

SIP Cheat Sheet

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	Requests (Methods)
ACK	INVITE -> Final response -> ACK
BYE	Terminate a session
CANCEL	Terminate during call setup
INFO	Carry application information within SIP signaling path
INVITE	Sent to set up a call
MESSAGE	Used to carry txt (SMS / IM)
NOTIFY	Disclose state of presence service
OPTIONS	Analogous to a SIP ping
PRACK	Makes 1xx messages reliable
PUBLISH	Reports state of a service
REFER	Used for transfers & features
REGISTER	Reports UA's location (IP)
SUBSCRIBE	Request to receive presence event
UPDATE	Used in early dialog to modify SDP

Responses		
1xx	(Provisional) Setup / More to come	
2xx	(Final) Success	
3 x x	(Final) Call Forwarding	
4xx	(Final) Request Failure	
5xx	(Final) Server Failure	
6хх	(Final) Global Failure	

	RTP Mapping
0	PCMU / 8000
1-2	Reserved
3	GSM / 8000
4	G723 / 8000
5	DVI4 / 8000
6	DVI4 / 16000
7	LPC / 8000
8	PCMA / 8000
9	G722 / 8000
10	L16 / 44100 (2chan)
11	L16 / 44100
12	QCELP / 8000
13	CN / 8000
14	MPA / 90000
15	G728 / 8000
16	DVI4 / 11025
17	DVI4 / 22050
18	G729 / 8000
19-24	Res or unassigned
25	CelB / 90000
26	JPEG / 90000
27	Unassigned

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RTP Mapping (continued)		
28	nv / 90000	
29 - 30	Unassigned	
31	H261 / 90000	
32	MPV / 90000	
33	MP2T / 90000	
34	H263 / 90000	
35 - 71	Unassigned	
72-76	Reserved for RTCP conflict avoidance	
77 - 95	Unassigned	
96 - 127	Dynamic	

Common Terms
Session Initiation Protocol
User Agent (SIP phone is a UA)
User Agent Client (sends requests)
User Agent Server (sends responses)
Session Border Controller (edge security)
Routes (relays) SIP requests and responses
Accepts SIP UA registration (tracking UA location)
Back-To-Back-UA (separates SIP dialogs)
Real-Time Transport Protocol (voice and / or video)
Session Description Protocol (describes RTP setup)

SIP Call Example
INVITE + SDP 100 < 180 / 183 < 200 + SDP ACK>
<===========>
< BYE 200>



