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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.

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- [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".
- [6] 3GPP TR 24.930: "Signalling flows for the session setup in the IP Multimedia core network Subsystem (IMS) based on Session Initiation Protocol (SIP) and Session Description Protocol (SDP); Stage 3".
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- [8] 3GPP TS 29.162: "Interworking between the IM CN subsystem and IP networks".
- [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".
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- [14] IETF RFC 3966: "The tel URI for Telephone Numbers".
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- [25] IETF RFC 4244: "An Extension to the Session Initiation Protocol (SIP) for Request History Information".
- [26] IETF RFC 6050: "A Session Initiation Protocol (SIP) Extension for the Identification of Services".
- [27] IETF RFC 4168: "The Stream Control Transmission Protocol (SCTP) as a Transport for the Session Initiation Protocol (SIP)".
- [28] Void.
- [29] 3GPP TS 32.260: "Telecommunication management; Charging management; IP Multimedia Subsystem (IMS) charging".
- [30] 3GPP TS 22.173: "IP Multimedia Core Network Subsystem (IMS); Multimedia Telephony Service and supplementary services; Stage 1".
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- [32] 3GPP TS 24.607: "Originating Identification Presentation (OIP) and Originating Identification Restriction (OIR) using IP Multimedia (IM); Core Network (CN) subsystem".
- [33] 3GPP TS 24.616: "Malicious Communication Identification (MCID) using IP Multimedia (IM) Core Network (CN) subsystem".
- [34] IETF RFC 3323: "A Privacy Mechanism for the Session Initiation Protocol (SIP)".
- [35] 3GPP TS 23.003: "Numbering, addressing and identification".
- [36] 3GPP TS 24.610: "Communication HOLD (HOLD) using IP Multimedia (IM) Core Network (CN) subsystem".
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- [40] IETF RFC 3312: "Integration of resource management and Session Initiation Protocol (SIP)".
- [41] IETF RFC 4032: "Update to the Session Initiation Protocol (SIP) Preconditions Framework".
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- [49] IETF RFC 6432: "Carrying Q.850 Codes in Reason Header Fields in SIP (Session Initiation Protocol) Responses".
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- [52] IETF RFC 4028: "Session Timers in the Session Initiation Protocol (SIP)".
- [53] IETF RFC 3892: "The Session Initiation Protocol (SIP) Referred-By Mechanism".
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- [55] IETF RFC 3911: "The Session Initiation Protocol (SIP) 'Join' Header".
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- [57] IETF RFC 5079: "Rejecting Anonymous Requests in the Session Initiation Protocol (SIP)".
- [58] IETF RFC 4458: "Session Initiation Protocol (SIP) URIs for Applications such as Voicemail and Interactive Voice Response (IVR)".
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- [69] IETF RFC 5368: "Referring to Multiple Resources in the Session Initiation Protocol (SIP)".
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- [101] 3GPP TS 24.239: "Flexible Alerting (FA) using IP Multimedia (IM) Core Network (CN) subsystem".
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- [103] 3GPP TS 24.654: "Closed User Group (CUG) using IP Multimedia (IM) Core Network (CN) subsystem".
- [104] Void.
- [105] 3GPP TS 24.605: "Conference (CONF) using IP Multimedia (IM) Core Network (CN) subsystem".
- [106] 3GPP TS 24.147: "Conferencing using the IP Multimedia (IM) Core Network (CN) subsystem".
- [107] Void.
- [108] Void.
- [109] 3GPP TS 24.642: "Completion of Communications to Busy Subscriber (CCBS) Completion of Communications by No Reply (CCNR) using IP Multimedia (IM) Core Network (CN) subsystem".
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- [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: "Anonymous Communication Rejection (ACR) and Communication Barring (CB) using IP Multimedia (IM) Core Network (CN) subsystem".
- [115] Void.
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- [118] Void.
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- [121] 3GPP TS 24.292: "IP Multimedia Core Network subsystem Centralized Services; Stage 3".
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- [123] Void.
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- [128] Void.
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- [137] IETF RFC 3550: "RTP: A Transport Protocol for Real-Time Applications".
- [138] IETF RFC 768: "User Datagram Protocol".
- [139] IETF RFC 3551: "RTP Profile for Audio and Video Conferences with Minimal Control".
- [140] IETF RFC 3556: "Session Description Protocol (SDP) Bandwidth Modifiers for RTP Control Protocol (RTCP) Bandwidth".
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- [143] IETF RFC 4916: "Connected Identity in the Session Initiation Protocol (SIP)".
- [144] Void.

3 Definitions, symbols and abbreviations

3.1 Definitions

For the purposes of the present document, the terms and definitions given in 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 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 TS 22.228 [9].

IP multimedia session: as specified in 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.

non-roaming II-NNI: the II-NNI between IMS home networks.

roaming II-NNI: the II-NNI between a visited IMS network and the IMS home network.

For the purposes of the present document, the following terms and definitions given in 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 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 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
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 TS 23.002 [3] and 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 TS 24.229 [5]. Example call flows are provided in TR 24.930 [6].

The general interconnection model is shown in Figure 4.1.

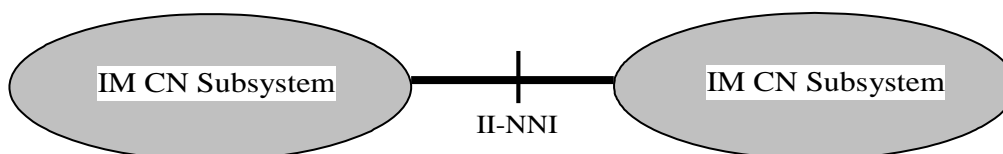


Figure 4.1: Interconnection Model for IM CN subsystems

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 TS 24.229 [5], in TS 24.292 [121] and in TS 29.162 [8].

IP Version interworking is described within 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 TS 23.228 [4] showing the Inter-IMS Network to Network Interface (II-NNI) between two IM CN subsystem networks.

