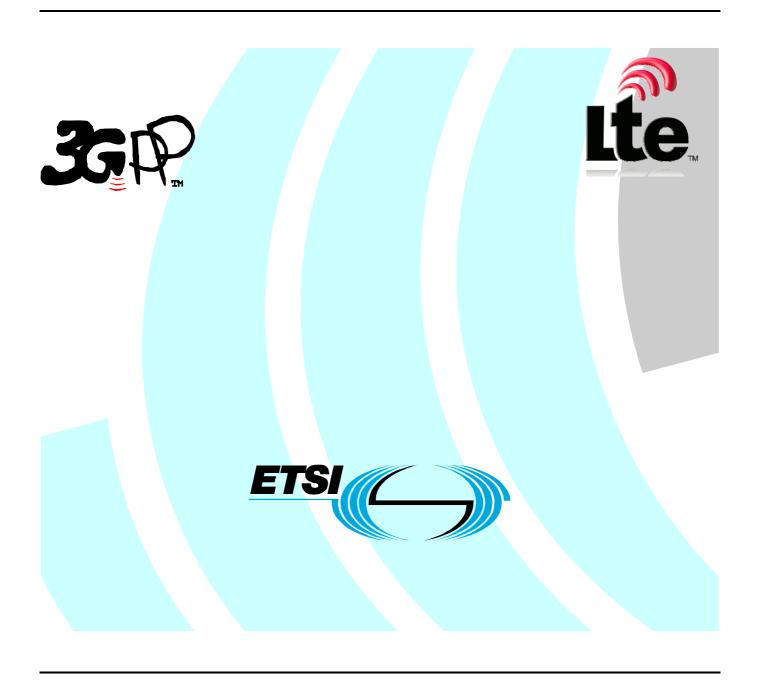
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ETSI

650 Route des Lucioles F-06921 Sophia Antipolis Cedex - FRANCE

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

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1 Scope

The objective of this document is to address the Inter-IMS Network to Network Interface (II-NNI) consisting of Ici and Izi reference points between IMS networks in order to support end-to-end service interoperability.

The present document will address the issues related to control plane signalling (3GPP usage of SIP and SDP protocols, required SIP header fields) as well as other interconnecting aspects like security, numbering/naming/addressing and user plane issues as transport protocol, media and codecs actually covered in a widespread set of 3GPP specifications. A profiling of the Inter-IMS Network to Network Interface (II-NNI) is also provided.

Charging aspects will be addressed as far as SIP signalling is concerned.

2 References

The following documents contain provisions which, through reference in this text, constitute provisions of the present document.

- References are either specific (identified by date of publication, edition number, version number, etc.) or non-specific.
- For a specific reference, subsequent revisions do not apply.
- For a non-specific reference, the latest version applies. In the case of a reference to a 3GPP document (including a GSM document), a non-specific reference implicitly refers to the latest version of that document *in the same Release as the present document*.

[1]	3GPP TR 21.905: "Vocabulary for 3GPP Specifications".
[2]	IETF RFC 791: "Internet Protocol".
[3]	3GPP TS 23.002: "Network architecture".
[4]	3GPP TS 23.228: "IP Multimedia Subsystem (IMS); Stage 2".
[5]	3GPP TS 24.229: "Internet Protocol (IP) multimedia call control protocol based on Session Initiation Protocol (SIP) and Session Description Protocol (SDP); Stage 3".
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[7]	IETF RFC 2460: "Internet Protocol, Version 6 (IPv6) Specification".
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[9]	3GPP TS 22.228: "Service requirements for the IP multimedia core network subsystem".
[10]	3GPP TS 33.210: "3G security; Network Domain Security (NDS); IP network layer security".
[11]	3GPP TS 26.114: "IP Multimedia Subsystem (IMS); Multimedia Telephony; Media handling and interaction".
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[14]	IETF RFC 3966: "The tel URI for Telephone Numbers".
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[45]	IETF RFC 3608: "Session Initiation Protocol (SIP) Extension Header Field for Service Route Discovery During Registration".
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[102] Void.

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subsystem".

[104] Void.

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subsystem".

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[108] Void.

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subsystem".

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[111] Void.

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(CN) subsystem".

[113]	3GPP TS 24.608: "Terminating Identification Presentation (TIP) and Terminating Identification Restriction (TIR) using IP Multimedia (IM); Core Network (CN) subsystem".
[114]	3GPP TS 24.611: "Communication Barring (CB) using IP Multimedia (IM); Core Network (CN) subsystem".
[115]	Void.
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[119]	Void.
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Editor's note: Th	e above document cannot be formally referenced until it is published as an RFC.
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[130]	Void.
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Editor's note: The above document cannot be formally referenced until it is published as an RFC.

3 Definitions, symbols and abbreviations

3.1 Definitions

For the purposes of the present document, the terms and definitions given in 3GPP TR 21.905 [1] and the following apply. A term defined in the present document takes precedence over the definition of the same term, if any, in 3GPP TR 21.905 [1].

example: text used to clarify abstract rules by applying them literally.

IM CN subsystem: (IP Multimedia CN subsystem) comprises of all CN elements for the provision of IP multimedia applications over IP multimedia sessions, as specified in 3GPP TS 22.228 [9].

IP multimedia session: as specified in 3GPP TS 22.228 [9] an IP multimedia session is a set of multimedia senders and receivers and the data streams flowing from senders to receivers. IP multimedia sessions are supported by the IP multimedia CN Subsystem and are enabled by IP connectivity bearers (e.g. GPRS as a bearer). A user can invoke concurrent IP multimedia sessions.

For the purposes of the present document, the following terms and definitions given in 3GPP TS 23.292 [120] apply:

MSC Server enhanced for ICS

3.2 Symbols

For the purposes of the present document, the following symbols apply:

Ici Reference Point between an IBCF and another IBCF or I-CSCF belonging to a different IM CN

subsystem network

Izi Reference Point between a TrGW and another TrGW or media handling node belonging to a

different IM CN subsystem network

Mi Reference Point between a BGCF and CSCF

Mm Reference Point between a CSCF/BGCF/IMS ALG and an IP multimedia network.

Mw Reference Point between a CSCF and another CSCF

Mx Reference Point between a CSCF/BGCF/MSC Server enhanced for ICS and IBCF

3.3 Abbreviations

For the purposes of the present document, the abbreviations given in 3GPP TR 21.905 [1] and the following apply. An abbreviation defined in the present document takes precedence over the definition of the same abbreviation, if any, in 3GPP TR 21.905 [1].

ACR Anonymous Communication Rejection

B2BUA Back 2 Back User Agent

BGCF Breakout Gateway Control Function

CAT Customized Alerting Tone CB Communication Barring

CCBS Completion of Communications to Busy Subscriber

CCNR Communication Completion on No Reply

CDIV Communication Diversion

CDIVN Communication Diversion Notification

CRS Customized Ringing Signal ECT Explicit Communication Transfer

FA Flexible Alerting
HOLD Communication HOLD
CW Communication Waiting

IBCF Interconnection Border Control Function ICB Incoming Communication Barring

ICS IMS Centralized Services
I-CSCF Interrogating CSCF

II-NNI Inter-IMS Network to Network Interface

IM Instant Messaging

IMS-ALG IMS Application Level Gateway

MCID Malicious Communication IDentification
MRFC Media Resource Function Controller
MSRP Message Session Relay Protocol
MWI Message Waiting Indication

NA(P)T-PT Network Address (Port-Multiplexing) Translation-Protocol Translation

NNI Network to Network Interface
OCB Outgoing Communication Barring
OIP Originating Identification Presentation
OIR Originating Identification Restriction

P-CSCF Proxy CSCF

PNM Personal Network Management

PRES Presence

TIP Terminating Identification Presentation
TIR Terminating Identification Restriction

TrGW Transition Gateway

4 Overview

Interconnection between two different IM CN subsystems shall be guaranteed in order to support end-to-end service interoperability. For this purpose, Inter-IMS Network to Network Interface (II-NNI) between two IM CN subsystem networks is adopted, according to the assumptions coming from 3GPP TS 23.002 [3] and 3GPP TS 23.228 [4].

Aiming to support the delivery of IMS services between two separated IM CN subsystems, protocol interconnection has to occur:

- at a control plane level, in order that IMS procedures can be supported. In this case the adopted reference point is the Ici; and
- at a user plane level, where media streams are exchanged over the Izi reference point.

The management of IP multimedia sessions is acted by using SIP. The transport mechanism for both SIP session signalling and media transport is IPv4 (IETF RFC 791 [2]) or IPv6 (IETF RFC 2460 [7]). The 3GPP profile of SIP defining the usage of SIP within the IM CN subsystem is specified in 3GPP TS 24.229 [5]. Example call flows are provided in 3GPP TR 24.930 [6].

The general interconnection model is shown in Figure 4.1.

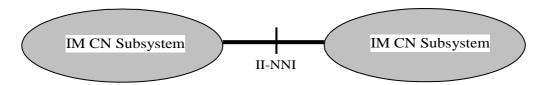


Figure 4.1: Interconnection Model for IM CN subsystems

The possible functional entities involved in the signalling plane interconnection (IBCF, I-CSCF, P-CSCF, BGCF and MSC Server enhanced for ICS) and in the user plane interconnection (TrGW) are specified in 3GPP TS 24.229 [5], in 3GPP TS 24.292 [121] and in 3GPP TS 29.162 [8].

