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https://www.dumpsfiles.com/files/Cisco/350-801/Cisco.350-801.v2025-10-28.q393	

NEW QUESTION: 1

Which two protocols should be configured for the cisco unity connection and cisco UCM integration?

- A. SIP
- B. H.323
- C. MGCP
- D. RTP
- E. SCCP

Answer: (SHOW ANSWER)

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/11x/design/guide/b_11xcucdg/b_11xcucdg_chapter_00.html#ID-2342-00000133

NEW QUESTION: 2

Refer to the exhibit.

When a UC Administrator is troubleshooting DTMF negotiated by this SIP INVITE, which two messages should be examined next to further troubleshoot the issue? (Choose two.)

- A. UPDATE
- B. PRACK
- C. REGISTER
- D. NOTIFY
- E. SUBSCRIBE

Answer: D,E (LEAVE A REPLY)

NEW QUESTION: 3

A DTMF mismatch is occurring between an MGCP gateway registered FXS port and a Cisco Unified Communications Manager SIP trunk. Which media resource can be leveraged to interwork this mismatch?

- A. Conference Bridge
- B. Trusted Relay Point
- C. Media Termination Point
- D. Annunciator

Answer: [\(SHOW ANSWER\)](#)

Section: Cisco IOS XE Gateway and Media Resources

NEW QUESTION: 4

An engineer is configuring Cisco Jabber for Windows and must implement desk phone control mode for some of the users. Which access control group must be configured for those users to enable this functionality?

- A. Standard CTI Allow Reception of SRTP Key Material
- B. Standard CTI Enabled
- C. Standard CTI Secure Connection
- D. Allow Control of Device from CTI

Answer: D [\(LEAVE A REPLY\)](#)

NEW QUESTION: 5

When configuring Cisco Unified Communications Manager, which configuration enables phones to automatically register to a Cisco Unified Communications publisher when the connection to the subscriber is lost?

- A. Cisco Unified CM Group
- B. Device Pool
- C. SRST
- D. Route Group

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

NEW QUESTION: 6

How does Cisco Unified Communications Manager perform a digit analysis on-hook versus off-hook for an outbound call from a Cisco IP phone that is registered to Cisco Unified CM?

- A. On-hook, by pressing the digits and entering "#" to process the call, Unified CM performs a digit-by-digit analysis; off-hook, Unified CM analyses all digits as a string.
- B. On-hook, unified CM performs a digit-by-digit analysis; off-hook, Unified CM considers all digit were dialed and dos not wait for additional digits.
- C. On-hook, Unified CM considers all digits were dialed and does not wait for additional digits; off-hook, Unified CM performs a digit-by-digit analysis.
- D. On-hook, no digit analysis is performed; off-hook, Unified CM requires the "#" to start the digit analysis.

Answer: C [\(LEAVE A REPLY\)](#)

NEW QUESTION: 7

Multiple route patterns match a number. How does Cisco Unified Communications Managers determine which pattern to use?

- A. the one that comes first in numerical order
- B. the one with the longest match
- C. the one with the closest match
- D. the one that discards everything PreDot

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

Section: Call Control

Explanation/Reference: <https://www.cisco.com/c/en/us/support/docs/voice-unified-communications/unified-communications-manager-callmanager/13920-call-routing.html#bcr>

NEW QUESTION: 8

An engineer is troubleshooting a Primary Rate Interface that fails to come up on a H.323 production voice gateway. Layer 1 and Layer 2 configurations were validated successfully Which Cisco IOS command must be run to debug the issue with the least impact on network performance?

- A. debug vtsp session
- B. debug all
- C. debug isdn q931
- D. debug ccsip messages

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

NEW QUESTION: 9

Refer to the exhibit.

The SDP offer/answer has been completed successfully but there is no DTMF when users press keys. What is the cause of the issue?

- A. DTMF was not negotiated on the call.
- B. DTMF was negotiated properly in these messages.
- C. G.729 rather than G.711ulaw was negotiated.
- D. Payload type 110 was negotiated rather than type 101.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 10

Which Cisco IM and Presence service handles failover and state changes in the cluster?

- A. Cisco XCP Connection Manager
- B. XCP router
- C. XCP Sync Agent
- D. Cisco Server Recovery Manager

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 11

Regarding SIP integrations with Cisco Unified Communications Manager, if the Cisco Unity Connection is configured to listen for incoming IPv4 and IPv6 traffic, how should the addressing mode be set up in the Cisco Unity Connection?

- A. Set up is not required.
- B. Set up for each group to use IPv4 and IPv6.
- C. Set up media ports for each port group to use IPv4.
- D. Set up IPv4 and IPv6 in Cisco Unified CM.

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

Section: Collaboration Applications

NEW QUESTION: 12

Refer to the exhibit. An engineer configured NTP on Cisco UCM and runs the `utils ntp status` command from Cisco UCM CLI and receives an error message. How is this issue mitigated and fixed?

- A. Ensure that Cisco UCM can communicate with the NTP server on port 53 and have a Windows NTP server configured with NTPv3.
- B. Use an external NTP server and allow communication on port 53.
- C. Configure the NTP server to answer requests on port 443.
- D. Ensure that Cisco UCM can communicate with the NTP server on port 123 and have a Linux NTP server configured with NTPv4.

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

NEW QUESTION: 13

In the cisco expressway solution, which two features does mobile and Remote access provide? (Choose two)

- A. the ability to register third-party SIP or H.323 devices on Cisco UCM without requiring VPN
- B. VPN-based enterprise access for a subset of Cisco Unified IP Phone models
- C. the ability for remote users and their devices to access and consume enterprise collaboration applications and services
- D. Secure reverse proxy firewall traversal connectivity
- E. the ability of Cisco IP Phones to access the enterprise through VPN connection

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 14

On a Cisco Catalyst Switch, which command is required to send CDP packets on a switch port that configures a Cisco IP phone to transmit voice traffic in 802.1Q frames, tagged with the voice VLAN ID 221?

- A. `Device(config-if)# switchport trunk allowed vlan 221`
- B. `Device(config-if)# switchport vlan voice 221`
- C. `Device(config-if)# switchport voice vlan 221`
- D. `Device(config-if)# switchport access vlan 221`

Answer: (SHOW ANSWER)

Section: Infrastructure and Design

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/switches/lan/catalyst2960/software/release/12-2_40_se/configuration/guide/scg/swvoip.pdf

NEW QUESTION: 15

An administrator needs to stop the leading 9 from outbound calls on an IOS Voice Gateway and ensure that the system handles 911 emergency calls.

Which configuration is needed to accomplish this task?

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

Answer: D (LEAVE A REPLY)

The first rule will match 9911 and send it as 911. If that rule is not matched, rule 2 will match "9anything" and send it without the leading 9.

NEW QUESTION: 16

A SIP phone has been configured in the system with MAC address 0030 96D2 D5CB The phone retrieves the configuration file from the Cisco UCM. Which naming format is the file that is downloaded?

- A. SEP0030406706705.cnf
- B. SIP0030406706705.cnf
- C. SEP003096D2D5CB.cnf.xml
- D. SIPOO3096D2D5CB.cnf.xml

Answer: D (LEAVE A REPLY)

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NEW QUESTION: 17

Refer to the exhibit. An administrator accesses the terminal of a Cisco UCM and starts a packet capture.

Which two commands must the administrator use on Cisco UCM to generate DNS traffic?
(Choose two.)

- A. show version active
- B. utils ntp status
- C. show cdp neighbor
- D. utils diagnose module validate Network
- E. utils diagnose test

Answer: D,E ([LEAVE A REPLY](#))

NEW QUESTION: 18

You are adding regions in Cisco Unified Communications Manager. Which codec(s) are selected when a call is placed if you set up the max audio bit rate to use 8 kbps?

- A. G.722
- B. G.711ulaw and G.711alaw
- C. G.729 and G.711ualw
- D. G.729

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

NEW QUESTION: 19

Refer to the exhibit.

An engineer is troubleshooting a codec negotiation issue where both endpoints that are involved in the call support the codecs listed in the exhibit. Which audio codec is selected if a call between two endpoints in Region 1 is placed?