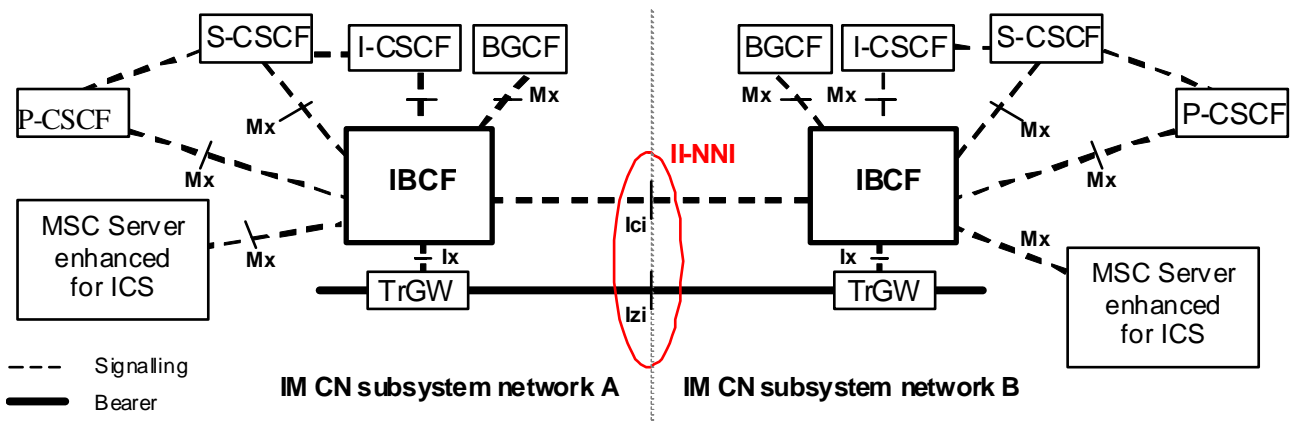


Figure 5.1.1: Inter-IMS Network to Network Interface between two IM CN subsystem networks

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 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 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 TS 23.228 [4] and specified in 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 TS 24.229 [5].

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

5.2.2 Transition Gateway (TrGW)

According to 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 TS 29.162 [8].

Further details are described in 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 TS 24.229 [5] with modifications as described in the following subclauses.

6.1.1.2 SIP methods

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 TS 24.229 [5] and endorsed for this document:

Table 6.1: Supported SIP methods

Item	Method	Ref.	II-NNI	
			Sending	Receiving
1	ACK request	IETF RFC 3261 [13]	m	m
2	BYE request	IETF RFC 3261 [13]	m	m
3	BYE response	IETF RFC 3261 [13]	m	m
4	CANCEL request	IETF RFC 3261 [13]	m	m
5	CANCEL response	IETF RFC 3261 [13]	m	m
5A	INFO request	IETF RFC 6086 [39]	o	o
5B	INFO response	IETF RFC 6086 [39]	o	o
8	INVITE request	IETF RFC 3261 [13]	m	m
9	INVITE response	IETF RFC 3261 [13]	m	m
9A	MESSAGE request	IETF RFC 3428 [19]	o	o
9B	MESSAGE response	IETF RFC 3428 [19]	o	o
10	NOTIFY request	IETF RFC 3265 [20]	c1	c1
11	NOTIFY response	IETF RFC 3265 [20]	c1	c1
12	OPTIONS request	IETF RFC 3261 [13]	m	m
13	OPTIONS response	IETF RFC 3261 [13]	m	m
14	PRACK request	IETF RFC 3262 [18]	m	m
15	PRACK response	IETF RFC 3262 [18]	m	m
15A	PUBLISH request	IETF RFC 3903 [21]	c1	c1
15B	PUBLISH response	IETF RFC 3903 [21]	c1	c1
16	REFER request	IETF RFC 3515 [22]	o	o
17	REFER response	IETF RFC 3515 [22]	o	o
18	REGISTER request	IETF RFC 3261 [13]	c2	c2
19	REGISTER response	IETF RFC 3261 [13]	c2	c2
20	SUBSCRIBE request	IETF RFC 3265 [20]	c1	c1
21	SUBSCRIBE response	IETF RFC 3265 [20]	c1	c1
22	UPDATE request	IETF RFC 3311 [23]	m	m
23	UPDATE response	IETF RFC 3311 [23]	m	m
c1:	In 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:	In the above table, m, o and c and n/a have the meanings indicated in Table 6.3			

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 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 subclause 5.10.2 of 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 subclause 5.10.3 of TS 24.229 [5].

Additionally, in case there is no trust relationship between the two IM CN subsystems connected by II-NNI, the IBCF applies the procedures described in subclause 4.4 of TS 24.229 [5], before forwarding the SIP signalling.

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 CN 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	Reference	Trust relationship	Not trust relationship
1	P-Asserted-Identity	IETF RFC 3325 [44]	As specified in TS 24.229 [5], subclause 4.4 (NOTE 5)	As specified in TS 24.229 [5], subclause 4.4 (NOTE 5)
2	P-Access-Network-Info	IETF RFC 3455 [24]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
3	Resource-Priority	IETF RFC 4412 [78]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
4	History-Info	IETF RFC 4244 [25]	As specified in TS 24.229 [5], subclause 4.4	As specified in subclause 4.3.3 of RFC 4244 [25] and in TS 24.229 [5], subclause 4.4
5	P-Asserted-Service	draft-drage-sipping-service-identification [26]	As specified in TS 24.229 [5], subclause 4.4 (NOTE 3)	As specified in TS 24.229 [5], subclause 4.4 (NOTE 3)
6	P-Charging-Vector	IETF RFC 3455 [24]	As specified in TS 24.229 [5], subclause 5.10	As specified in TS 24.229 [5], subclause 5.10
7	P-Charging-Function-Addresses (NOTE 4)	IETF RFC 3455 [24]	As specified in TS 24.229 [5], subclause 5.10	As specified in TS 24.229 [5], subclause 5.10
8	P-Profile-Key (NOTE 2)	IETF RFC 5002 [64]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
9	P-Private-Network-Indication (NOTE 1)	IETF RFC 7316 [84]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
10	P-Served-User (NOTE 1, NOTE 2)	IETF RFC 5502 [85]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
11	Reason (in a response)	IETF RFC 6432[49]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
12	P-Early-Media	IETF RFC 5009 [74]	As specified in TS 24.229 [5], subclause 4.4	As specified in TS 24.229 [5], subclause 4.4
<p>NOTE 1: For a roaming II-NNI, a trust relationship with respect to this header field is required.</p> <p>NOTE 2: This header field is only applicable on a roaming II-NNI.</p> <p>NOTE 3: In addition, value-dependent operator policies may be applied.</p> <p>NOTE 4: This header field is not applicable at II-NNI.</p> <p>NOTE 5: The handling of the URI parameters "cpc" and "oli", defined in TS 24.229 [5] subclause 7.2A.12, is specified in TS 24.229 [5], subclause 4.4.</p>				