IP Version interworking is described within 3GPP TS 29.162 [8].

5 Reference model for interconnection between IM CN subsystems

5.1 General

Figure 5.1 illustrates the architecture diagram given in 3GPP TS 23.228 [4] showing the Inter-IMS Network to Network Interface (II-NNI) between two IM CN subsystem networks.

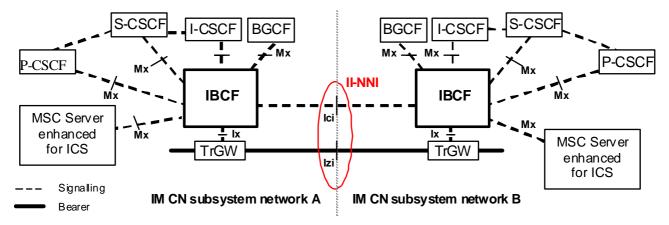


Figure 5.1.1: Inter-IMS Network to Network Interface between two IM CN subsystem networks

The protocols over the two reference points Ici and Izi make up the Inter-IMS Network to Network Interface.

The Ici reference point allows IBCFs to communicate with each other in order to provide the communication and forwarding of SIP signalling messaging between IM CN subsystem networks. The Izi reference point allows TrGWs to forward media streams between IM CN subsystem networks.

IMS roaming performed by using II-NNI is considered, when the IBCFs are inserted at the network borders.

Whenever the Inter-IMS Network to Network Interface is used to interconnect two IM CN subsystem networks belonging to different security domains, security procedures apply as described in 3GPP TS 33.210 [10].

5.2 Functionalities performed by entities at the edge of the network

5.2.1 Interconnection Border Control Function (IBCF)

An IBCF provides application specific functions at the SIP/SDP protocol layer in order to perform interconnection between IM CN subsystem networks by using Ici reference point. According to 3GPP TS 23.228 [4], IBCF can act both as an entry point and as an exit point for a network.

The functionalities of IBCF are indicated in the 3GPP TS 23.228 [4] and specified in 3GPP TS 24.229 [5]. They include:

- network topology hiding;
- application level gateway (for instance enabling communication between IPv6 and IPv4 SIP applications, or between a SIP application in a private IP address space and a SIP application outside this address space);
- controlling transport plane functions;
- controlling media plane adaptations;

- screening of SIP signalling information;
- selecting the appropriate signalling interconnect; and
- generation of charging data records.

Based on local configuration, the IBCF performs transit routing functions as specified in 3GPP TS 24.229 [5].

The IBCF acts as a B2BUA when it performs IMS-ALG functionality.

5.2.2 Transition Gateway (TrGW)

According to 3GPP TS 23.002 [3], the TrGW is located at the network borders within the media path and is controlled by an IBCF. Forwarding of media streams between IM CN subsystem networks is applied over Izi reference point.

The TrGW provides functions like network address/port translation and IPv4/IPv6 protocol translation. NAT-PT binds addresses in IPv6 network with addresses in IPv4 network and vice versa to provide transparent routing between the two IP domains without requiring any changes to end points. NA(P)T-PT provides additional translation of transport identifier (TCP and UDP port numbers). The approach is similar to that one described also in 3GPP TS 29.162 [8].

Further details are described in 3GPP TS 23.228 [4].

6 Control plane interconnection

6.1 Definition of Inter-IMS Network to Network Interconnection

6.1.1 SIP methods and header fields

6.1.1.1 General

The functional entity closest to the border of an II-NNI (see reference model in Clause 5) shall provide the capabilities specified for that network element in Annex A.2 of 3GPP TS 24.229 [5] with modifications as described in the following sub-clauses.

6.1.1.2 SIP methods

3GPP TS 24.229 [5] defines the methods allowing an IBCF to interconnect to an IBCF placed in another IM CN subsystem.

The following SIP Methods are supported on the II-NNI as defined in table 6.1.

The following table is based on Table A.5 and Table A.163 of 3GPP TS 24.229 [5] and endorsed for this document:

Table 6.1: Supported methods

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Item	Method	Ref.	II-NNI		
			Sending	Receiving	
1	ACK request	[13]	m	m	
2	BYE request	[13]	m	m	
3	BYE response	[13]	m	m	
4	CANCEL request	[13]	m	m	
5	CANCEL response	[13]	m	m	
5A	INFO request	[28]	0	0	
5B	INFO response	[28]	0	0	
8	INVITE request	[13]	m	m	
9	INVITE response	[13]	m	m	
9A	MESSAGE request	[19]	0	0	
9B	MESSAGE response	[19]	0	0	
10	NOTIFY request	[20]	c1	c1	
11	NOTIFY response	[20]	c1	c1	
12	OPTIONS request	[13]	m	m	
13	OPTIONS response	[13]	m	m	
14	PRACK request	[18]	m	m	
15	PRACK response	[18]	m	m	
15A	PUBLISH request	[21]	c1	c1	
15B	PUBLISH response	[21]	c1	c1	
16	REFER request	[22]	0	0	
17	REFER response	[22]	0	0	
18	REGISTER request	[13]	c2	c2	
19	REGISTER response	[13]	c2	c2	
20	SUBSCRIBE request	[20]	c1	c1	
21	SUBSCRIBE response	[20]	c1	c1	
22	UPDATE request	[23]	m	m	
23	UPDATE response	[23]	m	m	
c1:		n case of roaming scenario, the support of the method is m, else o.			
c2:	In case of roaming scenario, the support of the method is m, else n/a.				
NOTE:	NOTE: In the above table, m, o and c and n/a have the meanings indicated in Table 6.3				

The methods described in Table 6.1 shall be passed transparently on the II-NNI.

6.1.1.3 SIP header fields

6.1.1.3.0 General

The IBCF shall provide the capabilities to manage and modify SIP header fields according to section 5.10 and Annex A of 3GPP TS 24.229 [5] with modifications as described in the following sub-clauses.

6.1.1.3.1 Trust and no trust relationship

The IBCF acting as exit point applies the procedures described in clause 5.10.2 of 3GPP TS 24.229 [5] before forwarding the SIP signalling to the IBCF acting as entry point. The IBCF acting as entry point applies the procedures described in clause 5.10.3 of 3GPP TS 24.229 [5].

Additionally, in case there is no trust relationship between the two IM CN subsystems connected by II-NNI, the IBCF acting as exit point applies the procedures described in clause 4.4 of 3GPP TS 24.229 [5], before forwarding the SIP signalling to the next IBCF acting as an entry point. The IBCF acting as an entry point applies the procedures described in the section 5.10.3 of 3GPP TS 24.229 [5].

These procedures may be utilized on a per header field basis to realize overall trust as well as per service level screening of header fields. Trust relationships and trust domains may be defined by inter-operator agreements for individual services and/or individual SIP header fields.

The management of the SIP header fields (if present) over II-NNI in case of a presence or not of a trust relationship between the two interconnected IM subsystems is wrapped up in the following table.

Table 6.2: Management of SIP header fields over II-NNI in presence or not of a trust relationship

Item	Header field	Trust relationship	Not trust relationship
1	P-Asserted-Identity	As specified in 3GPP TS 24.229 [5], clause 4.4 and Annex J.2	As specified in 3GPP TS 24.229 [5], clause 4.4 and Annex J.2
2	P-Access-Network-Info	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
3	Resource-Priority	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
4	History-Info	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in Clause 4.3.3 of RFC 4244 [25] and in 3GPP TS 24.229 [5], clause 4.4
5	P-Asserted-Service	As specified in 3GPP TS 24.229 [5], clause 4.4 (NOTE 3)	As specified in 3GPP TS 24.229 [5], clause 4.4 (NOTE 3)
6	P-Charging-Vector (see RFC 3455 [24])	As specified in 3GPP TS 24.229 [5], clause 5.10	As specified in 3GPP TS 24.229 [5], clause 5.10
7	P-Charging-Function-Addresses (see RFC 3455 [24])	As specified in 3GPP TS 24.229 [5], clause 5.10	As specified in 3GPP TS 24.229 [5], clause 5.10
8	P-Profile-Key (NOTE 2)	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
9	P-Private-Network-Indication (NOTE 1)	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
10	P-Served-User (NOTE 1, NOTE 2)	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
11	Reason (in a response)	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4
12	P-Early-Media	As specified in 3GPP TS 24.229 [5], clause 4.4	As specified in 3GPP TS 24.229 [5], clause 4.4

NOTE 1: For a roaming II-NNI between a home IMS and a visited IMS, a trust relationship with respect to this header field is required.

6.1.1.3.2 Derivation of applicable SIP header fields from 3GPP TS 24.229 [5]

For any method in Table 6.1, the SIP header fields applicable on the II-NNI are detailed in the corresponding method tables for the UA role and proxy role sending behaviour in Annex A of 3GPP TS 24.229 [5]. Unless other information is specified in the normative part of the present specification, the applicability of header fields at the II-NNI can be derived for each method from the corresponding tables in Annex A of 3GPP TS 24.229 [5] as follows:

- All header fields not present in the corresponding tables in Annex A of 3GPP TS 24.229 or marked as "n/a" in both the "RFC status" and "profile status" columns for the UA role and proxy role sending behaviour of that tables are not applicable at the II-NNI.

NOTE 1: Operators could choose to apply header fields for new SIP extensions on an II-NNI based on bilateral agreements, but this is outside the scope of the present specification.

