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Universal Mobile Telecommunications System (UMTS);
LTE;
Interworking between the IP Multimedia (IM) Core Network (CN)
subsystem and Circuit Switched (CS) networks
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650 Route des Lucioles
F-06921 Sophia Antipolis Cedex - FRANCE

Tel.: +33 4 92 94 42 00 Fax: +33 4 93 65 47 16

Siret N° 348 623 562 00017 - NAF 742 C
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Contents

Intellectual Property Rights	2
Foreword.....	2
Modal verbs terminology.....	2
Foreword.....	13
1 Scope	14
2 References	14
3 Definitions and abbreviations.....	19
3.1 Definitions	19
3.2 Abbreviations	19
4 General	21
4.1 General interworking overview	21
5 Network characteristics	21
5.1 Key characteristics of ISUP/BICC based CS networks	21
5.2 Key characteristics of IM CN subsystem	22
6 Interworking with CS networks	22
6.1 Interworking reference model	22
6.1.1 Interworking reference points and interfaces	22
6.1.2 Interworking functional entities	23
6.1.2.1 Signalling Gateway Function (SGW)	23
6.1.2.2 Media Gateway Control Function (MGCF)	23
6.1.2.3 IP Multimedia - Media Gateway Function (IM-MGW)	23
6.2 Control plane interworking model.....	23
6.3 User plane interworking model	23
7 Control plane interworking	23
7.1 General	24
7.2 Interworking between CS networks supporting ISUP and the IM CN subsystem	24
7.2.1 Services performed by network entities in the control plane	25
7.2.1.1 Services performed by the SS7 signalling function	25
7.2.1.2 Services of the SGW	25
7.2.1.3 Services of the MGCF.....	25
7.2.1.4 Services of the SIP signalling function	25
7.2.2 Signalling interactions between network entities in the control plane.....	25
7.2.2.1 Signalling between the SS7 signalling function and MGCF	25
7.2.2.1.1 Signalling from MGCF to SS7 signalling function	25
7.2.2.1.2 Signalling from SS7 signalling function to MGCF	26
7.2.2.1.3 Services offered by SCTP and M3UA.....	26
7.2.2.2 Signalling between the MGCF and SIP signalling function.....	26
7.2.3 SIP-ISUP protocol interworking	26
7.2.3.1 Incoming call interworking from SIP to ISUP at I-MGCF	26
7.2.3.1.1 Sending of IAM.....	26
7.2.3.1.2 Coding of the IAM	27
7.2.3.1.2A Coding of the IAM when Number Portability is supported.....	38
7.2.3.1.2B Coding of the IAM for Carrier Routeing.....	40
7.2.3.1.3 Sending of COT.....	41
7.2.3.1.3A Sending of SAM.....	41
7.2.3.1.4 Sending of 180 ringing	42
7.2.3.1.4A Sending of 183 Session Progress for early media scenarios.....	45
7.2.3.1.4B Sending of 181Call is being forwarded	47
7.2.3.1.4C Sending of 183 Session Progress for overlap signalling using the in-dialog method	48
7.2.3.1.5 Sending of the 200 OK (INVITE).....	48
7.2.3.1.6 Sending of the Release message (REL).....	50

7.2.3.1.7	Coding of the REL.....	51
7.2.3.1.8	Receipt of the Release Message	52
7.2.3.1.9	Receipt of RSC, GRS or CGB (H/W oriented)	54
7.2.3.1.9a	Receipt of REFER	55
7.2.3.1.10	Autonomous Release at I-MGCF	55
7.2.3.1.11	Internal through connection of the bearer path.....	56
7.2.3.2	Outgoing Call Interworking from ISUP to SIP at O-MGCF.....	56
7.2.3.2.1	Sending of INVITE	56
7.2.3.2.1a	Sending of INVITE without determining the end of address signalling.....	59
7.2.3.2.2	Coding of the INVITE.....	61
7.2.3.2.2A	Coding of the INVITE when Number Portability is supported	71
