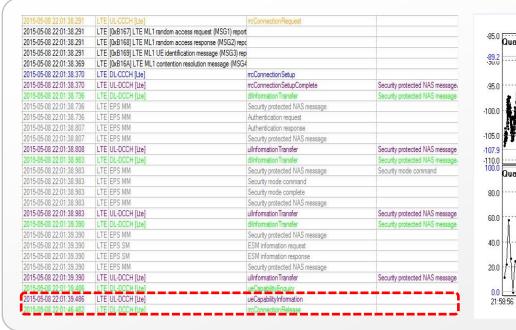
Test Date	LDC Building 2 nd floor 03-06-2015 11:34:53 - Reproduction test
Reason	MME doesn't receive E-RAB Setup response from eNB MME sends "deactivate EPS bearer context request (ESM 31)" to UE
Suggestion	Need eNB system trace to clarify eNB or Backhaul issue

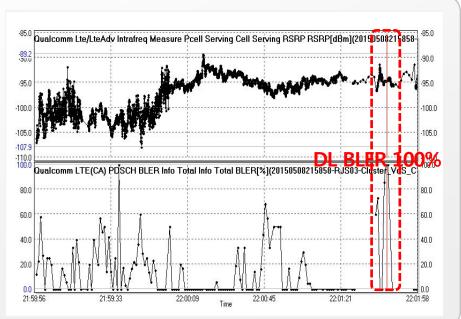


04 Event Cases - Attach

Due to poor RF condition, UE cannot receive RRC reconfiguration message including activate default EPS bearer context request

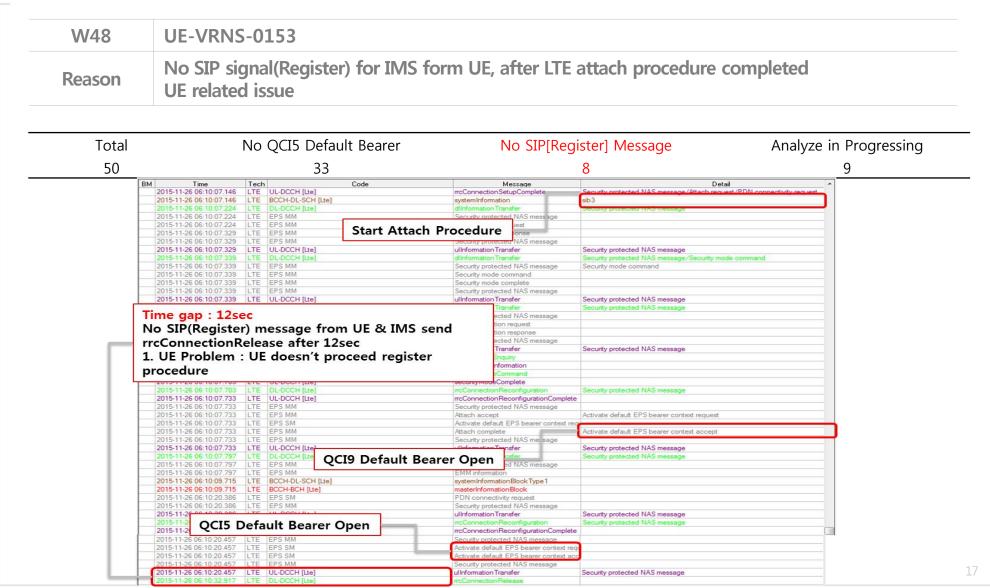
Cluster / Date	TN-DGGR-0100 08-05-2015 (20150508215858-RJS03-Cluster_VoS_C1_InProg_TN-VDGR0100_F4788193.drm)
Reason	eNB sends initial context setup failure (failure-in-radio-interface-procedure) to MME when RRC timer (2sec) expires
Suggestion	RF Issue





04 Event Cases - Attach

(UE-VRNS-0153)



04 Event Cases - Attach

W47	KO-KLKT-0077
Reason	no procedure for 2 nd Register UE, after 1 st Register done. IMS stack in UE doesn't work properly