- A. G.711u
- B. G.722
- C. G.729
- D. G.711a

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 20

An engineer must configure switch port 5/1 to send CDP packets to configure an attached Cisco IP phone to trust tagged traffic on its access port.

Which command is required to complete the configuration?

Router# configure terminal

Router(config)# interface gigabitethernet 5/1

Router(config-if)# description Cube E42.421-5965

- A. platform qos trust extend cos 5
- B. platform qos trust extend
- C. platform qos extend trust
- D. platform qos trust extend cos 3

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

NEW QUESTION: 21

Refer to the exhibit. An engineer configures ISDN on a voice gateway. The provider confirms that the PRI is configured with 10 channels the engineer ordered and is working from the provider side, but the engineer cannot get a B-channel to carry voice. The rest of the configuration for the serial interface and voice network is functioning correctly. Which actions must be taken to carry voice?

- A. The engineer must activate the controller card on the voice gateway before configuring the device.
- B. The engineer must manually revert the order of using the channels.
- C. The engineer used a T1 interface but must use an E1 interface.
- D. The pri-group timeslots command must be 0-9 for the 10 channels because all values on a router start with 0.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 22

A customer has Cisco Unity Connections that is integrated with LDAP. As a Unity Connection administrator, you have received a request to change the first name for VM user. Where must the change be performed?

- A. Cisco Unity Connection
- B. Cisco Unified Communications Manager end user
- C. Cisco IM and Presence
- D. Active Directory

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

NEW QUESTION: 23

Refer to the exhibit.

Refer to the exhibit. A collaboration engineer is configuring the Cisco UCM IM and Presence Service. Which two steps complete the configuration of the SIP trunk security profile? (Choose two.)

- A. Check the box to accept replaces header.
- B. Check the box to allow charging header.
- C. Check the box to enable application-level authorization.
- D. Check the box to accept unsolicited notification.
- E. Check the box to transmit security status.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 24

An employee of company ABC just quit. The IT administrator deleted the employee's userid from the active directory at 10 a.m. on March 4th. The nightly sync occurs at 10 p.m. daily. The IT

administrator wants to troubleshoot and find a way to delete the userid as soon as possible How is this issue resolved?

- A. Wait until 3:15 a.m. on March 5th for garbage collection to remove the user from Cisco UCM
- B. Wait until 3:15 a.m. on March 6th for garbage collection to remove the user from Cisco UCM.
- C. Wait until 10 pm on March 5th when the user is automatically removed from Cisco UCM.
- D. Wait until 10 pm on March 4th when the user is automatically removed from Cisco UCM

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

NEW QUESTION: 25

Refer to the exhibit.

What is a possible cause of the PRI issue?

- A. The framing is configured incorrectly.
- B. The cable is unplugged.
- C. The controller shut down.
- D. The clock source is incorrect.

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

NEW QUESTION: 26

What is the difference between Cisco Unified Border Element and a conventional Session Border Controller?

- A. DTMF interworking
- B. Address hiding
- C. Voice policy
- D. SIP security

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

NEW QUESTION: 27

An engineer must configure codec on a Cisco Unified Border Element to prefer the G.711 ulaw and use G.711 codec as the next The engineer logs In to the CUBE, enters the dial-peer configuration level, and runs the voice-class codec 100 command. Which set of commands completes the configuration?

- A. voice class codec 100 codec g711alaw preference 1 codec a7Hulaw preference 2
- B. voice class codec 11j codec <?7iulaw preferred codec g7iialaw
- C. vice class codec 100
codec preference 1 g7llulaw
codec preference 2 o711alaw
- D. voice class codec ::: codec g711ulaw g711alaw

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

The following commands are used to configure the codec on a Cisco Unified Border Element to prefer the G.

711 ulaw and use G.711 alaw as the next codec:

Code snippet

```
voice class codec 100
```

```
codec preference 1 g711ulaw
```

```
codec preference 2 g711alaw
```

The voice class codec 100 command creates a new voice class with the ID of 100. The codec preference 1 g711ulaw command sets the preference for the G.711 ulaw codec to 1. The codec preference 2 g711alaw command sets the preference for the G.711 alaw codec to 2.

NEW QUESTION: 28

A Cisco IP Phone 7841 that is registered to a Cisco Unified Communications Manager with default configuration receives a call setup message.

Which codec is negotiated when the SDP offer includes this line of text?

```
M=audio 498181 RTP/AVP 0 8 97
```

A. G.711ulaw

B. iLBC

C. G.711alaw

D. G.722

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

Those are the codecs.

They are RTP payload types, and in preference order.

Eg 8=g711alaw, 0=g711ulaw, 18=g729, 97=this one is special, it's actually what they want to Mark RTP-NTP (RFC2833) DTMF relay as.

Reference:

<https://community.cisco.com/t5/ip-telephony-and-phones/sip-sdp-no-codec-specified/td-p/3913990>

https://en.m.wikipedia.org/wiki/RTP_payload_formats

NEW QUESTION: 29

Given this H.323 gateway configuration and using Cisco best practices, how must the called party transformation pattern be configured to ensure that a proper ISDN type of number is set?

A.

B.

C.

D.

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

NEW QUESTION: 30

Refer to the exhibit. A collaboration engineer needs to replace the original, single NTP server that was configured during the initial install of a Cisco UCM server.

What is the first step to accomplish this task?

A. Stop the NTP service on Cisco UCM.

B. Delete the original NTP server from Cisco UCM

- C. Enable NTP authentication for the new NTP server on Cisco UCM.
- D. Restart the NTP service on Cisco UCM.

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

NEW QUESTION: 31

An administrator configures international calling on a Cisco UCM cluster and wants to minimize the number of route patterns that are needed. Which route pattern enables the administrator to match variable-length numbers?

- A. 9.011@
- B. 9.011#
- C. 9.011*
- D. 9.011!

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab12/collab12/dialplan.html

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NEW QUESTION: 32

There is a saturated link that has traffic shaping configured. How is incoming traffic processed?

- A. Traffic is compressed so that the traffic fits within the bandwidth of the link.
- B. Excess traffic is queued for later transmission.
- C. Excess traffic is dropped.
- D. Excess traffic is queued, and then dropped after the timer expires.

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

NEW QUESTION: 33

A company hosts a conference call with no local users.

How does the administrator stop the conference from continuing?

- A. changes the codecs that are supported on the conference resource
- B. modifies the Drop Ad Hoc Conference service parameter
- C. modifies the Block OffNet to OffNet Transfer service parameter
- D. removes the transcoder

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

NEW QUESTION: 34

Which two configuration elements are part of the Cisco Unified Communications Manager toll-fraud prevention? (Choose two.)

- A. SIP trunk security profile.
- B. Partition
- C. Calling search space
- D. SUBSCRIBE calling search space
- E. Feature control policy

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

<https://www.ciscopress.com/articles/article.asp?p=2218297&seqNum=10>

NEW QUESTION: 35

An engineer implements QoS in the enterprise network. Which command can to verify the correct classification and marking on a cisco IOS switch?

- A. show class-map interface GigabitEthernet 1/0/1
- B. show policy-map interface GigabitEthernet 1/0/1
- C. show access-lists
- D. show policy-map

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

NEW QUESTION: 36

When a remote office location is set up with limited bandwidth resources, which codec carries the most voice calls?

- A. G.711
- B. G.722
- C. G.723
- D. G.729

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

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab10/collab10/media.html

NEW QUESTION: 37

Refer to the exhibit.

You deploy Mobile and Remote Access for Jabber and discover that Jabber for Windows does not register to cisco Unified Communications Manager while outside of the office. What is a cause of this issue?

- A. The DNS record should be changed from _collab-edge_tcp example.com.
- B. Server 4.2.2.2 is not a valid DNS server.
- C. The DNS record should be created for _cisco-uds_tcp example.com.
- D. The DNS record type should be changed from SRV to A.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 38

Which two access layer switches provide support to provide high-quality voice and take advantage of the full voice feature set. To provide high-quality voice and take advantage of the full voice feature set, which two access layer switches provide support? Choose two

- A.** Implement IP RTP header compression on the serial interface to reduce the bandwidth required per voice call on point-to-point links.
- B.** Deploy RSVP to improve VoIP QoS only where it can have a positive impact on quality and functionality where there is limited bandwidth and frequent network congestion.
- C.** Use 802.1Q trunking and 802.1p for proper treatment of Layer 2 CoS packet marking on ports with phones connected.
- D.** Map audio and video streams of video calls (AF41 and AF42) to a class-based queue with weighted random early detection.
- E.** Use multiple egress queues to provide priority queuing of RTP voice packet streams and the ability to classify or reclassify traffic and establish a network trust boundary.