7.2.3.2.2B	Coding of the INVITE for Carrier Routeing	74
7.2.3.2.3	Receipt of CONTINUITY	74
7.2.3.2.4	Sending of ACM and awaiting answer indication	75
7.2.3.2.5	Coding of the ACM.....	77
7.2.3.2.6	Sending of the Call Progress message (CPG).....	80
7.2.3.2.7	Coding of the CPG	82
7.2.3.2.7a	Receipt of 200 OK (INVITE).....	83
7.2.3.2.7b	Internal through connection of the bearer path.....	83
7.2.3.2.8	Sending of the Answer Message (ANM).....	83
7.2.3.2.9	Coding of the ANM.....	84
7.2.3.2.10	Sending of the Connect message (CON)	86
7.2.3.2.11	Coding of the CON.....	86
7.2.3.2.11A	Receipt of a reINVITE request.....	86
7.2.3.2.12	Receipt of Status Codes 4xx, 5xx or 6xx.....	86
7.2.3.2.13	Receipt of a BYE.....	89
7.2.3.2.14	Receipt of the Release Message	89
7.2.3.2.15	Receipt of RSC, GRS or CGB (H/W oriented)	89
7.2.3.2.16	Autonomous Release at O-MGCF.....	89
7.2.3.2.17	Special handling of 580 precondition failure received in response to either an INVITE or UPDATE	90
7.2.3.2.18	Sending of CANCEL.....	90
7.2.3.2.19	Receipt of SIP redirect (3xx) response.....	91
7.2.3.2.20	Sending of INFO for overlap signalling using the in-dialog method	91
7.2.3.3	Timers	92
7.3	Interworking between CS networks supporting BICC and the IM CN subsystem.....	92
7.3.1	Services performed by network entities in the control plane	93
7.3.2	Signalling interactions between network entities in the control plane.....	93
7.3.2.1	Signalling between the SS7 signalling function and MGCF.....	93
7.3.2.1.1	Signalling from MGCF to SS7 signalling function	93
7.3.2.1.2	Signalling from SS7 signalling function to MGCF	94
7.3.2.1.3	Services offered by STC, SCTP and M3UA	94
7.3.2.2	Signalling between the MGCF and SIP signalling function.....	94
7.3.3	SIP-BICC protocol interworking	94
7.3.3.1	Incoming call interworking from SIP to BICC at I-MGCF.....	94
7.3.3.1.1	Sending of IAM.....	94
7.3.3.1.2	Coding of IAM	95
7.3.3.1.2A	Coding of the IAM when Number Portability is supported.....	96
7.3.3.1.2B	Coding of the IAM for Carrier Routeing	96
7.3.3.1.3	Sending of COT.....	96
7.3.3.1.3A	Sending of SAM	96
7.3.3.1.4	Sending of 180 Ringing.....	96
7.3.3.1.4A	Sending of 183 Session Progress for early media scenarios.....	96
7.3.3.1.4B	Sending of 181 Call is being forwarded	96
7.3.3.1.4C	Sending of 183 Session Progress for overlap signalling using the in-dialog method	96
7.3.3.1.5	Sending of the 200 OK (INVITE)	97
7.3.3.1.6	Sending of the Release message (REL).....	97
7.3.3.1.7	Coding of the REL.....	97
7.3.3.1.8	Receipt of the Release Message	97
7.3.3.1.9	Receipt of RSC, GRS or CGB (H/W oriented)	97
7.3.3.1.9a	Receipt of REFER	97
7.3.3.1.9b	Autonomous Release at I-MGCF	97

7.3.3.1.10	Internal through connection of the bearer path.....	97
7.3.3.1.11	Out of Band DTMF	97
7.3.3.2	Outgoing Call Interworking from BICC to SIP at O-MGCF	97
7.3.3.2.1	Sending of INVITE	97
7.3.3.2.1a	Sending of INVITE without determining the end of address signalling.....	98
7.3.3.2.2	Coding of the INVITE.....	98
7.3.3.2.2A	Coding of the INVITE when number portability is supported	99
7.3.3.2.2B	Coding of the INVITE for Carrier Routeing	99
7.3.3.2.3	Sending of UPDATE.....	99
7.3.3.2.4	Sending of ACM and Awaiting Answer indication.....	100
7.3.3.2.5	Coding of the ACM	100
7.3.3.2.6	Sending of the Call Progress message (CPG).....	100
