Polycom VVX 500 or VVX 600 - 20 second audio delay to external party

Last Modified on 01/31/2022 1:48 pm EST

Issue:

On a Wave system, when a Polycom VVX 500 or VVX 600 user is connected to a caller on an external SIP trunk, the external user may not hear them for about 20 seconds.

Affected calls have one or more of the following attibutes...

- Call recording System, Queue, or ad-hoc
- Conference
- · Queue call

Cause:

This issue has been observed on the these specific VVX phones when Video calling is enabled, whether or not a camera is present.

If reviewing a packet capture, the video codecs will be noted in the SDP of an INVITE packet as below:

```
> Frame 3716: 1406 bytes on wire (11248 bits), 1406 bytes captured (11248 bits) on interface 0
> Ethernet II, Src: Polycom 80:79:25 (00:04:f2:80:79:25), Dst: AxiomTec 52:31:74 (00:60:e0:52:31:74)
> Internet Protocol Version 4, Src: 10.120.142.38, Dst: 10.120.142.10
> User Datagram Protocol, Src Port: 5060, Dst Port: 5060

✓ Session Initiation Protocol (INVITE)

  > Request-Line: INVITE sip:Conference@10.120.142.10 SIP/2.0
  > Message Header

✓ Message Body

    Session Description Protocol

          Session Description Protocol Version (v): 0
        > Owner/Creator, Session Id (o): - 1643650001 1643650001 IN IP4 10.120.142.38
          Session Name (s): Polycom IP Phone
        > Connection Information (c): IN IP4 10.120.142.38
        > Bandwidth Information (b): AS:512
        > Time Description, active time (t): 0 0
          Session Attribute (a): sendrecv
        > Media Description, name and address (m): audio 2230 RTP/AVP 0 8 18 101
        > Media Attribute (a): rtpmap:0 PCMU/8000
        > Media Attribute (a): rtpmap:8 PCMA/8000
        > Media Attribute (a): rtpmap:18 G729/8000
        > Media Attribute (a): fmtp:18 annexb=no
        Media Attribute (a): rtpmap:101 telephone-event/8000
        Media Description, name and address (m): video 2232 RTP/AVP 109 34
        > Media Attribute (a): rtpmap:109 H264/90000
        > Media Attribute (a): fmtp:109 profile-level-id=42800d; packetization-mode=0
        Media Attribute (a): rtpmap:34 H263/90000
          Media Attribute (a): fmtp:34 CIF=1;QCIF=1;SQCIF=1
          [Generated Call-ID: 3eb4b0b3f12a5dbe880d263381807925]
```

Workaround:

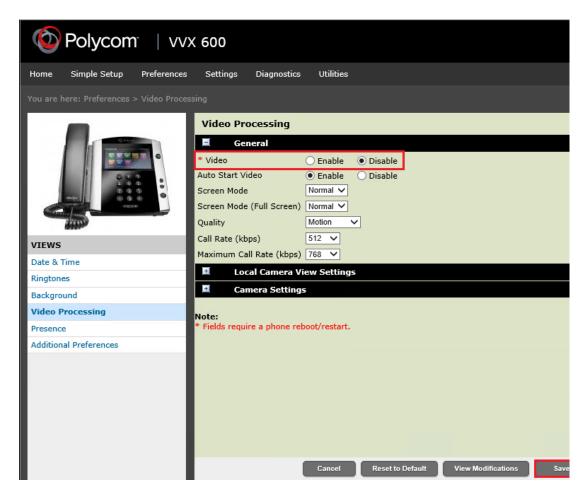
The easiest method to workaround this issue is to simply disable Video Processing on these phones as shown below.

- 1. Discover the phone's IP address. Go to Home -> Settings -> Status -> Network -> TCP/IP Parameters.
- 2. Use a web browser and navigate to http:// or https:// (which response may be determined by the firmware version of the phone) and login under the 'Admin' user.

- If the password is unknown, try to contact Support as a next step.
- 3. Once logged in go to Preferences and click on 'Video Processing'.



4. For the 'Video' option, click 'Disable' and then 'Save' at the bottom of the screen. The phone will reboot.



5. Once the phone has rebooted / re-registered, calls using the features noted in the 'Issue' section can be retested to verify they are working as expected.

If still having issues at this point, please contact Support for additional assistance.