- All header fields which are marked as "o" in at least one of the "RFC status" or the "profile status" profile columns for the sending behaviour in the corresponding UA role and proxy role tables in Annex A of 3GPP TS 24.229 [5] and as "n/a" or "o" in the other such columns are applicable at II-NNI based on bilateral agreement between operators.

NOTE 2: This header field is only applicable on an II-NNI between a home IMS and a visited IMS.

NOTE 3: In addition, value-dependent operator policies may be applied.

- All header fields which are marked as "m" in at least one of the "RFC status" or the "profile status" columns for the sending behaviour in the corresponding UA role or proxy role table in Annex A of 3GPP TS 24.229 [5] and as "n/a", "o", or "m" in the other such columns are applicable at the II-NNI.
- If conditions are specified, they are also applicable at the II-NNI and the above rules are applicable to the "n/a", "o" and "m" values within the conditions.

NOTE 2: In the above rules, the RFC profile columns are taken into account in order to enable interworking with non-3GPP networks,

An informative summary of SIP header fields to be used over the II-NNI is proposed in Annex A.

6.1.1.3.3 Applicability of SIP header fields on a roaming II-NNI between home IMS and visited IMS

The following SIP header fields are only applicable on a roaming II-NNI between a home IMS and a visited IMS:

- Proxy-Authenticate
- Proxy-Authorization

6.1.1.3.4 Applicability of SIP header fields on an II-NNI between home IMS networks

The following SIP header fields are not applicable on an II-NNI between home IMS networks:

- P-Called-Party-ID
- P-Preferred-Service
- P-Profile-Key
- P-Served-User
- P-Visited-Network-ID
- WWW-Authenticate

6.1.1.4 Notations of the codes

In the table 6.1 the status codes "m", "o", "c" and "n/a" have the following meanings:

Notation

code

m

n/a

<integer>

Same as for sending side.

discarded by the IBCF.

Same as for sending side.

It is impossible to use/support the

message. This message will be

Sending side **Notation name** Receiving side The message shall be supported at II-Supporting receiving a SIP message at mandatory the II-NNI means that this message shall Supporting sending a SIP message at be forwarded to the serving network. It the II-NNI means that this message shall does not imply that network elements be sent over the II-NNI if received from inside the served network or user the serving network. It does not imply equipment connected to this network are that network elements inside the serving supporting this message. network or user equipment connected to this network shall support this message.

Table 6.3: Key to notation codes for SIP messages

6.1.1.5 Modes of signalling

Overlap signalling may be used if agreement exists between operators to use overlap and which method to be used, otherwise enbloc shall be used at the II-NNI.

The message may or may not be

It is impossible to use/support the

message.

supported at II-NNI. The support of the method is provided based on bilateral agreement between the operators.

The requirement on the message ("m",

"o" or "n/a") depends on the support of other optional or conditional items. <integer> is the identifier of the conditional expression.

6.1.2 SDP protocol

optional

not applicable

conditional

6.1.2.1 General

The functional entity closest to the border of an II-NNI (see reference model in Clause 5) shall provide the capabilities specified for that network element in Annex A.3 of 3GPP TS 24.229 [5].

6.1.3 Major capabilities

This subclause contains the major capabilities to be supported over the II-NNI.

The table 6.1.3.1 specifies which capabilities are applicable for II-NNI. The profile status codes within table 6.1.3.1 are defined in table 6.1.3.2.

For the "Basic SIP" capabilities part of table 6.1.3.1, the last column "Profile status over II-NNI" specifies the general status of applicability of the IETF RFC 3261 [13] main mechanisms described in the 2nd column "Capability over the Ici"

For the "Extensions to basic SIP" capabilities part, the last column "Profile status over II-NNI" specifies the general status of applicability of the RFC referenced in the 2^{nd} column "Capability over the Ici".

If necessary, the applicability of RFCs at the II-NNI level is further detailed in the present Technical Specification.

The columns "Reference item in 3GPP TS 24.229 [5] for the profile status" provide informative references for comparison purposes into the UA and Proxy role major capabilities tables in 3GPP TS 24.229 [5], where the capabilities are defined via additional references.

Table 6.1.3.1: Major capabilities over II-NNI

Item	Capability over the Ici	Reference item in 3GPP TS 24.229 [5] for the profile status		Profile status over II-NNI
		UA Role (NOTE 1)	Proxy role (NOTE 2)	
	Basic SIP (IETF RFC 3261 [13])			
1	registrations	1, 2, 2A	-	c2
2	initiating a session	2B, 2C, 3, 4	-	m
3	terminating a session	5	3	m
4	General proxy behaviour	-	4, 5, 14, 15, 19F	n/a
5	Forking of initial requests	9,10	6	m
6	support of indication of TLS connections in the Record-Route header	-	7, 8	n/a
7	usage of http authentication	7, 8, 8A	8A	c2
8	Timestamped requests (Timestamp header field)	6	-	m
9	Presence of date in requests and responses (Date header field)	11	9	m
10	Presence of alerting information data (Alert-info header field)	12	10	0
11	Support and handling of the Require header field for REGISTER and other requests or responses for methods other than REGISTER	-	11, 12, 13	m
12	Support and reading of the Supported and Unsupported header fields	-	16, 17, 18	m
13	Support of the Error-Info header field in 3xx - 6xx responses	-	19	0
14	Support and handling of the Organization header field	-	19A, 19B	m
15	Support and handling of the Call-Info header field	-	19C, 19D	m
16	Support of the Contact header field in 3xx response Extensions to basic SIP	-	19E	m
17	draft-ietf-sipcore-info-events-08 [39]: SIP INFO method and package framework	13	20	0
17A	draft-ietf-sipcore-info-events-08 [39]: legacy INFO usage	13A	20A	0
18	IETF RFC 3262 [18]: reliability of provisional responses in SIP (PRACK method)	14	21	m
19	IETF RFC 3515 [22]: the SIP REFER method	15	22	0
20	IETF RFC 3312 [40] and RFC 4032 [41]: integration of resource management and SIP (Preconditions framework)	16	23	0
21	IETF RFC 3311 [23]: the SIP UPDATE method	17	24	m
22	IETF RFC 3313 [42]: SIP extensions for media authorization (P-Media-Authorization header field)	19	26	n/a
23	IETF RFC 3265 [20]: SIP specific event notification (SUBSCRIBE/NOTIFY methods)	20, 21, 22, 23	27, 28	c1
24	IETF RFC 3327 [43]: session initiation protocol extension header field for registering non-adjacent contacts (Path header field)	24	29	c2
25	IETF RFC 3325 [44]: private extensions to the Session Initiation Protocol (SIP) for network asserted identity within trusted networks	25	30, 30A, 30B, 30C	c4
26	IETF RFC 3325 [44]: the P-Preferred-Identity header field extension	-	-	n/a
27	IETF RFC 3325 [44]: the P-Asserted-Identity header field extension	-	-	c4
28	IETF RFC 3323 [34]: a privacy mechanism for the Session Initiation Protocol (SIP) (Privacy header field)	26, 26A, 26B, 26C, 26D, 26E, 26F, 26G, 26H	31, 31A, 31B, 31C, 31D, 31E, 31F, 31G, 31H	m
29	IETF RFC 3428 [19]: a messaging mechanism for the Session Initiation Protocol (SIP) (MESSAGE method)	27	33	0
30	IETF RFC 3608 [45]: session initiation protocol extension header field for service route discovery during registration (Service-Route header field)	28	32	c2
31	IETF RFC 3486 [46]: compressing the session initiation protocol	29	34	n/a
32	IETF RFC 3455 [24]: private header extensions to the	30	35	0