6.1.1.3.2 Derivation of applicable SIP header fields from 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 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 TS 24.229 [5] as follows:

- All header fields not present in the corresponding tables in Annex A of TS 24.229 [5] 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 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.
- 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 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

The following SIP header fields are only applicable on a roaming II-NNI:

- Authentication-Info
- Authorization
- P-Associated-URI
- P-Called-Party-ID
- P-Preferred-Service
- P-Profile-Key
- P-Served-User
- P-Visited-Network-ID
- Path
- Proxy-Authenticate
- Proxy-Authorization
- Service-Route
- WWW-Authenticate

6.1.1.3.4 Applicability of SIP header fields on a non-roaming II-NNI

Void

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:

Table 6.3: Key to notation codes for SIP messages

Notation code	Notation name	Sending side	Receiving side
m	mandatory	The message shall be supported at II-NNI. Supporting sending a SIP message at the II-NNI means that this message shall be sent over the II-NNI if received from the serving network. It does not imply that network elements inside the serving network or user equipment connected to this network shall support this message.	Supporting receiving a SIP message at the II-NNI means that this message shall be forwarded to the serving network unless the operator's policy is applied as defined in subclause 5.10.1 of TS 24.229 [5]. It does not imply that network elements inside the served network or user equipment connected to this network are supporting this message.
o	optional	The message may or may not be supported at II-NNI. The support of the message is provided based on bilateral agreement between the operators.	Same as for sending side.
n/a	not applicable	It is impossible to use/support the message.	It is impossible to use/support the message. This message will be discarded by the IBCF.
c <integer>	conditional	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.	Same as for sending side.

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.

6.1.2 SDP protocol

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 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 2nd 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 TS 24.229 [5] for the profile status" provide informative references for comparison purposes into the UA and Proxy role major capabilities tables in 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 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	n/a
5	Managing several responses due to forking	9,10	6	m
6	support of indication of TLS connections in the Record-Route header	-	7, 8	n/a
7	Support of 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	o
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	o
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	-	19E	m
16A	Proxy reading the contents of a body or including a body in a request or response	-	19F	n/a
	Extensions to basic SIP			
17	IETF RFC 6086 [39]: SIP INFO method and package framework	13	20	o
17A	IETF RFC 6086 [39]: legacy INFO usage	13A	20A	o
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	o
20	IETF RFC 3312 [40] and RFC 4032 [41]: integration of resource management and SIP (Preconditions framework)	16	23	o
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	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	o
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 session initiation protocol for the 3rd-Generation Partnership Project (3GPP)	30	35	o
32A	IETF RFC 3325 [44]: act as first entity within the trust domain for asserted identity	30A	30A	n/a
32B	IETF RFC 3325 [44]: act as entity within trust network that can route outside the trust network	30B	30B	n/a
32C	IETF RFC 3325 [44]: 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	c1
39	IETF RFC 3329 [47]: security mechanism agreement for the session initiation protocol	37	47	n/a
39A	TS 24.229 [5] subclause 7.2A.7: Capability Exchange for Media Plane Security	37A	47A	n/a
40	IETF RFC 3326 [48]: the Reason header field for the session initiation protocol	38	48	o
41	IETF RFC 6432 [49]: carrying Q.850 codes in reason header fields in SIP (Session Initiation Protocol) responses	38A	48A	c4
42	IETF RFC 3581 [50]: an extension to the session initiation protocol for symmetric response routing	39	49	o
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	o
48	IETF RFC 3911 [55]: the Session Initiation Protocol (SIP) "Join" header	45	55	o
49	IETF RFC 3840 [56]: the callee capabilities	46	56	o
50	IETF RFC 4244 [25]: an extension to the session initiation protocol for request history information (History-Info header field)	47	57	o
51	IETF RFC 5079 [57]: Rejecting anonymous requests in the session initiation protocol	48	58	o
52	IETF RFC 4458 [58]: session initiation protocol URIs for applications such as voicemail and interactive voice response (NOTE 3)	49	59	o
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 (URN) for Emergency and Other Well-Known Services	52	62	n/a
56	IETF RFC 5627 [62]: obtaining and using GRUUs in the Session Initiation Protocol (SIP)	53	63	c1
57	Void			
58	IETF RFC 4168 [27]: the Stream Control Transmission Protocol (SCTP) as a Transport for the Session Initiation Protocol (SIP)	55	65	o
59	IETF RFC 5002 [64]: the SIP P-Profile-Key private header field extension	56	66, 66A, 66B	c3