7.3.3.2.7	Coding of the CPG	100
7.3.3.2.7a	Receipt of 200 OK (INVITE).....	100
7.3.3.2.7b	Internal through connection of the bearer path.....	100
7.3.3.2.8	Sending of the Answer Message (ANM).....	100
7.3.3.2.9	Coding of the ANM.....	100
7.3.3.2.10	Sending of the Connect message (CON)	100
7.3.3.2.11	Coding of the CON.....	100
7.3.3.2.11A	Receipt of re-INVITE requests.....	101
7.3.3.2.12	Receipt of Status Codes 4xx, 5xx or 6xx.....	101
7.3.3.2.13	Receipt of a BYE.....	101
7.3.3.2.14	Receipt of the Release Message	101
7.3.3.2.15	Receipt of RSC, GRS or CGB (H/W oriented)	101
7.3.3.2.16	Out of Band DTMF	101
7.3.3.2.17	Sending of CANCEL.....	101
7.3.3.2.18	Autonomous Release at O-MGCF.....	101
7.3.3.2.19	Special handling of 580 precondition failure received in response to either an INVITE or UPDATE	101
7.3.3.2.20	Receipt of SIP redirect (3xx) response	101
7.3.3.2.21	Sending of INFO for overlap signalling using the in-dialog method	102
7.3.3.3	Timers	102
7.4	Supplementary Services	102
7.4.1	Calling line identification presentation/restriction (CLIP/CLIR)	102
7.4.2	Connected line presentation and restriction (COLP/COLR).....	102
7.4.2.0	General	102
7.4.2.1	Incoming Call Interworking From SIP to BICC/ISUP At The I-MGCF.....	102
7.4.2.1.1	INVITE to IAM interworking (SIP to ISUP/BICC calls).....	102
7.4.2.1.2	ANM/CON to 200 OK (INVITE)	102
7.4.2.2	Outgoing Call Interworking from BICC/ISUP to SIP at O-MGCF	103
7.4.2.2.1	IAM to INVITE interworking (ISUP to SIP calls)	103
7.4.2.2.2	1XX to ANM or CON interworking.....	104
7.4.2.2.3	200 OK (INVITE) to ANM/CON interworking.....	104
7.4.3	Direct Dialling In (DDI)	105
7.4.4	Malicious call identification	105
7.4.5	Subaddressing (SUB).....	105
7.4.5.1	General	105
7.4.5.2	Interworking at I-MGCF	105
7.4.5.3	Interworking at O-MGCF	106
7.4.6	Call Forwarding Busy (CFB)/ Call Forwarding No Reply (CFNR) / Call Forwarding Unconditional (CFU).....	107
7.4.6.1	General	107
7.4.6.2	Interworking at the O-MGCF.....	108
7.4.6.2.1	General	108
7.4.6.2.2	Interworking SIP to ISUP	108
7.4.6.2.3	Interworking ISUP to SIP	111
7.4.6.3	Interworking at the I-MGCF	113
7.4.6.3.1	General	113
7.4.6.3.2	Interworking from SIP to ISUP	113
7.4.6.3.3	Interworking from ISUP to SIP	115
7.4.7	Call Deflection (CD).....	121
7.4.8	Explicit Call Transfer (ECT)	121

7.4.9	Call Waiting	121
7.4.10	Call Hold.....	121
7.4.10.1	Session hold initiated from the IM CN subsystem side.....	121
7.4.10.2	Session hold initiated from the CS network side.....	122
7.4.11	Call Completion on busy subscriber	123
7.4.12	Completion of Calls on No Reply (CCNR)	124
7.4.13	Terminal Portability (TP).....	124
7.4.14	Conference calling (CONF) / Three-Party Service (3PTY).....	124
7.4.15	Void	124
7.4.16	Closed User Group (CUG)	124
7.4.17	Multi-Level Precedence and Pre-emption (MLPP).....	124
7.4.18	Global Virtual Network Service (GVNS).....	124
7.4.19	International telecommunication charge card (ITCC)	124
7.4.20	Reverse charging (REV).....	125
7.4.21	User-to-User Signalling (UUS).....	125
7.4.21.0	General	125
7.4.21.1	User-to-User Signalling (UUS) service 1 (implicit).....	125
7.4.21.1.0	General	125
7.4.21.1.1	Void	125
7.4.21.1.2	User-to-user information Interworking from SIP to ISUP	125
7.4.21.1.3	User-to-user information Interworking from ISUP to SIP	125
7.4.21.2	User-to-User Signalling (UUS) service 1 (explicit).....	126
