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What Is 300-815 Exam and Who Is Its Targeted Audience?

This 300-815 Implementing Cisco Advanced Call Control as well as Mobility Services exam checks the applicant's understanding of various advanced level call control as well as mobility features. Particularly, such an exam is intended for those with a firm grasp of various signaling and media protocols, gateway technologies, Cisco Unified Border element, CM Call Control, and other related concepts. As for its details, the exam needs to be finished within 90 minutes. Also, the applicant requires an account on Pearson VUE to schedule the exam. Thus, the candidate should select Proctored Exams and enter 300-815 as the test number on the Pearson VUE website. As you keep in mind, securing a passing score in the Cisco 300-815 test will push the entrant a step closer to earning the CCNP Collaboration endorsement.

Earning the Cisco 300-815 certification validates an individual's expertise in advanced call control and mobility services and opens up various career opportunities in the field of networking and collaboration. It is an essential certification for IT professionals who are responsible for designing, deploying, and managing Cisco call control and mobility solutions in an enterprise environment. With the rapid growth of cloud-based collaboration and communication services, the demand for IT professionals with advanced knowledge and skills in Cisco call control and mobility services is expected to increase significantly in the coming years.

Latest 300-815 Demo | New 300-815 Test Format

Nowadays most people are attracted to the Implementing Cisco Advanced Call Control and Mobility Services (300-815) certification and take it seriously because they know that it is the future. But they can't figure out where to prepare for Implementing Cisco Advanced Call Control and Mobility Services (300-815) certification exam. After observing the problems of the students PracticeDump provides them with the best Implementing Cisco Advanced Call Control and Mobility Services (300-815) Questions so they don't get depressed anymore and pass the Implementing Cisco Advanced Call Control and Mobility Services (300-815) exam on the first try. The Implementing Cisco Advanced Call Control and Mobility Services (300-815) is designed after consulting with a lot of professionals and getting their reviews.

Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q11-Q16):

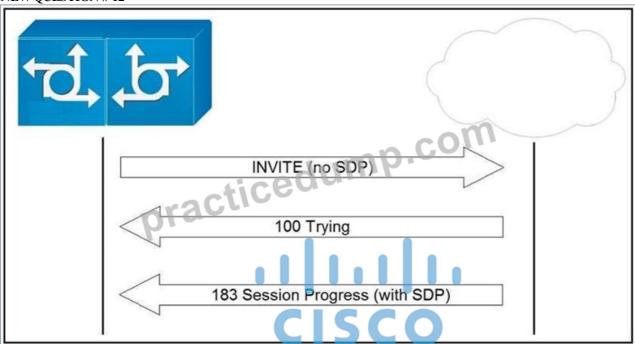
NEW QUESTION #11

A single site reports that when they dial select numbers, the call connects, but they do not get audio. The administrator finds that the calls are not routing out of the normal gateway but out of another site's gateway due to a TEHO configuration. What is the next step to diagnose and solve the issue?

- A. Verify that the route pattern is not blocking calls to the destination number.
- B. Verify that IP routing is correct between the gateway and the IP phone.
- C. Verify that the dial peer of the gateway has the correct destination pattern configured.
- D. Verify that the route pattern has the correct calling-party transformation mask

Answer: C

NEW QUESTION #12



Refer to the exhibit. An administrator is troubleshooting why users are not hearing audio when dialing long distance numbers across their Cisco Unified Border Element. The customer's carrier has a requirement that dialing long distance requires an access code to be entered. Looking at the exhibit, what two actions can be taken to correct signaling? (Choose two.)

- A. Enable Mid-Call Signaling Consumption.
- B. Enable the supplementary-service media-renegotiate command.
- C. Enanle PRACK.
- D. Enable Early Offer on the Cisco Unified Border Element.

• E. Enable Media Flow Around

Answer: C,D

Explanation:

Section: Cisco Unified Border Element

NEW QUESTION #13

Which two statements about Cisco Unified Communications Manager Extension Mobility are true? (Choose two.)

- A. A device profile has most of the same attributes as a physical device.
- B. An autogenerated device profiles can be loaded on a device at the same time as a user profile.
- C. Devices can be configured to allow more than one user to be logged in at the same time.
- D. A device can adopt a user profile even when no user is logged in.
- E. After an autogenerated device profile is created, you can associate it with one or more users.

Answer: A,D

NEW QUESTION #14

Which two descriptions of the Standard Local Route Group deployment are true? (Choose two.)

- A. can be associated only under the route list
- B. can be associated under the route group
- C. chooses the route group that is configured under the device pool of the called-party device
- D. can be assigned directly to the route pattern
- E. chooses the route group that is configured under the device pool of the calling-party device

Answer: A,E

NEW QUESTION #15

voice translation-rule 84 .usco rule 1 /^\{[2-9]..[2-9].....\$\)/ /\2/

Refer to the exhibit. Users report that outbound PSTN calls from phones registered to Cisco Unified Communications Manager are not completing. The local service provider in North America has a requirement to receive calls in 10-digit format. The Cisco Unified CM sends the calls to the Cisco Unified Border Element router in a globalized E.164 format. There is an outbound dial peer on Cisco Unified Border Element configured to send the calls to the provider. The dial peer has a voice translation profile applied in the correct direction but an incorrect voice translation rule applied, which is shown in the exhibit. Which rule modified DNIS in the format that the provider is expecting?

• A. rule 1 /

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