60	IETF RFC 5626 [65]: managing client initiated connections in SIP	57	67	c1
61	IETF RFC 5768 [66]: indicating support for interactive connectivity establishment in SIP	58	-	n/a
62	IETF RFC 5365 [67]: multiple-recipient MESSAGE requests in the session initiation protocol	59	69	o if 29, else n/a
63	IETF RFC 6442 [68]: Location Conveyance for the Session Initiation Protocol	60	70, 70A, 70B	m
64	IETF RFC 5368 [69]: referring to multiple resources in the session initiation protocol	61	71	o if 19, else n/a
65	IETF RFC 5366 [70]: conference establishment using request-contained lists in the session initiation protocol	62	72	o
66	IETF RFC 5367 [71]: subscriptions to request-contained resource lists in the session initiation protocol	63	73	o if 23, else n/a
67	IETF RFC 4967 [72]: dialstring parameter for the session initiation protocol uniform resource identifier	64	74	c2
68	IETF RFC 4964 [73]: the P-Answer-State header extension to the session initiation protocol for the open mobile alliance push to talk over cellular	65	75	o
69	IETF RFC 5009 [74]: the SIP P-Early-Media private header field extension for authorization of early media	66	76	c4
70	IETF RFC 4694 [75]: number portability parameters for the 'tel' URI	67, 67A, 67B	77, 77A, 77B	o
72	IETF RFC 4411 [77]: extending the session initiation protocol Reason header for preemption events	69	79	o
73	IETF RFC 4412 [78]: communications resource priority for the session initiation protocol (Resource-Priority header field)	70, 70A, 70B	80, 80A, 80B	o
74	IETF RFC 5393 [79]: addressing an amplification vulnerability in session initiation protocol forking proxies	71	81	m
75	IETF RFC 5049 [80]: the remote application identification of applying signalling compression to SIP	72	82	n/a
76	IETF RFC 5688 [81]: a session initiation protocol media feature tag for MIME application sub-types	73	83	c1
77	IETF RFC 6050 [26]: Identification of communication services in the session initiation protocol	74	84, 84A	o
78	IETF RFC 5360 [82]: a framework for consent-based communications in SIP	75, 75A, 75B	85	o
79	IETF RFC 7433 [83]: a mechanism for transporting user-to-user call control information in SIP	76	86	c1
79A	IETF RFC 7434 [83A]: interworking ISDN call control user information with SIP	76A	-	c1
80	IETF RFC 7316 [84]: The SIP P-Private-Network-Indication private header (P-Header)	77	87	c1
81	IETF RFC 5502 [85]: the SIP P-Served-User private header	78	88	c2
84	IETF RFC 6228 [88]: the 199 (Early Dialog Terminated) response code	81	91	m
85	IETF RFC 5621 [89]: message body handling in SIP	82	92	m
86	IETF RFC 6223 [90]: indication of support for keep-alive	83	93	o
87	IETF RFC 5552 [91]: SIP Interface to VoiceXML Media Services	84	94	n/a
88	IETF RFC 3862 [92]: common presence and instant messaging (CPIM): message format	85	95	o
89	IETF RFC 5438 [93]: instant message disposition notification	86	96	o
90	IETF RFC 5373 [94]: requesting answering modes for SIP (Answer-Mode and Priv-Answer-Mode header fields)	87	97, 97A	o
	Void			
92	IETF RFC 3959 [96]: the early session disposition type for SIP	89	99	o
93	Subclause 7.2A.16 of TS 24.229 [5]: delivery of Request-URI targets to user agents	90	100	o
94	IETF RFC 7329 [124]: A Session Identifier for the Session Initiation Protocol (SIP)	91	101	o
95	IETF RFC 6026 [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	o
97	IETF RFC 5954 [127]: essential correction for IPv6 ABNF	94	104	m

	and URI comparison in IETF RFC 3261 [13]			
98	IETF RFC 4488 [132]: suppression of session initiation protocol REFER method implicit subscription	95	105	c5
99	IETF RFC 7462 [133]: Alert-Info URNs for the Session Initiation Protocol	96	106	o
100	Subclause 3.1 of TS 24.229 [5]: multiple registrations	97	107	c2
101	IETF RFC 4538 [135]: request authorization through dialog Identification in the session initiation protocol (Target-Dialog header field)	99	109	o
c1: m in case of roaming II-NNI, else o c2: m in case of roaming II-NNI, else n/a c3: o in case of roaming II-NNI, 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 TS 24.229 [5].				
NOTE 2: The item numbering corresponds to the one provided in table A.162 in TS 24.229 [5].				
NOTE 3: A common URI namespace is required to apply this feature on the II-NNI.				

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.
o	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>	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.

6.2 Control Plane Transport

6.2.1 General

The control plane transport of the II-NNI shall comply with subclause 4.2A of 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 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 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	IETF RFC 3550 [137]	RTP: A Transport Protocol for Real-Time Applications	Mandatory
2	IETF RFC 768 [138]	User Datagram Protocol	Mandatory
3	IETF RFC 3551 [139]	RTP Profile for Audio and Video Conferences with Minimal Control	Mandatory
4	IETF RFC 3556 [140]	Session Description Protocol (SDP) Bandwidth Modifiers for RTP Control Protocol (RTCP) Bandwidth	Mandatory
5	IETF RFC 4585 [141]	Extended RTP Profile for Real-time Transport Control Protocol (RTCP) - Based Feedback (RTP/AVPF)	Optional (NOTE 1)
6	IETF RFC 793 [142]	Transmission Control Protocol	Optional (NOTE 2)

NOTE 1: used by MTSI, as indicated in TS 26.114 [11]
NOTE 2: used for MSRP service

8 Numbering, Naming and Addressing

8.1 Numbering, Naming and Addressing for SIP message

The following URI formats in SIP messages may be applied at the ICI as standardized in 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].

According to TS 24.229 [5], the IBCF acting as an exit or entry point in the IMS network supports these URI formats. These URI formats shall be supported at the roaming II-NNI. The SIP URI format shall be supported at the non-roaming II-NNI. The tel URI, IM URI and PRES URI formats may be supported at the non-roaming II-NNI based on agreement between operators. Other URI formats may be supported over the II-NNI depending on the operator 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 II-NNI 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: 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 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.

The optional "oli" and "cpc" tel URI parameters associated with a tel URI or a SIP URI with user=phone are described in TS 24.229 [5] and can be part of the P-Asserted-Identity header field. Depending on operator agreements, those URI parameters may be supported at the non-roaming II-NNI.

The "sos" SIP URI parameter associated with a URI in the Contact header field of a REGISTER request or 200 OK response to REGISTER request is described in TS 24.229 [5]. The "sos" SIP URI parameter shall be supported at the roaming II-NNI.

The "rn" and "npdi" number portability parameters for the tel URI and the SIP URI with user=phone as described within IETF RFC 4694 [75] can be part of the Request-URI. Depending on operator agreements these parameters may be exchanged over the non-roaming II-NNI.