7.4.21.3	User-to-User Signalling (UUS) service 2 (explicit).....	126
7.4.21.4	User-to-User Signalling (UUS) service 3 (explicit).....	126
7.4.22	Multiple Subscriber Number (MSN)	126
7.4.23	Anonymous Call rejection	126
7.4.23.1	ISUP-SIP protocol interworking at the I-MGCF.....	126
7.4.23.2	SIP-ISUP protocol interworking at the O-MGCF.....	126
7.5	IMS Supplementary Services	126
7.5.1	Originating Identification Presentation (OIP) and Originating Identification Restriction (OIR).....	126
7.5.2	Terminating Identification Presentation (TIP) and Terminating Identification Restriction (TIR).....	127
7.5.2.1	General	127
7.5.2.2	Interworking at the O-MGCF.....	127
7.5.2.3	Interworking at the I-MGCF	128
7.5.2.4	Timer.....	129
7.5.3	(void)	129
7.5.4	Communication Diversion (CDIV).....	129
7.5.4.1	General	129
7.5.4.2	Interworking at the O-MGCF.....	130
7.5.4.2.1	General	130
7.5.4.2.2	Call forwarding within the ISUP Network appeared.....	134
7.5.4.3	Interworking at the I-MGCF	136
7.5.5	Communication Hold (HOLD).....	143
7.5.6	Conference call (CONF).....	143
7.5.6.1	General	143
7.5.6.2	Subscribing for the conference event package	143
7.5.6.3	Interworking the notification.....	143
7.5.7	Anonymous Communication Rejection (ACR) and Communication Barring (CB).....	144
7.5.8	Message Waiting Indication (MWI)	144
7.5.9	Malicious Communication Identification (MCID)	144
7.5.9.0	General	144
7.5.9.1	Interworking at the O-MGCF.....	145
7.5.9.1.0	General	145
7.5.9.1.1	Interworking of the MCID XML Request schema with the ISUP MCID request indicators	145
7.5.9.1.2	Interworking of the ISUP MCID response indicators with the MCID XML Response schema ...	145
7.5.9.1.3	Interworking of the ISUP Calling Party Number in an Identification Response with the OrigPartyIdentity within the MCID XML Response schema.....	146
7.5.9.1.4	Interworking of the ISUP Generic Number in an Identification Response with the GenericNumber within the MCID XML Response schema	146
7.5.9.2	Interworking at the I-MGCF	146
7.5.9.2.1	General	146
7.5.9.2.2	Interworking of identification Request	146

7.5.9.2.3	Interworking of identification Response.....	146
7.5.10	Closed User Group (CUG)	147
7.5.10.0	General	147
7.5.10.1	Interworking at the I-MGCF	147
7.5.10.2	Interworking at the O-MGCF.....	148
7.5.11	CCBS/CCNR	149
7.5.11.0	General	149
7.5.11.1	Interworking at the I-MGCF	149
7.5.11.2	Interworking at the O-MGCF.....	151
7.5.12	Communication Waiting (CW).....	153
7.5.12.0	General	153
7.5.12.1	Interworking at the I-MGCF	153
7.5.12.2	Interworking at the O-MGCF.....	153
8	User plane interworking	153
8.1	Interworking between IM CN subsystem and bearer independent CS network	153
8.1.1	Transcoder-less Mb to Nb Interworking	154
8.1.1.1	Initialisation	154
8.1.1.2	Time alignment	154
8.1.1.3	Rate control	155
8.1.1.4	Frame quality indication	155
8.1.1.5	Framing	155
8.1.1.6	Transcoding.....	155
8.1.1.7	Discontinuous transmission	155
8.1.1.8	Timing and sequence information.....	156
8.2	Interworking between IM CN subsystem and TDM-based CS network.....	156
8.3	Transcoding requirements	156
8.4	Diffserv code point requirements	156
8.5	DTMF handling	157
9	MGCF – IM-MGW Interaction	157
9.1	Overview	157
9.2	Mn signalling interactions	157
9.2.1	Network model	157
9.2.2	Basic IM CN subsystem originated session.....	158
