📞 SIP Phone A | 📡 SIP Proxy | 📞 SIP Phone B

📁 **sip_basic_do_183.pcapng**

📡 SIP INVITE sip:123456789@10.180.110.58:5060 SIP/2.0

Frame 1 | 2019-05-02T20:19:24.216984Z

📡 Via	SIP/2.0/TCP 10.180.110.48:5060;branch=z9hG4bK15de53673925
🔗 Call-ID	91d09d00-10001-1277-93f492a@10.180.110.48
🔢 CSeq	101 INVITE
📞 Contact	<sip:1234@10.180.110.48:5060;transport=tcp>;+u.sip!devicename.ccm.cisco.com="SEP2C31246A214B"

⏳ SIP 100 Trying

Frame 2 | 2019-05-02T20:19:24.219974Z

🎵 SIP 183 Session Progress

💡 183 Session Progress — early media with SDP answer; RTP flows before call is answered

🌐 Connection	IN IP4 10.180.110.58
🎵 Media	audio 8350 RTP/AVP 18 0 8 9 4 96 15 3 19
🌐 Connection	IN IP4 10.180.110.58

✅ SIP 200 OK

💡 200 OK — call answered or registration accepted; SDP answer completes codec negotiation; ACK required to complete INVITE 3-way handshake

📞 Contact	<sip:123456789@10.180.110.58:5060;transport=tcp>
🌐 Connection	IN IP4 10.180.110.58
🎵 Media	audio 8350 RTP/AVP 18 0 8 9 4 96 15 3 19
🌐 Connection	IN IP4 10.180.110.58

📡 SIP ACK sip:123456789@10.180.110.58:5060;transport=tcp SIP/2.0

Frame 7 | 2019-05-02T20:19:27.838945Z

📡 Via	SIP/2.0/TCP 10.180.110.48:5060;branch=z9hG4bK15df350cfbcb
🔗 Call-ID	91d09d00-10001-1277-93f492a@10.180.110.48
🔢 CSeq	101 ACK
👤 Owner	CiscoSystemsCCM-SIP 142076 1 IN IP4 10.180.110.48
🌐 Connection	IN IP4 10.10.214.57
🎵 Media	audio 23874 RTP/AVP 0 101
📋 Attr	rtpmap:0 PCMU/8000
📋 Attr	rtpmap:101 telephone-event/8000
📋 Attr	fmtp:101 0-15

🎵 RTP

Frame 8 | 2019-05-02T20:19:27.857942Z

Setup frame	4
Marker	False
🎵 Codec	ITU-T G.729 (18)
Sequence number	5961
Timestamp	401708646
🔢 SSRC	0x00007ad2 (31442)

🎵 RTP

Frame 14 | 2019-05-02T20:19:27.959941Z

Setup frame	4
Marker	True
🎵 Codec	ITU-T G.711 PCMU (0)
Sequence number	61902
Timestamp	4260423999
🔢 SSRC	0xebcdda73 (3956136563)