Answer: C,E ([LEAVE A REPLY](#))

NEW QUESTION: 39

Refer to the exhibit.

A call to an international number has failed. Which action corrects this problem?

- A.** Add the bearer-cap speech command to the voice port.
- B.** Assign a transcoder to the MRGL of the gateway.
- C.** Strip the leading 011 from the called party number
- D.** Add the isdn switch-type primart-dms100 command to the serial interface.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 40

What dialled numbers match this Cisco UCM route pattern?

1[23]XX

- A.** 1230 through 1239 only
- B.** 1200 through 1300 only
- C.** 1200 through 1399 only
- D.** 12300 through 12399 only

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 41

According to the QoS Baseline Model, drag and drop the applications from the left onto the correct per-Hop Behaviour values on the right.

Answer:

NEW QUESTION: 42

Refer to the exhibit. Cisco Unified Border Element is attempting to establish a call with Subscriber 1, but the call fails. Cisco Unified Border Element then retries the same call with Subscriber 2, and the call proceeds normally. Which action resolves the issue?

- A. Verify that the Run On All Active Unified CM Nodes checkbox is enabled.
- B. Verify that the correct calling search space is selected for the Inbound Calls section.
- C. Verify that the Significant Digits field for Inbound Calls is set to All.
- D. Verify that the PSTN Access checkbox is enabled.

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

Run On All Unified CM Nodes allows Subscribers NOT configured within the Call Manager Group that is applied to Route Lists and/or SIP Trunks to process calls.

There is a whole rabbit whole with how this feature works when only applied to one or the other but per best practice, unless there is a VERY specific reason, you want this check box enabled on ALL Route Lists and SIP Trunks.

NEW QUESTION: 43

What are two key features of the Expressway series? (Choose two.)

- A. VPN connect toward the internal UC resources
- B. B2B calls
- C. IP to PSTN call connectivity
- D. Device registration over the internet
- E. SIP header modification

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

<https://www.cisco.com/c/en/us/products/collateral/unified-communications/expressway-series/datasheet-c78-737605.html>

NEW QUESTION: 44

Refer to the exhibit. When a UC Administrator is troubleshooting DTMF negotiated by this SIP INVITE, which two messages should be examined next to further troubleshoot the issue? (Choose two.)

- A. NOTIFY
- B. SUBSCRIBE
- C. PRACK
- D. REGISTER
- E. UPDATE

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

NEW QUESTION: 45

A remote office has a less-than-optimal WAN connection and experiences packet loss, delay, and jitter. Which VoIP codec should be used in this situation?

- A. G.722.1
- B. iLBC

C. G.711alaw

D. G.729A

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

NEW QUESTION: 46

Refer to the exhibit. A call from +1 613 555 1234 that is sent out through the Nice Gateways should result in a calling party of 001 613 555 1234 with the numbering type "international". Which configuration accomplishes this goal?

A. \+.! pre-dot 00 international

B. \+ 0011 pre-dot 1 international

C. \+1 ! none pre-dot 001 international

D. 613XXXXXXX none +011 international

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab12/collab12/dialplan.html

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NEW QUESTION: 47

Callers from a branch report getting busy tones intermittently when trying to reach colleagues in other office branches during peak hours An engineer collects Cisco CallManager service traces to examine the situation. The traces show:

What should be fixed to resolve the issue?

A. codec configuration

B. region configuration

C. geolocation configuration

D. class of service configuration

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

NEW QUESTION: 48

Refer to the exhibit. An engineer is integrating Cisco UCM with Cisco Unity Connection via SIP. Which two boxes should be checked to complete the configuration of the SIP trunk security profile? (Choose two.)

A. to accept replaces header

B. to enable application-level authorization

- C. to enable digest authentication
- D. to accept out-of-dialog refer
- E. to accept presence subscription

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

NEW QUESTION: 49

Multiple route patterns match a number. How does Cisco Unified Communications Manager determine which pattern to use?

- A. the one with the closed match
- B. the one that comes first in numerical order
- C. the one with the longest match
- D. the one that discards everything PreDot

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 50

A company deploys centralized cisco ucm architecture for a hub location and two remote sites.

* The company has only one ITSP connection at the hub connection, and ITSP supports only G.711 calls

* Remote site A has a 1-Gbps fiber connection to the hub connection and calls to and from remote side A use G.711 codec

* Remote site B has a 1 T1 connection to the hub location and calls to and from remote site B use G.729 codec Based on the provided guidance, a Cisco voice engineer must design media resource management for the customer. What is the method that needs to be followed?

- A. configure the hardware transcoder on the hub location router.
- B. configure the hardware transcoder on the site A router.
- C. configure the hardware transcoder on the site B router.
- D. configure the software transcoder on Cisco UCM to support voice calls to and from both remote sites.

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

NEW QUESTION: 51

Refer to the exhibit. An engineer is troubleshooting this video conference issue:

A video call between a Cisco 9971 in Region1 and another Cisco 9971 in Region1 works.

As soon as the Cisco 9971 in Region1 conferences in a Cisco 8945 in Region2, the Region1 endpoint cannot see the Region2 endpoint video.

What is the cause of this issue?

- A. Maximum Session Bit Rate for Immersive Video Calls is too low.
- B. Maximum Audio Bit Rate must be increased.
- C. Cisco 8945 does not have a camera connected.
- D. Maximum Session Bit Rate for Video Calls is too low.

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

NEW QUESTION: 52

An incoming off-net call to a user fails. An engineer notices that the off-net call is G.711, but the phone accepts only G.729. which media resource on a Cisco Unified Border Element and Cisco Unified Communications Manager must the engineer configure to Manage the engineer configure to manage the codec negotiation?

- A. MOH
- B. CFB
- C. transcoder
- D. MTP

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

NEW QUESTION: 53

A user reports that when receiving an inbound call on their IP Phone from the PSTN they are unable to transfer this call to another PSTN number What is the reason for this failure?

- A. The IP phone is configured with the wrong region
- B. The gateway is configured with the wrong device pool
- C. The incoming calling search space of the SIP trunk does not include the partition of the line in the IP phone
- D. The service parameter related to Offnet to Offnet Call Transfer is set to TRUE

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 54

An engineer is setting up a system with voice and video endpoints using auto-QoS policy on the switches. Which DSCP values are expected for interactive voice and video?

- A. EF AND CS3
- B. EF AND AF21
- C. EF AND CS6
- D. EF AND AF41

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

NEW QUESTION: 55

An engineer must configure a DHCP server to save automatic bindings on a remote host that has an IP address of 172.16.1.1. The IP address range of 172.16.1.100 to 172.16.1.108 must be blocked from the DHCP server. Which set of commands must the engineer run?

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

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 56

What is the major difference between the two possible Cisco IM and presence high-availability modes?

A. Balanced mode provides user load balancing and user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

B. Balanced mode provides user load balancing and user failover only for manually generated failovers.

Active/standby mode provides an unconfigured standby node in the event of an outage, but it does not provide load balancing.

C. Balanced mode provides user load failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, and is also provides load balancing.

D. Balanced mode does not provide user load balancing, but it provides user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, but is does not provide load balancing.

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

NEW QUESTION: 57

Refer to the exhibit.

Refer to the exhibit. Calls to Cisco Unity Connection are failing across Cisco Unified Border Element when callers try to select a menu prompt Why is this happening and how is it fixed?

A. Cisco Unity Connection is configured on G.729 only. Cisco Unity Connection must be reconfigured to support iLBC.

B. Cisco Unified Border Element is not sending any support for DTMF. OTMF configuration must be added to the appropriate dial peer.