9.2.2.1	BICC forward bearer establishment.....	158
9.2.2.1.1	IM-MGW selection	158
9.2.2.1.2	CS network side bearer establishment.....	158
9.2.2.1.3	IM CN subsystem side termination reservation.....	158
9.2.2.1.4	IM CN subsystem side session establishment	159
9.2.2.1.5	Through-connection	159
9.2.2.1.6	Codec handling.....	159
9.2.2.1.7	Failure handling in MGCF	159
9.2.2.1.8	Message sequence chart.....	160
9.2.2.2	BICC backward bearer establishment	162
9.2.2.2.1	IM-MGW selection	162
9.2.2.2.2	IM CN subsystem side termination reservation.....	162
9.2.2.2.3	IM CN subsystem side session establishment	162
9.2.2.2.4	CS network side bearer establishment.....	163
9.2.2.2.5	Through-connection	163
9.2.2.2.6	Codec handling.....	163
9.2.2.2.7	Failure handling in MGCF	163
9.2.2.2.8	Message sequence chart.....	163
9.2.2.3	ISUP	164
9.2.2.3.1	IM-MGW selection	164
9.2.2.3.2	IM CN subsystem side termination reservation.....	165
9.2.2.3.3	IM CN subsystem side session establishment	165
9.2.2.3.4	CS network side circuit reservation.....	165
9.2.2.3.5	Through-connection	165
9.2.2.3.6	Continuity check.....	166
9.2.2.3.7	Codec handling.....	166
9.2.2.3.8	Voice processing function	166

9.2.2.3.9	Failure handling in MGCF	166
9.2.2.3.10	Message sequence chart.....	166
9.2.3	Basic CS network originated session	168
9.2.3.1	BICC forward bearer establishment	168
9.2.3.1.1	IM-MGW selection	168
9.2.3.1.2	IM CN subsystem side termination reservation.....	168
9.2.3.1.3	IM CN subsystem side session establishment	168
9.2.3.1.4	CS network side bearer establishment	168
9.2.3.1.5	Called party alerting	169
9.2.3.1.6	Called party answer	169
9.2.3.1.7	Through-Connection.....	169
9.2.3.1.8	Codec handling.....	169
9.2.3.1.9	Failure handling in MGCF	169
9.2.3.1.10	Message sequence chart.....	169
9.2.3.2	BICC Backward bearer establishment	171
9.2.3.2.1	IM-MGW selection	171
9.2.3.2.2	CS network side bearer establishment	171
9.2.3.2.3	IM CN subsystem side termination reservation.....	171
9.2.3.2.4	IM CN subsystem side session establishment	172
9.2.3.2.5	Called party alerting	172
9.2.3.2.6	Called party answer	172
9.2.3.2.7	Through-Connection.....	172
9.2.3.2.8	Codec handling.....	172
9.2.3.2.9	Failure handling in MGCF	173
9.2.3.2.10	Message sequence chart.....	173
9.2.3.3	ISUP	175
9.2.3.3.1	IM-MGW selection	175
9.2.3.3.2	CS network side circuit reservation.....	175
9.2.3.3.3	IM CN subsystem side termination reservation.....	175
9.2.3.3.4	IM CN subsystem side session establishment	175
9.2.3.3.5	Called party alerting	176
9.2.3.3.6	Called party answer	176
9.2.3.3.7	Through-Connection.....	176
9.2.3.3.8	Continuity Check.....	176
9.2.3.3.9	Codec handling.....	176
9.2.3.3.10	Voice Processing function	177
9.2.3.3.11	Failure handling in MGCF	177
9.2.3.3.12	Message sequence chart.....	177
9.2.3.4	Handling of Forking	179
9.2.3.4.1	Detection of Forking	179
9.2.3.4.2	IM CN subsystem side session establishment	179
9.2.3.4.3	IM CN subsystem side session establishment completion	180
9.2.3.4.4	Message sequence chart.....	180
9.2.4	Session release initiated from IM CN subsystem side	183
9.2.4.1	BICC	183
9.2.4.1.1	Session release in the IM CN subsystem side.....	183
9.2.4.1.2	Session release in the CS network side.....	183
9.2.4.1.3	Message sequence chart.....	183
9.2.4.2	ISUP	184
9.2.4.2.1	Session release in the IM CN subsystem side.....	184