	session initiation protocol for the 3rd-Generation Partnership	1	1	
	Project (3GPP)			
32A	IETF RFC 3325: act as first entity within the trust domain for asserted identity	30A	30A	n/a
32B	IETF RFC 3325: act as entity within trust network that can route outside the trust network	30B	30B	n/a
32C	IETF RFC 3325: act as entity passing on identity transparently independent of trust domain	30C	30C	n/a
33	IETF RFC 3455 [24]: the P-Associated-URI header field extension	31	36	c2
34	IETF RFC 3455 [24]: the P-Called-Party-ID header field extension	32	37	c2
35	IETF RFC 3455 [24]: the P-Visited-Network-ID header field extension	33	38, 39	c2
36	IETF RFC 3455 [24]: the P-Access-Network-Info header field extension	34	41, 42, 43	c4
37	IETF RFC 3455 [24]: the P-Charging-Function-Addresses header field extension	35	44, 44A	n/a
38	IETF RFC 3455 [24]: the P-Charging-Vector header field extension	36	45, 46	c2
39	IETF RFC 3329 [47]: security mechanism agreement for the session initiation protocol	37	47	n/a
39A	draft-dawes-dispatch-mediasec-parameter-02 [134]: mediasec header field parameter for marking security mechanisms related to media	37A	47A	n/a
40	IETF RFC 3326 [48]: the Reason header field for the session initiation protocol	38	48	0
41	draft-jesske-dispatch-reason-in-responses-02 [49]: use of the Reason header field in Session Initiation Protocol (SIP) responses	38A	48A	c4
42	IETF RFC 3581 [50]: an extension to the session initiation protocol for symmetric response routeing	39	49	0
43	IETF RFC 3841 [51]: caller preferences for the session initiation protocol (Accept-Contact, Reject-Contact and Request-Disposition header fields)	40, 40A, 40B, 40C, 40D, 40E, 40F	50, 50A, 50B, 50C, 50D, 50E, 50F	m
44	IETF RFC 3903 [21]: an event state publication extension to the session initiation protocol (PUBLISH method)	41	51	c1
45	IETF RFC 4028 [52]: SIP session timer (Session-Expires and Min-SE headers)	42	52	m
46	IETF RFC 3892 [53]: the SIP Referred-By mechanism	43	53	m
47	IETF RFC 3891 [54]: the Session Initiation Protocol (SIP) "Replaces" header	44	54	0
48	IETF RFC 3911 [55]: the Session Initiation Protocol (SIP) "Join" header	45	55	0
49	IETF RFC 3840 [56]: the callee capabilities	46	56	0
50	IETF RFC 4244 [25]: an extension to the session initiation protocol for request history information (History-Info header field)	47	57	0
51	IETF RFC 5079 [57]: Rejecting anonymous requests in the session initiation protocol	48	58	0
52	IETF RFC 4458 [58]: session initiation protocol URIs for applications such as voicemail and interactive voice response (NOTE 3)	49	59	0
53	IETF RFC 4320 [59]: Session Initiation Protocol's (SIP) non-INVITE transactions	50	61	m
54	IETF RFC 4457 [60]: the P-User-Database private header field extension	51	60	n/a
55	IETF RFC 5031 [61]: a uniform resource name for services	52	62	n/a
56	IETF RFC 5627 [62]: obtaining and using GRUUs in the Session Initiation Protocol (SIP)	53	63	c1
57	draft-patel-dispatch-cpc-oli-parameter-03 [63]: an extension to the session initiation protocol for request cpc information	54	64	c4
58	IETF RFC 4168 [27]: the Stream Control Transmission Protocol (SCTP) as a Transport for the Session Initiation Protocol (SIP)	55	65	0
59	IETF RFC 5002 [64]: the SIP P-Profile-Key private header	56	66, 66A,	c3

	field extension		66B	
60	IETF RFC 5626 [65]: managing client initiated connections in	57	67	c1
00	SIP	37	07	C1
61	IETF RFC 5768 [66]: indicating support for interactive	58	†-	n/a
01	connectivity establishment in SIP	30		11/4
62	IETF RFC 5365 [67]: multiple-recipient MESSAGE requests	59	69	o if 29, else
02	in the session initiation protocol		00	n/a
63	draft-ietf-sipcore-location-conveyance-03 [68]: SIP location	60	70, 70A,	m
00	conveyance (Geolocation header)		70B	'''
64	IETF RFC 5368 [69]: referring to multiple resources in the	61	71	o if 19, else
04	session initiation protocol	01	/ '	n/a
65	IETF RFC 5366 [70]: conference establishment using	62	72	0
00	request-contained lists in the session initiation protocol	02	12	
66	IETF RFC 5367 [71]: subscriptions to request-contained	63	73	o if 23, else
00	resource lists in the session initiation protocol	00	7.5	n/a
67	IETF RFC 4967 [72]: dialstring parameter for the session	64	74	c2
01	initiation protocol uniform resource identifier	04	/	02
68	IETF RFC 4964 [73]: the P-Answer-State header extension	65	75	0
00	to the session initiation protocol for the open mobile alliance		10	
	push to talk over cellular			
69	IETF RFC 5009 [74]: the SIP P-Early-Media private header	66	76	m
00	field extension for authorization of early media			
70	IETF RFC 4694 [75]: number portability parameters for the	67, 67A,	77, 77A,	0
70	"tel" URI	67B	77B	
71	draft-yu-tel-dai-09 [76]: DAI Parameter for the 'tel' URI	68	78	0
72	IETF RFC 4411 [77]: extending the session initiation protocol	69	79	0
12	Reason header for preemption events		13	
73	IETF RFC 4412 [78]: communications resource priority for	70, 70A,	80, 80A,	0
73	the session initiation protocol? (Resource-Priority header	70, 70A, 70B, 70C,	80B, 80C,	
	field)	70D, 70E,	80D, 80E,	
	neid)	70E, 70E, 70F, 70G	80F, 80G	
74	IETF RFC 5393 [79]: addressing an amplification	70F, 70G	81	m
14	vulnerability in session initiation protocol forking proxies	' '	01	'''
75	IETF RFC 5049 [80]: the remote application identification of	72	82	n/a
75	applying signalling compression to SIP	12	02	11/a
76	IETF RFC 5688 [81]: a session initiation protocol media	73	83	c1
70	feature tag for MIME application sub-types	13	03	CI
77	draft-drage-sipping-service-identification-04 [26]:	74	84, 84A	0
11	Identification of communication services in the session	' 4	04, 04/	0
	initiation protocol			
78	IETF RFC 5360 [82]: a framework for consent-based	75, 75A,	85	0
70	communications in SIP?	75, 75A, 75B	00	
79	draft-johnston-sipping-cc-uui-09 [83]: transporting user to	76	86	c1
13	user information for call centers using SIP?	10	00	
80	draft-vanelburg-dispatch-private-network-ind-01 [84]: The	77	87	c1
00	SIP P-Private-Network-Indication private-header (P-Header)	' '	07	01
81	IETF RFC 5502 [85]: the SIP P-Served-User private header	78	88	c2
83	draft-dawes-sipping-debug-02 [87]: the P-Debug-ID header	80	90	0
00	extension	30	30	
84	draft-ietf-sipcore-199-02 [88]: the 199 (Early Dialog	81	91	m
04	Terminated) response code)	01	اقا	'''
85	IETF RFC 5621 [89]: message body handling in SIP	82	92	m
86	draft-holmberg-sip-keep-04 [90]: indication of support for	83	93	m o
00	keep-alive	03	93	١
0.7		0.4	0.4	2/2
87	IETF RFC 5552 [91]: SIP Interface to VoiceXML Media Services	84	94	n/a
88	IETF RFC 3862 [92]: common presence and instant	85	95	
00		00	90	0
89	messaging (CPIM): message format IETF RFC 5438 [93] : instant message disposition notification	86	96	
				0
90	IETF RFC 5373 [94]: requesting answering modes for SIP	87	97, 97A	0
91	(Answer-Mode and Priv-Answer-Mode header fields)	88	00	n/o
91	draft-patel-ecrit-sos-parameter-09 [95]: SOS URI parameter	00	98	n/a
00	for marking SIP requests related to emergency calls	90	00	1
92	IETF RFC 3959 [96]: the early session disposition type for	89	99	0
00	SIP	00	100	2/2
93	draft-rosenberg-sipcore-target-uri-delivery-01 [97]: delivery of	90	100	n/a
	Request-URI targets to user agents		1	

94	draft-kaplan-sip-session-id-02 [124]: The Session-ID header	91	101	0
95	draft-ietf-sipcore-invfix-01 [125]: correct transaction handling for 200 responses to Session Initiation Protocol INVITE requests	92	102	m
96	IETF RFC 5658 [126]: addressing Record-Route issues in the Session Initiation Protocol (SIP)	93	103	0
97	draft-ietf-sip-ipv6-abnf-fix-05 [127]: essential correction for IPv6 ABNF and URI comparison in IETF RFC 3261 [13]	94	104	m
98	IETF RFC 4488 [132]: suppression of session initiation protocol REFER method implicit subscription	95	105	c5
99	draft-liess-dispatch-alert-info-urns-02 [133]: Alert-Info URNs for the Session Initiation Protocol	96	106	0

- c1: m in case of roaming NNI between home and visited IMS, else o
- c2: m in case of roaming NNI between home and visited IMS, else n/a
- c3: o in case of roaming NNI between home and visited IMS, else n/a
- c4: m in case of trust relationship between the interconnected networks, else n/a
- c5: m in the case the REFER request is supported, else n/a
- NOTE 1: The item numbering corresponds to the one provided in Table A.4 in [5]
- NOTE 2: The item numbering corresponds to the one provided in Table A.162 in [5]
- NOTE 3: A common URI namespace is required to apply this feature on the II-NN

Table 6.1.3.2: Key to notation codes for major capabilities

Notation code	Notation name	Explanation
m	mandatory	The capability shall be supported at II-NNI. SIP message relating to this capability shall be sent over the II-NNI if received from the serving network, unless they also make use of other unsupported capabilities. SIP headers or other information elements relating to this capability shall be passed over the II-NNI if received from the sending side. This does not imply that network elements inside the serving network or served
		network or user equipment connected to these networks shall support this capability.
0	optional	The capability may or may not be supported at II-NNI. The support of the capability is provided based on bilateral agreement between the operators.
n/a	not applicable	It is impossible to use/support the capability at the II-NNI.
c <integer></integer>	conditional	The support of the capability ("m", "o" or "n/a") depends on the support of other optional or conditional items. <integer> is the identifier of the conditional expression.</integer>

6.2 Control Plane Transport

6.2.1 General

The control plane transport of the II-NNI shall comply with Clause 4.2A of 3GPP TS 24.229 [5].

Support of SCTP as specified in IETF RFC 4168 [27] is optional for an IBCF connected by II-NNI. Nevertheless this option is favourable if the operators would like to improve reliability over the Ici.

