# Configure Jabber to Use Custom Audio and Video Port Range on CUCM 11.5.1

## Contents

Introduction Prerequisites Requirements Components Used Configure Verify Troubleshoot

## Introduction

This document describes the procedure to configure Cisco Jabber to use custom audio and video port range on Cisco Unified Communications Manager (CUCM) 11.5.1.

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# Prerequisites

#### Requirements

Cisco recommends that you have knowledge of:

- Cisco Jabber
- Cisco Unified Communication Manager (CUCM)

#### **Components Used**

The information in this document is based on these software versions:

- Cisco Jabber for Windows 11.9.x
- Cisco Unified Communications Manager 11.5.x

The information in this document was created from the devices in a specific lab environment. All of the devices used in this document started with a cleared (default) configuration. If your network is live, ensure you understand the potential impact of any command.

# Configure

You can create a new SIP Profile or simply copy the Standard SIP Profile and modify the port range on your copied profile as follows:

- Navigate to Device > Device Settings > SIP Profile on the Cisco Unified CM Administration page
- Click on Find button to display all SIP profiles
- Click on the Standard SIP Profile and click on the Copy button
- Scroll down to the Media Port Ranges field and click on the radio button titled **Separate Port Ranges for Audio and Video**
- Specify the Start Audio Port, the Stop Audio Port, the Start Video Port and the Stop Video Port

Parameters used in Phone	
Timer Invite Expires (seconds)*	180
Timer Register Delta (seconds)*	5
Timer Register Expires (seconds)*	3600
Timer T1 (msec)*	500
Timer T2 (msec)*	4000
Retry INVITE*	6
Retry Non-INVITE*	10
Media Port Ranges	O Common Port Range for Audio and Video
	Separate Port Ranges for Audio and Video
Start Audio Port*	16384
Stop Audio Port*	16393
Start Video Port *	16394
Stop Video Port*	16403

Note: The Start port needs to be an even number and the Stop port needs to be odd number

## Verify

Place a call from Jabber and confirm in the Jabber log that the ports specified in the outbound INVITE Session Description Protocol (SDP) message falls within the configured port range.

```
CSeq: 101 INVITE
User-Agent: Cisco-CSF
Contact: <sip:1553c615-14d6-39e8-0399-
af6fdd5ee447@10.66.87.207:50268;transport=tcp>;+u.sip!devicename.ccm.cisco.com="CSFwstest1";vide
o;bfcp
Expires: 180
Accept: application/sdp
Allow: ACK, BYE, CANCEL, INVITE, NOTIFY, OPTIONS, REFER, REGISTER, UPDATE, SUBSCRIBE, INFO
Remote-Party-ID: "1000" <sip:1000@dmaccorm-ucmpub.myothertestdomain.net>;party=calling;id-
type=subscriber;privacy=off;screen=yes
Supported: replaces, join, sdp-anat, norefersub, resource-priority, extended-refer, X-cisco-
callinfo,X-cisco-serviceuri,X-cisco-escapecodes,X-cisco-service-control,X-cisco-srtp-fallback,X-
cisco-monrec, X-cisco-config, X-cisco-sis-7.0.0, X-cisco-xsi-8.5.1
Allow-Events: kpml, dialog
Recv-Info: conference
Recv-Info: x-cisco-conference
Content-Length: 2246
Content-Type: application/sdp
Content-Disposition: session; handling=optional
```

v=0o=Cisco-SIPUA 20391 0 IN IP4 10.66.87.207 s=SIP Call b=AS:4000 t=0 0 a=cisco-mari:v1 a=cisco-mari-rate m=audio 16390 RTP/AVP 114 9 104 105 0 8 18 111 101 C=TN TP4 10.66.87.207 a=rtpmap:114 opus/48000/2 a=rtpmap:9 G722/8000 a=rtpmap:104 G7221/16000 a=fmtp:104 bitrate=32000 a=rtpmap:105 G7221/16000 a=fmtp:105 bitrate=24000 a=rtpmap:0 PCMU/8000 a=rtpmap:8 PCMA/8000 a=rtpmap:18 G729/8000 a=fmtp:18 annexb=no a=rtpmap:111 x-ulpfecuc/8000 a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us a=fmtp:111 max\_esel=1420;m=8;max\_n=32;FEC\_ORDER=FEC\_SRTP a=rtpmap:101 telephone-event/8000 a=fmtp:101 0-15 a=sendrecv m=video 16398 RTP/AVP 126 97 111 c=IN IP4 10.66.87.207 b=TIAS:4000000 a=rtpmap:126 H264/90000 a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;maxfs=3601;max-rcmd-nalu-size=32000 a=imageattr:126 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,q=1.00] a=content:main a=label:11 a=rtpmap:97 H264/90000 a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601 a=imageattr:97 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,g=1.00] a=rtpmap:111 x-ulpfecuc/8000 a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us a=fmtp:111 max\_esel=1420;m=8;max\_n=32;FEC\_ORDER=FEC\_SRTP a=rtcp-fb:\* ccm fir a=rtcp-fb:\* ccm tmmbr a=rtcp-fb:\* nack pli a=recvonly m=video 16394 RTP/AVP 126 97 111 c=IN IP4 10.66.87.207 b=TIAS:4000000 a=rtpmap:126 H264/90000 a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;maxfs=3601;max-rcmd-nalu-size=32000 a=content:slides a=label:12 a=rtpmap:97 H264/90000 a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601 a=rtpmap:111 x-ulpfecuc/8000 a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us a=fmtp:111 max\_esel=1420;m=8;max\_n=32;FEC\_ORDER=FEC\_SRTP a=rtcp-fb:\* ccm fir a=rtcp-fb:\* ccm tmmbr a=rtcp-fb:\* nack pli a=sendrecv m=application 5904 UDP/BFCP \* c=IN IP4 10.66.87.207 a=floorctrl:c-s

```
a=confid:3
a=floorid:2 mstrm:12
a=userid:3
a=setup:actpass
a=connection:new
a=sendrecv
m=application 39878 RTP/AVP 125
c=IN IP4 10.66.87.207
a=rtpmap:125 H224/4800
a=rtcp:39879
a=sendrecv
```