9.2.4.2.2	Session release in the CS network side.....	184
9.2.4.2.3	Message sequence chart.....	184
9.2.5	Session release initiated from CS network side	184
9.2.5.1	BICC	184
9.2.5.1.1	Session release in the CS network side.....	184
9.2.5.1.2	Session release in the IM CN subsystem side.....	185
9.2.5.1.3	Message sequence chart.....	185
9.2.5.2	ISUP	185
9.2.5.2.1	Session release in the CS network side.....	185
9.2.5.2.2	Session release in the IM CN subsystem side.....	185
9.2.5.2.3	Message sequence chart.....	186
9.2.6	Session release initiated by MGCF.....	186

9.2.6.1	BICC	186
9.2.6.1.1	Session release in the CS network side.....	186
9.2.6.1.2	Session release in the IM CN subsystem side.....	186
9.2.6.1.3	Message sequence chart.....	186
9.2.6.2	ISUP.....	187
9.2.6.2.1	Session release in the CS network side.....	187
9.2.6.2.2	Session release in the IM CN subsystem side.....	187
9.2.6.2.3	Message sequence chart.....	187
9.2.7	Session release initiated by IM-MGW.....	188
9.2.7.1	BICC	188
9.2.7.1.1	Session release in the CS network side.....	188
9.2.7.1.2	Session release in the IM CN subsystem side.....	188
9.2.7.1.3	Message sequence chart.....	188
9.2.7.2	ISUP	189
9.2.7.2.1	Session release in the CS network side.....	189
9.2.7.2.2	Session release in the IM CN subsystem side.....	189
9.2.7.2.3	Message sequence chart.....	190
9.2.8	Handling of RTP telephone events	190
9.2.8.1	Sending DTMF digits out-of-band to CS CN (BICC).....	191
9.2.8.2	Sending and receiving DTMF digits inband to/from CS CN (ISUP or BICC)	193
9.2.8.3	Receiving DTMF digits out-of-band from CS CN (BICC)	194
9.2.9	Session hold initiated from IM CN subsystem	196
9.2.10	Session hold initiated from CS network	197
9.3	Mn Signalling procedures	198
9.3.1	Procedures related to terminations towards the IM CN Subsystem.....	198
9.3.1.1	Reserve IMS connection point	199
9.3.1.2	Configure IMS resources	199
9.3.1.3	Reserve IMS Connection point and configure remote resources	200
9.3.1.4	Release IMS termination	201
9.3.1.5	Detect IMS RTP Tel event	202
9.3.1.6	Notify IMS RTP Tel event.....	202
9.3.1.7	Void.....	202
9.3.1.8	Send IMS RTP Tel event	202
9.3.1.9	Stop IMS RTP Tel event	202
9.3.1.10	Termination heartbeat indication	202
9.3.1.11	IMS Bearer Released.....	202
9.3.1.12	End IMS RTP Tel event	203
9.3.1.13	IMS Send Tone	203
9.3.1.14	IMS Stop Tone	203
9.3.1.15	IMS Tone Completed.....	203
9.3.2	Procedures related to a termination towards an ISUP network.....	203
9.3.2.1	Reserve TDM circuit.....	203
9.3.2.2	Change TDM through-connection	204
9.3.2.3	Activate TDM voice-processing function	204
9.3.2.4	Send TDM tone.....	204
9.3.2.5	Stop TDM tone.....	204
9.3.2.6	Play TDM announcement	204
9.3.2.7	TDM announcement completed	204
9.3.2.8	Stop TDM announcement	204
9.3.2.9	Continuity check	204
9.3.2.10	Continuity check verify.....	205
9.3.2.11	Continuity check response	205
9.3.2.12	Release TDM termination	205
9.3.2.13	Termination Out-of-Service	206
9.3.2.14	Termination heartbeat indication	206
9.3.2.15	Bearer Released.....	206
9.3.2.16	TDM tone completed	206
9.3.3	Procedures related to a termination towards a BICC network.....	206
9.3.4	Non-call related procedures	207
9.3.5	Multiple IP Realms	208

Annex A (informative):	Summary of differences items between TS 29.163 and ITU-T Q.1912.5	210
A.1	List of differences.....	210