7 User plane Interconnection

7.1 Media and Codec

For "end-to-end" media session involving the II-NNI, the SIP/SDP codec negotiation procedure can be applied between IM CN subsystems using different media codecs. It is possible that the end-to-end codec negotiation could fail because no common codec could be supported by the UEs, in particular for voice services.

To enhance interoperability, the IBCF, the MRFC, or other IMS network entities can interfere with the end-to-end codec negotiation to offer additional codec(s) available via transcoding, or to remove codecs. The IBCF can configure an attached TrGW to transcode, and the MRFC can configure an attached MRFP to transcode.

Codecs applicable at the NNI may be a subject of interworking agreements.

NOTE: Possible codecs which could be used at the II-NNI are described in 3GPP TS 26.114 [11] and ETSI TS 181 005 [12].

However, to avoid that transcoding is performed several times, applicable codecs at the NNI should be restricted as little as possible.

NOTE: Transcoding can be performed in an IMS network serving an SDP offerer or in an IMS network serving an SDP answerer. To avoid that transcoding is performed multiple times, inter-operator agreements can clarify if it is preferred that IMS network serving an SDP offerer or IMS network serving an SDP answerer modify an SDP offer to offer transcoding.

If the IBCF performs media transcoding control, it shall apply the related procedures in 3GPP TS 24.229 [5].

7.2 User Plane Transport

The user plane transport of the II-NNI may use the protocols listed in Table 7.2.1. The used protocols to transport media are negotiated by means of SDP offer/answer.

Table 7.2.1: Supported transport-level RFCs to be described in SIP/SDP messages

Item	RFC	Title	Support
1	RFC 3550	RTP: A Transport Protocol for Real-Time Applications	Mandatory
2	RFC 768	User Datagram Protocol	Mandatory
3	RFC 3551	RTP Profile for Audio and Video Conferences with Minimal Control	Mandatory
4	RFC 3556	Session Description Protocol (SDP) Bandwidth Modifiers for RTP Control Protocol (RTCP) Bandwidth	Mandatory
5	RFC 4585	Extended RTP Profile for Real-time Transport Control Protocol (RTCP) - Based Feedback (RTP/AVPF)	Optional (NOTE 1)
6	RFC 793	Transmission Control Protocol	Optional (NOTE 2)
	sed by MTSI, as	s indicated in 3GPP TS 26.114 [11] service	

8 Numbering, Naming and Addressing

The following URI formats in SIP messages may be applied at the Ici as standardized in 3GPP TS 24.229 [5]:

- SIP URI defined in IETF RFC 3261 [13];
- tel URI defined in IETF RFC 3966 [14];
- IM URI defined in IETF RFC 3860 [15];
- PRES URI defined in IETF RFC 3859 [16].

Moreover, in case of MSRP sessions passing through the II-NNI, the MSRP URI may be also used at the Ici in the SDP exchange, following the formats defined in IETF RFC 4975 [17].

The IBCF shall support these URI formats. Other URI formats may be supported over the II-NNI depending on the operators" agreements.

A global number as defined in IETF RFC 3966 [14] shall be used in a tel-URI or in the user portion of a SIP URI with the user=phone parameter when conveyed via a non-roaming interface in the Request-URI and in the P-Asserted-Identity header field, except when agreement exists between the operators to also allow other kinds of numbers.

NOTE 1: In a SIP URI the user portion of the Request-URI represents a telephone number only if the SIP URI includes the user=phone parameter.

- NOTE 2: Agreements can exist between operators to allow non-global number (e.g. national service numbers. business trunking numbers, or private numbers) at a non-roaming II-NNI. A SIP URI with such a number, a user=phone parameter, and a phone-context parameter agreed between the operators can then be used.
- NOTE 3: 3GPP TS 24.229 [5] allows to restrict the number within a SIP Request-URI with user=phone parameter at a non-roaming II-NNI to be a global number (i.e. E.164 in international format) via an appropriate Application Server. Suitable configuration by the operator is needed to achieve the desired modification of the format.
- NOTE 4: The allowed phone number formats in the P-Asserted-Identity header field of a served user are configured by the operator. According to 3GPP TS 23.003 [35], international E.164 format is used within a P-Asserted-Identity header field.
- NOTE 5: The global number format usage within a SIP Request-URI with the user=phone parameter at a non-roaming II-NNI allows the terminating network to find the called subscriber, via HSS interrogation, without any further number translation and thus improves the success of the interconnection between IMS Operators.

9 IP Version

The network elements interconnected by means of the II-NNI may support IPv4 only, IPv6 only or both.

The support of one or both of the IP versions is an operator option and should be based on bilateral agreement.

In case IPv4 and IPv6 networks are interconnected, the involved IBCFs and TrGWs shall apply the IP version interworking procedures as indicated in 3GPP TS 29.162 [8].

10 Security

The supported security mechanisms for IP signalling transport over II-NNI interfaces are described in 3GPP TS 33.210 [10].

11 Charging

The accounting information to be supported over the Ici is described in 3GPP TS 32.260 [29]. It shall be configurable by the operator to use or not the accounting mechanisms provided by the IBCF.

Supplementary services associated with the IMS multimedia telephony communication service

12.1 General

In order to assure the end-to-end service interoperability through the Inter-IMS Network to Network Interface (II-NNI), associated supplementary services of the multimedia telephony communication service may be supported on the II-NNI between the two IMS networks. If they are supported, the related procedures from the 3GPP TS 22.173 [30], the protocol details from the 3GPP TS 24.173 [31] and specifications referenced in the later specification shall be applied with the following restrictions due to the crossing of the II-NNI.

12.2 Malicious Communication IDentification (MCID)

Service specific requirements in accordance with 3GPP TS 24.616 [33] shall be supported over the II-NNI.

The INFO request and the 200 (OK) response to the INFO request containing the application/vnd.etsi.mcid+xml body defined in 3GPP TS 24.616 [33] may be supported at the II-NNI.

If a network terminating the dialog supports MCID, the terminating network shall only deliver the MCID request in the mcid+xml body, as specified in the 3GPP TS 24.616 [33], if an agreement to use the MCID supplementary service according to the 3GPP TS 24.616 [33] exists with the network originating the dialog and if the INVITE request received by the terminating network does not contain the information of the originating party.

NOTE: The IBCF and the AS in the terminating network interact to deliver the MCID request only if an agreement to use the MCID supplementary service exists, as specified in 3GPP TS 24.616 [33] and 3GPP TS 24.229 [5].

The originating network and the terminating network shall have a bilateral agreement to support transportation of the minimum information specified in subclause 4.5.2.5.0 of the 3GPP TS 24.616 [33] between the networks.

12.3 Originating Identification Presentation (OIP) and Originating Identification Restriction (OIR)

Service specific requirements in accordance with 3GPP TS 24.607 [32] shall be supported over the II-NNI.

The P-Asserted-Identity header field and the Privacy header field with values "id", "user", "none", "header" and "critical" shall be supported at the II-NNI.

- NOTE 1: P-Asserted-Identity header fields are intended for end-to-end operation. Removal of such header fields will impact the intended end-to-end operation between the end users. Where a trust relationship exists on the P-Asserted-Identity header field between the two IMS networks, this header field cannot be altered when passing through the II-NNI according to 3GPP TS 24.229 [5]. Where no trust relationship exists on the P-Asserted-Identity header field between the two IMS networks, the P-Asserted-Identity header field will be removed by the IBCF of the originating network prior passing through the II-NNI according to the 3GPP TS 24.229 [5]. The IBCF determines whether to remove the P-Asserted-Identity header field according to procedures in 3GPP TS 24.229 [5] referencing IETF RFC 3325 [44].
- NOTE 2: The From header field cannot be altered when passing through the II-NNI and will be passed transparently by the IBCF. If a request is received by the terminating network and the application of the OIR service is required with the value "user" for the Privacy header field then the From header field will be anonymised in accordance with IETF RFC 3323 [34] by the terminating network.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.4 Terminating Identification Presentation (TIP) and Terminating Identification Restriction (TIR)

Service specific requirements in accordance with 3GPP TS 24.608 [113] shall be supported over the II-NNI.

The P-Asserted-Identity header field and the Privacy header field with values "id", "user", "none", "header" and "critical" shall be supported at the II-NNI.

NOTE: P-Asserted-Identity header fields are intended for end-to-end operation. Removal of such header fields will impact the intended end-to-end operation between the end users. Where a trust relationship exists on the P-Asserted-Identity header field between the two IMS networks, this header field cannot be altered when passing through the II-NNI according to 3GPP TS 24.229 [5].

Where no trust relationship exists on the P-Asserted-Identity header field between the two IMS networks, the P-Asserted-Identity header field will be removed by the IBCF of the originating network prior passing through the II-NNI according to the 3GPP TS 24.229 [5]. The IBCF determines whether to remove the P-Asserted-Identity header field according to procedures in 3GPP TS 24.229 [5] referencing IETF RFC 3325 [44].

12.5 Anonymous Communication Rejection (ACR)

Service specific requirements in accordance with 3GPP TS 24.611 [107] shall be supported over the II-NNI.

Procedures as described in subclause 12.21.2 are used to provide announcements.

The response code 433 (Anonymity Disallowed) shall be supported at the II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.6 Communication DIVersion (CDIV)

Service specific requirements in accordance with 3GPP TS 24.604 [117] shall be supported over the II-NNI.