NOTE 6: The "rn" and "npdi" parameters can be used to address the entry point of the terminating operator depending on national rules for number portability.

The "isub" tel URI parameter for the tel URI and the SIP URI with user=phone as described within IETF RFC 3966 [14] can be part of the Request-URI, To header field and P-Asserted-Identity header field. Depending on operator agreements, this URI parameter may be exchanged over the II-NNI.

8.2 Numbering, Naming and Addressing for SDP

The following URI format in the SDP exchange may be applied at the Ici as standardized in TS 24.229 [5]:

- MSRP URI for a session of instant messages defined in IETF RFC 4975 [17].

This URI format shall be supported at the roaming II-NNI and may be supported at the non-roaming II-NNI based on agreement between operators. Other URI formats may be supported over the II-NNI depending on the operators' agreements.

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 TS 29.162 [8].

10 Security

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

11 Charging

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

12 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), the associated supplementary services of the multimedia telephony communication service may be supported on the II-NNI between the two IMS networks.

The MMTel communication service is identified by means of the media feature tag +g.3gpp.icsi-ref set to "urn:urn-7:3gpp-service.ims.icsi.mmtel". The media feature tag can appear in the Contact header field, the Accept-Contact header field and the P-Asserted-Service header field.

The support of each associated supplementary service is based on agreement between operators.

If a supplementary service is supported, the related procedures from the TS 22.173 [30], the protocol details from the TS 24.173 [31] and specifications referenced in the later specification shall be applied with the requirements in the relevant subclause due to the crossing of the II-NNI.

12.2 Malicious Communication IDentification (MCID)

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

The P-Asserted-Identity header field shall be supported at the II-NNI.

The INFO request and the 200 (OK) response to the INFO request containing the "application/vnd.etsi.mcid+xml" MIME body defined in 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 "application/vnd.etsi.mcid+xml" MIME body, as specified in the TS 24.616 [33], if an agreement to use the MCID supplementary service according to the 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 TS 24.616 [33] and 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 TS 24.616 [33] between the networks.

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

Service specific requirements in accordance with 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 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 TS 24.229 [5]. The IBCF determines whether to remove the P-Asserted-Identity header field according to procedures in TS 24.229 [5] subclause 4.4.2 referencing IETF RFC 3325 [44].

The option tag "from-change" in the Supported header field should be supported at II-NNI.

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 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 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 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 TS 24.229 [5]. The IBCF determines whether to remove the P-Asserted-Identity header field according to procedures in TS 24.229 [5] subclause 4.4.2 referencing IETF RFC 3325 [44].

The option tag "from-change" defined in IETF RFC 4916 [143], in the Supported header field should be supported at II-NNI.

12.5 Anonymous Communication Rejection (ACR)

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

The P-Asserted-Identity header field and the Privacy header field shall be supported at the II-NNI.

Procedures as described in subclause 12.21.4 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 TS 24.238 [100] shall be supported at the roaming II-NNI.

12.6 Communication DIVersion (CDIV)

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

NOTE 1: The support of the Diversion header field not adopted in TS 24.604 [117] requires bilateral agreement between the operators.

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

The Privacy header field with a priv-value set to "history" included in the hi-targeted-to-uri or as a standalone header field shall be supported at the II-NNI.

The History-Info header field as described by TS 24.604 [117] and a "cause" SIP URI parameter with cause values as defined by the IETF RFC 4458 [58] shall be supported over the II-NNI.

NOTE 2: The networks can have an internal limit in the number of allowed diversions, as described in TS 24.604 [117], section 4.5.2.6.1. To ensure efficiency of this control operators can indicate in their bilateral agreements their own number of allowed communication diversions, the parameter that is used for counting, and the network behavior when the internal limit is reached.

The response code 181 (Call Is Being Forwarded) shall be supported at the II-NNI.

The MESSAGE request procedure for indication of communication diversion to the diverting user as specified in TS 24.604 [117] and TS 24.229 [5] should be supported at the roaming II-NNI.

NOTE 3: The content of the MESSAGE request is operator specific.

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

12.7 Communication Waiting (CW)

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

The "application/vnd.3gpp.cw+xml" MIME body defined in TS 24.615 [37] in the INVITE request 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 TS 24.610 [36] shall be supported over the II-NNI.

NOTE: The support of an alternative method not adopted in TS 24.610 [36] 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 TS 24.606 [112] shall be supported over the II-NNI.

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

The application/simple-message-summary+xml MIME body described in 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 TS 24.611 [114] shall be supported over the II-NNI.

Procedures as described in subclause 12.21.4 are used to provide announcements.

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

A Reason header field as described in TS 24.611 [114] included in the BYE request shall be supported at the II-NNI.

SIP based user configuration as described in 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 TS 24.611 [114] shall be supported over the II-NNI.

Procedures as described in subclause 12.21.4 are used to provide announcements.

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

SIP based user configuration as described in 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 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 12.21.3 and subclause 12.21.4 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 TS 24.229 [5] shall be supported at the roaming II-NNI.

As a network option the special REFER request handling procedures according to TS 24.628 [38] 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 TS 24.229 [5].

The SUBSCRIBE and NOTIFY methods according to IETF RFC 3265 [20] and 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 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.

The Date header field in the 486 (Busy Here) response to the INVITE request shall be supported at the roaming II-NNI.

SIP based user configuration as described in 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 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 II-NNI.

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 12.21.3 and subclause 12.21.4 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 TS 24.229 [5] shall be supported at the roaming II-NNI.

As a network option the special REFER request handling procedures according to TS 24.628 [38] 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 TS 24.229 [5].

The SUBSCRIBE and NOTIFY methods according to IETF RFC 3265 [20] and 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 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.

The Date header field in the 480 (Temporarily Unavailable) response to the INVITE request shall be supported at the roaming II-NNI.

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

12.13 Explicit Communication Transfer (ECT)

Service specific requirements in accordance with 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 TS 24.629 [116] and the NOTIFY method containing an " application/sipfrag " MIME body 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 TS 24.629 [116] and the NOTIFY method containing an " application/sipfrag " MIME body shall be supported at the roaming II-NNI for call transfer with third party call control.

