

Cisco.300-815.v2020-12-11.q21

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NEW QUESTION: 1

What is first preference condition matched in a SIP-enabled incoming dial peer?

- A. incoming called-number
- B. answer-address
- C. target carrier-id
- D. incoming uri

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 2

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: ([SHOW ANSWER](#))

NEW QUESTION: 3

Which call pickup feature allows users to pick up incoming calls in a group that is associated with their own group?

- A. BLF Call Pickup
- B. Directed Call Pickup
- C. Group Call Pickup
- D. Other Group Pickup

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 4

Which two configuration parameters are prerequisites to set Native Call Queuing on Cisco Unified Communications Manager? (Choose two.)

- A. Cisco RIS data collector service must be running on the same server as the Cisco CallManager service.
- B. The phone button template must have the Queue Status Softkey configured.
- C. Cisco IP Voice Media Streaming Service must be activated on at least one node in the cluster.
- D. The maximum number of callers allowed in queue must be 10.
- E. A unicast music on hold audio source must be configured.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 5

When you troubleshoot H.323 call setup, which message informs you that the called party is being notified about the call?

- A. RINGING
- B. ALERTING
- C. CONNECT
- D. PROCEEDING

Answer: B ([LEAVE A REPLY](#))

NEW QUESTION: 6

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 7

Which top-level IOS command is needed to begin the configuration of a Cisco Unified Communications Manager Express gateway to enable phones to be registered via SIP?

- A. allow-connections sip to sip
- B. voice service voip
- C. voice register dn
- D. voice register global

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 8

Which two types of distribution algorithm are within a line group? (Choose two.)

- A. bottom up
- B. random
- C. highest preference
- D. circular
- E. top down

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 9

A user reports that when they attempt to log out from the Cisco Extension Mobility service by pressing the Services button, they cannot log out. What is the most likely cause of this issue?

- A. The user device profile is not subscribed to the Cisco Extension Mobility service.
- B. The Cisco Extension Mobility service has not been configured on the phone.
- C. There might be a significant delay between the button being pressed and the Cisco Extension Mobility service recognizing it. It would be best to check network latency.
- D. The user device profile has not been assigned to the user.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 10

For s SIP to SIP call flow, when does Cisco Unified Border Element require transcoding resources for DTMF?

- A. interworking between an OOB method and RFC2833 for flow-through calls
- B. interworking between h245-alpha numeric and sip-kpml
- C. interworking between an OOB method and RFC2833 for flow-around calls
- D. interworking between h245-signal and rtp-nte

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 11

Where on Cisco Unified Communications Manager do you configure the standard local route group for a group of devices?

- A. Call Routing > Emergency Location > Emergency Location (ELIN) Groups
- B. System > Device Pool
- C. Call Routing > Route/Hunt > Local Route Group Names
- D. System > Location Info

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 12

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 /^+\([1].*\)/ /011\1/
- B. rule 1 /^+1\([2-9]..[2-9].....\$\)/ \0/
- C. rule 1 /^+1\([2-9]..[2-9].....\$\)/ \1/
- D. rule 1 /^\([2-9]..[2-9].....\$\)/ \1/

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 13

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

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 14

In Cisco Unified Communications Manager, which tool do you use to check SIP traces?

- A. CCSIP
- B. OS Administration Page
- C. RTMT
- D. MTP

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 15

Which two types of authentication are supported for the configuration of Intercluster Lookup Service?

(Choose two.)

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 16

Refer to the exhibit.

A user reports that when they call a specific phone number, no one answers the call, but when they call from a mobile phone, the call is answered. The engineer troubleshooting the issue is expecting the far-end gateway to cut through audio on the 183 Session Progress SIP message. Which SIP Profile configuration element is necessary for the Cisco Unified Communications Manager to send acknowledgement of provisional responses?

- A. On the SIP Profile, the configuration parameter SIP Rel1XX Options must be set to Send PRACK for all 1xx Messages.
- B. Early Offer for G Clear Calls must be enabled.
- C. Accept Audio Codec Preferences in Received Offer must be set to On.
- D. Allow Passthrough of Configured Line Device Caller Information must be enabled.

Answer: ([SHOW ANSWER](#))

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NEW QUESTION: 17

Which configuration element of a hunt group allows for changing Calling Party Transformations settings?

- A. route group
- B. line group
- C. hunt list
- D. hunt pilot

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 18

You see the voice register pool 1 command in your Cisco Unified Communications Manager Express configuration. Which configuration is occurring in this section?

- A. configuration for SIP registrar service
- B. configuration for a single SIP phone
- C. configuration for a pool of SIP phones (similar to device pool on Cisco Unified Communications Manager)
- D. configuration items common for all SIP phones

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 19

When configuring hunt groups, where do you add the individual directory numbers that will be part of the group?

- A. hunt pilot
- B. route group
- C. hunt list
- D. line group

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 20

The administrator of ABC company is troubleshooting a one-way audio issue for a call that uses H.323 protocol (slow-start mode). The administrator requests that you provide the IP and port information of the Real-Time Transport Protocol traffic that had the one-way audio call.

You gather the H.225 and H.245 messages for one of the one-way audio calls. Where can you find the RTP IP and port information for both sides? (Note: This call flow has not invoked any media resources like MTP or transcoders).

- A. H.245 Open Logical Channel Ack
- B. H.225 Connect
- C. H.245 Open Logical Channel
- D. H.245 Terminal Capability Set

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 21

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. Phone_A sends a SIP-REFER message to the Cisco Unified Communications Manager with Phone_C information in the Refer-To section.
- C. 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.
- D. 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.
- E. Phone_B sends a SIP-REFER message to the Cisco Unified CM with Phone_C information in the Refer-To section.

Answer: ([SHOW ANSWER](#))

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