NOTE 1: The support of the Diversion header field not adopted in 3GPP TS 24.604 requires bilateral agreement between the operators.

Procedures as described in subclause 12.21.2 are used to provide announcements.

The Privacy header field with value "history" shall be supported at the II-NNI.

The History-Info header field as described by 3GPP TS 24.604 [117] and the Cause-Codes as defined by the IETF RFC 4458 [118] shall be supported over the II-NNI.

The response code 181 (Call Is Being Forwarded) shall be supported at the II-NNI. The SUBSCRIBE requests with the event package name "comm-div-info" and the NOTIFY request procedure as specified in IETF RFC 3265 [20] and 3GPP TS 24.229 [5] shall be supported at the roaming II-NNI if CDIVN is provided.

The MESSAGE request procedure as specified in IETF RFC 3428 [19] and 3GPP TS 24.229 [5] should be supported at the roaming II-NNI if CDIVN is provided.

NOTE 2: The content of the MESSAGE request is operator specific.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.7 Communication Waiting (CW)

Service specific requirements in accordance with 3GPP TS 24.615 [37] shall be supported over the II-NNI.

The INVITE method containing the application/vnd.3gpp.cw+xml body defined in 3GPP TS 24.615 [37] shall be supported at the roaming II-NNI.

The Alert-Info header field set to "urn:alert:service:call-waiting" in a 180 (Ringing) response shall be supported at the II-NNI.

As a network option, in case of expiry of the CW timer, the response code 480 (Temporarily Unavailable) including a Reason header field set to cause 19 shall be supported at the non-roaming II-NNI.

Procedures as described in subclause 12.21.2 are used to provide announcements.

12.8 Communication HOLD (HOLD)

Service specific requirements in accordance with 3GPP TS 24.610 [36] shall be supported over the II-NNI.

NOTE: The support of an alternative method not adopted in 3GPP TS 24.610 requires bilateral agreement between the operators and is outside the scope of the present document.

Procedures as described in subclause 12.21.3 are used to provide announcements.

12.9 Message Waiting Indication (MWI)

Service specific requirements in accordance with 3GPP TS 24.606 [112] shall be supported over the II-NNI.

The SUBSCRIBE request with the event package name "message-summary" and NOTIFY request procedures according to IETF RFC 3265 [20] and 3GPP TS 24.229 [5] shall be supported at the roaming II-NNI.

The "simple-message-summary+xml" body described in 3GPP TS 24.606 [112] in the NOTIFY request shall be supported at the roaming II-NNI.

12.10 Communication Barring (CB)

12.10.1 Incoming Communication Barring (ICB)

Service specific requirements in accordance with 3GPP TS 24.611 [114] shall be supported over the II-NNI.

Procedures as described in subclause 12.21.2 are used to provide announcements.

The response code 603 (Decline) including a Reason header field as described in 3GPP TS 24.611 [114] shall be supported at the II-NNI.

The BYE request including a Reason header field as described in 3GPP TS 24.611 [114] shall be supported at the II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.10.2 Outgoing Communication Barring (OCB)

Service specific requirements in accordance with 3GPP TS 24.611 [114] shall be supported over the II-NNI.

Procedures as described in subclause 12.21.2 are used to provide announcements.

The response code 603 (Decline) including a Reason header field as described in 3GPP TS 24.611 [114] shall be supported at the roaming II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.11 Completion of Communications to Busy Subscriber (CCBS)

Service specific requirements in accordance with 3GPP TS 24.642 [109] shall be supported over the II-NNI.

The response code 486 (Busy Here) containing a Call-Info header field with a "purpose" header field parameter set to "call-completion" and the "m" parameter set to "BS" shall be supported at the non-roaming II-NNI.

For invoking and revoking of the CCBS supplementary service, announcement procedures shall be used to provide announcements and inband-interaction procedures as described in subclause 12.21.2 shall be supported at the roaming II-NNI.

The response code 199 (Early Dialog Terminated) shall be supported at the roaming II-NNI.

Basic call procedures and in case of a call-completion recall initiated by a REFER request, normal REFER method handling procedures according to 3GPP TS 24.229 [5] shall supported at the roaming II-NNI.

As a network option the special REFER request handling procedures according to 3GPP TS 24.628 [110] should be supported at the roaming II-NNI.

NOTE 1: 3rd party call control procedures can be used when the REFER request is not supported at the II-NNI.

NOTE 2: A REFER request can be rejected by IBCF based on operator policy as specified by 3GPP TS 24.229 [5].

The SUBSCRIBE and NOTIFY methods according to IETF RFC 3265 [20] and 3GPP TS 24.229 [5] containing the event package name "call-completion" and the Call-Info header field with a purpose parameter set to 'call-completion' and the m parameter set to "BS" shall be supported at the non-roaming interface II-NNI.

The Request-URI with the "m" SIP URI parameter with a value set to "BS" and the Call-Info header field with a purpose parameter set to 'call-completion' and the "m" parameter set to "BS" in the INVITE method shall be supported at the non-roaming II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.12 Completion of Communications by No Reply (CCNR)

Service specific requirements in accordance with 3GPP TS 24.642 [109] shall be supported over the II-NNI.

The response code 180 (Ringing) containing a Call-Info header field with a purpose parameter set to 'call-completion' and the "m" parameter set to "NR" shall be supported at the non-roaming interface.

For invoking and revoking of the CCNR supplementary service, announcement procedures shall be used to provide announcements and inband-interaction procedures as described in subclause 12.21.2 shall be supported at the roaming II-NNI.

The response code 199 (Early Dialog Terminated) shall be supported at the roaming II-NNI.

Basic call procedures and in case of a call-completion recall initiated by a REFER request, normal REFER method handling procedures according to 3GPP TS 24.229 [5] shall supported at the roaming II-NNI.

As a network option the special REFER request handling procedures according to 3GPP TS 24.628 [110] should be supported at the roaming II-NNI.

NOTE 1: 3rd party call control procedures can be used when the REFER request is not supported at the II-NNI.

NOTE 2: A REFER request can be rejected by IBCF based on operator policy as specified by 3GPP TS 24.229 [5].

The SUBSCRIBE and NOTIFY methods according to IETF RFC 3265 [20] and 3GPP TS 24.229 [5] containing the event package name "call-completion" and the Call-Info header field with a purpose parameter set to 'call-completion' and the m parameter set to "NR" shall be supported at the non-roaming interface II-NNI.

The Request-URI with the "m" SIP URI parameter with a value set to "NR" and the Call-Info header field with a purpose parameter set to 'call-completion' and the "m" parameter set to "NR" in the INVITE method shall be supported at the non-roaming II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.13 Explicit Communication Transfer (ECT)

Service specific requirements in accordance with 3GPP TS 24.629 [116] shall be supported over the II-NNI.

The REFER method, the Referred-By header field and the Replaces header field as specified in 3GPP TS 24.629 [116] shall be supported at the II-NNI for call transfer without third party call control.

The REFER method, the Referred-By header field and the Replaces header field as specified in 3GPP TS 24.629 [116] shall only be supported at the roaming II-NNI for call transfer with third party call control.

12.14 Customized Alerting Tone (CAT)

Service specific requirements in accordance with 3GPP TS 24.182 [129] shall be supported over the II-NNI.

The P-Early-Media header field in as described in 3GPP TS 24.182 [129] shall be supported at the II-NNI.

The response code 183 (Session Progress) including a P-Early-Media header field shall be supported over the II-NNI.

The response code 199 (Early Dialog Terminated) shall be supported over the II-NNI.

The Supported header field and the Require header field with "early-session" option-tag may be supported at the II-NNI

An SDP MIME body with the Content-Disposition set to "early-session" as specified in IETF RFC 3959 [96] may be supported at II-NNI.

The SIP INFO mechanism for DTMF transport, as defined in 3GPP TS 24.229 [5] shall be supported at the II-NNI.

NOTE: For telephone-event based DTMF transport, the DTMF digits are sent as media and not visible in the control plane.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.15 Customized Ringing Signal (CRS)

Service specific requirements in accordance with 3GPP TS 24.183 [98] shall be supported over the II-NNI.

An Alert-Info header field in the initial INVITE request containing an URI followed by a URN "urn:alert:service:crs" shall be supported at the II-NNI.

A Contact header field in the initial INVITE request containing a "g.3gpp.crs" feature tag may be supported at the II-NNI.

The Supported header field and the Require header field with "early-session" option-tag may be supported at the II-NNI

An SDP MIME body with the Content- Disposition set to "early-session" as specified in IETF RFC 3959 [96] may be supported at II-NNI.

The SIP INFO mechanism for DTMF transport, as defined in 3GPP TS 24.229 [5] may be supported at the II-NNI.

NOTE: For telephone-event based DTMF transport, the DTMF digits are sent as media and not visible in the control plane.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.16 Closed User Group (CUG)

Service specific requirements in accordance with 3GPP TS 24.654 [103] shall be supported over the II-NNI.

The application/vnd.etsi.cug+xml MIME as specified 3GPP TS 24.654 [103] shall be supported in INVITE requests at the II-NNI.

NOTE: If no agreement between the originating network and the terminating network exists to support the CUG supplementary service the INVITE request is rejected as described in IETF RFC 5621 [89] when the "handling" parameter in the Content-Disposition of the vnd.etsi.cug+xml MIME body is set to "required".