C. The Cisco Unity Connection Call Handler is configured for a "Release to Switch" transfer type Transfers must be disabled for the Cisco Unity Connection Call Handler

D. Cisco Unified Border Element is sending the incorrect media IP address. The IP address of the loopback interface must be advertised in the SDP

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

NEW QUESTION: 58

An administrator troubleshoots call flows and suspects that there are issues with the dial plan.

Which tool enables a quick analysis of the dial plan and provides call flows of dialled digits?

A. Digit Analysis Analyzer

B. Cisco Dial Plan Analyzer

C. Dialed Number Analyzer

D. Dial Plan Analyzer

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

NEW QUESTION: 59

Refer to the exhibit. An administrator accesses the terminal of a Cisco UCM and starts a packet capture. Which two commands must the administrator use on Cisco UCM to generate DNS traffic? (Choose two.)

- A. show cdp neighbor
- B. utils ntp status
- C. show version active
- D. utils diagnose module validate_network
- E. utils diagnose lest

Answer: D,E ([LEAVE A REPLY](#))

NEW QUESTION: 60

Which two steps are required for bulk configuration transactions on the Cisco UCM database utilizing BAT? (Choose two.)

- A. A data file in Abstract Syntax Notation One format must be uploaded to Cisco UCM
- B. A server template must be created in Cisco UCM
- C. A data file in comma-separated values format must be uploaded to Cisco UCM
- D. A data file in Extensible Markup Language format must be uploaded to Cisco UCM
- E. A device template must be created in Cisco UCM

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 61

Refer to the exhibit. In this cisco UCM setup configured for EARLY offer, what is the codec preference line in the initial SIP INVITE SDP?

- A. m=audio 16444 RTP/AVP 9 18 101
- B. m=audio 16444 RTP/AVP 0 8 18 101
- C. m=audio 16444 RTP/AVP 9 0 8 18 101
- D. m=audio 16444 RTP/AVP 18 101

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

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NEW QUESTION: 62

An administrator would like to set several Cisco Jabber configuration parameters to only apply to mobile clients (iOS and Android). How does the administrator accomplish this with Cisco Jabber 12.9 and Cisco UCM 12.5?

- A. Create a user profile in Jabber Policies.
- B. Upload the jabber-config.enc file to TFTP
- C. Assign the desired configuration file to "Mobile" Jabber Client Configuration in the Service Profile.
- D. Deploy jabber-config-user.xml on iOS and Android devices.

Answer: [\(SHOW ANSWER\)](#)

NEW QUESTION: 63

What is the maximum one-way latency and jitter for voice before quality issues arise?

- A. ≤ 150 ms Latency and ≤ 30 ms jitter
- B. ≤ 300 ms Latency and ≤ 60 ms jitter
- C. ≤ 100 ms Latency and ≤ 10 ms jitter
- D. ≤ 90 ms Latency and ≤ 30 ms jitter

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

NEW QUESTION: 64

Refer to the exhibit.

What is the result of applying these two rules to a voice translation-profile for use with an ISDN PRI on a Cisco Voice gateway?

- A. Any Zero is stripped from the numeric phone number.
- B. The ISDN Plan is modified to the administrator's defined value.
- C. The ISDN Type is modified to the administrator's defined value.
- D. The leading Plus is stripped from the numeric phone number.

Answer: C [\(LEAVE A REPLY\)](#)

NEW QUESTION: 65

What is the purpose of a hybrid Local Gateway?

- A. to handle calls between Webex Calling and Cisco Calling Plans
- B. to handle calls between Webex Calling and Cloud Connected PSTN
- C. to handle calls between Cisco IJCM and Webex Calling
- D. to handle calls between the Public Switched Telephone Network and Webex Calling

Answer: D [\(LEAVE A REPLY\)](#)

A hybrid local gateway handles calls between the Public Switched Telephone Network (PSTN) and Webex Calling. It is commonly deployed on the customer's premises but can also be hosted by a partner. The local gateway registers with Webex Calling and handles all calls between the PSTN and Webex Calling. It gives customers the flexibility to bring their own service provider or continue using their existing provider for a smooth and effective transition to the cloud.

NEW QUESTION: 66

What are two functions of a dial-plan? (Choose two.)

- A. endpoint features
- B. ring cadence
- C. digit manipulation
- D. calling privileges
- E. call recording

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 67

Where is the default for maximum session Bit Rate for a region configured?

- A. Enterprise parameters configuration
- B. Region configuration
- C. Enterprise Phone configuration
- D. Service parameter configuration

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

NEW QUESTION: 68

An administrator configures the voicemail feature in a Cisco collaboration deployment. The user mailboxes must be configured when the Cisco Unity Connection server is configured. Which action accomplishes this task?

- A. Configure an SCCP integration with Cisco UCM.
- B. Configure an active directory to sync the users who will have a voicemail box.
- C. Configure an AXL server to access the Cisco UCM users.
- D. Configure a SIP integration with Cisco UCM to sync users.

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

NEW QUESTION: 69

What is a QoS requirement for VoIP calls?

- A. 150 bps per phone of guaranteed bandwidth
- B. 150 ms of one-way latency from mouth to ear
- C. 15% packet loss
- D. 150 ms of jitter

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 70

What are two functions of Cisco Expressway in the Collaboration Edge? (Choose two)

- A. Expressway-E provides a perimeter network that separates the enterprise network from the Internet.
- B. The Expressway-C and Expressway-E pair can enable connectivity from the corporate network to the PSTN via a T1/E1 trunk.

- C. Expressway-E provides a VPN entry point for Cisco IP phones with a Cisco AnyConnect client using authentication based on certificates.
- D. The Expressway-C and Expressway-E pair can interconnect H 323-to-SIP calls for voice.
- E. Expressway-C provides encryption for Mobile and Remote Access but not for business-to-business communications.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 71

When a new SIP phone is registered to Cisco Unified communications Manager, it keeps failing and showing an "unprovisioned" error message in the phone display. Which problem is a possible cause of this issue?

- A. The DHCP settings are incorrectly and the phone does not have an alternate TFTP defined.
- B. Auto-registration is disabled on the Cisco Unified Communications Manager nodes and the phone device does not have a DN configured.
- C. The DN configuration for this phone is shared with SCCP phone, which is not supported.
- D. The DN assigned to the phone is already in use by another SIP phone.
- E. The phone cannot download and install the latest firmware.

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

NEW QUESTION: 72

The chief officer at a company must reduce collaboration infrastructure costs by onboarding all on-premises equipment to the cloud by using CISCO Webex Control Hub. Administrators need the ability to manage upgrades and set up hot desking for on-premises devices.

Which action must be taken before on boarding devices by using the Control Hub?

- A. Configure tie Control Hub organization ID on the devices
- B. Acquire a license for each device.
- C. Allow HTTP traffic from each device to Control Hub.
- D. Upgrade all the devices to software version CE9.15 or later

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

This is a prerequisite for using the Device Connector tool, which allows you to onboard and register several devices simultaneously to the Webex Control Hub¹. The Device Connector tool creates a workspace, an activation code, and activates all of your devices in one go¹. This way you don't need to be physically present in the same room to activate the devices.

The other options are not required before onboarding devices by using the Control Hub:

* Configuring the Control Hub organization ID on the devices is not necessary, as the Device Connector tool will send the device information to your Webex organization and generate activation codes for them

1.

* Acquiring a license for each device is not necessary, as you can assign licenses to users and devices after they are registered to the Webex Control Hub².

* Allowing HTTP traffic from each device to Control Hub is not necessary, as HTTPS connectivity is required for the Device Connector tool to communicate with the devices1.

NEW QUESTION: 73

Which DSCP marking is represented as 101110 in an IP header?

- A. CS3
- B. AF31
- C. AF41
- D. EF

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

NEW QUESTION: 74

What is a software-based media resource that is provided by the Cisco IP Voice Streaming Application?

- A. Video conference bridge
- B. transcoder
- C. annunciator
- D. auto-attendant

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

An annunciator is a software function of the Cisco IP Voice Media Streaming Application that provides the ability to stream spoken messages or various call progress tones from the system to a user.

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab11/collab11/media.html#32774

NEW QUESTION: 75

An engineer must configure a SIP integration between cisco UCM and cisco Unity Connection. These components are configured already:

Drag and drop the components from the left into the order they must be configured on the right.

