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Cisco Cisco Voice over IP (CVOICE)

Practice Exam: 642-436 Exams

Exam Number/Code: 642-436

Exam Name: Cisco Voice over IP (CVOICE)

Questions and Answers: 77 Q&As

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Title : Cisco Voice over IP (CVOICE)

1. The SJ local zone contains a gatekeeper that controls two gateways, SJ1 and SJ2. Both gateways provide access to area code 408. Which two command strings should be entered into the gatekeeper to give the SJ2 gateway priority over the SJ1 gateway? (Choose two.)

- A. zone prefix SJ 408 gw-priority 6 SJ1
- B. zone prefix SJ 408 gw-priority 6 SJ2
- C. zone prefix SJ 408 gw-priority 10 SJ1
- D. zone prefix SJ 408 gw-priority 10 SJ2
- E. zone prefix SJ 408 gw-priority 0 SJ2, 10 SJ1
- F. zone prefix SJ 408 gw-priority 6 SJ1, 10 SJ2

Answer: AD

2. Refer to the exhibit. When extension 201-555-1000 dials 404-555-1200, how are the digits manipulated in R1 so that they are presented correctly at R2?

- A. The outbound VoIP dial peer is matched and all digits are sent.
- B. The digits 404-555 are stripped off before matching the outbound POTS dial peer.
- C. The digits 404-555 are stripped off by the connection trunk and R2 receives only 1200.
- D. R1 collects the 1200 and prepends the tie-line digits 404555. That number is matched to a VoIP dial peer and sent to the appropriate address.

Answer: A

3. At what point does the MGCP call agent release the setup of the call path to the residential gateways?

- A. after the call agent has been notified that an event occurred at the source residential gateway
- B. after the call agent has been notified of an event and has instructed the source residential gateway to create a connection
- C. does not release call path setup
- D. after the call agent has sent a connection request to both the source and destination and has relayed a modify-connection request to the source so that the source and destination can set up the call path
- E. after the call agent has forwarded session description protocol information to the destination from the source and has sent a modify connection to the destination and a create-connection request to the source

Answer: D

4. Refer to the exhibit. A QoS strategy has already been deployed on the LAN. Choose three WAN QoS best practices that should be used over the WAN link. (Choose three.)

- A. Implement NBAR.
- B. Implement admission control.
- C. Mark voice traffic as EF in DSCP.
- D. Mark voice traffic highest priority in 802.1p.
- E. Use cRTP to maximize bandwidth utilization.
- F. Configure access switches to trust traffic from IP phones.

Answer: BCE

5. Refer to the exhibit. Highland Park Property Development is integrating a Cisco Unified Communications Manager Express system with the existing PBX via an E1 QSIG trunk. After the initial configuration, no calls can be placed

from IP phones to PBX phones. How can this problem be resolved?

- A. Increase the ISDN T302 timer to allow more time for call setup.
- B. Add the command `isdn negotiate-bchan` to the serial interface.
- C. Add the command `isdn contiguous-bchan` to the serial interface.
- D. Change the channel selection order from descending to ascending.

Answer: B

6. Refer to the exhibit. Users are not able to complete a call from 678-555-1212 to 770-555-1111. What is the correct diagnosis for the problem?

- A. incorrect destination-pattern in router 1
- B. incorrect POTS dial-peer statement in router 2
- C. incorrect session-target statement in router 2
- D. incorrect port statement in router 1 pots dial peer
- E. missing no digit-strip on the voip dial peer in router 1

Answer: A

7. When setting up a VoIP call, what is the first thing a gateway router tries to match to a dialed number?

- A. call leg
- B. IP route
- C. session target
- D. destination pattern

Answer: D

8. Refer to the exhibit for IP addresses and telephone numbers. You are working with a customer opening a small sales office in Atlanta. You want the user in Atlanta to be able to dial into the PBX in New York over the IP WAN. The New York PBX uses ground start, a two-wire operation, and DTMF dialing.

Choose the correct FXO port configuration commands for New York.

- A. `voice-port 1/0/0`
`signal ground-start`
`operation 2-wire`
`dial-type dtmf`
- B. `voice-port 1/1/1`
`destination 2015551212`
`signal ground-start`
`operation 2-wire`
`type 1`
`dial-type dtmf`
- C. `voice port 1/0/0`
`session target ipv4:172.16.1.1`
`destination 2015551212`
`signal ground-start`
`operation 2-wire`
`dial-type dtmf`
- D. `voice port 1/0/0`
`session target ipv4:172.16.1.1`
`source 2015551212`
`signal wink-start`
`operation 2-wire`
`dial-type dtmf`

Answer: A

9. Refer to the exhibit. You have been asked to configure a dial peer on R2 that will match only the extensions of the four telephones attached. Which dial-peer statement will you use?

- A. dial-peer voice 1 pots
destination-pattern 5552.[0-5]0
- B. dial-peer voice 1 pots
destination pattern 5552[5-6].0
- C. dial-peer voice 1 pots
destination-pattern 555[2-5][56]
- D. dial-peer voice 1 pots
destination-pattern 5552[5-6][05]0

Answer: D

10. Which dial plan characteristic shows the most obvious improvement by dropping a number translation step?

- A. availability
- B. post-dial delay
- C. scalability
- D. hierarchical design

Answer: B

11. Refer to the exhibit. Your customer wants to converge the existing PBX network with the IP network. The three remote offices have various types of PBXs. The customer is using a combination of tie-lines and trunks to connect the PBXs today.

Which kind of connection should be implemented to allow calls to be placed from 201-555-1000 to 727-555-1000 so that when the call is completed, network resources are returned for other uses?

- A. PLAR
- B. trunk
- C. tie-line
- D. answer-mode

Answer: C

12. Which path selection mechanism lets you choose either the even or odd channels first?

- A. hunt groups
- B. trunk groups
- C. tailend hopoff
- D. Call Admission Control

Answer: B

13. Refer to the exhibit.

Lighthouse Equine Management would like to investigate converging voice and data on their existing T1 Frame Relay WAN link between New York and Atlanta. Currently the following list of applications are consuming no more bandwidth than what is listed on this segment of the network.

T1 link 1536 kbps
e-mail 75 kbps
internet 200 kbps
Oracle 500 kbps
FTP 250 kbps
Total 1025 kbps

The customer has allocated 25% of the WAN link for routing updates and other overhead. They would like to increase the number of samples encapsulated in each PDU to 40 ms. You have calculated 6 bytes of overhead for Frame Relay, no cRTP, and the use of the G.711 codec.

How many simultaneous calls could be placed on this link?

- A. 0 calls
- B. 1 call
- C. 2 calls
- D. no more than 5 calls

E. no more than 10 calls

F. no more than 20 calls

Answer: B

14. Refer to the exhibit. The Carmichael caller dials the site access code for Merrimack (6) followed by the four-digit extension number of the destination phone (0124). If the call is going to go across the IP WAN, which action will have to be taken?

A. Translate 50124 to 5125550124.

B. Strip the site access code and send four digits.

C. Strip the site access code and prepend 1512555.

D. Do nothing because the site access code matches the last five digits of the target number.

E. Strip the site access code, send four digits, then prepend the access code when it reaches the Merrimack gateway.

Answer: B

15. Which option is true concerning the MGCP call agent?

A. acts only as a recorder of call details

B. provides only call signaling and call setup

C. manages all aspects of the call and voice stream

D. monitors the quality of each call after setup

Answer: B

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