

300-815^{Q&As}

Implementing Cisco Advanced Call Control and Mobility Services (CLACCM)

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QUESTION 1

Which IOS command creates a SIP-enabled dial peer?

- A. voice dial-peer 20 sip
- B. dial-peer voice 20 voip
- C. dial-peer voice 20 pots
- D. dial peer voice 20 sip
- Correct Answer: B

Reference: https://www.ciscopress.com/articles/article.asp?p=664148andseqNum=6

QUESTION 2

Where is the dtmf-relay command configured on Cisco Unified Border Element?

- A. in the voice-class VoIP configuration
- B. in the VoIP dial peer
- C. in global SIP configuration
- D. in the VoIP or POTS dial peers
- Correct Answer: B

Reference: https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube/configuration/cube-book/dtmf-relay.html

QUESTION 3

Which two types of distribution algorithm are within a line group? (Choose two.)

- A. random
- B. circular
- C. highest preference
- D. top down
- E. bottom up
- Correct Answer: BD

Reference: https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/9_0_1/ccmcfg/CUCM_BK_CDF59AFB_00_admin-guide_chapter_0100011.html



QUESTION 4

Which elements does Cisco cloud mobility for collaboration include?

- A. Cisco Webex Collaboration Cloud Services
- B. Cisco Collaboration Cloud
- C. Cisco Collaboration Cloud and Cisco Webex Collaboration Cloud Services
- D. Cisco Collaboration Cloud and Cisco Mobile and Remote Access Collaboration Cloud Services

Correct Answer: D

QUESTION 5

MTP Preferred Originating Codec*	711ulaw	~
BLF Presence Group*	Standard Presence group	\sim
SIP Trunk Security Profile*	Non Secure SIP Trunk Profile	~
Rerouting Calling Search Space	<none></none>	\sim
Out-Of-Dialog Refer Calling Search Space	<none></none>	~
SUBSCRIBE Calling Search Space	<none></none>	~
SIP Profile*	Standard SIP Profile	~
DTMF Signaling Method*	RFC 2833	~

Refer to the exhibit. An administrator just implemented SIP trunking on their Cisco UCM and reports that calls using the SIP trunk are using Media Termination Point resources unnecessarily. Which action resolves the issue?

A. Change DTMF Signaling Method to "No Preference".

- B. Disable SIP Rel1XX Options.
- C. Change to a range that does not result in MTP.
- D. Change DTMF Signaling Method to "RFC 4733".

Correct Answer: C



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