

300-815^{Q&As}

Implementing Cisco Advanced Call Control and Mobility Services
(CLACCM)

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QUESTION 1

```
46282041.005 |09:18:16.331 |AppInfo |DET-RegionsServer::matchCapabilities-- savedOption=3,
PREF_NONE, regionA=(null) regionB=(null) latentCaps(A=0, B=0) kbps=8, capACount=1, capBCount=7

46282041.006 |09:18:16.331 |AppInfo |DET-MediaManager-(1698821)::checkAudioPassThru,
param(bPostMTPAllocation=0,chkTrp=1), capCount(1,7), mtpPT=1, aPT=2

46282041.007 |09:18:16.331 |AppInfo |DET-MediaManager-(1698821)::preCheckCapabilities,
region1=RTP_Reg, region2=SJ_Reg, Pty1 capCount=1 (Cap,ptime)=(4,20), Pty2 capCount=7 (Cap,ptime)=
(4,20) (2,20) (6,20) (11,20) (12,20) (15,20) (16,20)

46282041.008 |09:18:16.331 |AppInfo |DET-RegionsServer::matchCapabilities-- savedOption=0,
PREF_NONE, regionA=(null) regionB=(null) latentCaps(A=0, B=0) kbps=8, capACount=1, capBCount=7

46282041.009 |09:18:16.331 |AppInfo |RegionsServer: applyCodecFilterIfNeeded - no codecs remained
after filtering so restored original 0 caps
```

Refer to the exhibit. All calls from site A to site B are failing, and the issue has been identified as a media negotiation problem. Which configuration change resolves this issue?

- A. Increase the bandwidth allowance between the RTP_Reg and SJ_Reg regions to 64 kbps.
- B. Enable Early Offer on the SIP trunk.
- C. Create a new audio codec preference list with G.711 U-law 64k as the highest priority and apply it to RTP_Reg and SJ_Reg.
- D. Disable G.722 on all devices at both sites.

Correct Answer: C

QUESTION 2

A user reports when they press the services key they do not receive a user ID and password prompt to assign the phone extension. Which action resolves the issue?

- A. Create the default device profiles for all phone models that are used.
- B. Subscribe the phone to the Cisco Extension Mobility service.
- C. Create the end user and associate it to the device profile.
- D. Assign the extension as a mobile extension.

Correct Answer: B

QUESTION 3

A customer routes PSTN calls to ITSP through a SIP trunk on Cisco UCM that forwards and receives calls to and from ITSP. ITSP is set to send an E.164 number when the customer's extension is four digits. Which action should be taken to route the incoming calls to four-digit extensions?

- A. Configure a voice translation profile to map the E.164 number to four digits and assign it to the incoming dial-peer on Cisco Unified Border Element.
- B. Set the Significant Digits to 8 on the SIP trunk.
- C. Set the Significant Digits to 4 on the SIP trunk.
- D. Configure a voice translation rule to map the E.164 number to four digits and assign it to the incoming dial-peer on Cisco Unified Border Element.

Correct Answer: C

QUESTION 4

Refer to the exhibit.

```
!  
dial-peer voice 10 voip  
  description Inbound  
  session protocol sipv2  
  incoming called-number 2000  
  dtmf-relay rtp-nte  
  no vad  
!  
dial-peer voice 20 voip  
  description Outbound  
  destination-pattern 2.  
  session protocol sipv2  
  session target ipv4:192.168.100.101  
  voice-class sip options-keepalive  
  dtmf-relay rtp-nte  
!  
  
CUBE#show dial-peer voice summary  
dial-peer hunt 0
```

TAG	TYPE	MIN	OPER	PREFIX	DEST-PATTERN	PRE	PASS	SESS-SER-GRP\ SESS-TARGET	OUT	STAT	PORT	KEEPALIVE	VRF
10	voip	up	up			0	sys						NA
20	voip	up	up		2.	0	sys	ipv4:192.168.100.101				busyout	NA

A call made through the Cisco Unified Border Element to pilot 2000 is failing. What is causing the call to fail?

- A. The Cisco Unified Border Element is not receiving a response to its OPTION keepalives.
- B. The destination pattern is incorrect for the dialed number.
- C. VAD was not disabled on the outgoing dial peer.
- D. No codecs are configured on the dial peers.

Correct Answer: D

QUESTION 5

Where on Cisco UCM does the administrator configure the standard local route group for a group of devices?

- A. System > Location Info
- B. Call Routing > Route/Hunt > Local Route Group Names
- C. System > Device Pool
- D. Call Routing > Emergency Location > Emergency Location (ELIN) Groups

Correct Answer: B

Reference: <https://www.uccollabing.com/configuring-standard-local-route-group-cucm/>

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