2015-11-03 12:54:50.005	LTE	EPS MM	Security protected NAS message	
2015-11-03 12:54:50.005	LTE	EPS MM	EMM information	
2015-11-03 12:54:50.802	LTE	EPS SM	PDN connectivity request	
2015-11-03 12:54:50.802	LTE	EPS MM	Security protected NAS message	
2015-11-03 12:54:50.802	LTE	UL-DCCH [Lte]	ulInformation Transfer	Security protected NAS message
2015-11-03 12:54:50.909	LTE	DL-DCCH [Lte]	rrcConnectionReconfiguration	Security protected NAS message
2015-11-03 12:54:50.909	LTE	UL-DCCH [Lte]	rrcConnectionReconfigurationComplete	
2015-11-03 12:54:50.909	LTE	EPS MM	Security protected NAS message	
2015-11-03 12:54:50.909	LTE	EPS SM	Activate default EPS bearer context request	
2015-11-03 12:54:50.909	LTE	[0xB0E4] LTE NAS ESM bearer Co		IP address and QoS to be used by
2015-11-03 12:54:50.909	LTE	[0xB0E5] LTE NAS ESM bearer co		IMS are allocated to UE
2015-11-03 12:54:50.909	LTE	[0xB0E5] LTE NAS ESM bearer co		TIVIS are anotated to of
2015-11-03 12:54:50.909	LTE	[0xB0E4] LTE NAS ESM bearer Co		
2015-11-03 12:54:50.909	LTE	EPS SM	Activate default EPS bearer context accept	

- UE didn't send the 2nd Register as a response to 401 Unauthorized
 It seems that IMS stack in UE doesn't work properly

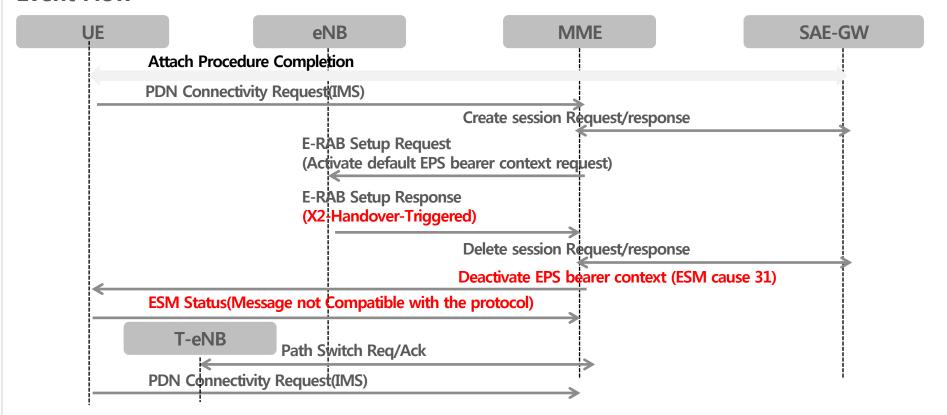
2015-11-03-12-54-52-346		тср	
2015-11-03 12:54:52:346	TCP Tx-SIP	TCP	25741, 60313, 5060, Seq:2025401588, Ack:214281898, REGISTER sip:ims.mnc873.mcc408
2015-11-03 12:54:52.410	TCP Rx-SIP	TCP	17443, 5060, 60313, Seq:214281898, Ack:2025402615
2015-11-03 12:54:52 494	TCP Rx-SIP	TCP	17444 5060 60313 Sep;214281898 Ack;2025402615 SIP/2 0 401 Unauthorized
2015-11-03 12:54:52.494	TCP Tx-SIP	TCP	25742, 60313, 5060, Seq:2025402615, Ack:214283128
2015-11-03 12:54:52.594	TCP Tx-SIP	TCP	25743, 60313, 5060, Seq:2025402615, Ack:214283128
2015-11-03 12:55:02.267	TCP Tx-ETC	TCP	39366, 53395, 15600, Seq:1328928166, Ack:0
2015-11-03 12:55:02.353	TCP Rx-ETC	TCP	31600, 15600, 53395, Seq:1959046023, Ack:1328928167
2015-11-03 12:55:02.353	TCP Tx-ETC	TCP	39367, 53395, 15600, Seq:1328928167, Ack:1959046024
2015-11-03 12:55:02.353	TCP Tx-ETC	TCP	39368, 53395, 15600, Seq:1328928167, Ack:1959046024
2015-11-03 12:55:02.353	TCP Tx-ETC	TCP	39369, 53395, 15600, Seq:1328928183, Ack:1959046024
2015-11-03 12:55:02.467	TCP Tx-ETC	TCP	39370, 53395, 15600, Seq:1328928204, Ack:1959046024
2015-11-03 12:55:02.467	TCP Tx-ETC	TCP	39371, 53395, 15600, Seq:1328928420, Ack:1959046024
2015-11-03 12:55:02.467	TCP Tx-ETC	TCP	39372, 53395, 15600, Seq:1328928457, Ack:1959046024

