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QUESTION 1

Users report silence on the line when they try to connect to external voice numbers. The company is using Cisco 8865 phones with Cisco UCM connected to a gateway with two T1 circuits. The MGCP server seems to be responding. Which action must the engineer take to begin to investigate the media for this type of issue?

A. Run RTMP on the Cisco router.

B. Verify that the MGCP service is running under serviceability in Cisco UCM.

C. Run the debug mgcp media command on the Cisco router.

D. Run the debug H.323 gateway command on the Cisco router.

Correct Answer: C

QUESTION 2

What is a description of RTP timestamps or sequence numbers?

A. The sequence number is used to detect losses.

B. Timestamps increase by the time "carrying" by a packet.

C. Sequence numbers increase by four for each RTP packet transmitted.

D. The timestamp is used to place the incoming audio and video packets in the correct timing order (playout delay compensation).

Correct Answer: D

Reference: https://www.cs.columbia.edu/~hgs/rtp/faq.html

QUESTION 3

Refer to the exhibit.

Region Configuration   Related Links: Back To Find/List ▼   Go

🖫 Save   ✖ Delete   🔄 Reset   ✏ Apply Config   ➕ Add New

Region Information

Name* Region A

Region Relationships

Region	Audio Codec Preference List	Maximum Audio Bit Rate	Maximum Session Bit Rate for Video Calls	Maximum Session Bit Rate for Immersive Video Calls
Region B	Use System Default (Factory Default low loss)	8 kbps (G.729)	None	None
Region C	Use System Default (Factory Default low loss)	16 kbps (iLBC, G.728)	Use System Default (384 kbps)	Use System Default (2000000000 kbps)
Region D	Use System Default (Factory Default low loss)	64 kbps (G.722, G.711)	Use System Default (384 kbps)	Use System Default (2000000000 kbps)
NOTE: Regions not displayed	Use System Default	Use System Default	Use System Default	Use System Default