The 403 (Forbidden) response, the 603 (Decline) response and the 500 (Server Internal Error) response shall be supported at II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.17 Personal Network Management (PNM)

Service specific requirements in accordance with 3GPP TS 24.259 [99] shall be supported over the II-NNI.

The Contact header field of the REGISTER request containing the g.3gpp.iari_ref feature tag with the value urn:urn-7:3gpp-application.ims.iari.pnm-controller shall be supported at the roaming II-NNI.

The Accept-Contact header field containing a g.3gpp.iari_ref feature with the value urn:urn-7:3gpp-application.ims.iari.pnm-controller shall be supported at the II-NNI.

The History-Info header field and Supported header field containing the "histinfo" option tag as described by 3GPP TS 24.259 [99] shall be supported at II-NNI.

12.18 Three-Party (3PTY)

Service specific requirements in accordance with 3GPP TS 24.605 [105] shall be supported over the II-NNI.

NOTE 1: The requirements below can be relaxed by bilateral agreements between operators.

The requirements for the 3PTY supplementary service are the same as for the CONF supplementary service specified in subclause 12.19 with the following additional requirement:

- If a REFER request is supported at the II-NNI, a Replaces header field in the header portion of the SIP URI of the Refer-to header field of the REFER request shall also be supported at II-NNI.

NOTE 2: Subclause 12.19 describes the conditions for the support of the REFER request.

12.19 Conference (CONF)

Service specific requirements in accordance with 3GPP TS 24.605 [105] shall be supported over the II-NNI.

NOTE 1: The requirements below can be relaxed by bilateral agreements between operators.

The REFER request shall be supported at the roaming II-NNI in the direction from visited to home network. Based on inter-operator agreement, the REFER request may be supported at II-NNI between home networks, and at the roaming II-NNI in the direction from home network to visited network.

NOTE 2: If the REFER request is not supported at the II-NNI between home networks, or at the roaming II-NNI in the direction from home network to visited network, an attempt of an UE to send the REFER directly to peers to invite them to a conference without involvement of the conference focus can fail over such an II-NNI. However such failures can also occur if a peer is located in a circuit switched network, or if a peer does not support the REFER method. An operator can avoid such failures by configuring an AS to convert the REFER to an INVITE, as detailed in 3GPP TS 24.628 [38]. Information on security risks associated with the REFER request is provided within the "security consideration" of IETF RFC 3515 [22].

NOTE 2: A REFER request can be rejected by IBCF based on operator policy as specified by 3GPP TS 24.229 [5].

The "application/resource-lists+xml" body shall be supported at the roaming II-NNI.

The Referred-By header field in the INVITE request shall be supported at the II-NNI.

The "isfocus" feature parameter indicated in Contact header field of the INVITE request and in the 200 (OK) response shall be supported at the II-NNI.

The SUBSCRIBE request including the "conference" event package name in the Event header field and the NOTIFY request procedures according to 3GPP TS 24.147 [106] shall be supported at the II-NNI.

12.20 Flexible Alerting (FA)

Service specific requirements in accordance with 3GPP TS 24.239 [101] shall be supported over the II-NNI.

The 486 (Busy Here) response code shall be supported at the II-NNI.

SIP based user configuration as described in 3GPP TS 24.238 [100] shall be supported at the roaming II-NNI.

12.21 Announcements

12.21.1 General

Announcements may be provided during the establishment of a communication session or during an established communication session. Both of them shall be managed over the II-NNI.

12.21.2 Providing announcements during the establishment of a communication session

Procedures as described in 3GPP TS 24.628 [38] are used to provide announcements.

The P-Early-Media header authorizing early media as defined in IETF RFC 5009 [12] during the establishment of a communication fields shall be supported at the II-NNI.

The Alert-Info header in the 180 (Ringing) response to the INVITE request during the establishment of a communication, should be supported at the II-NNI.

NOTE: The IBCF can decide to remove the Alert-Info header field if required by local policy.

12.21.3 Providing announcements during an established communication session

Procedures as described in 3GPP TS 24.628 [38] are used to provide announcements.

In case of provision of an announcement to a user over the II-NNI during an established communication, the Call-Info header field in a re-INVITE request should be supported at the II-NNI.

NOTE 1: An alternative method to provide announcements is to use the existing media stream.

NOTE 2: The IBCF can decide to remove the Call-Info header field if required by local policy.

12.22 Advice of Charge (AOC)

Service specific requirements in accordance with 3GPP TS 24.647 [122] shall be supported over the II-NNI.

The Accept header field with "application/vnd.etsi.aoc+xml" shall be supported at the roaming II-NNI.

The INVITE method containing "application/vnd.etsi.aoc+xml" MIME shall be supported at the roaming II-NNI.

1xx provisional responses and the 200 (OK) response to the initial INVITE request containing an "application/vnd.etsi.aoc+xml" MIME body shall be supported at the roaming II-NNI.

The INFO method containing an "application/vnd.etsi.aoc+xml" MIME body shall be supported at the roaming II-NNI.

The response code 504 (Server Time-out) shall be supported at the II-NNI.

A BYE request including Reason header field with a reason value with the protocol set to "SIP" and the cause set to "504" and a reason value with the protocol set to "Q.850" and the cause set to "31" shall be supported at the II-NNI.

The BYE request or the final response to the BYE request including an "application/vnd.etsi.aoc+xml" MIME body shall be supported over the roaming II-NNI.

Annex A (informative): Summary of SIP header fields

A summary of the SIP header fields to be used in case of interconnection by using II-NNI is proposed in Table A.1.

The starting point is the sending behaviour described for proxy and UA roles in Annex A of 3GPP TS 24.229 [5]. In case of misalignment between Table A.1 and the behaviour described in 3GPP TS 24.229 [5], the behaviour in 3GPP TS 24.229 [5] has the precedence. In case a header field is not described in Table A.1 and it is described in [5], the description in 3GPP TS 24.229 [5] is applicable over II-NNI.

The notation of the codes used for the SIP headers listed in table A.1 has a different meaning to the one proposed for the SIP messages. The definition of these terms is provided in table A.2.

Table A.1: Supported header fields

Item	Header field	Ref.	II-NNI
1	Accept	[5]	m
2	Accept-Contact	[5]	m
3	Accept-Encoding	[5]	m
4	Accept-Language	[5]	m
5	Alert-Info	[5]	0
6	Allow	[5]	m
7	Allow-Events	[5]	_ m
9	Authentication-Info Authorization	[5]	m
9 9a	Answer-Mode	[5]	m o
10	Call-ID	[5] [5]	m
11	Call-Info	[5]	m
12	Contact	[5]	m
13	Content-Disposition	[5]	m
14	Content-Encoding	[5]	m
15	Content-Language	[5]	m
16	Content-Length	[5]	m
17	Content-Type	[5]	m
18	Cseq	[5]	m
19	Date	[5]	m
20	Error-Info	[5]	0
21	Expires	[5]	m
22	Event	[5]	m
23	From	[5]	m
24	Geolocation	[5]	m
25	History-Info	sub-clause	m in case of a trust relationship between the interconnected
		6.1.1.3.1	networks, else n/a
		(Table 6.2,	
-00	1.5.1.7	item 4)	
26 27	In-Reply-To Join	[5]	0
27a	Max-Breadth	[5] [5]	0 n/a
28	Max-Forwards	[5]	m
29	Min-Expires	[5]	m
30	MIME-Version	[5]	m
31	Min-SE	[5]	m
32	Organization	[5]	m
33	P-Access-Network-Info	sub-clause	m in case of a trust relationship between the interconnected
		6.1.1.3.1	networks, else n/a
		(Table 6.2,	
		item 2)	
33a	P-Answer-state	[5]	0
34	P-Asserted-Identity	sub-clause	m in case of a trust relationship between the interconnected
		6.1.1.3.1	networks, else n/a
		(Table 6.2,	
25	D Assembled Committee	item 1)	un in annual attendant lateration to the second
35	P-Asserted-Service	sub-clause	m in case of a trust relationship between the
		6.1.1.3.1 (Table 6.2,	interconnected networks, else n/a
		item 5)	
35a	P-Associated-URI	[5]	m on roaming NNI between home and visited IMS, else n/a
36	P-Called-Party-ID	[5]	m on roaming NNI between home and visited IMS, else n/a
37	P-Charging-Function-	sub-clause	n/a
	Addresses	6.1.1.3.1	
		(Table 6.2,	
		item 7)	
38	P-Charging-Vector	sub-clause	m in case of a trust relationship between the interconnected
		6.1.1.3.1	networks, else n/a
		(Table 6.2,	
		item 6)	
38a	P-Debug-Id	[5]	0
39	P-Early-Media	[5]	m