04 Event Cases – Attach

In case of X2HO collision during activating IMS default bearer

Test Date	GJ-AMBD-0049, 2015-05-20 17:46:00, CLOT VoS Test(C2)
Reason	Failure to create IMS APN default bearer due to Handover-trigger in eNB

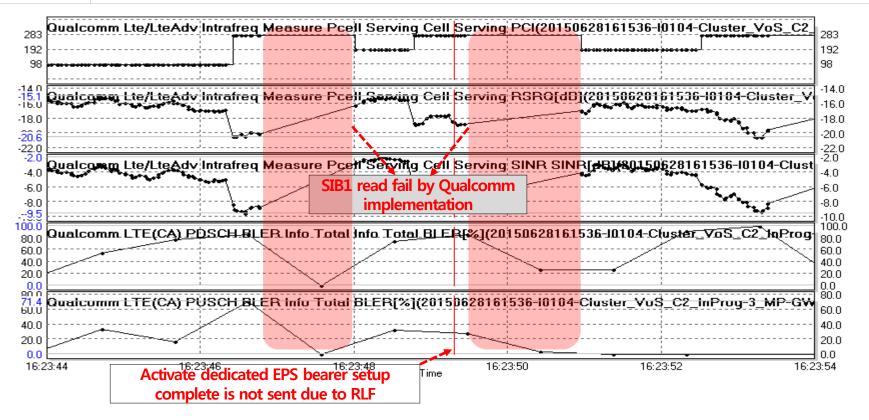
Event Flow



04. Event Cases – Call setup

VolTE call setup fail by receiving 503 service unavailable due to RF condition

Test Date	MP-GWLR-0717 2015-06-28 16:23:49 CLOT VoS Test(C2)
Reason	If there is no SIB1 message after completely receiving MIB then UE declares RLF by complementation



O4 Event Cases – Call setup

W47	HR-MHNG-0205				
Reason	UE does not retransmit " INVITE", after no 100Trying from IMS UE related issue				

Due to low attach success counts(85 call), despite only 1 call failure, KPI was failed.

- RF condition: Good, RSRP -78.54 dBm, SINR 11.53 dB
- Issue: There is no '100 trying' from IMS, then UE does not retransmit "INVITE", until timer expires.

- UE should retransmit "INVITE" before timer is expired, if there is no "100 trying" from IMS