Annex B (normative):	Codec Negotiation between a BICC CS network and the IM CN subsystem.....	211
B.1	Introduction	211
B.2	Control plane interworking	211
B.2.1	Incoming call interworking from SIP to BICC at I-MGCF.....	211
B.2.1.1	Sending of IAM	211
B.2.1.2	Sending of SDP answer	211
B.2.2	Outgoing call interworking from BICC to SIP at O-MGCF	211
B.2.2.1	Sending of INVITE.....	211
B.2.2.2	Responding to serving node initiating codec negotiation	212
B.2.3	Mid-call interworking from SIP to BICC at I-MGCF or O-MGCF	212
B.2.3.1	Receipt of SDP offer.....	212
B.2.3.2	Generating SDP answer	213
B.2.4	Mid-call interworking from BICC to SIP at I-MGCF or O-MGCF	213
B.2.4.1	Receipt of mid-call codec negotiation request.....	213
B.2.4.2	Responding to serving node initiating mid-call codec negotiation	213
B.2.4.3	Receipt of codec modification request.....	214
B.2.5	Codec parameter translation between BICC CS network and the IM CN subsystem	214
B.2.5.1	Codec parameters for 3GPP AMR-NB codecs	215
B.2.5.2	Codec parameters for 3GPP AMR-WB codecs	216
B.2.5.3	Codec parameters for 3GPP non-AMR codecs	218
B.2.5.4	Codec parameters for ITU-T codecs.....	218
B.3	MGCF – IM-MGW interaction during interworking of codec negotiation.....	219
B.3.1	Basic IM CN subsystem originated session	219
B.3.1.1	BICC forward bearer establishment	220
B.3.1.1.1	IM-MGW selection	220
B.3.1.1.2	CS network side bearer establishment.....	220
B.3.1.1.3	IM CN subsystem side session establishment	220
B.3.1.1.4	Through-connection	220
B.3.1.1.5	Codec handling	221
B.3.1.1.6	Failure handling in MGCF	221
B.3.1.1.7	Message sequence chart	221
B.3.2	Basic CS network originated session.....	223
B.3.2.1	BICC forward bearer establishment.....	223
B.3.2.1.1	IM-MGW selection	223
B.3.2.1.2	IM CN subsystem side termination reservation	223
B.3.2.1.3	IM CN subsystem side session establishment	223
B.3.2.1.4	CS network side bearer establishment.....	223
B.3.2.1.5	Called party alerting	224
B.3.2.1.6	Called party answer.....	224
B.3.2.1.7	Through-Connection	224
B.3.2.1.8	Codec handling	224
B.3.2.1.9	Failure handling in MGCF	224
B.3.2.1.10	Message sequence chart	224
B.3.3	CS network initiated mid-call codec negotiation.....	227
B.3.4	IM CN subsystem initiated mid-call codec negotiation.....	228
Annex C (normative):	Interworking of CPC parameter	229
C.1	Interworking SIP to ISUP.....	229
C.2	Interworking ISUP to SIP.....	230
Annex D:	Void	231

Annex E (normative):	Multimedia interworking between the IP Multimedia Core Network (CN) Subsystem (IMS) and Circuit Switched (CS) networks	232
E.1	Basic Multimedia calls interworking between the IMS and CS Networks scenarios	232
E.2	Control plane interworking	232
E.2.1	General	232
E.2.2	Functionalities required in the MGCF for multimedia calls support.....	233
E.2.3	IM CN subsystem originated session	233
E.2.3.1	Preconditions used at IMS side.....	233
E.2.3.1.1	Interactions between H.245 or MONA and SIP/SDP	233
E.2.3.2	Preconditions not used at IMS side.....	235
E.2.3.2.1	Interactions between H.245 or MONA and SIP/SDP	235
E.2.3.3	Fallback to speech at session establishment	237
E.2.4	CS network originated session	237
E.2.4.1	Interactions between SIP/SDP and H.245 or MONA	237
E.2.4.1.1	Normal Call setup	237
E.2.4.1.2	Call setup if multimedia call can not be recognized in an unambiguous manner.....	239