Item	Header field	Ref.	II-NNI
40	P-Media-Authorization	[5]	n/a
41	P-Preferred-Identity	[5]	n/a
42	P-Preferred-Service	[5]	m on roaming NNI between home and visited IMS, else n/a
43	P-Private-Network-Indication	sub-clause 6.1.1.3.1 (Table 6.2, item 9)	m on roaming NNI between home and visited IMS, else o
44	P-Profile-Key	sub-clause 6.1.1.3.1 (Table 6.2, item 8)	o on roaming NNI between home and visited IMS, else n/a
45	P-Served-User	sub-clause 6.1.1.3.1 (Table 6.2, item 10)	m on roaming NNI between home and visited IMS, else n/a
46	P-User-Database	[5]	n/a
47	P-Visited-Network-ID	[5]	m on roaming NNI between home and visited IMS, else n/a
47a	Path	[5]	m on roaming NNI between home and visited IMS, else n/a
48	Priority	[5]	0
48a	Priv-Answer-Mode	[5]	0
49	Privacy	[5]	m
50	Proxy-Authenticate	[5]	m on roaming NNI between home IMS and visited IMS, else n/a
51	Proxy-Authorization	[5]	m on roaming NNI between home IMS and visited IMS, else n/a
52	Proxy-Require	[5]	m
53	Reason	[5] and sub- clause 6.1.1.3.1 (Table 6.2, item 11)	o when in a request. When in a response, m in case of a trust relationship between the interconnected networks, else n/a
54	Record-Route	[5]	m
55	Referred-By	[5]	m
56	Reject-Contact	[5]	m
57	Replaces	[5]	0
58	Reply-To	[5]	0
59	Request-Disposition	[5]	m
60	Require	[5]	m
61	Resource-Priority	sub-clause 6.1.1.3.1 (Table 6.2, item 3)	m in case of a trust relationship between the interconnected networks, else n/a
62	Route	[5]	m
63	Security-Client	[5]	n/a
64	Security-Verify	[5]	n/a
65	Server	[5]	0
65a	Service-Route	[5]	m on roaming NNI between home and visited IMS, else n/a
65b	Session-ID	[5]	0
66	Session-Expires	[5]	m
67	Subject	[5]	0
68	Supported	[5]	m
69	Timestamp	[5]	m
70	To	[5]	m
71	Trigger-Consent	[5]	m
71a	Unsupported	[5]	m
72	User-Agent	[5]	m
73	User-to-User	[5]	0
74	Via	[5]	m
75	Warning	[5]	O see an accoming NINII between bears and visited IMC also n/s
76	WWW-Authenticate	[5]	m on roaming NNI between home and visited IMS, else n/a

Table A.2: Key to notation codes for SIP header fields

Notation code	Meaning
m	The SIP header field is applicable at II-NNI. Supporting a SIP header field at the II-NNI means that this header field is passed through the IBCF. It does not imply that network elements inside the serving and served networks or user equipment connected to these networks shall support this header field, where 3GPP TS 24.229 [5] is applied. If specified in 3GPP TS 24.229, the IBCF modifies the SIP header field.
0	The applicability of SIP header field at II-NNI depends on bilateral agreement between the operators.
n/a	It is impossible to use the SIP header field at the II-NNI. This header field could be discarded by the IBCF.

Annex B: Change history

	I=0.0	I=0.0 -		_	Change history		
Date	TSG #	TSG Doc.	CR	Rev	Subject/Comment	Old	New
4/05/2008					TS Skeleton (C3-080779)	-	0.0.0
07/07/2008					Added agreed text of C3-080991, C3-081158 and C3-081208	0.0.0	0.1.0
28/08/2008					Added agreed text of C3-081282 and C3-081672	0.1.0	0.2.0
01/09/2008					Version 1.0.0 created for presentation to TSG by MCC	0.2.0	1.0.0
17/10/2008					Added agreed text of C3-081721 and C3-082105	1.0.0	1.1.0
20/11/2008					Added agreed text of C3-082303, C3-082446, C3-082447 and C3-082611	1.0.0	1.2.0
26/11/2008					v 2.0.0 was produced by MCC for Approval in CT#42	1.2.0	2.0.0
13/12/2008	TSG#42				V 8.0.0 was produced by MCC	2.0.0	8.0.0
03/2008	TSG#43	CP-090087	002	3	Charging requirements on II-NNI	8.0.0	8.1.0
03/2008	TSG#43	CP-090087	004	1	Modification of the REFER method status	8.0.0	8.1.0
03/2008	TSG#43	CP-090087	007	2	NNI header tables	8.0.0	8.1.0
05/2009	TSG#44	CP-090341	008	4	Use of E.164 number at the II-NNI	8.1.0	8.2.0
05/2009	TSG#44	CP-090341	009	4	Correction to SIP headers table	8.1.0	8.2.0
09/2009	TSG#45	CP-090576		1	Removal of left-over text from TS drafting phase and update of a reference	8.2.0	8.3.0
09/2009	TSG#45	CP-090576	018	2	Applicability of SIP headers for roaming II-NNI	8.2.0	8.3.0
09/2009	TSG#45	CP-090576		1	Application level gateway usage to enable communication from	8.2.0	8.3.0
00/2000	T90#45	CD 000570	020	2	private IP address space Codecs at the NNI	9 2 0	020
09/2009	TSG#45	CP-090576		3		8.2.0	8.3.0
09/2009	TSG#45	CP-090584		4	Major capabilities on II-NNI	8.3.0	9.0.0
09/2009	TSG#45	CP-090584		4	Management of SIP headers over II-NNI in presence of trust or no trusted relationship (V1)	8.3.0	9.0.0
09/2009	TSG#45	CP-090584	015	4	Requirements for the end-to-end interoperability of supplementary services	8.3.0	9.0.0
09/2009	TSG#45	CP-090584	016	3	Deletion of the note about the normalization of phone numbers	8.3.0	9.0.0
12/2009	TSG#46	CP-090854	021	5	Requirements for HOLD service over II-NNI	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	022	5	Requirements for CW service over II-NNI	9.0.0	9.1.0
12/2009	TSG#46	CP-090844		3	Aligning references to P-Asserted-Service	9.0.0	9.1.0
12/2009	TSG#46	CP-090844			Annex A header updated with Answer-Mode, Priv-Answer-Mode and P-Answer-State	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	031	6	Filling of the table about major capabilities on II-NNI	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		1	Customized Ringing Signal (CRS) modification	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		2	Completing the Personal Network Management (PNM)	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	034	1	supplementary service Aligning existing supplementary services	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		1	Completing the Flexible Alerting (FA) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		1	Completing the Closed User Group (CUG) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		3	Completing the Closed Oser Group (COG) supplementary service Completing the Three-Party (3PTY) and Conference (CONF)	9.0.0	9.1.0
					supplementary services		
12/2009	TSG#46	CP-090854	038	3	Completing the Anonymous Communication Rejection (ACR) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	039	3	Completing Completion of Communications to Busy Subscriber (CCBS) and Completion of Communications by No Reply (CCNR) supplementary services	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	040		Completing Message Waiting Indication (MWI) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	041	1	Completing the Terminating Identification Presentation (TIP) and Terminating Identification Restriction (TIR) needs to be completed.	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	042	3	Completing the Communication Barring (CB) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	043	2	Completing Explicit Communication Transfer (ECT)	9.0.0	9.1.0
12/2009	TSG#46	CP-090854		3	Completing Communication DIVersion (CDIV) supplementary	9.0.0	9.1.0
12/2000	T90#46	CD 000054	046	2	Services	0.0.0	0.1.0
12/2009	TSG#46	CP-090854		2	Deletion of an editors note on OIR service	9.0.0	9.1.0
12/2009	TSG#46	CP-090844		3	Annex A header updated	9.0.0	9.1.0
03/2010	TSG#47	CP-100077		2	Format of Request URI	9.1.0	9.2.0
03/2010	TSG#47	CP-100077	055	2	MSC Server enhanced for ICS missing in architecture	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	058		AOC added to supplementary services	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	059		CPC and OLI IETF reference update	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	060	3	CPC and OLI and trust domain	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	061	1	Modifying CUG interactions	9.1.0	9.2.0
03/2010 03/2010	TSG#47	CP-100087 CP-100087	062 063	4	Correcting minor errors Updating Major Capability according to latest 24.229	9.1.0	9.2.0 9.2.0

03/2010	TSG#47	CP-100087	064	1	Major Capabilities revision 2	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	067	1	Table 6.2 update	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	068	1	Modification of OIP/OIR paragraph description (R9 29.165)	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	071	1		9.1.0	9.2.0
			_		References numbers update (R9 29.165)		
06/2010	TSG#48	CP-100319	073	1	The Session-ID added to annex A	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	077	2	Correcting incorrect interface name	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	079	2	Supported URI formats also based on operator agreements	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	080	2	Customized Alerting Tone (CAT)	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	081	2	Customized Ringing Signal (CRS)	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	082	1	Correcting miscellaneous errors	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	084	2	Applicability of "critical" privacy value	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	085		Making Alert-Info not mandatory	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	086	1	Draft-johnston-sipping-cc-uui reference updated to new version	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	087	2	Correcting references for removal of P-Asserted-Identity	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	091	2	Correction of Mayor Capabilities	9.2.0	9.3.0
06/2010	TSG#48	CP-100319	093	2	Usage of "Refer" for conferencing and 3PTY	9.2.0	9.3.0
09/2010	TSG#49	CP-100553	095	2	Support of REFER checked by entry IBCF	9.3.0	9.4.0
09/2010	TSG#49	CP-100545	098	2	Removing unnecessary normative text	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	100	2	CCBS and CCNR and the m-parameter	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	103	1	Correcting references	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	105		Adding missing "history" privacy value to CDIV	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	108	3	MCID and identity information transfer clarification	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	110	2	Correcting name of the Authenticate header field and the applicability over II-NNI	9.3.0	9.4.0
09/2010	TSG#49	CP-100553	112	1	Aligning Major Capabilities Table with 24.229	9.3.0	9.4.0

History

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