- not able to analyze further because Tektronix log data doesn't exist

Result: one abnormal case occurred, No relation with RF condition

Abno	ormal	case		No	rmal	case		retransmit	X 100 Trying
2015-11-13 16:01:07.50	Tx Voice Call	Call Setup Status	Intent.ACTION CALL	2015-11-13 16:04:39.39	Tx Voice Call		IntentACTION_CALL		INVITE
2015-11-13 16:01:07.914	TCP Tx-SIP	TCP	18060, 45422, 5060, Seq:1042729266, Ack:1378043782, INVITE tel:55555;pho	2015-11-13 16:04:40.517	TCP Tx-SIP		64980, 60548, 5060, Seq:3183853442, Ack:3608402396, INVITE tel:55555;phoi		
	17.50/.000.500	1.750			TCP Tx-SIP TCP Rx-SIP	TCP TCP	64981, 60548, 5060, Seq:3183854810, Ack:3608402396, INVITE tel:55555;pho		
015-11-13 16:01:07.914	TCP Tx-SIP	TCP	18061, 45422, 5060, Seq:1042730634, Ack:1378043782, INVITE tel:55555;pho	2015-11-13 16:04:40.588	TCP Rx-SIP	TCP	30445, 5060, 60548, Seq:3608402396, Ack:3183855278 30446, 5060, 60548, Seq:3608402396, Ack:3183855278, SIP/2.0 100 Trying	RRC F	Reconf.
015-11-13 16:01:08.112	TCP Rx-SIP	TCP	38795, 5060, 45422, Seq:1378043782, Ack:1042730634	2015-11-13 16:04:40.645	TCP Tx-SIP		64982, 60548, 5060, Seq:3183855278, Ack:3608402649		er (QCI 1)
015-11-13 16:01:08.429	TCP Tx-SIP	TCP	18062 45422 5060 Seq:1042730634 Ack:1378043782 00	2015-11-13 16:04:40.833 LT	E DL-DCCH [Lie]	rrcConnectionReconfig	Security protected NAS message		I
	10.700.000.700.00			1-13 16:04:40.843 LT		Security protected NA			SIP: 183 Session progress (SDP Answe
15-11-13 16:01:09.165	TCP Tx-SIP	UE does	not get 100 Trying from I	1-13 16:04:40.843 LT		rrcConnectionReconfig			
15-11-13 16:01:10.532	TCP Tx-SIP		3 , 3	1-13 16:04:40.844 LT 1-13 16:04:40.844 LT	E EPS SM E EPS SM	Activate dedicated EP Activate dedicated EP			SIP Prack
015-11-13 16:01:11.749	TCP Rx-SIP	LIE door	not retransmit "INVITE"	1-13 16:04:40.844 LT		Security protected NA			
				1-13 16:04:40.844 LT			Security protected NAS message		
15-11-13 16:01:11.749	TCP Tx-SIP	TCP	18065, 45422, 5060, Seq:1042/31098, Ack:13/8043/83	2015-11-13 16:04:40.891	TCP Rx-SIP		30447, 5060, 60548, Seg:3608402649, Ack:3183855278, SIP/2.0 183 Session F		
15-11-13 16:01:11.794	TCP Rx-SIP	TCP	0, 5060, 45422, Seg:1378043782, Ack:0	2015-11-13 16:04:40.891	TCP Tx-SIP		64983, 60548, 5060, Seq:3183855278, Ack:3608404001		
15-11-13 16:01:11.794	TCP Rx-SIP	TCP	0, 5060, 45422, Seg:1378043782, Ack:0	2015-11-13 16:04:40.962	TCP Tx-SIP		64984, 60548, 5060, Seq:3183855278, Ack:3608404001, PRACK sip:hr1pcfx02(SIP 200 OK (Prack)
	1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1			2015-11-13 16:04:41.083	TCP Rx-SIP		30448, 5060, 60548, Seq:3608404001, Ack:3183856264		180 Ringing
15-11-13 16:01:11.794	TCP Rx-SIP	TCP	0, 5060, 45422, Seq:1378043782, Ack:0	2015-11-13 16:04:41.083	TCP Rx-SIP		30449, 5060, 60548, Seq:3608404001, Ack:3183856264, SIP/2.0 200 OK		
15-11-13 16:01:11.794	TCP Rx-SIP	TCP	0, 5060, 45422, Seg:1378043782, Ack:0	2015-11-13 16:04:41.093 2015-11-13 16:04:41.126	TCP Tx-SIP TCP Rx-SIP		64985, 60548, 5060, Seq:3183856264, Ack:3608404393 30450, 5060, 60548, Seq:3608404393, Ack:3183856264, SIP/2,0 200 OK		4
	1100 100 100 100 V			2015-11-13 16:04:41.127	TCP Tx-SIP		64986, 60548, 5060, Seq;3806404333, ACX;3163836264, SIF/2.0 200 OR		SIP 200 OK (Invite)
15-11-13 16:01:11.794	TCP Rx-SIP	TCP	0, 5060, 45422, Seq:1378043783, Ack:0	2015-11-13 16:04:41.176	TCP Tx-SIP		64987, 60548, 5060, Seq.3183856264, Ack:3608405389, ACK sip:hr1pcfx0201q	complete 🧶	f*
				2015-11-13 16:04:41.243	TCP Rx-SIP		30451, 5060, 60548, Seq:3608405389, Ack:3183857150		
				2015-11-13 16:04:43.72	Tx Voice Call		First RTP	1	l .
				2015-11-13 16:04:43.72	Tx Voice Call	Traffic Start	751.55 750.5		
				2015 11 12 16:04:52 75	Tu Voice Call	C			

