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Cisco 300-815 exam is an excellent way for candidates to showcase their knowledge and skills in implementing advanced call control and mobility services. Implementing Cisco Advanced Call Control and Mobility Services certification exam is highly respected in the industry and can help candidates advance their career prospects. With proper preparation and training, candidates can successfully pass the exam and earn their certification.

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 Exam covers a wide range of topics, including call routing, dial plans, media resources, call control protocols, mobility services, and Cisco Unified Border Element (CUBE). Candidates are expected to have a deep understanding of Cisco Unified Communications solutions, as well as hands-on experience in implementing and troubleshooting these technologies. Successful completion of 300-815 exam demonstrates the candidate's ability to design and deploy advanced collaboration solutions that meet the needs of modern enterprises.

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Cisco Implementing Cisco Advanced Call Control and Mobility Services Sample Questions (Q27-Q32):

NEW QUESTION # 27
Refer to the exhibit. An administrator configured a new SIP trunk between Cisco UCM and Cisco Unified Border Element. Calls that use this new trunk are failing, and debugs from the Cisco Unified Border Element do not show any signaling received from Cisco UCM. The administrator verified that OPTIONS is enabled on the SIP profile configured on the trunk on Cisco UCM. The route pattern matched for this call is correctly configured to send calls to the Cisco Unified Border Element using this new trunk. What is the cause of this failure?

- A. The called party number is not complete because it contains only 7 digits
- B. A connectivity issue between Cisco UCM and Cisco Unified Border Element prevents Cisco UCM from receiving a response to the OPTIONS it is sending to Cisco Unified Border Element
- C. There are multiple matches for the dialed number, and Cisco UCM is trying to send the call to SEP2834A2824611 rather than Cisco Unified Border Element
- D. The calling search space on the calling line/device does not contain the partition of the route pattern

Answer: B

NEW QUESTION # 28
An engineer is troubleshooting Cisco Device Mobility and find that the phone has roamed to a building that is assigned to a different device pool but has not changed its device pool accordingly.
What action resolves the issue?

- A. Set correct Location under Current Device Mobility Settings
- B. Set Device CSS under Current Device Mobility Settings.
- C. Enable SRST under Current Device Mobility Settings
- D. Set the correct subnet under Device Mobility Info.

Answer: D

NEW QUESTION # 29
An administrator troubleshoots call failure in a new deployment and finds that the SIP INVITE messages sent to the service provider contain a diversion header with the user's 4-digit directory number. These 4-digit directory numbers range from 1000 to 9999. The service provider is rejecting the calls because it requires that the diversion header contain 10 digits. Which command on the Cisco Unified Border Element resolves this issue for all users?
A)
B)

- A. Option B
- B. Option C
- C. Option D
- D. Option A

Answer: C

NEW QUESTION # 30
Which action is correct with respect to toll fraud prevention configuration in the Cisco Unified Communications Manager Express?

- A. Configure the command no ip address trusted authenticate under "voice service voip".
- B. Enable Secondary Dial tone on Analog and Digital FXO Ports.
- C. Configure IP Address Trusted Authentication for Incoming VoIP Calls.
- D. Configure Direct Inward Dial for Incoming ISDN Calls with overlap dialing.

Answer: C

Explanation:
Section: CME/SRST Gateway Technologies
Explanation/Reference: https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucme/admin/configuration/manual/cmeadm/cmetoll.html#concept_ECC4F4E7ED0F45C594B703EEF34762F2

NEW QUESTION # 31

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/