Media Resource Group Information

Name* MRG2

Description Media Resource Group 2

Devices for this Group

Available Media Resources**
```
MOH_5
MTP_3
MTP_4
MTP_5
XCODER
```

✓ ∧

Selected Media Resources*
```
ANN_2 (ANN)
CFB_2 (CFB)
MOH_2 (MOH)
MTP_2 (MTP)
```

```
#CallManager SDL Logs
|AppInfo  |DET-MediaManager- (22401) : :preCheckCapabilities, caps mismatch!
Xcoder Reqd. kbps(8),
    filtered A[capCount=0 {Cap,ptime}=], B[capCount=4 {Cap,ptime} = {11,220}
{12,220} {15,220} {9,270} ]
    allowMTP=0 numXcoderRequired=1 xcodingSide=1
|SdlSig        |MrmAllocateMtpResourceErr   |waitResourcesAllocated
    |MediaManager (6,100,144,22401)   MediaResourceManager (6, 100, 142, 1)
```
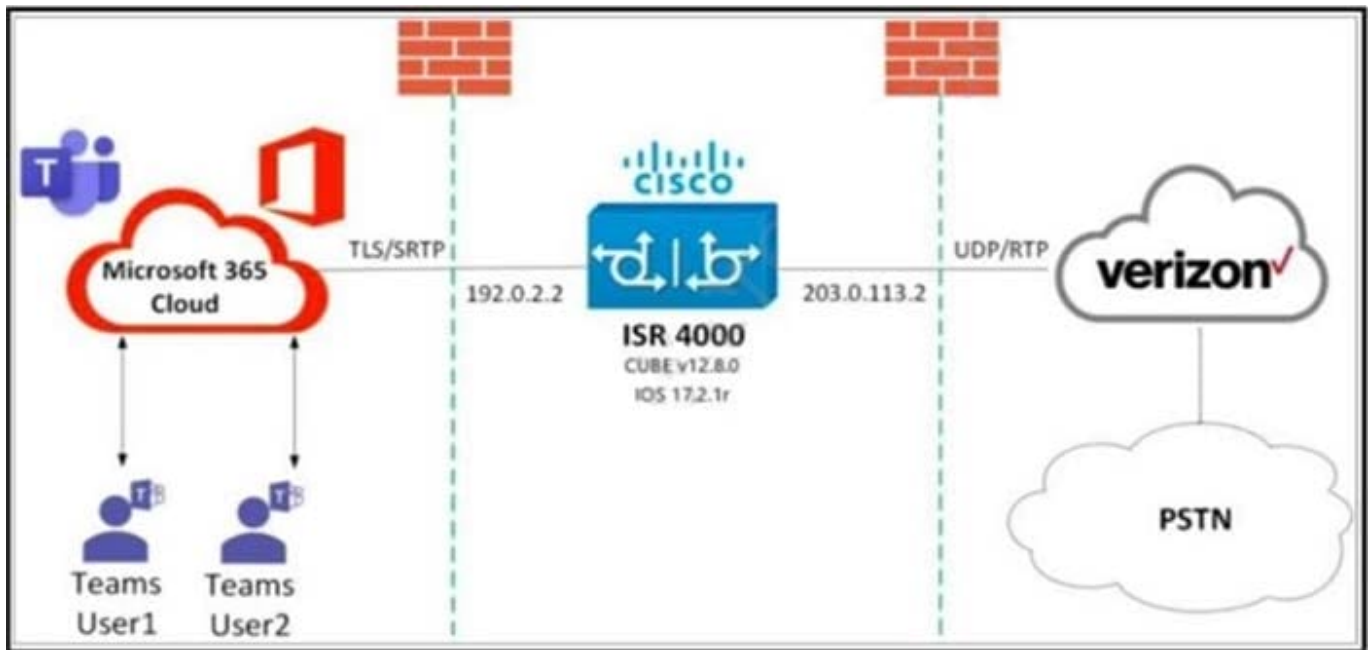
Regions have been configured for all major branches based on the available circuit bandwidth. Some calls from Region A endpoints to Region B endpoints are failing to connect. How is the issue resolved?

A. Update the calling search space for affected endpoints to none.

B. Update all regions to 8 kbps maximum audio bitrate.

C. Increase the number of available media termination points.

D. Add a media resource to transcode between available capabilities.

Correct Answer: D

Reference: https://www.cisco.com/c/en/us/support/docs/unified-communications/unified-communications-manager-callmanager/200788-Call-Recording-Basic-Configuration-and-T.html

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QUESTION 4

Refer to the exhibit. A company is using Microsoft Teams with Cisco Unified Border Element integration, but the administrator sees a one-way audio issue with Microsoft Teams. The administrator must modify the SIP profile to send the proper information on the SDP for IP address for media to match the internal and external interface. Which set of commands resolves the issue?

A. voice class sip-profiles 1 request INVITE sdp-header Connection-Info modify "2\.76\.1" "2.78.1"

B. voice class sip-profiles 1 request INVITE sdp-header Session-Owner modify "27\.0\.0" "27.3.0"

C. voice class sip-profiles 1 request INVITE sdp-header Connection add "2\.76\.1" "2.78.1"

D. voice class sip-profiles 1 request INVITE sdp-header Audio-Attribute modify "2\.76\.1" "2.71.1"

Correct Answer: C

QUESTION 5

A company has users that are logged in to hunt groups. However, there is a requirement for hunt group configurations to provide an option to turn on audible ringtones when calls to a line group arrive at a phone that is logged out and on a break. This ringtone alerts a logged-out user that there is an incoming call to a hunt list to which the line is a member, but the call does not ring at the phone of that line group member because of the logged-out status. Which action meets this requirement?

A. Configure the HLog softkey on the phone so that while a user is logged off, it plays an audible tone when a call is missed.

B. Set the service parameter Enterprise Feature Access number for hunt group logout and set up an access number.

C. Set the service parameter Party Entrance Tone to "True."

D. Configure the service parameter hunt group logoff notification and specify the name of the ringtone file.

Correct Answer: D

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