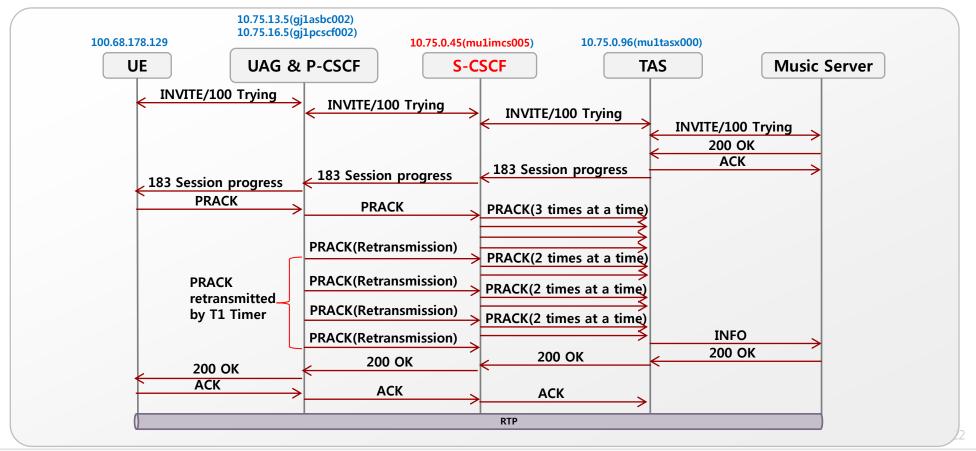
RRC Reconf. AM Bearer (QCI 5) **IMS**

SIP Invite (SDP offer)

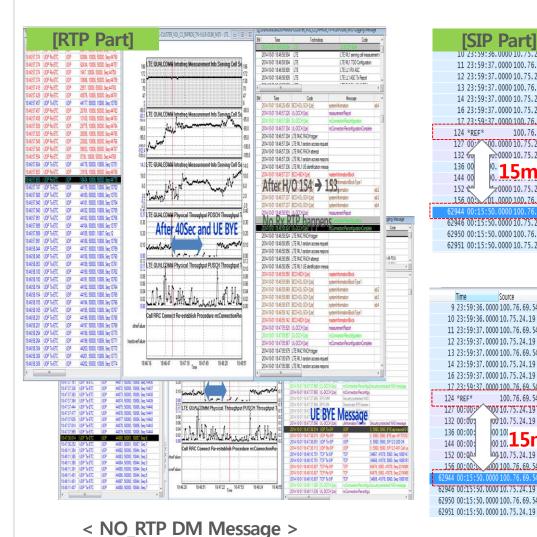
04. Event Cases – Call setup

200OK for PRACK is delayed

Test Date	GJ-GDGR-0632 15-05-2016 17:55:27 ~ 17:55:50					
Reason	S-CSCF didn't send PRACK normally to TAS during 8 sec, so 2000K for PRACK is delayed IMC had problem between 12th~ 20th May, that problem is resolved by removing its routing table.					