Answer:

Explanation:

A screenshot of a computer program AI-generated content may be incorrect.

Summary of Steps:

Create a SIP Trunk Security Profile.

Create a SIP Profile.

Create the SIP Trunk in Cisco UCM.

Configure Route Patterns for routing calls to Unity Connection.

Configure SIP ports in Cisco Unity Connection.

Set the Voicemail Pilot Number in Cisco UCM.

Configure the Voicemail Ports in Cisco UCM.

Assign Voicemail Profile to the appropriate users.

NEW QUESTION: 76

What are two characteristics of priority queueing? (Choose two.)

- A. It guarantees bandwidth for user-defined traffic classes.
- B. It applies strict priority to user-defined traffic classes.
- C. It allocates an equal share of bandwidth to each flow.
- D. It places traffic into one of four queues.
- E. It ensures that critical traffic queues are serviced first.

Answer: D,E (LEAVE A REPLY)

https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/qos_conmgt/configuration/xe-16-5/qos-conmgt-xe-16-5-book/qos-conmgt-overview.html#:~:text=to%2064%20classes,-4%20queues,-Kind%20of%20Service

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NEW QUESTION: 77

Which datastore and protocol is used for saving back-up files within the Disaster Recovery System of Cisco UCM?

- A. local disk on the Cisco UCM server
- B. remote disk on the SFTP server
- C. remote disk on a CIFS share
- D. remote disk on an NFS share

Answer: B (LEAVE A REPLY)

Section: Infrastructure and Design

Explanation/Reference: <https://www.learnCisco.net/courses/cucm-basics/serviceability-backup-and-restore/backing-up-and-restoring-cucm.html>

NEW QUESTION: 78

Refer to the exhibit.

What causes the PRI issue?

- A. The controller shut down
- B. The cable is unplugged
- C. The framing is configured incorrectly
- D. The clock source is incorrect.

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

The show controller t1 command shows that the T1 interface is up but the line protocol is down. This indicates that the physical layer is working but the data link layer is not. The most likely cause of this is that the cable is unplugged.

NEW QUESTION: 79

Which action prevent toll fraud in Cisco Unified Communication Manager?

- A. Configure ad hoc conference restriction
- B. Allow off-net to off-net transfer
- C. Implement route patterns in Cisco Unified CM
- D. Implement toll fraud restriction in the Cisco IOS router

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

NEW QUESTION: 80

What is a capability of the call forwarding feature in a Cisco Webex dial plan?

- A. It provides the ability to forward a Webex meeting to another user.
- B. It forwards calls when a line is busy by enabling the call waiting feature.
- C. It forwards calls to the voicemail of a mobile phone when there is no answer.
- D. It chooses a tone to be played when forwarding all calls.

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

NEW QUESTION: 81

Location-based call admission control is used in Cisco UCM to control the amount of bandwidth allowed between predefined locations. What must be considered when calculating the number of calls that can be made?

- A. A limit of 900 kb is the total amount of bandwidth used for the sum of calls of the defined codec including headers and media traffic
- B. A limit of 900 kb is the total amount of bandwidth used for the sum of the media and signaling in the calls The system allows for an actual of 20% overhead of IP packet headers This is to compare the use of ISDN lines with no IP overhead.
- C. A limit of 900 kb is the total amount of actual used RTP traffic so the oversubscription can be done for multiple calls adding its maximum RTP stream requirements.
- D. A limit of 900 kb is the total amount of bandwidth used for the sum of calls. The use of compression does not impact the amount of calls because only the codec definition is considered when calculating the CAC in Cisco UCM

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

NEW QUESTION: 82

An engineer wants to manually deploy a Cisco Webex DX80 video endpoint to an end user. Which type of provisioning can be configured on the endpoint?

- A. CMS

- B. CUC
- C. CUBE
- D. Edge

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 83

Cisco Prime Collaboration Manager is using Cisco Discovery Protocol to discover devices in an automated setup. What must be used to get alerts and faults via traps from devices at regular polling intervals?

- A. XML
- B. CDP
- C. SNMP
- D. HTTPS

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 84

Which application traffic does the DiffServ AF41 class according to the Cisco Collaboration System Solution Reference Network Design?

- A. signalling
- B. video call
- C. audio call
- D. messaging

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 85

Refer to the exhibit. An engineer is confining class of control for a user in Cisco UCM. Which change will ensure that the user is unable to call 2143?

- A. Set the users device CSS to <None>
- B. Change line CSS to only contain Partition_B
- C. Set the user's line CSS to <None>
- D. Change line partition to Partition_A

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

NEW QUESTION: 86

What is a valid class included in the 8-Class QoS Strategy in a VoIP network?

- A. Assured Forwarding
- B. Broadcast Video
- C. Multimedia Conferencing
- D. Real-Time Interactive

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

Explanation/Reference: <https://www.ciscopress.com/articles/article.asp?p=2756478&seqNum=8>

NEW QUESTION: 87

Refer to the exhibit. An engineer configures a VoIP dial peer on a Cisco gateway. Which codec is used?

- A. G.711ulaw
- B. No codec is used (missing codec command).
- C. G.711alaw
- D. G.729r8

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

<https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/dialpeer/configuration/15-mt/vd-15-mt-book/vd-dp-cfg-examp.pdf>

NEW QUESTION: 88

Refer to the exhibit.

When a UC Administrator is troubleshooting DTMF negotiated by this SIP INVITE, which two messages should be examined next to further troubleshoot the issue? (Choose two.)

- A. UPDATE
- B. SUBSCRIBE
- C. PRACK
- D. NOTIFY
- E. REGISTER

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 89

What is the cisco UCM service parameter default value for DSCP for Telepresence calls ?

- A. EF
- B. CS3
- C. CS4
- D. AF41

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

NEW QUESTION: 90

Which DTMF relay method configured on a SIP dial-peer will ensure that a media resource is not invoked by Unified CM for calls to UCCX IVRs?

- A. dtmf-relay slp-kpml
- B. dtmf-relay rtp-nte
- C. sdtmf-relay h245-signal
- D. dtmf-relay sip-notify

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

https://www.cisco.com/c/dam/en/us/td/docs/ios-xml/ios/voice/cube_uccx/cube_uccx_interop_bestprac_guide.pdf

NEW QUESTION: 91

Which DSCP value and PHB equivalent are the default for audio calls?

- A. 32 and CS4
- B. 48 and EF
- C. 34 and AF41
- D. 32 and AF41

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

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NEW QUESTION: 92

A Cisco IP Phone 7841 that is registered to a Cisco Unified Communications Manager with default configuration receives a call setup message.

Which codec is negotiated when the SDP offer includes this line of text?

M=audio 498181 RTP/AVP 0 8 97

- A. G.711alaw
- B. G.711ulaw
- C. iLBC
- D. G.722

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

NEW QUESTION: 93

What dialed number match this cisco UCM route pattern?

1[23]XX

- A. 1200 through 1300 only
- B. 12300 through 12399 only
- C. 1230 through 1239 only
- D. 1200 through 1399 only

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 94

A company has an excessive number of call transfers to local and long-distance PSTN from Cisco Unity Connection voicemail. Which action in the Cisco Unity Connection restriction table resolves this issue?

- A. Create a custom restriction table ***** and block it.
- B. Create a custom restriction table ?????????? and block it.
- C. Block PSTN patterns on Default Transfer, Default Outdial, and Default System Transfer.
- D. Implement password complexity on voicemail boxes to prevent accounts from being compromised.

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

NEW QUESTION: 95

A user dials 9011841234567 to reach Vietnam. Which steps send the call to the PSTN provider as

011841234567?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 96

Refer to the exhibit.

Refer to the exhibit. When making a call to a Mobile and Remote Access client, what are the combinations of protocol on each of the different sections A-B-C?

- A. IP TCP/TLS (A) + SIP TCP/TLS (B) + SIP TLS (C)
- B. SIP TCP/TLS (A) + SIP TCP/TLS (B) + SIP TCP/TLS (C)
- C. SIP TLS (A) + SIP TLS (B) + SIP TLS (C)
- D. SIP TCP/TLS (A) + SIP TLS (B) + SIP TLS (C)

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

NEW QUESTION: 97

Refer to the exhibit. A Cisco UCM user with directory number 4401 dials 5507, and the call is routed to a Cisco Unified Border Element. Which IP address will the call be sent to?

