



```
Negotiated Codec : g711ulaw, bytes :160
Negotiated DTMF relay : rtp-nte
Negotiated NTE payload : 101 (tx), 101 (rx)
Negotiated CN payload : 0
Media Src Addr/Port : [10.10.10.1]:16666
Media Dest Addr/Port : [10.10.10.2]:21046

*Nov 11 19:29:14.810: //556/622600508211/SIP/Media/sipSIPSetStreamInfo: 0 Active Streams
*Nov 11 19:29:14.810: //556/622600508211/SIP/Media/sipSIPSetStreamInfo: Number of active streams is zero (0)!
*Nov 11 19:29:14.810: //556/622600508211/SIP/Media/sipSIPSetStreamInfo:
caps_stream_count=0,caps_stream[0].stream_type=0x0FFF,caps_stream_list.xmitFunc=
*Nov 11 19:29:14.810: //556/622600508211/SIP/Media/sipSIPSetStreamInfo: ?unknown??, caps_stream_list.context=
*Nov 11 19:29:14.810: //556/622600508211/SIP/Media/sipSIPSetStreamInfo: 0x0 (gccb)
*Nov 11 19:29:14.810: //558/622600508211/SIP/Media/sipSIPUpdCallWithSdpInfo:
Stream type : voice+dtmf
Media line : 1
State : STREAM_ADDING (2)
Stream address type : 1
Callid : 558
Negotiated Codec : g711ulaw, bytes :160
Negotiated DTMF relay : rtp-nte
Negotiated NTE payload : 101 (tx), 101 (rx)
Negotiated CN payload : 0
Media Src Addr/Port : [10.10.10.1]:16666
Media Dest Addr/Port : [10.10.10.2]:21046

*Nov 11 19:29:14.814: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8B7786D0
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: sipSIPUpdateRtpSession for m-line 1
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: rtp_session info
laddr = 10.10.10.1, lport = 16666, raddr = 10.10.10.2, rport=17056, do_rtcp=TRUE
src_callid = 556, dest_callid = 558, stream type = voice-only, stream direction = SENDRECV
media_ip_addr = 10.10.10.12, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:29:14.818: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: No rtp session, creating a new one
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPCreateRtpSession: stun is disabled
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: Voice quality monitoring is not enabled for this RTP session due to sdp passthru enabled
*Nov 11 19:29:14.818: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: VQM: gccb=0x0, gccb->callid=0, cc->ccCallID=556
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPGetNewLocalMediaDirection:
New Remote Media Direction = SENDRECV
Present Local Media Direction = SENDRECV
New Local Media Direction = SENDRECV
retVal = 0

*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: sipSIPUpdateRtpSession for m-line 1
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: rtp_session info
laddr = 10.10.10.1, lport = 16666, raddr = 10.10.10.2, rport=17046, do_rtcp=TRUE
src_callid = 558, dest_callid = 556, stream type = voice+dtmf, stream direction = SENDRECV
media_ip_addr = 10.10.10.2, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: RTP session already created - update
*Nov 11 19:29:14.822: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8B7786D0
*Nov 11 19:29:14.822: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: VQM: gccb=0x0, gccb->callid=0, cc->ccCallID=558
*Nov 11 19:29:14.822: //558/622600508211/SIP/Media/sipSIPGetNewLocalMediaDirection:
New Remote Media Direction = SENDRECV
Present Local Media Direction = SENDRECV
New Local Media Direction = SENDRECV
retVal = 0

*Nov 11 19:29:14.850: //558/622600508211/SIP/Media/act_update_nat_media: NEW: rtp session to be updated with the new src info, old addr:port [10.10.10.2]:21046, newaddr:port [10.10.10.2]:21046
*Nov 11 19:29:14.850: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8B7786D0
*Nov 11 19:29:14.850: //558/622600508211/SIP/Media/act_update_nat_media: NEW: rtp session to be updated with the new src info, old addr:port [10.10.10.2]:17166, newaddr:port [10.10.10.2]:17166
*Nov 11 19:29:14.902: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8D6F25F8
*Nov 11 19:29:34.322: //556/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8D6F25F8
*Nov 11 19:29:34.322: //558/622600508211/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8B7786D0
*Nov 11 19:29:34.338: //558/622600508211/SIP/Media/sipSIPHandleDestroyRtpSession: stream:8D6F25F8
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPCopyStunConfigFromPeerToCCB: Firewall traversal is not enabled
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPReserveRtpPort: reserved port 16670 for stream 1
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPAddSdpMediaPayload: Preferred method of dtmf relay is: 6, with payload: 101
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPProcessRtpSessions: Processing stream state = STREAM_IDLE
*Nov 11 19:31:36.042: //565/861001948218/SIP/Media/sipSIPProcessRtpSessions: No active streams.
