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## Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q216-Q221):

### NEW QUESTION # 216

Refer to the exhibit. An engineer is troubleshooting an issue where inbound Calls are failing after they transferred. The provider reports that update is not supported, and this is causing the calls to fail. Which command should resolve this issue?

- A. no update-callerId
- B. no contact-passig
- C. rel1 xx require "100rel"
- D. no midcall-signaling passthru

Answer: A

Explanation:

The exhibit shows a SIP message from a caller named "Amy" to a caller named "Bob". The message contains an "UPDATE" request, which is used to modify the caller ID of an existing call.

However, the provider does not support the UPDATE message, so the call is failing.

The no update-callerId command will disable the sending of UPDATE messages. This will resolve the issue, as the call will no longer be failing due to the unsupported message.

#### NEW QUESTION # 217

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 sip-notify
- C. dial-peer voice 2 voip  
dtmf-relay h245-alphanumeric
- D. dial-peer voice 2 pots  
dtmf-relay h245-alphanumeric

Answer: C

#### NEW QUESTION # 218

Which call pickup feature allows users to pick up incoming calls in a group that is associated with their own group?

- A. Directed Call Pickup
- B. Group Call Pickup
- C. Other Group Pickup
- D. BLF Call Pickup

Answer: C

Explanation:

[https://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucm/admin/4\\_1\\_3/ccmsys/a07cpick.html#wp1022865](https://www.cisco.com/en/US/docs/voice_ip_comm/cucm/admin/4_1_3/ccmsys/a07cpick.html#wp1022865)

#### NEW QUESTION # 219

A company has an SRST gateway running an IOS XE image. The company plans to enable the IPv6 addressing companywide. To enable the IPv6 in a unified SRST gateway to support SIP phones, what are two supported supplementary features for an IPv6 fallback scenario? (Choose two.)

- A. T.38 fax relay
- B. SIP trunk
- C. transcoding
- D. secure SIP lines
- E. three-way conference

Answer: A,E

Explanation:

The following supplementary services are supported as part of IPv6 in Unified SRST IP Phones:

Hold/Resume

Call Forward

Call Transfer

Three-way Conference (with BIB conferencing only)

Line to T1/E1 Trunk and Trunk to Line with Supplementary Service Features Fax to and from PSTN (IPv4 ATA to ISDN T1/E1) for both T.38 Fax Relay and Fax Passthrough



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