- A. 10.0.0.2
- B. 10.0.0.3
- C. 10.0.0.4
- D. 10.0.0.1

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

Section: Infrastructure and Design

NEW QUESTION: 98

A high-speed network is often configured with a five-class QoS model. Which classes are used in the model?

- A. voice, video, signaling critical data, and best-effort
- B. real-time, call-signaling, critical data, best-effort, and scavenger
- C. real-time, signaling, critical data, best-effort, and drop-class
- D. call-signaling, real-time, critical data, best-effort, and drop-class

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

https://www.cisco.com/en/US/technologies/tk543/tk759/technologies_white_paper0900aecd80295a9b.pdf

NEW QUESTION: 99

When configuring Cisco Unified Communications Manager, which configuration enables phones to automatically register to a Cisco Unified Communications publisher when the connection to the subscriber is lost?

- A. Route Group
- B. Cisco Unified CM Group
- C. SRST
- D. Device Pool

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 100

When a remote office location is set up with limited bandwidth resources, which codec would allow the most voice calls with the limited bandwidth?

- A. G.722
- B. G.711
- C. G.723
- D. G.729

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

Cisco recommends using G.729a as the low-bandwidth codec because it is supported on all Cisco Unified IP Phone models as well as most other Cisco Unified Communications devices, therefore it can eliminate the need for transcoding.

NEW QUESTION: 101

An administrator must implement toll fraud prevention on cisco UCM using these parameters:

- Enable Forced Authorization Code 112211.
- Set an authorization level of 3 for the route pattern 8005551212.
- Require no access code to dial 10-digit numbers.

How must the route pattern be implemented?

- A. Pattern = 3.800xxxxxx
- B. Pattern = 8005551212.1122113
- C. Pattern = 8005xxxxxx

D. Pattern = 1122113.8005551212

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

<https://www.cisco.com/c/en/us/support/docs/voice-unified-communications/unified-communications-manager-callmanager/81541-fac-config-ex.html>

NEW QUESTION: 102

In the cisco expressway solution, which two features does mobile and Remote access provide? (Choose two)

- A. secure reverse proxy firewall traversal connectivity
- B. the ability to register third-party SIP or H 323 devices on Cisco UCM without requiring VPN
- C. the ability of Cisco IP Phones to access the enterprise through VPN connection
- D. the ability for remote users and their devices to access and consume enterprise collaboration applications and services
- E. VPN-based enterprise access for a subset of Cisco Unified IP Phone models

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 103

As a voice engineer, which two recommendations do you make to your company to optimize Cisco Unified Communications Manager configuration to reduce the number of toll fraud incidents? (Choose two.)

- A. Classify all route patterns as on-net and prohibit on-net to on-net call transfers in Cisco Unified CM service parameters.
- B. Inbound CSS on any gateway typically should have access to internal destinations and PSTN destinations.
- C. Classify all route pattern as off-net and prohibit off-net to off-net call transfers in Cisco Unified CM service parameters.
- D. Classify all route patterns as on-net or off-net and prohibit off-net call transfers in Cisco Unified CM Service parameters.
- E. Inbounds CSS on any gateway typically should have access to internal destinations only and not PSTN destinations.

Answer: C,E ([LEAVE A REPLY](#))

NEW QUESTION: 104

What are the last two bits of a DS field in DiffServe Byte used for?

- A. AFxy
- B. ECN
- C. INC
- D. RMI

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

Reference:

https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/qos_dfsrv/configuration/15-mt/qos-dfsrv-15-mt-book/qos-dfsrv.html

NEW QUESTION: 105

Which SNMP service must be active manually after Cisco UCM is installed?

- A. Connection SNMP Agent
- B. Host Resources Agent
- C. Cisco CallManager SNMP
- D. SNMP Master Agent

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 106

An engineer must configure a route pattern that can route all +E.164 globalized international numbers for the plans of all countries. Which Cisco UCM configuration accomplish this task?

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

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

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<https://www.trainingdump.com/Cisco/350-801-practice-exam-dumps.html> (364 Q&As Dumps, **40%OFF Special Discount: Exam-Tests**)

NEW QUESTION: 107

Refer to the exhibit. Which two codec permutations should be transcoded by this dspfarm? (Choose two)

- A. G.729br8 to G.711alaw
- B. iLBC to G.711ulaw
- C. G.729ar8 to G.711alaw
- D. G.729r8 to G.711ulaw
- E. G.722 to G.729r8

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 108

An engineer is integrating Unity Connection with Cisco UCM. Which two actions must be configured so that recording and playback from the IP phones works at all times, including peak traffic hours? (Choose two.)

- A. Increase the number of voice ports.
- B. If it's a Unity Connection Cluster, ensure that replication is fine and not in split-brain mode.
- C. The phone system to which the phones are registered in Unity Connection has the Default Trap Switch check box enabled.
- D. Add dedicated dial-out ports with the allow trap connections setting selected.
- E. Ensure that you have set up SIP Digest Authentication on the SIP trunk security profile.

Answer: ([SHOW ANSWER](#))

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/14/integration/cucme_sccp/b_14cucintcucmesccp.html

NEW QUESTION: 109

A collaboration engineer must optimize the dial plan within Cisco UCM. There are multiple remote sites. And each site has its own route patterns and local gateways. What should the engineer do on the cisco UCM to optimize the dial plan?

- A. Implement ILS with GDPR so the dial plan can dynamically replicate across clusters.
- B. Create a centralized dial plan with a Cisco UCM Session Management Edition cluster or a Cisco gatekeeper.
- C. Configure a Standard Local Route Group to use a single route pattern for all calls within the cluster.
- D. Leverage Cisco UCM Express as SRST so the phones can have more features of each remote site

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

NEW QUESTION: 110

Which two steps should be taken to provision after the Self-Provisioning feature was configured for end users?

- A. Ask the Cisco UCM administrator to associate the phone to an end user.
- B. Dial the hunt pilot extension and associate the phone to an end user.
- C. Plug the phone into the network.
- D. Dial the self-provisioning IVR extension and associate the phone to an end user.
- E. Enter settings menu on the phone and press *,*,# (star, star, pound).

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 111

How many minutes does it take for automatic fallback to occur in a Presence Redundancy Group if the primary node lost a critical service?

- A. 5 min
- B. 10 min

C. 30 min

D. 60 min

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

Enable Automatic Fallback. This parameter specifies whether to do automatic fallback. In the event of a failover, the IM and Presence Service moves users automatically from the backup node to the primary node thirty minutes after the primary node returns to a healthy state.

Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/im_presence/configAdminGuide/11_5_1/cup0_b_config-and-admin-guide-1151su5/cup0_b_imp-system-configuration-1151su5_chapter_0100.html

NEW QUESTION: 112

A present redundancy group is deployed, and an engineer with ID012345678 initiates a manual fallback. Which statement about Cisco Server Recovery Manager is true?

A. disconnects all users that had been failed over

B. restarts the Cisco Presence Engine

C. restarts critical on the secondary node

D. disconnects all users that had been failed over, and the users must log in again.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 113

What is a function of a call handler in Cisco Unity Connection?

A. It collects information from callers by playing a series of questions and recording the answers.

B. It controls outgoing calls by allowing an administrator to specify the numbers that Cisco Unity Connection dials to transfer calls, notify users of messages, and deliver faxes

C. It answers calls, take messages, and provide menus of options.

D. It provides access to a corporate directory by playing an audio list that users and outside callers use to reach users and leave messages.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 114

Refer to the exhibit. An administrator configured a codec preference list with 0,122 and ILBC codecs. Which change must the administrator make in the dial-peer section of the configuration to use this list?

A. add codec preference 20

B. add voice-class codec 20

C. add session codec 20

D. add voice-codecs 20

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 115

Which command is used in a Cisco IOS XE TDM gateway to configure the voice T1/E1 controller to provide clocking?

