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Implementing Cisco Advanced Call Control and Mobility Services
(CLACCM)

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QUESTION 1

A customer reports audio quality issues between video endpoints in the HQ location in California and one of the branches in Texas. Which two actions must RTCP take to troubleshoot this issue? (Choose two.)

A. Allow for VAD to be used for calls using the G7.29 codec, which reduces the usage of the WAN bandwidth and saves around 30% of bandwidth per call

B. Configure the rtcp keepalive command in Cisco Unified Border Element to generate reports, which can be reviewed using the debug voip rtcp packet command.

C. Encrypt the media to stop rogue devices from replying and putting that traffic back on the WAN, which avoids any extra bandwidth and ensures the quality of the calls.

D. Gather statistics on a media connection and information such as packets sent, lost packets, jitter, feedback, and round-trip delay. This information can help isolate the type of audio quality issues and the direction of the affected traffic.

E. Compress the headers of RTP traffic to lower the bandwidth consumption over the WAN, which allows more calls with less bandwidth consumed.

Correct Answer: BD

QUESTION 2

An engineer is configuring a Cisco UCM solution. The requirements state that some users in one location will receive calls from a number during work hours 09AM to 5PM, and another group will get the calls from the same number outside this defined timeslot. Users will also change the outgoing number when reaching out to customers based on the same time-of-day routing rules. Which feature is needed to allow for this type of configuration?

A. partitions

B. CSS

C. route groups

D. route lists

Correct Answer: B

QUESTION 3

Customers that call into a company\\\'s IVR report that when they try to select an option, none of the prompts work. The administrator determines that the calls are coming in across an H.323 gateway. While analyzing the dial peer that points toward Cisco UCM, the administrator notices that no DTMF method is configured. Which command resolves this issue?

A. dial-peer voice 2 voip dtmf-relay sip-kpml

B. dial-peer voice 2 voip dtmf-relay h245-alphanumeric

C. dial-peer voice 2 voip dtmf-relay sip-notify

D. dial-peer voice 2 pots dtmf-relay h245-alphanumeric

Correct Answer: B

QUESTION 4

An engineer is configuring a Cisco Collaboration system for SIP endpoints and must enable Survivable Remote Site Telephony for these endpoints. Which code completes this configuration on the SRST gateway?

A. call-manager-fallback max-conferences 8 gain –6 ip source-address 10.10.10.100 port 2000 max-ephones 100 max-dn 200

B. telephony-service max-conferences 8 gain –6 ip source-address 10.10.10.100 port 2000 max-ephones 100 max-dn 200

C. voice service voip default mode secure address hiding allow-connections sip to sip sip registrar

D. voice register global default mode no allow-hash-in-dn max-dn 100 max-pool 200

Correct Answer: A

QUESTION 5

Refer to the exhibit.

```
!
dial-peer voice 100 voip
  description Outbound to CUCM
  translation-profile outgoing CUCM
  session protocol sipv2
  session target ipv4:192.168.100.200
  voice-class sip transport switch udp tcp
  voice-class sip conn-reuse
  voice-class sip rel1xx disable
  voice-class sip session refresh
  voice-class sip midcall-signaling block
  voice-class sip early-media update block
  dtmf-relay rtp-nte
  codec g711ulaw
  no vad
!
```

An engineer is troubleshooting an issue where inbound calls to Cisco UCM with early media fail to establish. While investigating the issue, the engineer finds that Cisco UCM is set to require a PRACK, but the Cisco Unified Border Element is not sending it. Which command is causing this issue?

A. voice-class midcall-signaling block

B. voice-class sip rel1xx disable

C. voice-class sip early-media update block

D. voice-class sip conn-reuse

Correct Answer: B