PGW 2200 Softswitch Error Resolution for MGCP Hung Calls

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Introduction

This document explains the items linked to hung calls on the gateway for the Call Control Cisco PGW 2200 Softswitch solution, in combination with a scenario to help you troubleshoot. Currently, the Cisco IOS® gateway does not have the ability to correlate the service processing element (SPE) (which is explained in the document Understanding NextPort SPE Versions) with a digital service zero (DSO) and a Media Gateway Control Protocol (MGCP) connection. Without Cisco IOS debugs, it is not possible to map a DSO to a digital signal processor (DSP) with the Cisco IOS command **show tdm mapping** for MGCP-based call types. Cisco bug ID CSCdz47711 (registered customers only) is introduced to fix this situation for the AS5350, AS5400, and AS5850 Cisco IOS gateways.

Prerequisites

Requirements

Readers of this document should have knowledge of these topics:

- Cisco Media Gateway Controller Software Release 9 Documentation
- Release Notes for the Cisco Media Gateway Controller Software Release 9.3(2)
- Release Notes for the Cisco Media Gateway Controller Software Release 9.4(1)

Components Used

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

- Cisco PGW 2200 software releases 9.3(2) and 9.4(1)
- Cisco IOS gateway release 12.3 and 12.3T

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, make sure that you understand the potential impact of any command.

Conventions

For more information on document conventions, refer to the Cisco Technical Tips Conventions.

Resolve MGCP Hung Call Errors

If you experience a hung MGCP call scenario, the use of debugs is not useful. Also, for a live system, it is difficult to correlate the synchronous payload envelope (SPE) with a DS0 and MGCP connection. If you want to correlate the DS0 and DSP for an active call, this document provides an explanation.

Before you begin, on the PGW 2200, ensure that the MgcpBehavior setting (use Man-Machine Language [MML]) has a value that equals 2 for the Cisco IOS gateway. Refer to the document XECfgParm.dat File Parameters for more information.

PGW 2200 version 9.1(5):

- If MgcpBehavior equals 1 (gateways that are not based on Cisco IOS Software, such as Cisco Voice Interworking Service Module [VISM] and Cisco MGX) upon receipt of the 501 error code, the PGW 2200 sets the circuit to a state to prevent further use. Refer to the document Components and Properties for more information.
- If the MgcpBehavior equals 2 (Cisco IOS gateway), upon receipt of the 501 error code, the PGW 2200 sets the circuit to a state to prevent further use. Upon receipt of the 502 error code in response to the first Create Connection (CRCX), the PGW 2200 sends the MGCP Delete Connection (DLCX) message, followed by another MGCP CRCX message. If another 502 error code is returned by the Cisco IOS gateway, the call is released. The assumption is that the circuit is again usable. See the document <u>Components and Properties</u> for more information.

PGW 2200 version 9.2(2) and later:

- If the MgcpBehavior equals 1 (for the VISM and MGX), upon receipt of the 501 error code, the PGW 2200 sets the circuit to a state to prevent further use.
- If the MgcpBehavior equals 2 (Cisco IOS gateway), upon receipt of the 501 error code, the PGW 2200 sets the circuit to a state to prevent further use. Upon receipt of the 502 error code (for the first MGCP CRCX message), the PGW 2200 sends an MGCP DLCX message followed by another MGCP CRCX message. If the PGW 2200 receives another 502 error code, the call is released. The circuit is set to a state to prevent further use. At the same time, the circuit is included in a list of circuits on which a background (mini) audit is performed. This audit sends a forced MGCP DLCX message for all the circuits in the mini audit list to try to bring the circuit state in synchronization with the PGW 2200.

The MGCP response time-out is treated like a transient failure GW_HELD condition, and the MGCP DLCX message retries every minute. Only the receipt of the Restart in Progress (RSIP) (graceful/forced) message, MGCP error code 500, or one of the special 501/502 error codes causes a permanent failure if the MgcpBehavior property is set appropriately. Be aware that error code 500 always causes a failure, regardless of MgcpBehavior, because it equates to "endpoint unknown."