- A. Clock source internal
- B. Clock source network
- C. Cisco IOS XE TDM gateway T1/E1 controller cannot provide clocking.
- D. Clock source line

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

NEW QUESTION: 116

Refer to the exhibit. A customer wants the SIP 200 OK shown to advertise codecs in the following order:

After correcting the codec preferences. What should the audio payload show in the SIP Traces?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 117

What is the traffic classification for voice and video conferencing?

- A. Voice is classified as CoS 5, and video conferencing is CoS 4.
- B. Video conferencing is classified as CoS 1, and voice is CoS 2.
- C. Voice and video conferencing are both classified is CoS 3.
- D. Voice is classified as CoS 4, and video conferencing is CoS 5.

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

NEW QUESTION: 118

Refer to the exhibit. An administrator accesses the terminal of a Cisco UCM and starts a packet capture. Which two commands must the administrator use on Cisco UCM to generate DNS traffic? (Choose two.)

- A. show version active
- B. utils diagnose lest
- C. utils diagnose module validate_network
- D. utils ntp status
- E. show cdp neighbor

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

NEW QUESTION: 119

What are two differences between media flow-around and media flow-through on cisco unified Border element? (Choose two.)

- A.** When using media flow-around, the call signaling goes through the Cisco Unified Border Element, but media is not passed through it
- B.** When using media flow-through, call signaling goes through the Cisco Unified Border Element but media does not
- C.** When using media flow-around, both call signaling and media do not go through the Cisco Unified Border Element
- D.** When using media flow-through, the call signaling goes through the Cisco Unified Border Element, but media is not passed through it.
- E.** When using media flow-through, the call signaling and media are passed through the Cisco Unified Border Element

Answer: A,E ([LEAVE A REPLY](#))

NEW QUESTION: 120

Drag and drop the SNMPv3 message types from the left onto the corresponding definitions on the right.

Answer:

NEW QUESTION: 121

Refer to the exhibit.

An engineer is troubleshooting a codec negotiation issue where both endpoints that are involved in the call support the codecs listed in the exhibit. Which audio codec is selected if a call between two endpoints in Region 1 is placed?

- A.** G.729
- B.** G.711u
- C.** G.722
- D.** G.711a

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

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NEW QUESTION: 122

A company has an excessive number of call transfers to local and long-distance PSTN from Cisco Unity Connection voicemail Which action in the Cisco Unity Connection restriction table resolves this issue?

- A. Implement password complexity on voicemail boxes to prevent accounts from being compromised
- B. Create a custom restriction table ***** and Wock it
- C. Create a custom restriction table ??????????? and block it
- D. Block PSTN patterns on Default Transfer. Default Outdial. and Default System Transfer

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 123

A Cisco UCM administrator sets up new route patterns to support phones in four different locations, all with local gateways. The administrator wants to use the same route pattern for all four locations. How must the system be configured to achieve this goal?

- A. Use standard local route groups.
- B. Use transforms in the route groups.
- C. Add a CSS to each local gateway.
- D. Use CSS alternate routing rules.

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

NEW QUESTION: 124

Refer to the exhibit. Unanswered calls do not reach the voicemail associated with the phones. Instead, callers receive the default greeting.

Which action fixes the configuration?

- A. Reboot Cisco Unity Connection
- B. Check the box "Redirecting Diversion Header Delivery - Outbound"
- C. Review the conversation manager logs on Cisco Unity Connection
- D. Check the box "Redirecting Diversion Header Delivery - Outbound", then reset the trunk

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 125

An administrator must make a pattern to route calls to two different destinations john.doe@company.com and jane.doe@company.com.

Which type of patterns are needed in the Cisco UCM, and what must the pattern look like?

- A. A regular route pattern with the URI feature enabled in the configuration page.

The pattern must look like this: MATCH(*.doe@company.com)

- B. A SIP route pattern that looks like this: company.com

- C. A regular route pattern with the URI feature enabled in the configuration page.

The pattern must look like this: (*@company.com)

- D. A SIP route pattern that looks like this: *@company.com

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 126

An engineer must configure a DHCP server to save automatic bindings on a remote host that has an IP address of 172.16.1.1.

The IP address range of 172.16.1.103 to 172.16.1.106 must be blocked from the DHCP server.

Which set of commands must the engineer run?

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

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

NEW QUESTION: 127

What are two reasons that AF41 is marked for the audio and video channels of a video call? (Choose two.)

- A. to prioritize video over other high -priority traffic classes
- B. to give video calls a higher priority than AP41 in the QoS policy
- C. to allow high-definition quality calls over low-speed links
- D. to preserve lip synchronization between the audio and video channels
- E. to provide separate classes for audio calls and video calls

Answer: ([SHOW ANSWER](#)**)**

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab10/collab10/netstruc.html

NEW QUESTION: 128

Refer to the exhibit. A collaboration engineer adds a redundant NTP server to an existing Cisco Collaboration solution. On the Cisco UCM OS Administration page, the new NTP server shows as "Not Accessible." Which action resolves this issue?

- A. Delete and re-add the new NTP server via the Cisco UCM command-line interface
- B. Configure the "reach" value as "377" for the new NTP server.
- C. Start the NTP service on the new NTP server
- D. Restart NTPD on the Cisco UCM server.

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

NEW QUESTION: 129

Refer to the exhibit.

Refer to the exhibit. A collaboration engineer adds a redundant NTP server to an existing Cisco Collaboration solution. On the Cisco UCM OS Administration page, the new NTP server shows as "Not Accessible." Which action resolves this issue?

- A. Configure the "reach" value as "377" for the new NTP server.
- B. Restart NTPD on the Cisco UCM server.
- C. Start the NTP service on the new NTP server
- D. Delete and re-add the new NTP server via the Cisco UCM command-line interface

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 130

An engineer configures Class-Based Weighted Fair Queuing, but must still configure Low Latency Queuing to prioritize voice audio traffic and guarantee 5 Mbps of bandwidth. Which Cisco IOS command must be run in the voice class map?

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

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

NEW QUESTION: 131

Which two conditions must a user meet to provision a new device using the self-provisioning feature?

(Choose two.)

- A. The user must have the appropriate universal device template linked to the linked to the user profile.
- B. The user must have a primary extension.
- C. At least DNs must be assigned to the user device.
- D. The user must be part of "standard CCM Super User".
- E. The user must have at least one user device profile assigned.

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

NEW QUESTION: 132

You are adding regions in Cisco Unified Communications Manager. Which codec(s) are selected when a call is placed if you set up the max audio bit rate to use 8 kbps?

- A. G.711ulaw and G.711alaw
- B. G.729 and G.711ulaw
- C. G.722
- D. G.729

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 133

A customer reports that the Cisco UCM toll-fraud prevention does not work correctly, and the customer is receiving charges for unapproved international calls as a result. Which two configuration changes resolve the issues? (Choose two.)

- A. Use Cisco Unified Border Element to debug the calls.
- B. Disable call forwarding on the phone.
- C. Make the calls route through a firewall.
- D. Mark patterns as off-net or on-net.
- E. Modify the Block OffNet to OffNet Transfer service parameter.

Answer: D,E ([LEAVE A REPLY](#))

NEW QUESTION: 134

Refer to the exhibit.

What is a possible cause of the PRI issue?

- A. The controller shut down.
- B. The clock source is incorrect.
- C. The framing is configured incorrectly.
- D. The cable is unplugged.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 135

A user forwards a corporate number to an international number. What are two methods to prevent this forwarded call? (Choose two.)

- A. Set the call Classification to OnNet for the international route pattern.
- B. Block international dial patterns in the SIP trunk CSS.
- C. Set call forward All CSS to restrict international dial patterns.
- D. Configure all forced code on all router.
- E. Configure a Forced Authorization Code on the international route pattern.

Answer: C,E ([LEAVE A REPLY](#))

NEW QUESTION: 136

According to the QoS Baseline Model, drag and drop the applications from the left onto the correct per-Hop Behaviour values on the right.

Answer:

https://www.cisco.com/c/en/us/td/docs/solutions/Enterprise/WAN_and_MAN/QoS_SRND/QoS-SRND-Book/QoSIntro.html

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NEW QUESTION: 137

Refer to the exhibit.

Refer to the exhibit. Endpoint A calls endpoint B. What is the only audio codec that is used for the call?