### Troubleshoot

If the port specified in the INVITE SDP is not within the configured range, review the jabber.log file after a restart of the client to determine that the custom port range has been applied.

```
CSeq: 101 INVITE
User-Agent: Cisco-CSF
Contact: <sip:1553c615-14d6-39e8-0399-
af6fdd5ee447@10.66.87.207:50268;transport=tcp>;+u.sip!devicename.ccm.cisco.com="CSFwstest1";vide
o;bfcp
Expires: 180
Accept: application/sdp
Allow: ACK, BYE, CANCEL, INVITE, NOTIFY, OPTIONS, REFER, REGISTER, UPDATE, SUBSCRIBE, INFO
Remote-Party-ID: "1000" <sip:1000@dmaccorm-ucmpub.myothertestdomain.net>;party=calling;id-
type=subscriber;privacy=off;screen=yes
Supported: replaces, join, sdp-anat, norefersub, resource-priority, extended-refer, X-cisco-
callinfo,X-cisco-serviceuri,X-cisco-escapecodes,X-cisco-service-control,X-cisco-srtp-fallback,X-
cisco-monrec,X-cisco-config,X-cisco-sis-7.0.0,X-cisco-xsi-8.5.1
Allow-Events: kpml,dialog
Recv-Info: conference
Recv-Info: x-cisco-conference
Content-Length: 2246
Content-Type: application/sdp
Content-Disposition: session; handling=optional
v=0
o=Cisco-SIPUA 20391 0 IN IP4 10.66.87.207
s=SIP Call
b=AS:4000
t=0 0
a=cisco-mari:v1
a=cisco-mari-rate
m=audio 16390 RTP/AVP 114 9 104 105 0 8 18 111 101
c=IN IP4 10.66.87.207
a=rtpmap:114 opus/48000/2
a=rtpmap:9 G722/8000
a=rtpmap:104 G7221/16000
a=fmtp:104 bitrate=32000
a=rtpmap:105 G7221/16000
a=fmtp:105 bitrate=24000
a=rtpmap:0 PCMU/8000
a=rtpmap:8 PCMA/8000
a=rtpmap:18 G729/8000
a=fmtp:18 annexb=no
a=rtpmap:111 x-ulpfecuc/8000
a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us
a=fmtp:111 max_esel=1420;m=8;max_n=32;FEC_ORDER=FEC_SRTP
a=rtpmap:101 telephone-event/8000
a=fmtp:101 0-15
```

a=sendrecv m=video 16398 RTP/AVP 126 97 111 c=IN IP4 10.66.87.207 b=TIAS:4000000 a=rtpmap:126 H264/90000 a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;maxfs=3601;max-rcmd-nalu-size=32000 a=imageattr:126 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,g=1.00] a=content:main a=label:11 a=rtpmap:97 H264/90000 a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601 a=imageattr:97 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,q=1.00] a=rtpmap:111 x-ulpfecuc/8000 a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us a=fmtp:111 max\_esel=1420;m=8;max\_n=32;FEC\_ORDER=FEC\_SRTP a=rtcp-fb:\* ccm fir a=rtcp-fb:\* ccm tmmbr a=rtcp-fb:\* nack pli a=recvonly m=video 16394 RTP/AVP 126 97 111 c=IN IP4 10.66.87.207 b=TIAS:4000000 a=rtpmap:126 H264/90000 a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;maxfs=3601;max-rcmd-nalu-size=32000 a=content:slides a=label:12 a=rtpmap:97 H264/90000 a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601 a=rtpmap:111 x-ulpfecuc/8000 a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us a=fmtp:111 max\_esel=1420;m=8;max\_n=32;FEC\_ORDER=FEC\_SRTP a=rtcp-fb:\* ccm fir a=rtcp-fb:\* ccm tmmbr a=rtcp-fb:\* nack pli a=sendrecv m=application 5904 UDP/BFCP \* c=IN IP4 10.66.87.207 a=floorctrl:c-s a=confid:3 a=floorid:2 mstrm:12 a=userid:3 a=setup:actpass a=connection:new a=sendrecv m=application 39878 RTP/AVP 125 c=IN IP4 10.66.87.207 a=rtpmap:125 H224/4800 a=rtcp:39879 a=sendrecv If the custom port range is not applied as per above log snippet, review the Jabber device configuration XML file on the TFTP server with a web browser

(http://**<TFTP\_SERVER\_ADDRESS>**:6970/**<DEVICE\_NAME>**.cnf.xml).

