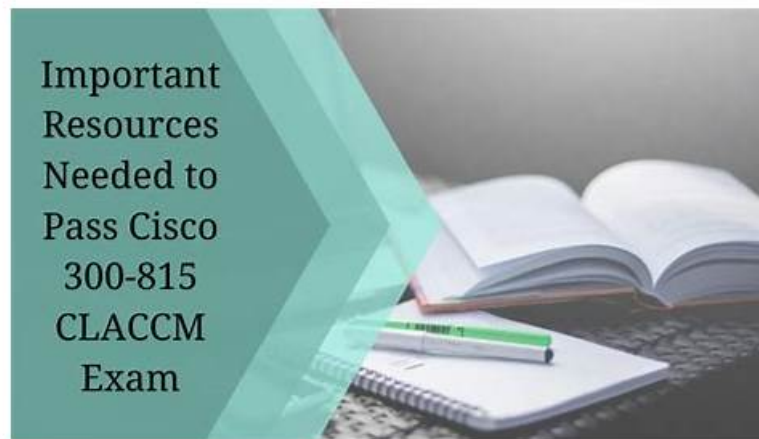


300-815 Latest Test Fee - 300-815 Valid Exam Simulator



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Topics of Cisco 300-815: Implementing Cisco Advanced Call Control and Mobility Services Exam

In the **300-815 exam dumps**, the following are also available:

1.0 Signaling and Media Protocols

- 1.2.b Call set up and tear down
- 1.2 Troubleshoot these H.323 protocol elements
- 1.1.c Mid-call signaling (hold/resume, call transfer, conferencing)
- 1.1.d Session timers
- 1.2.a DTMF
- 1.3 Troubleshoot media establishment
- 1.1 Troubleshoot these elements of a SIP conversation

2.0 CME/SRST Gateway Technologies

- 2.4.c Paging
- 2.5 Configure SIP SRST gateway
- 2.3 Implement toll fraud prevention
- 2.4.b Call park
- 2.2 Configure Cisco Unified CME dial plans
- 2.4 Configure these advanced Cisco Unified CME features

3.0 Cisco Unified Border Element

- 3.2.c Codec preference list
- 3.1.f Signaling and media bindings
- 3.1.d Dial peers
- 3.1.b Voice translation rules and profiles
- 3.2 Troubleshoot these Cisco Unified Border Element dial plan elements
- 3.2.f Signaling and media bindings
- 3.2.a DTMF
- 3.1.a DTMF

4.0 Call Control and Dial Planning

- 4.1 Configure these globalized call routing elements in Cisco Unified Communications Manager
- 4.1.d Transformation patterns
- 4.2 Troubleshoot these globalized call routing elements in Cisco Unified Communications Manager
- 4.2.a Translation patterns
- 4.1.e Standard local route group
- 4.1.g SIP trunking
- 4.1.c SIP route patterns
- 4.2.e Standard local route group

5.0 Cisco Unified CM Call Control Features

- 5.6.a Call park
- 5.3 Configure hunt groups
- 5.1 Troubleshoot Call Admission Control (exclude RSVP)
- 5.6.c Call pick-up
- 5.6.b Meet-me

6.0 Mobility

- 6.2.a Unified Mobility
- 6.2.c Device Mobility
- 6.1 Configure Cisco Unified Communications Manager Mobility
- 6.1.c Device Mobility
- 6.2 Troubleshoot Cisco Unified Communications Manager Mobility
- 6.2.b Extension Mobility

Target Audience and Prerequisites

Every Cisco certification path can be taken by the individuals with any level of technical background, and this option is not an exception. This means that you can go for the Cisco 300-815 test right away if you want to. However, it is important to understand that this is a professional-level exam, so it will be much easier for you if you have some prior experience. The applicants for this test are recommended to possess from 3 to 5 years of experience to get started with the certification process. They are advised to have a general understanding of networking, video, and voice technologies, know how to describe various codecs as well as transform analogue voice into digital streams, be familiar with the Cisco IOS XE command line, possess the relevant skills in configuring and modifying the requirements within Cisco Unified CM, and so on.

The most important task for you to perform before going for the Cisco 300-815 exam is to pass the Cisco 350-801 test first. This is the core exam, which will help you understand the basics of the domain you are planning to enter. And then you can proceed with preparation for the concentration test. Also, please note that if you don't want to obtain the CCNP Collaboration certificate, you can simply clear Cisco 300-815 and get the specialist-level certification.