The Refer-To URI header parameter in the REFER request containing the Require header field set to "replaces" shall be supported at the roaming II-NNI.

The Replaces header field in the INVITE request shall be supported at the non-roaming II-NNI.

12.14 Customized Alerting Tone (CAT)

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

The P-Early-Media header field in as described in 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 " application/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 TS 24.229 [5] may be supported at the II-NNI.

NOTE 1: For telephone-event based DTMF transport, the DTMF digits are sent as media and not visible in the control plane.

NOTE 2: Multiple methods for DTMF transport are defined in TS 24.182 [129].

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

12.15 Customized Ringing Signal (CRS)

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

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

A SDP "a=content" attribute with a "g.3gpp.crs" value in the PRACK request or the re-INVITE request 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 "application/sdp" MIME body with the Content- Disposition header field 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 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 TS 24.238 [100] shall be supported at the roaming II-NNI.

12.16 Closed User Group (CUG)

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

The "application/vnd.etsi.cug+xml" MIME body as specified 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 " application/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 TS 24.238 [100] shall be supported at the roaming II-NNI.

12.17 Personal Network Management (PNM)

Service specific requirements in accordance with 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 tag 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 TS 24.259 [99] shall be supported at II-NNI.

12.18 Three-Party (3PTY)

Service specific requirements in accordance with 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 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 the non-roaming II-NNI, 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 non-roaming II-NNI, 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 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 TS 24.229 [5].

The application/resource-lists+xml MIME 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 TS 24.147 [106] shall be supported at the II-NNI.

The Allow-Events header field with the value "conference" shall be supported at the roaming II-NNI and may be supported at the non-roaming II-NNI.

12.20 Flexible Alerting (FA)

Service specific requirements in accordance with 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 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, during an established communication session or when a communication request is rejected. All 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 TS 24.628 [38] are used to provide announcements.

The P-Early-Media header authorizing early media as defined in IETF RFC 5009 [74] 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 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.

12.21.4 Providing announcements when communication request is rejected

Procedures as described in TS 24.628 [38] are used to provide announcements when a communication request is rejected.

There are three methods defined in TS 24.628 [38] to provide the announcement:

- 1) sending an announcement as an early media;
- 2) return an Error-Info header field; and
- 3) accept the communication request and then provide the announcement.

NOTE 1: The II-NNI requirements for accepting the communication request and then provide the announcement is not within the scope of this subclause.

The P-Early-Media header field authorizing early media as defined in IETF RFC 5009 [74] and the Reason header field with the proper cause value shall be supported at the II-NNI.

NOTE 2: There are 2 methods to use early media for sending the announcement in-band. First method is the gateway model defined by IETF RFC 3960 [136], second method is described in TS 24.628 [38] Annex D.

The Error-Info header field in the 3xx, 4xx, 5xx or 6xx response to the INVITE request when rejecting the communication request, should be supported at the II-NNI.

NOTE 3: The IBCF can decide to remove the Error-Info header field if required by local policy.

12.22 Advice of Charge (AOC)

Service specific requirements in accordance with 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 an application/vnd.etsi.aoc+xml MIME body 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 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" in the BYE method shall be supported at the II-NNI.

An application/vnd.etsi.aoc+xml MIME body in the BYE request or the final response to the BYE request 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 TS 24.229 [5]. In case of misalignment between Table A.1 and the behaviour described in TS 24.229 [5], the behaviour in TS 24.229 [5] has the precedence. In case a header field is not described in Table A.1 and it is described in TS 24.229 [5], the description in 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
4a	Accept-Resource-Priority	[5]	o
5	Alert-Info	[5]	o
6	Allow	[5]	m
7	Allow-Events	[5]	m on roaming II-NNI, else o
8	Authentication-Info	[5]	m on roaming II-NNI, else n/a
9	Authorization	[5]	m on roaming II-NNI, else n/a
9a	Answer-Mode	[5]	o
10	Call-ID	[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]	o
21	Expires	[5]	m
21a	Flow-Timer	[5]	m on roaming II-NNI, else o
22	Event	[5]	m
23	From	[5]	m
24	Geolocation	[5]	m
24a	Geolocation-Error	[5]	m
24b	Geolocation-Routing	[5]	m
25	History-Info	subclause 6.1.1.3.1 (Table 6.2, item 4)	o
25a	Info-Package	[5]	o
26	In-Reply-To	[5]	o
27	Join	[5]	o
27a	Max-Breadth	[5]	m
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	subclause 6.1.1.3.1 (Table 6.2, item 2)	m in case of a trust relationship between the interconnected networks, else n/a
33a	P-Answer-state	[5]	o
34	P-Asserted-Identity	subclause 6.1.1.3.1 (Table 6.2, item 1)	m in case of a trust relationship between the interconnected networks, else n/a
35	P-Asserted-Service	subclause 6.1.1.3.1 (Table 6.2, item 5)	o
35a	P-Associated-URI	[5]	m on roaming II-NNI, else n/a
36	P-Called-Party-ID	[5]	m on roaming II-NNI, else n/a
37	P-Charging-Function-Addresses	subclause 6.1.1.3.1 (Table 6.2, item 7)	n/a
38	P-Charging-Vector	subclause	m on roaming II-NNI, else o