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPAddStream: Adding stream 1 of type voice+dtmf (callid 565) to the VOIP RTP library
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: sipSIPUpdateRtpSession for m-line 1
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: rtp_session info
laddr = 10.10.10.1, lport = 16670, raddr = 0.0.0.0, rport=0, do_rtcp=FALSE
src_callid = 565, dest_callid = -1, stream type = voice+dtmf, stream direction = RCVONLY
media_ip_addr = -, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: No rtp session, creating a new one
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: stun is disabled
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: Voice quality monitoring is not enabled for this RTP session due to sdp passthru enabled
*Nov 11 19:31:36.046: //565/861001948218/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:31:36.170: //565/861001948218/SIP/Media/sipSIPReplaceSdp: Main stream got changed & it's Flow Around
*Nov 11 19:31:36.170: //565/861001948218/SIP/Media/sipSIPUpdCallWithSdpInfo:
Preferred Codec : g711alaw, bytes :160
Preferred DTMF relay : rtp-nte
Preferred NTE payload : 101
Early Media : No
Delayed Media : No
Bridge Done : No
New Media : No
DSP DNLD Req'd : No

*Nov 11 19:31:36.174: //565/861001948218/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:31:36.174: //565/861001948218/SIP/Media/sipSIPUpdCallWithSdpInfo:
Stream type : voice+dtmf
Media line : 1
State : STREAM_ADDING (2)
Stream address type : 1
Callid : 565
Negotiated Codec : g711alaw, bytes :160
Negotiated DTMF relay : rtp-nte
Negotiated NTE payload : 101
Negotiated CN payload : 0
Media Src Addr/Port : [10.10.10.1]:16670
Media Dest Addr/Port : [10.10.10.2]:21114

*Nov 11 19:31:36.174: //565/861001948218/SIP/Media/sipSIPProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPAddStream: Adding stream 1 of type voice+dtmf (callid 565) to the VOIP RTP library
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: sipSIPUpdateRtpSession for m-line 1
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: rtp_session info
laddr = 10.10.10.1, lport = 16670, raddr = 10.10.10.2, rport=21114, do_rtcp=TRUE
src_callid = 565, dest_callid = 564, stream type = voice+dtmf, stream direction = SENDRECV
media_ip_addr = 10.10.10.2, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: RTP session already created - update
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8D6C478C
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: VQM: gccb=0x0, gccb->callid=0, cc->ccCallID=565
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPGetNewLocalMediaDirection:
New Remote Media Direction = SENDRECV
Present Local Media Direction = SENDRECV
New Local Media Direction = SENDRECV
retVal = 0

*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetStreamInfo: 1 Active Streams
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetStreamInfo: Adding stream type (voice+dtmf) from media
Line 1 codec g711alaw
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetStreamInfo:
caps_stream_count=1,caps_stream[0].stream_type=0x3,caps_stream_list.xmitFunc=
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetStreamInfo: void rtp_xmit,caps_stream_list.context=
*Nov 11 19:31:36.178: //565/861001948218/SIP/Media/sipSIPSetStreamInfo: 0x8D6F1880 (gccb)
*Nov 11 19:31:36.218: //565/861001948218/SIP/Media/act_update_nat_media: NEW: rtp session to be updated with the new src info, old addr:port [10.10.10.2]:21114, newaddr:port [10.10.10.2]:21114
*Nov 11 19:31:36.218: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8D6C478C
*Nov 11 19:31:43.474: //565/861001948218/SIP/Media/sipSIPUpdateRtpSession: stun is disabled for stream:8D6C478C
*Nov 11 19:31:43.490: //565/861001948218/SIP/Media/sipSIPHandleDestroyRtpSession: stream:8D6C478C
*Nov 11 19:45:41.206: //589/8E393E8227/SIP/Media/sipSIPCopyStunConfigFromPeerToCCB: Firewall traversal is not enabled,imageatr parse payload numtok not found,imageatr payload found, specific is 255
*Nov 11 19:45:41.210: //589/8E393E8227/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:45:41.210: //589/8E393E8227/SIP/Media/sipSIPSetMediaSrcAddr: Media src addr for stream 2 = 10.10.10.1
*Nov 11 19:45:41.210: //589/8E393E8227/SIP/Media/sipSIPDoMediaNegotiation: D0_Media_Negotiation: video m-line detected.