Cisco 300-815 exam measures candidates' abilities to configure, implement, and troubleshoot advanced call control and mobility services. 300-815 exam covers topics such as Cisco Unified Communications Manager (CUCM) call routing and signaling, device registration and mobility, media resources, and Quality of Service (QoS) policies. 300-815 exam also tests candidates' knowledge of Cisco Collaboration applications, including Cisco Unity Connection, Cisco Unified IM and Presence, and Cisco Expressway.

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Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q197-Q202):

NEW QUESTION # 197

Which IOS command creates a SIP-enabled dial peer?

- A. voice dial-peer 20 sip
- B. dial peer voice 20 sip
- **C. dial-peer voice 20 voip**
- D. dial-peer voice 20 pots

Answer: C

Explanation:

Reference:

<https://www.ciscopress.com/articles/article.asp?p=664148&seqNum=6>

NEW QUESTION # 198

Due to a shortage of physical interfaces on a device the administrator requires that a loopback for RTP is used. Which command is required when using a loopback interface for RTP?

- A. voice-class sip bind control source-interface Loopback0
- B. voice-class sip resources priority mode passthrough
- C. voice-class sip early-offer forced.
- **D. voice-class sip bind media source-interface Loopback0**

Answer: D

NEW QUESTION # 199

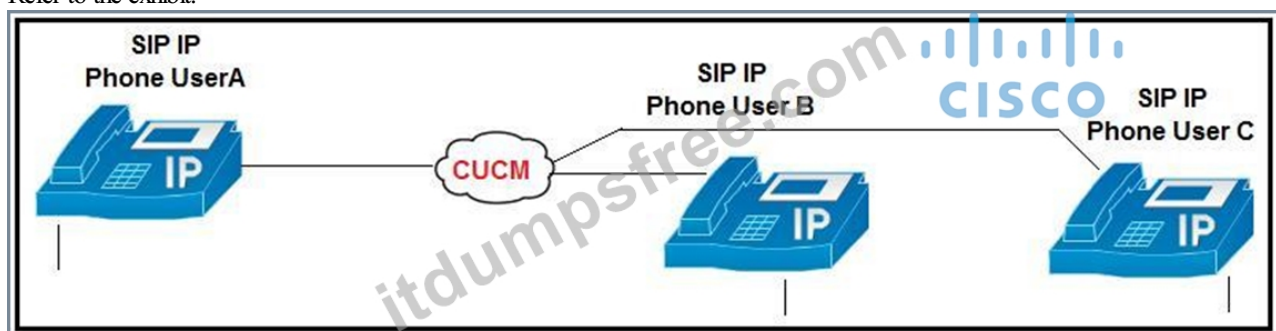
A support engineer is troubleshooting a voice network. When conducting a search for call setup details related to calling search space issues, which trace files should be investigated?

- A. Cisco IP Manager Assistant
- **B. CallManager traces**
- C. Call logs
- D. CTI Manager traces

Answer: B

NEW QUESTION # 200

Refer to the exhibit.



In an active SIP call between phone user A and phone user B, phone A initiates a call transfer to phone user C. Which two scenarios are correct? (Choose two.)

- **A. As soon as Phone_A presses the Transfer button for the first time, Phone_B hears the music on hold and the MOH audio is chosen from Phone_A Network Hold MOH Audio Source settings.**
- B. As soon as Phone_A presses the Transfer button for the first time, Phone_B hears the MOH and the MOH audio is chosen from Phone_A User Hold MOH Audio Source settings.
- C. Phone_B sends a SIP-REFER message to the Cisco Unified CM with Phone_C information in the Refer-To section.
- **D. Phone_A sends a SIP-REFER message to the Cisco Unified Communications Manager with Phone_C information in the**

- E. As soon as Phone_A presses the Transfer button for the first time, Phone_B hears the MOH and the MOH audio is chosen from Phone_B User Hold MOH Audio Source settings.

NEW QUESTION # 201

- A. ILS Max Number of Learned Objects in Database
- B. Global Data Service Parameter Limit
- C. ILS Active Learned Object Upper Limit
- D. Imported Dial Plan Replication Database Object Lower Limit

NEW QUESTION # 202

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