E.2.4.2	CS originated - IM CN transit - CS terminated	240
E.2.5	Service change.....	241
E.2.5.2.1	SCUDIF	241
E.2.5.2.1.1	IM CN subsystem originated change.....	241
E.2.5.2.1.2	CS network originated change.....	243
E.2.5.2.2	Non-SCUDIF case (ISUP or BICC without SCUDIF)	244
E.2.5.2.2.1	Change from multimedia to audio.....	244
E.2.5.2.2.2	Change from speech to multimedia	245
E.2.6	Call release	245
E.2.6.1	Call release initiated from the IM CN subsystem side.....	245
E.2.6.2	Call release initiated from the CS network side.....	246
E.2.6.3	Call release initiated from the interworking node	247
E.3	User plane interworking	248
E.3.1	Functionalities required in the IM-MGW for multimedia calls support.....	248
E.4	MGCF and IM-MGW interactions.....	248
E.4.1	Introduction	248
E.4.2	Mn signalling interactions	249
E.4.2.1	Introduction.....	249
E.4.2.2	H.248 Context Model	249
E.4.2.3	Transport of H.245 messages between the MGCF and IM-MGW	249
E.4.2.3.1	General	249
E.4.2.3.2	Transport from MGCF to IM-MGW.....	250
E.4.2.3.3	Transport from IM-MGW to MGCF.....	250
E.4.2.4	Call establishment procedure.....	250
E.4.2.5	Handling of H.245 indication message	253
E.4.2.5.1	Overview	253
E.4.2.5.2	Function Not Understood / Function Not Supported message	253
E.4.2.5.3	User Input Indication message	253
E.4.2.6	Handling of H.245 Command message	254
E.4.2.6.1	Overview	254
E.4.2.6.2	Flow control command	254
E.4.2.6.3	End Session Command	255
E.4.2.6.4	videoFastUpdatePicture	255
E.4.2.7	Mn Signalling Interactions to support the Media Oriented Negotiation Acceleration (MONA).....	255
E.4.2.7.1	Overview.....	255
E.4.2.7.2	Mn Interactions for MONA preference messages.....	257
E.4.2.7.3	Mn Interactions for MONA MPCs.....	259
E.4.2.7.4	Mn Interactions for MONA SPCs.....	261
E.4.2.7.4.1	General	261
E.4.2.7.4.2	Transport from MGCF to IM-MGW	261
E.4.2.7.4.3	Transport from IM-MGW to MGCF	262
E.4.2.7.4.4	Termination of SPC procedure	262
E.4.2.7.5	Mn Interactions for fallback from MONA procedures to standard H.324 setup.....	263

E.4.2.8	Interworking between RTCP messages and H.245 messages.....	264
E.4.2.8.1	Overview.....	264
E.4.2.8.2	IM CN subsystem originated RTCP messages.....	264
E.4.2.8.3	CS network originated H.245 messages.....	266
E.4.3	Mn Signalling procedures	266
E.4.3.1	Overview	266
E.4.3.2	Add Multiplex Termination	267
E.4.3.3	Configure Multiplex Termination.....	268
E.4.3.4	Signal H245 Message	269
E.4.3.5	Notify H245 Message	269
E.4.3.6	Notify MONA Preference Reception.....	270
E.4.3.7	Notify MONA Preference Completed	270
E.4.3.8	Signal SPC.....	271
E.4.3.9	Notify SPC.....	271
E.4.3.10	Notify MPC	272
E.4.3.11	Notify Detection of Legacy Interworking.....	272
E.4.3.12	Request RTCP-Interworking	273
E.4.3.13	Notify RTCP-Interworking.....	273
E.4.3.14	Signal-H.245-Interworking.....	274
E.4.3.15	Stop MPC	274
E.4.3.16	Stop SPC.....	274
E.4.3.17	Stop MONA Negotiation.....	275
Annex F (normative):	PSTN XML Scheme.....	277
F.1	Scope.....	277
F.2	MIME type	277
F.3	XML Schema definition.....	277
F.4	IANA registration template	282
Annex G (informative):	Void.....	285
Annex H (normative):	Interworking of Originating Line Information (OLI) parameter (network option).....	286
H.1	Interworking SIP to ISUP.....	286
H.2	Interworking ISUP to SIP.....	286
Annex I (informative):	Change history	287
History		290