*Nov 11 19:45:41.210: //589/8E393E8227/SIP/Media/sipSIPUpdCallWithSdpInfo:
Preferred Codec : g711alaw, bytes :160
Preferred DTMF relay : inband-voice
Preferred NTE payload : 101
Early Media : No
Delayed Media : No
Bridge Done : No
New Media : No
DSP DNLD Req'd : No
```



```
State : STREAM_ADDING (2)
Stream address type : 1
Callid : 591
Negotiated Codec : g711ulaw, bytes :160
Negot. Codec payload : 0 (tx), 0 (rx)
Negotiated DTMF relay : rtp-nte
Negotiated NTE payload : 101 (tx), 101 (rx)
Negotiated CN payload : 0
Media Srce Addr/Port : [10.10.10.1]:16676
Media Dest Addr/Port : [10.10.10.2]:20670

*Nov 11 19:45:41.354: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtpSession: stun is disabled for stream:8D6E4A28
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIAddStream: Adding stream 1 of type voice-only (callid 589) to the VOIP RTP library
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPISetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: sipSPIUpdateRtcpSession for m-line 1
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: rtcp_session info
laddr = 10.10.10.1, lport = 16672, raddr = 10.10.10.12, rport=20200, do_rtcp=TRUE
src_callid = 589, dest_callid = 591, stream type = voice-only, stream direction = SENDRECV
media_ip_addr = 10.10.10.12, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: No rtp session, creating a new one
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPICreateRtpSession: stun is disabled
*Nov 11 19:45:41.362: //-1/xxxxxxxxxxxx/SIP/Media/sipSPIUpdateRtcpSession: Voice quality monitoring is not enabled for this RTP session due to sdp passthru enabled
*Nov 11 19:45:41.362: //-1/xxxxxxxxxxxx/SIP/Media/sipSPIUpdateRtcpSession: VQM: gccb=0x0, gccb->callid=0, ccb->ccCallID=589
*Nov 11 19:45:41.362: //589/AE3093EE8227/SIP/Media/sipSPIGetNewLocalMediaDirection:
New Remote Media Direction = SENDRECV
Present Local Media Direction = SENDRECV
New Local Media Direction = SENDRECV
retVal = 0

*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIProcessRtpSessions: Processing stream state = STREAM_ADDING
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIAddStream: Adding stream 1 of type voice+dtmf (callid 591) to the VOIP RTP library
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPISetMediaSrcAddr: Media src addr for stream 1 = 10.10.10.1
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: sipSPIUpdateRtcpSession for m-line 1
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: rtcp_session info
laddr = 10.10.10.1, lport = 16676, raddr = 10.10.10.2, rport=20670, do_rtcp=TRUE
src_callid = 591, dest_callid = 589, stream type = voice+dtmf, stream direction = SENDRECV
media_ip_addr = 10.10.10.2, vrf tableid = 0 media_addr_type = 1 negotiated_bandwidth (kbps) = 0
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: RTP session already created - update
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: stun is disabled for stream:8D6E4A28
*Nov 11 19:45:41.362: //-1/xxxxxxxxxxxx/SIP/Media/sipSPIUpdateRtcpSession: VQM: gccb=0x0, gccb->callid=0, ccb->ccCallID=591
*Nov 11 19:45:41.362: //591/AE3093EE8227/SIP/Media/sipSPIGetNewLocalMediaDirection:
New Remote Media Direction = SENDRECV
Present Local Media Direction = SENDRECV
New Local Media Direction = SENDRECV
retVal = 0

*Nov 11 19:45:41.386: //591/AE3093EE8227/SIP/Media/act_update_nat_media: NEW: rtp session to be updated with the new src info, old addr:port [10.10.10.2]:20670 , newaddr:port [10.10.10.2]:20670
*Nov 11 19:45:41.386: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtpSession: stun is disabled for stream:8D6E4A28
*Nov 11 19:45:41.386: //589/AE3093EE8227/SIP/Media/act_update_nat_media: NEW: rtp session to be updated with the new src info, old addr:port [10.10.10.12]:20200 , newaddr:port [10.10.10.12]:20200
*Nov 11 19:45:41.438: //589/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: stun is disabled for stream:8EE45620
*Nov 11 19:45:47.322: //589/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: stun is disabled for stream:8EE45620
*Nov 11 19:45:47.322: //591/AE3093EE8227/SIP/Media/sipSPIUpdateRtcpSession: stun is disabled for stream:8D6E4A28
*Nov 11 19:45:47.326: //589/AE3093EE8227/SIP/Media/sipSPIHandleDestroyRtpSession: stream:8EE45620
*Nov 11 19:45:47.338: //591/AE3093EE8227/SIP/Media/sipSPIHandleDestroyRtpSession: stream:8D6E4A28
```