

Latest Cisco Questions Pdf–Pass-Sure Relevant 300-815 Questions

Cisco 300-815 Exam Detail	
Vendor	Cisco
Exam Code	300-815
Full Exam Name	Implementing Cisco Advanced Call Control and Mobility Services
Number of Questions	55-65
Sample Questions	Cisco 300-815 Sample Questions
Practice Exams	Cisco Certified Network Professional Collaboration Practice Test
Passing Score	Variable (750-850 / 1000 Approx.)
Time Limit	90 minutes
Languages	English

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Our Implementing Cisco Advanced Call Control and Mobility Services (300-815) practice exam highlights mistakes at the end of each attempt, allowing you to overcome them before it's too late. This kind of approach is great for complete and flawless Implementing Cisco Advanced Call Control and Mobility Services (300-815) test preparation. A free demo version is also available for satisfaction. This 300-815 software provides a real Implementing Cisco Advanced Call Control and Mobility Services (300-815) exam environment to help ease exam anxiety.

Cisco 300-815 (Implementing Cisco Advanced Call Control and Mobility Services) Certification Exam is a professional-level certification exam that is designed to test the knowledge and skills of IT professionals who are involved in implementing and managing advanced call control and mobility services using Cisco technologies. 300-815 exam is one of the six concentration exams that you can choose from when pursuing the Cisco Certified DevNet Professional Certification.

Cisco 300-815: Implementing Cisco Advanced Call Control and Mobility Services Exam Certification Path

The Exam is associated with the CCNP Service Provider and the Certifications of Cisco Accredited Professional-Service Provider VPN Services Implementation. This is the official condition for this examination. The course, Integrating Cisco Service Provider VPN Services, however, helps candidates to prepare for the test and is highly recommended.

Cisco 300-815 Exam is intended for professionals who work with Cisco Unified Communications Manager in a multi-site environment. 300-815 exam is also suitable for individuals who manage and monitor voice and video network devices, or those who design, deploy, and troubleshoot advanced Cisco collaboration solutions. Passing this certification exam validates your expertise in implementing, configuring, and troubleshooting advanced call control and mobility services, which is highly sought after in the IT industry.

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Relevant 300-815 Questions & 300-815 PDF

This updated Cisco 300-815 exam study material of Prep4sures consists of these 3 formats: Implementing Cisco Advanced Call Control and Mobility Services (300-815) PDF, desktop practice test software, and web-based practice exam. Each format of Prep4sures aids a specific preparation style and offers unique advantages, each of which is beneficial for strong Implementing Cisco Advanced Call Control and Mobility Services (300-815) exam preparation. The features of our three formats are listed below. You can choose any format as per your practice needs.

Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q199-Q204):

NEW QUESTION # 199

Refer to the exhibit.

An engineer is troubleshooting a call-establishment problem between Cisco Unified Border Element and Cisco UCM. Which command set corrects the issue?

- A. SIP binding in SIP configuration mode:
voice service voip sip
bind control source-interface GigabitEthernet0/0/0 bind media source-interface GigabitEthernet0/0/0
- B. SIP binding in dial-peer configuration mode:
dial-peer voice 100 voip
voice-class sip bind control source-interface GigabitEthernet0/0/0
voice-class sip bind media source-interface GigabitEthernet0/0/0
- C. SIP binding in SIP configuration mode:
voice service voip
sip
bind control source-Interface GigabitEthernet0/0/1 bind media source-Interface GigabitEthernet0/0/1
- D. SIP binding in dial-peer configuration mode:
dial-peer voice 300 voip
voice-class sip bind control source-interface GigabitEthernet0/0/1 voice-class sip bind media source-interface GigabitEthernet0/0/1

Answer: B

NEW QUESTION # 200

An organization configures a SIP trunk in Cisco UCM to connect to another system. These requirements must be met:

1. Use a specific IP address for SIP signaling.
2. Encrypt the signaling traffic.
3. Restrict which devices can use the SIP trunk.
4. Simplify SIP signaling.

Drag and drop the Cisco UCM configuration steps from the left onto the order on the right to achieve these goals.

Answer:

Explanation:

NEW QUESTION # 201

Refer to the exhibit.

Users report that when they dial to Cisco Unity Connection from an external network, they cannot enter any digits. Assuming only in-band DTMF is supported, what is a reason for this malfunction?

- A. The negotiated RTP port is outside of the range described by RFC, so inband DTMFs do not work.
- B. There is SIP Delayed Offer. DTMF is supported only in Early Offer.
- C. No DTMF is negotiated.
- D. The rtpmap:0 value for the negotiated codec is marking DTMF as inactive.

Answer: C

NEW QUESTION # 202

Refer to the exhibit.

The Cisco Unified Border Element receives an INVITE matching inbound dial peer 5002. The outbound dial peer supports only iLBC, and a Local Transcoding Interface is allocated. Based on the configuration and SDP from the INVITE message, which codec is chosen by Cisco Unified Border Element for the inbound call leg?

- A. G.711 A-law
- B. G.729r8
- C. G.729br8
- D. G.711 U-law