Note: With PGW 2200 release 9.5(2) and later, the PGW 2200 has implemented MGCP 1.0. This provides more robustness and better error-handling procedures.

Message	Cisco IOS Software (5xxx)
CRCX	502
Modify Connection (MDCX)	515
DLCX	250
Notification Request (RQNT)	400
Audit Endpoint (AUEP)	500

The reason for this is because the PGW 2200 has an audit mechanism to synchronize the channel states with the network element, such as a Cisco IOS gateway, with which it communicates. The audit program on the PGW 2200 runs at 4:00 a.m. (0400) every morning and does these actions in accordance with different scenarios:

- Scenario 1: When the channel state is BUSY on the PGW 2200 as well as the Cisco IOS gateway, there is no action.
- Scenario 2: When the channel state is IDLE on the PGW 2200 as well as the Cisco IOS gateway, an MGCP DLCX is sent to the Cisco IOS gateway for that endpoint. This clears any hung connection, if it exists.
- **Scenario 3:** When the channel state is BUSY on the PGW 2200 and IDLE on the Cisco IOS gateway, the PGW 2200 releases the call and sends a DLCX to the Cisco IOS gateway for the corresponding endpoint to synchronize the Cisco IOS gateway.
- Scenario 4: When the channel is IDLE on the PGW 2200 and BUSY on the Cisco IOS gateway, the PGW 2200 sends an MGCP DLCX to the Cisco IOS gateway for the corresponding endpoint to synchronize the Cisco IOS gateway. The PGW 2200 and Cisco IOS gateway audit procedure clears the channel on the Cisco IOS gateway. If the initial procedure that the Message Definition Language (MDL) invokes fails to bring the circuit to an idle condition, it invokes an engine interface to mark the endpoint as disabled and create an entry for the special hung/stranded endpoint audit mechanism of the engine. To change the MgcpBehavior value for the Cisco IOS gateway, change the MgcpBehavior property on the MGCPPATHs to 2.

```
mml> prov-sta::srcver="active",dstver="cisco1"
mml> prov-ed:sigsvcprop:name="sigmgcpto5xxx",MgcpBehavior="2"
mml> prov-cpy
```

Note: In some instances, a reload of the Cisco IOS gateway is requested to start from a clean situation again. Before doing this, some detail logging of the Cisco IOS gateway can help to solve the problem.

show Commands

The **show** commands discussed here can help with the verification and troubleshooting of a hung call.

Certain **show** commands are supported by the <u>Output Interpreter Tool</u> (<u>registered</u> customers only), which allows you to view an analysis of **show** command output.

The **show call active voice compact duration more?** command can help to find long-duration calls on the Cisco IOS gateway:

```
V5xxx-3\# show call active voice compact duration more ? 
 <1-2147483647> time in seconds 
 V5xxx-3\#
```

The **show call active voice brief** | **include duration 4d** command can also provide guidelines:

These **show** commands can help to determine the hung call:

- **show mgcp statistics**—Displays MGCP statistics about received and transmitted network messages.
- show mgcp connection—Displays information for active connections that are controlled by the MGCP.
- **show rtpspi statistics**—Displays the Real-Time Transport Protocol (RTP) Service Provider Interface (SPI) statistics.
- show ip socket—Displays IP socket information.
- show voice call summary—Displays a summary of all voice ports.
- show voice port summary—Displays summary configuration information about a specific voice port.
- **show vtsp call fsm**—Displays the complete history of all voice telephony service provider (VTSP) finite state machine (FSM) transitions.
- **show csm voice**—Displays the information related to the call switching module (CSM). The information is the CSM state that the machine is in for the call associated to that DSP channel, the start time of the call, the end time of the call, and the channel on the controller used by the call.**Note:** If it is an MGCP Signaling System 7 (SS7), then this command is not much use.
- show spe—Displays the SPE status.
- show spe voice summary—Displays the SPE voice status.
- **show port operational-status slot/port** (for the suspected DSP)—Displays information for all ports on the specified slot and SPE.
- show port voice log reverse *slot/port* (for the suspected DSP)—Displays information for all ports on the specified slot and SPE.