Item	Header field	Ref.	II-NNI
		6.1.1.3.1 (Table 6.2, item 6)	
39	P-Early-Media	subclause 6.1.1.3.1 (Table 6.2, item 12)	m in case of a trust relationship between the interconnected networks, else n/a
40	P-Media-Authorization	[5]	n/a
41	P-Preferred-Identity	[5]	n/a
42	P-Preferred-Service	[5]	m on roaming II-NNI, else n/a
43	P-Private-Network-Indication	subclause 6.1.1.3.1 (Table 6.2, item 9)	m on roaming II-NNI, else o
44	P-Profile-Key	subclause 6.1.1.3.1 (Table 6.2, item 8)	o on roaming II-NNI, else n/a
45	P-Served-User	subclause 6.1.1.3.1 (Table 6.2, item 10)	m on roaming II-NNI, else n/a
46	P-User-Database	[5]	n/a
47	P-Visited-Network-ID	[5]	m on roaming II-NNI, else n/a
47a	Path	[5]	m on roaming II-NNI, else n/a
47b	Permission-Missing	[5]	
48	Priority	[5]	o
48a	Priv-Answer-Mode	[5]	o
49	Privacy	[5]	m
50	Proxy-Authenticate	[5]	m on roaming II-NNI, else n/a
51	Proxy-Authorization	[5]	m on roaming II-NNI , else n/a
52	Proxy-Require	[5]	m
52a	RAck	[5]	m
53	Reason	[5] and subclause 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
54a	Recv-Info	[5]	o
55	Referred-By	[5]	m
55a	Refer-Sub	[5]	m in the case the REFER request is supported, else n/a
55b	Refer-To	[5]	m in the case the REFER request is supported, else n/a
56	Reject-Contact	[5]	m
57	Replaces	[5]	o
58	Reply-To	[5]	o
59	Request-Disposition	[5]	m
60	Require	[5]	m
61	Resource-Priority	subclause 6.1.1.3.1 (Table 6.2, item 3)	o
61a	Retry-After	[5]	m
62	Route	[5]	m
62a	RSeq	[5]	m
63	Security-Client	[5]	n/a
63a	Security-Server	[5]	n/a
64	Security-Verify	[5]	n/a
65	Server	[5]	o
65a	Service-Route	[5]	m on roaming II-NNI, else n/a
65b	Session-ID	[5]	o
66	Session-Expires	[5]	m
66a	SIP-ETag	[5]	m in the case the PUBLISH request is supported, else n/a
66b	SIP-If-Match	[5]	m in the case the PUBLISH request is supported, else n/a

Item	Header field	Ref.	II-NNI
67	Subject	[5]	o
67a	Subscription-State	[5]	m in the case the NOTIFY request is supported, else n/a
68	Supported	[5]	m
68a	Target-Dialog	[5]	o
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]	o
74	Via	[5]	m
75	Warning	[5]	o
76	WWW-Authenticate	[5]	m on roaming II-NNI, 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 TS 24.229 [5] is applied. If specified in TS 24.229, the IBCF modifies the SIP header field.
o	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

Change history							
Date	TSG #	TSG Doc.	CR	Rev	Subject/Comment	Old	New
09/2009	TSG#45	CP-090584	011	4	Major capabilities on II-NNI	8.3.0	9.0.0
09/2009	TSG#45	CP-090584	013	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	024	3	Aligning references to P-Asserted-Service	9.0.0	9.1.0
12/2009	TSG#46	CP-090844	026		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	032	1	Customized Ringing Signal (CRS) modification	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	033	2	Completing the Personal Network Management (PNM) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	034	1	Aligning existing supplementary services	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	035	1	Completing the Flexible Alerting (FA) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	036	1	Completing the Closed User Group (CUG) supplementary service	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	037	3	Completing the Three-Party (3PTY) and Conference (CONF) supplementary services	9.0.0	9.1.0
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	044	3	Completing Communication DIVersion (CDIV) supplementary services	9.0.0	9.1.0
12/2009	TSG#46	CP-090854	046	2	Deletion of an editors Note on OIR service	9.0.0	9.1.0
12/2009	TSG#46	CP-090844	047	3	Annex A header updated	9.0.0	9.1.0
03/2010	TSG#47	CP-100077	051	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	TSG#47	CP-100087	062		Correcting minor errors	9.1.0	9.2.0
03/2010	TSG#47	CP-100087	063	4	Updating Major Capability according to latest 24.229	9.1.0	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	References numbers update (R9 29.165)	9.1.0	9.2.0
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-uuu 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
12/2010	TSG#50	CP-100781	117	2	Aligning annex A with Major Capabilities Table	9.4.0	9.5.0
12/2010	TSG#50	CP-100775	120		Release version added to ETSI specification	9.4.0	9.5.0
12/2010	TSG#50	CP-100775	125		Correcting errors in SIP method subclause	9.4.0	9.5.0
12/2010	TSG#50	CP-100781	127		Duplications of rows in Major capability table	9.4.0	9.5.0
12/2010	TSG#50	CP-100775	132	1	IBCF and URI formats	9.4.0	9.5.0
12/2010	TSG#50	CP-100775	135	1	Roaming and non-roaming definition added	9.4.0	9.5.0
12/2010	TSG#50	CP-100775	139	2	Refer-Sub and Refer-To missing in annex A	9.4.0	9.5.0
12/2010	TSG#50	CP-100781	141	2	Supplementary services corrections	9.4.0	9.5.0
12/2010	TSG#50	CP-100781	153		MCID and ACR	9.4.0	9.5.0
03/2011	TSG#51	CP-110113	155		Correction on Major Capabilities over II-NNI	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	157	1	Addition of the Target-Dialog header field to the table of major capabilities and Annex A	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	159	2	Correcting major capabilities related to trust domain	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	163	1	Reference clause updates	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	167	2	Updates of the Major capability table	9.5.0	9.6.0
03/2011	TSG#51	CP-110107	171	1	Correcting header table in annex A	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	173	1	Reference update: RFC 6086	9.5.0	9.6.0
03/2011	TSG#51	CP-110107	177		Adding MMTel feature tag in clause 12	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	193	1	Alignment of INFO support for DTMF transport with TS 24.182	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	198	1	Correcting the management of SIP header field relating to trust relationship	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	211	1	Updating references in TS29.165	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	216	1	Removal av draft-patel-dispatch-cpc-oli-parameter from references	9.5.0	9.6.0
03/2011	TSG#51	CP-110113	218	1	Removal of draft-patel-ecrit-sos-parameter	9.5.0	9.6.0
03/2011	TSG#51				Editorial changes made by MCC	9.6.0	9.6.1
06/2011	TSG#52	CP-110403	237	3	Reference section updates (II-NNI)	9.6.1	9.7.0
06/2011	TSG#52	CP-110403	239	2	Invalid reference to annex J of 3GPP TS 24.229	9.6.1	9.7.0
06/2011	TSG#52	CP-110403	262	1	Allow-Event header field (II-NNI)	9.6.1	9.7.0
06/2011	TSG#52	CP-110403	265		Max-Breadth header field correction	9.6.1	9.7.0
06/2011	TSG#52	CP-110401	268	1	SIP headers alignment in II-NNI	9.6.1	9.7.0
06/2011	TSG#52	CP-110406	270	1	Removal of dial around indicator	9.6.1	9.7.0
06/2011	TSG#52	CP-110403	275	1	Forking definition on II-NNI	9.6.1	9.7.0
09/2011	TSG#53	CP-110612	285		Date header field missing in CCBS and CCNR	9.7.0	9.8.0
09/2011	TSG#53	CP-110612	288		Support of replaces missing in ECT	9.7.0	9.8.0
09/2011	TSG#53	CP-110612	290	1	Error-Info missing in announcements	9.7.0	9.8.0
09/2011	TSG#53	CP-110612	296		Reference section update (II-NNI)	9.7.0	9.8.0
12/2011	TSG#54	CP-110824	307		Aligning the references with Supported transport-level RFCs	9.8.0	9.9.0
12/2011	TSG#54	CP-110829	314	1	Reference updates	9.8.0	9.9.0
12/2011	TSG#54	CP-110829	319		Reference to wrong RFC for INFO method	9.8.0	9.9.0
12/2011	TSG#54	CP-110829	332		Reference update: Reason header in SIP responses	9.8.0	9.9.0
12/2011	TSG#54	CP-110829	344	2	RFC4694 passing of rn parameter	9.8.0	9.9.0
03/2012	TSG#55	CP-120204	341	5	Updating of UUS references	9.9.0	9.10.0
03/2012	TSG#55	CP-120060	351	1	Clarification on URI formats in SIP message	9.9.0	9.10.0
03/2012	TSG#55	CP-120060	368	1	IETF draft update - Location Conveyance	9.9.0	9.10.0
03/2012	TSG#55	CP-120060	371		Removing draft-rosenberg-sipcore-target-uri-delivery	9.9.0	9.10.0
03/2012	TSG#55	CP-120060	374		from-change in TIP-TIR missing	9.9.0	9.10.0
03/2012	TSG#55	CP-120060	388	2	CDIV limitation 29.165r9	9.9.0	9.10.0
06/2012	TSG#56	CP-120340	391	1	Reference update: draft-salud-alert-info-urns	9.10.0	9.11.0
06/2012	TSG#56	CP-120340	406	1	Usage of isub parameter over II-NNI	9.10.0	9.11.0
06/2012	TSG#56	CP-120340	417		Updating of UUS references	9.10.0	9.11.0
06/2012	TSG#56	CP-120332	422	2	The description of notation code for SIP message	9.10.0	9.11.0
09/2012	TSG#57	CP-120518	428		Correction on the description of "g.3gpp.crs"	9.11.0	9.12.0
09/2012	TSG#57	CP-120518	431		Correction on the description of "application/vnd.etsi.mcid+xml" MIME body	9.11.0	9.12.0
09/2012	TSG#57	CP-120518	438	1	Correcting reference	9.11.0	9.12.0
12/2012	TSG#58	CP-120825	476	1	Removal of draft-dawes-dispatch-mediasec-parameter draft	9.12.0	9.13.0

