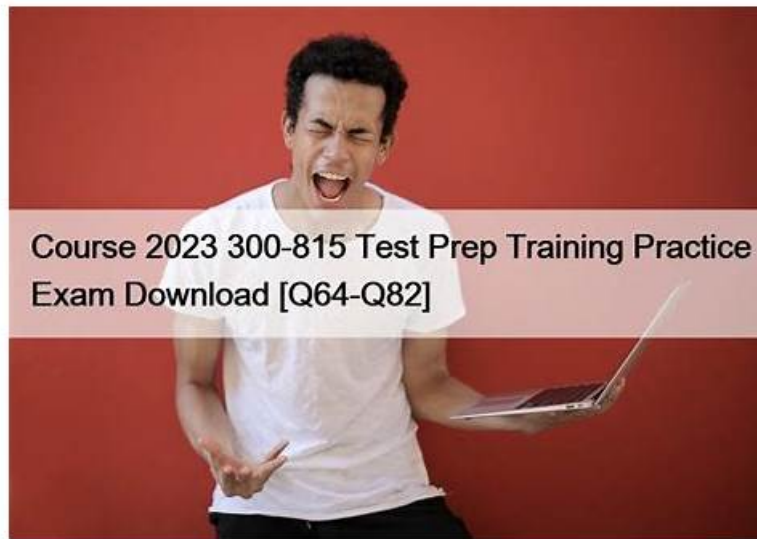


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Cisco 300-815 certification exam is one of the most popular exams for IT professionals who want to demonstrate their expertise in implementing advanced call control and mobility services. 300-815 exam is designed to test an individual's knowledge and skills in various areas, including Cisco Unified Communications Manager, Cisco Unity Connection, Cisco Unified IM and Presence, and Cisco Expressway.

Cisco 300-815 certification exam is designed to test the knowledge and skills of IT professionals in implementing advanced call control and mobility services using Cisco technology. Implementing Cisco Advanced Call Control and Mobility Services certification is ideal for those who want to enhance their career in the field of network engineering and become proficient in configuring, deploying, and managing advanced call control and mobility services solutions in a Cisco environment. 300-815 Exam covers topics such as Cisco Unified Communications Manager (CUCM) call routing, Cisco Unity Connection (CUC) call handlers, and Cisco Unified IM and Presence (IM&P) architecture.

To pass the Cisco 300-815 exam, candidates need to have a deep understanding of Cisco collaboration solutions, as well as hands-on experience in implementing and troubleshooting these solutions. 300-815 exam consists of multiple-choice questions, simulations, and hands-on labs that test the candidate's ability to configure and troubleshoot Cisco collaboration solutions.

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Cisco Implementing Cisco Advanced Call Control and Mobility Services

Sample Questions (Q114-Q119):

NEW QUESTION # 114

Management wants to change the initial announcements for one of the existing call hunt groups.

A new set of announcement audio file was provided. Which two configuration steps must the administrator take to accomplish this change? (Choose two.)

- A. Identify the configured announcement names to change under the MOH audio source section, then upload the new files to the respective announcements under the Announcement section.
- B. Identify the configured announcement names to change under the Announcement section, and assign the uploaded files to the Queuing section of the hunt pilot.
- C. Identify the MOH audio source ID associated to "Network Hold Source & Announcements" under the Queuing section of the hunt pilot.
- D. Identify the MOH audio source ID associated to one of the line group member's "User Hold MOH Audio Source".
- E. Identify the MOH audio source ID associated to one of the line group member's "Network Hold MOH Audio Source".

Answer: A,C

Explanation:

<https://community.cisco.com/t5/unified-communications-infrastructure/cucm-11-0-hunt-pilot-queue-custom-initial-announcement-not/td-p/4008782>

NEW QUESTION # 115

Which two statements are correct with respect to the Client Matter Code setting in the route pattern configuration? (Choose two.)

- A. If you check the Allow Overlap Sending check box, the Require Client Matter Code check box becomes disabled.
- B. The Client Matter Code feature does not support overlap sending because the Cisco Unified CM cannot determine when to prompt the user for the code.
- C. The Client Matter Code feature does support overlap sending because the Cisco Unified Communications Manager can determine when to prompt the user for the code.
- D. The Client Matter Code has the option to configure Authorization Level such as in the Forced Authorization Code.
- E. If you check the Allow Overlap Sending check box, you can also check the Require Client Matter Code check box.