The information in the series of **show** commands that follows references MGCP calls through AS5xxx gateways, which include <code>Call_ID©</code> information (highlighted in boldface) for this call. This is also important for when you want to troubleshoot. The MGCP endpoint can be found with the Cisco IOS Software **debug mgcp packet** command or with the Cisco Snooper application.

```
V5xxx-3# show mgcp connection
Endpoint Call_ID©) Conn_ID(I) (P)ort (M)ode (S)tate (CO)dec (E)vent[SIFL]
(R)esult[EA]

1. S3/DS1-0/1 C=2F,1,2 I=0x2 P=16628,17204 M=3 S=4,4
CO=2 E=0,0,0,0 R=0,0
```

Note: Check the M status, which is linked to the MGCP mode on <u>Troubleshoot Mute Calls on the Cisco PGW 2200</u>.

The **show call active voice brief** command provides information on transmit (Tx)/receive (Rx) packet information.

```
V5xxx-3# show call active voice brief
Telephony call-legs: 1
SIP call-legs: 0
H323 call-legs: 0
MGCP call-legs: 1
Multicast call-legs: 0
Total call-legs: 2
11DA : 37079hs.1 +-1 pid:0 Originate connecting
dur 00:00:00 tx:1198/189454 rx:113437/18149920
IP 10.48.84.217:17204 rtt:0ms pl:16000/1290ms lost:29/34/29 delay:30/25/110ms
g711alaw
media inactive detected:n media contrl rcvd:n/a timestamp:n/a
11DA: 37079hs.2 +0 pid:52 Originate active
 dur 00:37:50 tx:113437/18149920 rx:1198/189454
Tele 3/0:0 (1) [3/0.1] tx:2270655/3000/0ms g711alaw noise:-65
acom:90 I/0:-51/-45 dBm
Telephony call-legs: 1
SIP call-legs: 0
H323 call-legs: 0
MGCP call-legs: 1
Multicast call-legs: 0
Total call-legs: 2
v5xxx-3#
```

Issue the **show voip rtp connections** command to find out the Remote Gateway details. These include the Callid information for that call. (In this case, Callid is 1.)

```
v5xxx-3# show voip rtp connections
VoIP RTP active connections:
No. CallId dstCallId LocalRTP RmtRTP LocalIP RemoteIP
1 2 1 16628 17204 10.48.84.26 10.48.84.217
Found 1 active RTP connections
v5xxx-3#
```

The **show vtsp call fsm** command is a hidden Cisco IOS Software command and is only used for Cisco Technical Support and the Cisco Development team. With this command, you can look for the enclosures with the phrase "Invalid FSM". The **show vtsp call fsm** command displays the complete history of all VTSP FSM transitions. It is triggered automatically whenever any DSP problem occurs while **debug vtsp error** command-line interface (CLI) is turned on.