12/2012	TSG#58	CP-120825	480	1	Trust domain error corrections	9.12.0	9.13.0
03/2013	TSG#59	CP-130061	506	1	IMS debug removal of feature	9.13.0	9.14.0
03/2013	TSG#59	CP-130069	525	2	Correcting announcement when communication is rejected	9.13.0	9.14.0
03/2013	TSG#59	CP-130069	533	1	Clarification for CDIV NNI description	9.13.0	9.14.0
09/2013	TSG#61	CP-130544	563		Reference update of draft-vanelburg-dispatch-private-network-ind	9.14.0	9.15.0
09/2013	TSG#61	CP-130544	585	2	draft-avasara-dispatch-comm-div-notification ref update	9.14.0	9.15.0
12/2013	TSG#62	CP-130666	593		Reference update of draft-vanelburg-dispatch-private-network-ind	9.15.0	9.16.0
12/2013	TSG#62	CP-130666	597		Reference update: draft-kaplan-insipid-session-id	9.15.0	9.16.0
12/2013	TSG#62	CP-130666	610	1	Reference Update: draft-ietf-salud-alert-info-urns	9.15.0	9.16.0
03/2014	TSG#63	CP-140064	619	1	Reference update of draft-vanelburg-dispatch-private-network-ind	9.16.0	9.17.0
06/2014	TSG#64	CP-140358	629	-	Correcting the indication of communication diversion to the diverting user	9.17.0	9.18.0
06/2014	TSG#64	CP-140359	649	-	Removal of CDIVN	9.17.0	9.18.0
06/2014	TSG#64	CP-140360	643	1	Reference update of draft-vanelburg-dispatch-private-network-ind	9.17.0	9.18.0
06/2014	TSG#64	CP-140361	660	1	Reference update for "mp" header field parameter in History-Info header field	9.17.0	9.18.0
06/2014	TSG#64	CP-140361	664	-	Correcting the status of a major capability over II-NNI	9.17.0	9.18.0
12/2014	TSG#66	CP-140891	679	1	Reference update: RFC 7329 (draft-kaplan-insipid-session-id)	9.18.0	9.19.0
12/2014	TSG#66	CP-140891	691	1	Reference update from draft-vanelburg-dispatch-private-network-ind to RFC 7316	9.18.0	9.19.0
03/2015	TSG#67	CP-150098	709	2	Clarification on the II-NNI requirement for CDIV regarding Privacy header field	9.19.0	9.20.0
03/2015	TSG#67	CP-150098	714	1	Clarification on the II-NNI requirement regarding cause-param URI parameter	9.19.0	9.20.0
03/2015	TSG#67	CP-150099	733	1	Reference update for UUSIW	9.19.0	9.20.0
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