Answer: A,B

Explanation:

Section: Call Control and Dial Planning

Explanation/Reference: [https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmfeat/](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmfeat/CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100/CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100_chapter_010000.pdf)

[CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100/CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100_chapter_010000.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmfeat/CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100/CUCM_BK_F3AC1C0F_00_cucm-features-services-guide-100_chapter_010000.pdf)

NEW QUESTION # 116

Which two types of authentication are supported for the configuration of Intercluster Lookup Service? (Choose two.)

- A. username and secret key
- B. TLS certificates
- C. FQDN of the servers defined in DNS
- D. TokenID
- E. passwords

Answer: B,E

Explanation:

Section: Cisco Unified CM Call Control Features

Explanation/Reference: [https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1/](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1/sysConfig/11_5_1_SU1/cucm_b_system-configuration-guide-1151sul/cucm_b_system-configuration-guide-1151sul_chapter_011001.pdf)

[sysConfig/11_5_1_SU1/cucm_b_system-configuration-guide-1151sul/cucm_b_system-configuration-guide-1151sul_chapter_011001.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1/sysConfig/11_5_1_SU1/cucm_b_system-configuration-guide-1151sul/cucm_b_system-configuration-guide-1151sul_chapter_011001.pdf)

NEW QUESTION # 117

End users at a new site report being unable to hear the remote party when calling or being called by users at headquarters. Calls to and from the PSTN work as expected. To investigate the SIP signaling to troubleshoot the problem, which field can provide a hint for troubleshooting?

- A. c= line of SDP content
- B. o= line of SDP content
- C. Allow: header of the 200 OK response
- D. Contact: header of the 200 OK response

Answer: A

NEW QUESTION # 118

Refer to the exhibit.

```
02904115.001 |09:08:07.093 |AppInfo |SIPtcp - wait_Sd|Signal: Outgoing SIP TCP message to 10.1.1.102 on
port 50244 index 22157
[166156,NET]
ACK sip:1315932e-ff29-4203-23e3-ef940216638e@10.1.1.102:50244;transport=tcp SIP/2.0
Via: SIP/2.0/TCP 10.1.1.5:5060;branch=z9hG4bKb4fa1fa39d7e
From: <sip:1001@10.1.1.5>;tag=93016~bb788fb3-a1ef-4d03-a96d-651038e22050-28377951
To: <sip:1000@cucm1251.cisco.lab>;tag=7001b5dab46425b45ec6648a-25dc4f91
Date: Wed, 28 Jul 2021 13:08:07 GMT
Call-ID: cee27300-1ed10da7-b403-3251300e@10.1.1.5
User-Agent: Cisco-CUCM12.5
Max-Forwards: 70
CSeq: 103 ACK
Allow-Events: presence
Session-ID: 36ed016300105000a0002834a2824611;remote=49f3b76a00105000a0007001b5dab464
Content-Type: application/sdp
Content-Length: 412

v=0
o=CiscoSystemsCCM-SIP 93016 3 IN IP4 10.1.1.5
s=SIP Call
c=IN IP4 10.1.1.5
t=0 0
m=audio 4000 RTP/AVP 0
a=X-cisco-media:umoh
b=TIAS:64000
a=ptime:20
a=rtpmap:0 PCMU/8000
a=inactive
```

Refer to the exhibit. This message is sent to the device being placed on hold for the Music On Hold audio setup. The held party reports receiving dead air rather than music when the call was put on hold. The software Music On Hold server on Cisco UCM is used in this scenario. Assume that the audio leg between the Music On Hold server and the held device uses G.711, and the relevant region relationship is configured for 64 kbps. What is the cause of the issue?

- A. Cisco UCM is sending a=inactive to the held device.
- B. The Music On Hold server does not support G.711 and a transcoder could not be allocated for the call.
- C. The device that is placed on hold does not support G.711, and a transcoder could not be allocated for the call.
- D. The bandwidth configured for this region relationship is too low and must be increased to 96 kbps or higher.

Answer: A

NEW QUESTION # 119

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