CSeq: 101 INVITE User-Agent: Cisco-CSF Contact: <sip:1553c615-14d6-39e8-0399af6fdd5ee447@10.66.87.207:50268;transport=tcp>;+u.sip!devicename.ccm.cisco.com="CSFwstest1";vide o;bfcp Expires: 180 Accept: application/sdp

```
Allow: ACK, BYE, CANCEL, INVITE, NOTIFY, OPTIONS, REFER, REGISTER, UPDATE, SUBSCRIBE, INFO
Remote-Party-ID: "1000" <sip:1000@dmaccorm-ucmpub.myothertestdomain.net>;party=calling;id-
type=subscriber;privacy=off;screen=yes
Supported: replaces, join, sdp-anat, norefersub, resource-priority, extended-refer, X-cisco-
callinfo,X-cisco-serviceuri,X-cisco-escapecodes,X-cisco-service-control,X-cisco-srtp-fallback,X-
cisco-monrec,X-cisco-config,X-cisco-sis-7.0.0,X-cisco-xsi-8.5.1
Allow-Events: kpml,dialog
Recv-Info: conference
Recv-Info: x-cisco-conference
Content-Length: 2246
Content-Type: application/sdp
Content-Disposition: session; handling=optional
v=0
o=Cisco-SIPUA 20391 0 IN IP4 10.66.87.207
s=SIP Call
b=AS:4000
t = 0 \quad 0
a=cisco-mari;v1
a=cisco-mari-rate
m=audio 16390 RTP/AVP 114 9 104 105 0 8 18 111 101
c=IN IP4 10.66.87.207
a=rtpmap:114 opus/48000/2
a=rtpmap:9 G722/8000
a=rtpmap:104 G7221/16000
a=fmtp:104 bitrate=32000
a=rtpmap:105 G7221/16000
a=fmtp:105 bitrate=24000
a=rtpmap:0 PCMU/8000
a=rtpmap:8 PCMA/8000
a=rtpmap:18 G729/8000
a=fmtp:18 annexb=no
a=rtpmap:111 x-ulpfecuc/8000
a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us
a=fmtp:111 max_esel=1420;m=8;max_n=32;FEC_ORDER=FEC_SRTP
a=rtpmap:101 telephone-event/8000
a=fmtp:101 0-15
a=sendrecv
m=video 16398 RTP/AVP 126 97 111
c=IN IP4 10.66.87.207
b=TTAS:4000000
a=rtpmap:126 H264/90000
a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;max-
fs=3601;max-rcmd-nalu-size=32000
a=imageattr:126 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,q=1.00]
a=content:main
a=label:11
a=rtpmap:97 H264/90000
a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601
a=imageattr:97 recv [x=[32:1:1280],y=[18:1:720],par=1.7778,q=1.00]
a=rtpmap:111 x-ulpfecuc/8000
a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us
a=fmtp:111 max_esel=1420;m=8;max_n=32;FEC_ORDER=FEC_SRTP
a=rtcp-fb:* ccm fir
a=rtcp-fb:* ccm tmmbr
a=rtcp-fb:* nack pli
a=recvonly
m=video 16394 RTP/AVP 126 97 111
c=IN IP4 10.66.87.207
b=TIAS:4000000
a=rtpmap:126 H264/90000
a=fmtp:126 profile-level-id=42E01F;packetization-mode=1;level-asymmetry-allowed=1;max-
fs=3601;max-rcmd-nalu-size=32000
a=content:slides
```

```
a=label:12
a=rtpmap:97 H264/90000
a=fmtp:97 profile-level-id=42E01F;packetization-mode=0;level-asymmetry-allowed=1;max-fs=3601
a=rtpmap:111 x-ulpfecuc/8000
a=extmap:14/sendrecv http://protocols.cisco.com/timestamp#100us
a=fmtp:111 max_esel=1420;m=8;max_n=32;FEC_ORDER=FEC_SRTP
a=rtcp-fb:* ccm fir
a=rtcp-fb:* ccm tmmbr
a=rtcp-fb:* nack pli
a=sendrecv
m=application 5904 UDP/BFCP *
c=IN IP4 10.66.87.207
a=floorctrl:c-s
a=confid:3
a=floorid:2 mstrm:12
a=userid:3
a=setup:actpass
a=connection:new
a=sendrecv
m=application 39878 RTP/AVP 125
c=IN IP4 10.66.87.207
a=rtpmap:125 H224/4800
a=rtcp:39879
a=sendrecv
```