Note: You also can convert CallId = 1 into hex, which gives you id = 0x1.

```
id=0x1 state=S_CONNECT chan_id=3/0:0 (1) DSM state=S_DSM_BRIDGED
Stack 0:
State Transitions: timestamp (state, event) -> (state, event) ...
370.796 (S_SETUP_REQUEST, E_TSP_PROCEEDING) ->
370.796 (S_SETUP_REQ_PROC, E_TSP_CONNECT) ->
Event Counts (zeros not shown): (event, count)
(E_TSP_PROCEEDING, 2) : (E_TSP_CONNECT, 2) :
State Counts (zeros not shown): (state, count)
(S_SETUP_REQ_PROC, 2) : (S_SETUP_REQUEST, 2) :
----- DSM basic call state information -----
id=0x1 state=S_DSM_BRIDGED chan_id=0
Stack 0:
State Transitions: timestamp (state, event) -> (state, event) ...
370.796 (S_DSM_INIT, E_DSM_CC_GEN_TONE) ->
370.796 (S_DSM_INIT, E_DSM_CC_CALL_MODIFY) ->
370.796 (S_DSM_INIT, E_DSM_CC_BRIDGE) ->
370.800 (S_DSM_BRIDGING, E_DSM_CC_CAPS_IND) ->
370.800 (S_DSM_BRIDGING, E_DSM_CC_CAPS_ACK) ->
475.764 (S_DSM_BRIDGED, E_DSM_CC_GET_LEVELS) ->
2641.564 (S_DSM_BRIDGED, E_DSM_CC_GET_LEVELS) ->
Event Counts (zeros not shown): (event, count)
(E_DSM_DSP_GET_VP_DELAY, 496) : (E_DSM_DSP_GET_VP_ERROR, 496) : (E_DSM_DSP_GET_TX,
496) : (E_DSM_DSP_GET_RX, 496)
(E_DSM_DSP_GET_LEVELS, 2) : (E_DSM_CC_BRIDGE, 1) : (E_DSM_CC_GEN_TONE, 1) :
(E_DSM_CC_REQ_PACK_STAT, 496)
(E_DSM_CC_CAPS_IND, 1) : (E_DSM_CC_CAPS_ACK, 1) : (E_DSM_CC_CALL_MODIFY, 1) :
(E_DSM_CC_GET_LEVELS, 2)
State Counts (zeros not shown): (state, count)
(S_DSM_INIT, 3): (S_DSM_BRIDGING, 2): (S_DSM_BRIDGED, 2484):
v5xxx-3#
```

To find out on which DSP the call is being connected, issue the command **show tdm mapping** and link the details to the endpoint for which you are tracing. In this case, it is **S3**/DS1-**0**/1:

This is connected to SPE 1, port 1. Issue the **show spe** command to find out the Port and Call states.

```
v5xxx-3# show spe
Settings :
```

v5xxx-3#

In this case, you can find out if packets are still sent in and out on that SPE port if you issue the **show port operational-status 1/0** command (for the suspected DSP):

```
v5xxx-3# show port operational-status 1/0
Slot/SPE/Port -- 1/0/0
                                        : Voice service
Service Type
Voice Codec
                                                   : G.711 a-law
Voice Codec
Echo Canceler Length
Echo Cancellation Control
                                                 : 8 ms
                                                 : Echo cancellation - disabled
Echo update - enabled
                                                   Echo update - enabled
Non-linear processor - enabled
                                                    Echo reset coefficients - disabled
                                                    High pass filter enable - disabled
Digit detection enable
                                       : DTMF signaling - enabled
: Enabled
: Generate comfort noise
: OOB Digit relay - enabled
IB Digit relay - enabled
Voice activity detection
Comfort noise generation
Digit relay enable
                                    : 20 ms
: adaptive
Information field size
Playout de-jitter mode
Encapsulation protocol
                                                   : RTP
Input Gain
                                                   : 0.0 dB
                                                  : 0.0 dB
Output Gain
Tx/Rx SSRC
                                                 : 24/0
Current playout delay
Min/Max playout delay
Clock offset
Predictive concealment
Interpolative concealment
Silence concealment
                                                 : 30 ms
                                                 : 25/110 ms
                                                 : 180505398 ms
                                                  : 0 ms
                                                 : 1105 ms
                                                 : 0 ms
Silence concealment
Buffer overflow discards : 19
End-point detection errors : 23
: 944
Tx/Rx Voice packets
                                                 : 944/88273
```

```
Tx/Rx signaling packets
Tx/Rx comfort noise packets
--- Auration
                                         : 0/0
                                         : 11/0
                                         : 1767250/1767250 ms
                                         : 3000/16000 ms
Out of sequence packets
                                         : 0
Bad protocol headers
                                         : 0
                                          : 23
Num. of late packets
Num. of early packets
                                         : -45.2/-51.2 dBm
Tx/Rx Power
                                         : -44.3/-51.0 dBm
Tx/Rx Mean
VAD Background noise level
                                         : -65.8 dBm
ERL level
                                         : 27.7 dB
                                          : 90.1 dB
ACOM level
Tx/Rx current activity
Tx/Rx byte count
                                          : silence/silence
TX/RX current activity : silence/silence

Tx/Rx byte count : 151051/14123360

ECAN Background noise level : 0.0 dBm
                                         : 4144068239
Latest SSRC value
Number of SSRC changes
                                         : 1
Number of payload violations : 0
v5350-3#
```

Issue this command several times to provide details on the type of connection that is in combination with the Remote Gateway. Issue this command on the Local/Remote Gateway to find out the status.