Event Case – NO RTP



Source Destination 9 23:59:36.0000 100.76.69.54 10.75.24.19 1020 Request: REGISTER sip:ims.mnc869.mcc405.3gppnetwork.org 100.76.69.54 606 Status: 401 Unauthorized (0 bindings) 10 23:59:36.0000 10.75.24.19 SIP 11 23:59:37,0000 100,76,69,54 10,75,24,19 1117 Request: REGISTER sip:ims.mnc869.mcc405.3qppnetwork.org 100.76.69.54 12 23:59:37.0000 10.75.24.19 890 Status: 200 OK (1 bindings) 13 23:59:37.0000 100.76.69.54 10.75.24.19 955 Request: SUBSCRIBE sip:+917010075115@ims.mnc869.mcc405.3gppnetwork.org 14 23:59:37.0000 10.75.24.19 100.76.69.54 609 Status: 200 OK 100.76.69.54 16 23:59:37.0000 10.75.24.19 617 Request: NOTIFY sip:+917010075115@100.76.69.54:5060 10.75.24.19 17 23:59:37,0000 100,76,69,54 100.76.69.54 124 *REF 10.75.24.19 419 Request: INVITE tel:55555; phone-context=ims.mnc869.mcc405.3gppnetwork.org 127 00:00 00 10.75.24.19 100.76.69.54 369 Status: 100 Trying 100 10.75.24.19 100.76.69.54 132 00:00; 136 00:00 15min 50sec (Invite **Bye)** 00 0K 144 00:00 152 00:QQ 100, 76, 69, 54 000 100, 76, 69, 54 10.75.24.19 594 Request: ACK_sip:mavodi-0-112-cae-1-fffffff0-@10.75.24.19:5060:mavsipodi-0-11b-9-1-1a91 100.76.69.54 1085 Request: BYE sip:+917010075113@100.76.69.54:5060

498 Status: 200 OK

< 50sec NO_RTP Bye Case >

454 Status: 481 Call Leg/Transaction Does Not Exist

1020 Request: REGISTER sip:ims.mnc869.mcc405.3gppnetwork.org

1117 Request: REGISTER sip:ims.mnc869.mcc405.3gppnetwork.org |

617 Request: NOTIFY sip:+917010075115@100.76.69.54:5060

1085 Request: BYE sip:+917010075115@100.76.69.54:5060

454 Status: 481 Call Leg/Transaction Does Not Exist

955 Request: SUBSCRIBE sip:+917010075115@ims.mnc869.mcc405.3gppnetwork.org

419 Request: INVITE tel:55555; phone-context=ims.mnc869.mcc405.3gppnetwork.org

594 Request: ACK sip:mavodi-0-112-cae-1-fffffff0-@10.75.24.19:5060:maysipodi-0-11b-9-1-1a91

606 Status: 401 Unauthorized (0 bindings) |

890 Status: 200 OK (1 bindings)

609 Status: 200 OK

369 Status: 100 Trying

< 50sec NO RTP Bye Case >

1047 Status: 200 OK

1093 Status: 183 Session Progress

10.75.24.19

100.76.69.54

10.75.24.19

100, 76, 69, 54

10.75.24.19

100.76.69.54

10.75.24.19

100,76,69,54

100.76.69.54

10.75.24.19

100.76.69.54

SIP

11 23:59:37,0000 100,76,69,54

12 23:59:37.000010.75.24.19

13 23:59:37.0000 100.76.69.54

14 23:59:37.0000 10.75.24.19

62950 00:15:50.0000 100.76.69.54

62951 00:15:50.0000 10.75.24.19

62950 00:15:50.0000 100.76.69.54

100.76.69.54

00.000010.75.24.19

70000 10.75, 24.19

124 *REF*

O5 Appendix#1 - IMS function Blocks

- P-CSCF (Proxy CSCF): initial contact point of UE, SIP proxy between UE and IMS core
- S-CSCF (Serving CSCF): provide session setup, session tear-down, session control, routing function
- I-CSCF (Interrogating CSCF): I-CSCF is the contact point within an operator's network for all connections destined to a user of that network
- TAS (Telephony Application Server): TAS provide supporting for a minimum set of mandatory Multimedia Telephony services
- MRF (Media Resource Function): provide media plane processing independent of application types (eg transcoding, multiparty conference, network announcements/tone etc.
- IBCF/TrGW (Interconnection Border Control Function/Transaction Gateway): interconnection point to other PMNs.
- IMS-ALG/IMS-AGW (IMS Application Level Gateway/IMS Access Gateway): It provides functions for Gate Control & Local NAT, IP realm indication and availability, Remote NAT traversal support, Traffic Policing, QoS Packet Marking, IMS Media Plane Security, etc.
- MGCF/IMS-MGW (Media Gateway Control Function/IMS Media Gateway) : interconnecting point to circuit-switched network
- BGCF (Breakout Gateway Control Function): determining the next hop for routing SIP message (select proper MGCF for CS domain)
- SEG (Security Gateway): originate and terminate secure associations between eNB and EPS bye IPsec
- ENUM (E.164 Number Mapping): translate E.164 number to SIP URIs using DNS to enable message routing of IMS sessions
- Diameter Agent: enabling the seamless communication and control of information between network elements within LTE or IMS networks and across network borders

05 Appendix#2 - SIP code list

100 Trying

180 Ringing

181 Call Forwarding

182 Queued

183 Session Progress

199 Early Dialog Terminated

200 OK

202 Accepted

204 No Notification

300 Multiple Choices

301 Moved Permanently

302 Moved Temporarily

305 Use Proxy

380 Alternative Service

400 Bad Request

401 Unauthorized

402 Payment Requirement

403 Forbidden

404 Not Found

405 Bad Method

406 Not Acceptable

407 Proxy Authentication Required

408 Request Timeout

410 Gone

412 Conditional Request Failed

413 Request Entity Too Large

414 Request-URI Too Long

415 Unsupported Content

417 Unknown Resource-Priority

420 Bad Extensions

421 Extension Required

422 Session Interval Too Small

423 Interval Too Brief

424 Bad Location Information

428 Use Identity Header

429 Provide Referrer Identity

430 Flow Failed

433 Anonymity Disallowed

436 Bad Identity-Info

437 Unsupported Certificate

438 Invalid Identity Header

439 First Hop Lacks outbound Support

440 Max-Breadth Exceed

469 Bad Info Package

470 Consent Needed

480 Temporally Unavailable

481 Call/Transaction Does Not Exist

482 Loop detected

493 Undecipherable

494 security Agreement Required

483 Too Many Hops

484 Address Incomplete

485 Ambiguous

486 Busy Here

487 Request Terminated

488 Not Acceptable Here

489 Bad Event

491 Request Pending

500 Server Error

501 Not Implemented

503 Unavailable

504 Timeout

600 Busy Everywhere

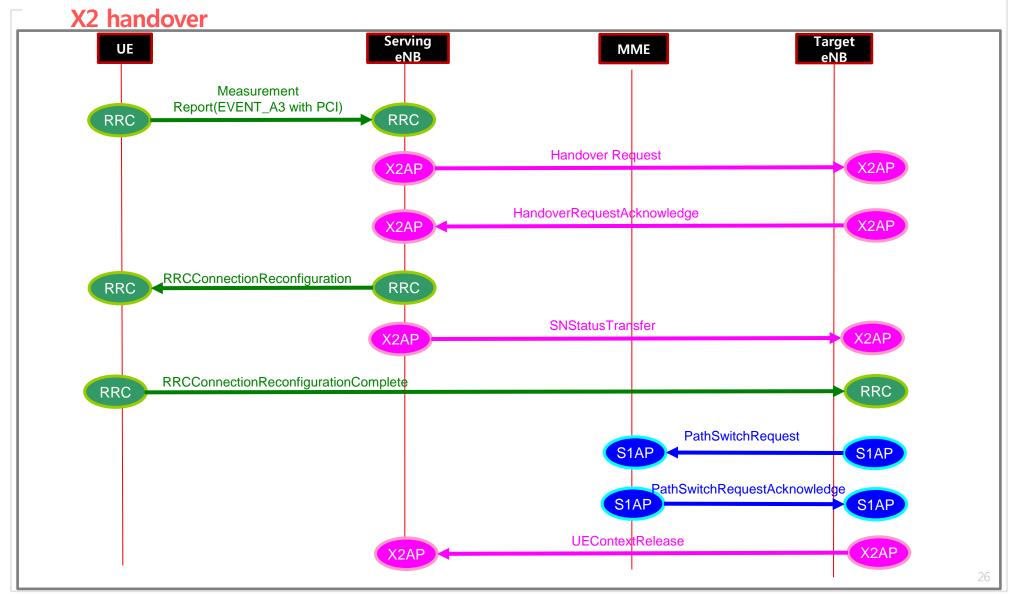
603 Decline

604 Doesn't Exist

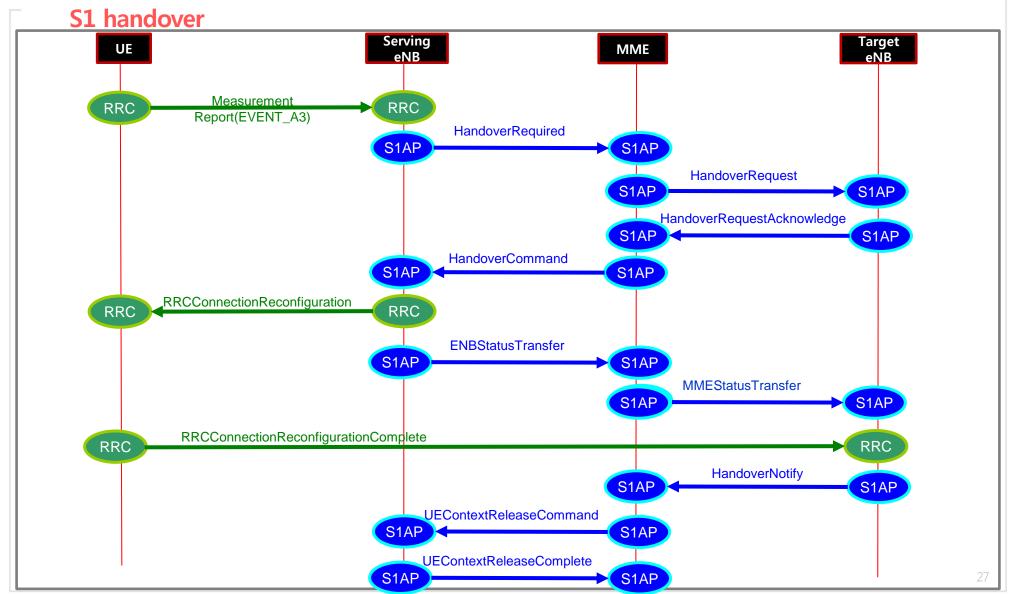
606 Not Acceptable

- 0

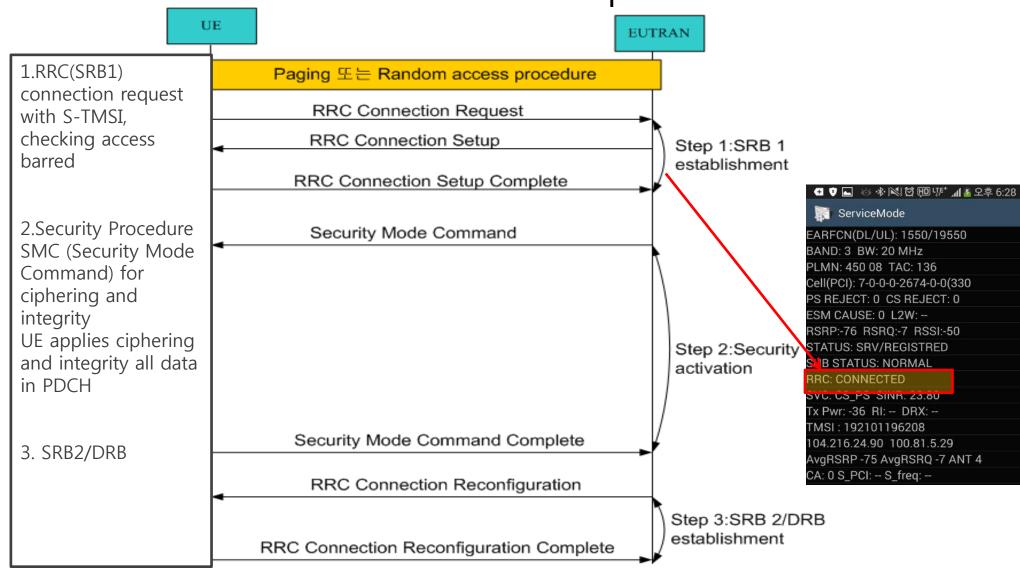
05 Appendix#3



O5 Appendix#4



RRC connection setup flow



Thank you



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