If you have a hung call, you can issue the **debug vtsp error** and **debug mgcp packet endpoint S3/DS1-0/1** commands. When you bring the MGCP endpoint down, the result is this debug message:

```
Apr 9 12:30:18.602: MGCP Packet received from 10.48.84.25:2427-DLCX 617 S3/DS1-0/1@v5300-3.cisco.com MGCP 0.1
C: 1C
I: 4D
R:
S:
X: 268
Apr 9 12:30:18.626: 250 617 OK
P: PS=128, OS=20241, PR=16615, OR=2658400, PL=4, JI=24, LA=0
```

These commands are also useful:

The **show mgcp statistics** command also provides details on the failed connection. Try to understand the failed field information. One of causes of the failed MGCP connection is the fact that endpoint reports are in transient mode and are temporarily unavailable when the PGW 2200

sends a CRCX. The PGW 2200 then releases with a temporary failure as a cause and attempts that endpoint again at a later time because it was only in transient mode. These SS7 circuit identification codes (CICs) do not have any MGCP connection. The reason for this situation is that the MGCP on the gateway returns a 400 MGCP error code (temporary failure for new CRCX messages sent by the Cisco IOS gateway).

```
v5xxx-3# show mgcp statistics
UDP pkts rx 306, tx 330
Unrecognized rx pkts 0, MGCP message parsing errors 0
Duplicate MGCP ack tx 0, Invalid versions count 0
CreateConn rx 0, successful 0, failed 0
DeleteConn rx 0, successful 0, failed 0
ModifyConn rx 0, successful 0, failed 0
DeleteConn tx 0, successful 0, failed 0
NotifyRequest rx 0, successful 0, failed 0
AuditConnection rx 0, successful 0, failed 0
AuditEndpoint rx 306, successful 305, failed 1
RestartInProgress tx 1, successful 1, failed 0
Notify tx 0, successful 0, failed 0
ACK tx 305, NACK tx 1
ACK rx 0, NACK rx 0
IP address based Call Agents statistics:
IP address 10.48.84.25, Total msg rx 306,
    successful 305, failed 1
System resource check is DISABLED. No available statistic
v5xxx-3#
```

Diagnose PGW 2200 Hung Calls

This section provides steps to isolate a hung SS7 CIC on the PGW 2200 in the way CIC "x" via the MML command **rtrv-tc:all** is stuck as call out on the PGW 2200. First, issue the MML **prt-call** command on this CIC.

For example, on an MGCP backhaul connection, if the bearer requested in the SETUP message is not available for that call, the PGW 2200 generates the alarm PRI: B-Channel Not Available and reports CP_ERR_CHAN_NOT_ACQ errors in platform.log. Other error messages can appear in platform.log, depending on the type of call scenario that you are running. For details, refer to the <u>Diagnosing Hung Calls</u> section of the document <u>Troubleshooting the Cisco MGC Node</u> for the PGW 2200.

There are three possible reasons for the nonavailability:

- 1. The bearer is not configured.
- 2. The bearer is not in service. (For example, it is in an Out-of-Service (OOS) state, it is in a locked/blocked state, or the MGCP disabled the endpoint.)
- 3. The bearer is busy (glare condition).

Perform these steps:

- 1. Note when the PGW 2200 reports errors for each call.
- 2. If you see errors at least three to five times in a single day on the same CIC (bearer), it is suspect.
- 3. Check the status of the CIC/bearer with the use of the rtrv-tr:all MML command. If it is idle,

the CIC is not hung.

4. If the SS7 CIC is busy, issue the **prt-call** command on that CIC.For more details on the **prt-call** MML command, issue the command **help:prt-call**.

```
mgc-bru-20 mml> help :prt-call
       MGC-01 - Media Gateway Controller 2004-11-29 19:32:35.998 GMT
        M RTRV
PRT-CALL -- Print Call
______
             Prints diagnostic information about hung calls to a log file.
  Purpose:
  [,EVT]
  Input
  Description: Target parameters are as follows:
              * sigPath -- Corresponding MML name for any of the
               following component types:
                - Signal path of in-band TDM up to MUX and then
                  time switched to TDM media and sent to Cisco MGC
                - Signal path of in-band TDM signaling up to CU
                  and then encapsulated and sent over IP to the Cisco MGC
<Press 'SPACE' for next page, 'Enter' for next line or 'q' to quit this output>
```

A print call file with the .prt extension is written in the /opt/CiscoMGC/var/trace directory.

5. Open the file and search for the string LcmOrigSmState. If you see both ${\tt OrigSmState}$ and

```
TermSmState as Relidle, you do not have a hung CIC.Example:
VAR LcmOrigSmState: STATE
{
   OsmRelidle
} [8]
VAR LcmTermSmState: STATE
{
   TsmRelidle
```

If either <code>origsmstate</code> or <code>Termsmstate</code> is not <code>RelIdle</code>, you have a likely suspect. Here are two examples of hung CIC print calls: **Example 1**:

```
VAR LcmOrigSmState: STATE
{
   OsmRelTerm3wAwaitConnDelInd
}[8]
VAR LcmTermSmState: STATE
{
   TsmRelTermInit
}[8]

Example 2:
VAR LcmOrigSmState: STATE
{
   OsmRelOrigInit
}[8]
VAR LcmTermSmState: STATE
{
   TsmRelIdle
}
```

If you reach the next step, you have identified a hung CIC.

6. Issue the **stp-call** MML command to clear the hung CIC.Issue the **grep Osm** *file_name.prt* command. You should get <code>osmRelIdle.Issue</code> the **grep Tsm** *file_name.prt* command. You should get <code>TsmRelIdle.If</code> you do not see <code>osmRelIdle</code> and <code>TsmRelIdle</code>, and if this condition

persists after you issue another **prt-call** command (may be part of transient), the CIC is likely hung.

7. If the issue of the **stp-call** command fails to clear the problem, issue the **kill-call** MML command. The **kill-call** command does not clear the connection in the MGCP gateway. Therefore, an MGCP audit is required if you issue the **kill-call** command. Perform the audit during a low-traffic period. For more details on the **kill-call** command, issue the **help**:**kill-call** command:

```
PGW2200A mml> help :kill-call
         MGC-01 - Media Gateway Controller 2004-11-29 19:34:52.084 GMT
          M RTRV
KILL-CALL -- Resolve a Stuck CIC
Purpose: Resolves a stuck or hung CIC (forcefully releases a bearer channel
associated with a single call instance that cannot be returned to
the idle state with the reset-cic or stp-call command) on the MGC.
Note: This command only releases bearer channels locally on the
MGC. No SS7 messages are sent to the remote call side (destination
MGW).
Syntax: kill-call:<sigpath_name>|<target>:CID=sip call id,confirm
kill-call:<sigpath_name>|<target>:[span= number,]confirm
kill-call:<sigpath_name>|<target>:[cic=<num>], [RNG=number,]com
kill-call:<dest_mgw>:span=<span>,bc=<bearer channel>,[RNG=numbm
Input * sigpath_name -- MML name of the SS7 or ISDN-PRI signal path
Description:
<Press 'SPACE' for next page, 'Enter' for next line or 'q' to quit this output>
```

8. Create a service request with <u>Cisco Technical Support</u> and submit the **prt-call** output for analysis.

Related Information

- Cisco PGW 2200 Softswitch Troubleshooting TechNotes
- Cisco Signaling Controllers Product Support
- Voice Technology Support
- Voice and IP Communications Product Support
- Troubleshooting Cisco IP Telephony
- Technical Support & Documentation Cisco Systems