- A. PCMA/8000
- B. G722/8000
- C. Telephone-event/8000
- D. G7221/16000

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

NEW QUESTION: 138

Which statement about Cisco Unified Communications Manager and Cisco IM and Presence backups is true?

- A. Backups should be scheduled during off-peak hours to avoid system performance issues.
- B. Backups are saved as .tar files and encrypted using the web administrator account.
- C. Backups are saved as unencrypted .tar files.
- D. Backups are not needed for subscriber Cisco Unified Communications Manager and Cisco IM and Presence servers.

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1_SU1/Administration/cucm_b_administration-guide-1151su1/cucm_b_administration-guide-1151su1_chapter_01010.html#CUCM_TK_S7FC26D5_00

NEW QUESTION: 139

Refer to the exhibit. Unanswered calls do not reach the voicemail associated with the phones. Instead, callers receive the default greeting.

Which action fixes the configuration?

- A. Check the box "Redirecting Diversion Header Delivery - Outbound", then reset the trunk
- B. Check the box "Redirecting Diversion Header Delivery - Outbound"
- C. Reboot Cisco Unity Connection
- D. Review the conversation manager logs on Cisco Unity Connection

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

NEW QUESTION: 140

A remote office has a less-than-optimal WAN connection and experiences packet loss, delay, and jitter. Which VoIP codec should be used in this situation?

- A. G.711ulaw
- B. iLBC
- C. G.722.1
- D. G.729A

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

Explanation/Reference: <https://community.cisco.com/t5/collaboration-voice-and-video/summary-of-cucm-supported-codecs/ta-p/3162905>

NEW QUESTION: 141

An engineer wants to manually deploy a Cisco Webex DX80 video endpoint to an end user. Which type of provisioning can be configured on the endpoint?

- A. CUBE
- B. CMS
- C. CUCM
- D. Edge

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

Explanation/Reference: <https://www.cisco.com/c/en/us/products/collateral/collaboration-endpoints/desktop-collaboration-experience-dx600-series/datasheet-c78-731879.html>

NEW QUESTION: 142

Where is the default for maximum session Bit Rate for a region configured?

- A. Service parameter configuration
- B. Enterprise parameters configuration
- C. Enterprise Phone configuration
- D. Region configuration

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 143

Where is urgent priority enabled to bypass the T302 timer?

- A. directory number
- B. transformation pattern
- C. CTI port
- D. route partition

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

NEW QUESTION: 144

Refer to the exhibit.

An engineer must modify the existing QoS policy-map statement to implement LLQ for voice traffic. Which change must the engineer make in the configuration?

- A. bandwidth 170 to reserve 170
- B. bandwidth 170 to percent 170
- C. bandwidth 170 to LLQ 170
- D. bandwidth 170 to priority 170

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

NEW QUESTION: 145

An engineer wants to manually deploy a CISCO Webex DX80 Video endpoint to a remote user. Which type of provisioning is configured on the endpoint?

- A. Cisco Unified Border Element
- B. Cisco Unity Connection

C. Cisco Meeting Server

D. Edge

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

The Cisco Webex DX80 Video endpoint can be provisioned in two ways:

- * Automatically, using the Cisco Unified Communications Manager (CUCM) or Cisco Video Communication Server (VCS)

- * Manually, using the Edge provisioning mode

The Edge provisioning mode is used when the endpoint is not connected to the CUCM or VCS. In this mode, the endpoint is configured with the necessary settings, such as the IP address, SIP/H.323 parameters, and time and date.

The Cisco Unified Border Element (Cisco UBE) is a network element that provides security and call control for IP telephony networks. The Cisco Unity Connection is a unified messaging system that provides voicemail, email, and fax services. The Cisco Meeting Server is a video conferencing system that provides high-quality video and audio conferencing.

NEW QUESTION: 146

When a call is delivered to a gateway, the calling and called party number must be adapted to the PSTN service requirements of the trunk group. If a call is destined locally, the + sign and the explicit country code must be replaced with a national prefix. For the same city or region, the local area code must be replaced by a local prefix as applicable. Assuming that a Cisco UCM has a SIP trunk to a New York gateway (area code 917), which two combinations of solutions localize the calling and called party for a New York phone user? (Choose two.)

A.

B.

C.

D.

E.

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 147

A SIP phone has been configured in the system with MAC address 0030:96D2:D5CB.

The phone retrieves the configuration file from the Cisco UCM.

Which naming format is the file that is downloaded?

A. SIP003096D2D5CB.cnf.xml

B. SEP003096D2D5CB.cnf.xml

C. SEP0030406706705.cnf

D. SIP0030406706705.cnf

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

All phones in CUCM are SEP, as in Selsius Ethernet Phone.

NEW QUESTION: 148

Refer to the exhibit. An engineer is troubleshooting why PSTN phones are not receiving the caller's name when called from a remote Cisco UCM site.

An ISDN PRI connection is being used to reach the PSTN.

What must the administrator select to resolve the issue?

- A. isdn supp-service name calling
- B. isdn outgoing display-ie
- C. isdn enable did
- D. isdn send display le

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

https://www.cisco.com/c/en/us/td/docs/ios/voice/command/reference/vr_book/vr_i2.html

NEW QUESTION: 149

Which characteristic of distributed class-based weighted fair queueing addresses jitter prevention?

- A. It minimizes jitter by implementing a priority queue for voice traffic
- B. It uses a priority queue for voice traffic to avoid jitter
- C. It provides additional granularity by allowing a user to define custom classes
- D. It provides additional granularity by allowing a user to create classes

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

NEW QUESTION: 150

Refer to the exhibit.

A customer submits this output, captured on a Cisco IOS router. Assuming that an MGCP gateway is configured with a ISDN BRI interface, which BRI changes resolve the issue?

- A)
- B)
- C)
- D)

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

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 151

An administrator needs to stop the leading 9 from outbound calls on an IOS Voice Gateway and ensure that the system handles 911 emergency calls Which configuration is needed to accomplish this task?

- A. Option C
- B. Option A

C. Option B

D. Option D

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

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NEW QUESTION: 152

An engineer is designing a load balancing solution for two Cisco Unified Border routers. The first router (cube1.abc.com) takes 60% of the calls and the second router (cube2.abc.com) takes 40% of the calls, Assume all DNS A records have been created.

Which two SRV records are needed for a load balanced solution? (Choose two.)

A. sip_udp.abc.com 60 IN SRV 1 40 5060 cube2.abc.com

B. sip_udp.abc.com 60 IN SRV 3 60 5060 cube2.abc.com

C. sip_udp.abc.com 60 IN SRV 60 1 5060 cube1.abc.com

D. sip_udp.abc.com 60 IN SRV 2 60 5060 cube1.abc.com

E. sip_udp.abc.com 60 IN SRV 1 60 5060 cube1.abc.com

Answer: A,E ([LEAVE A REPLY](#))

- Service: The symbolic name of the desired service
- Proto: The transport protocol of the desired service; this is usually either Transmission Control Protocol (TCP) or User Datagram Protocol (UDP)
- Name: The domain name for which this record is valid, it ends in a dot
- TTL: Standard DNS time to live field
- Class: Standard DNS class field (this is always IN)
- Priority: The priority of the target host, lower value means more preferred
- Weight: A relative weight for records with the same priority, higher value means more preferred
- Port: The TCP or UDP port on which the service is to be found
- Target: The canonical hostname of the machine that provides the service, it ends in a dot

NEW QUESTION: 153

An engineer is notified that the Cisco TelePresence MX800 that is registered in Cisco Unified communications Manager shows an empty panel, and the Touch 10 shows a corresponding icon with no action when pressed. Where does the engineer go to remove the inactive custom panel?

A. The SIP Trunk security profile page in CUCM Administration

B. The In-Room control Editor on the webpage of the MX800

- C. The software Upgrades page in CUCM OS Administration
- D. The phone configuration page in CUCM Administration

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

NEW QUESTION: 154

By default, IP phones upgrade their images using the HTTP Service from the TFTP Server integrated into one or more of the call processing platforms. Which Port / Transport Protocol does this service use?

- A. 6970 / UDP
- B. 6970 / TCP
- C. 443 /TCP
- D. 8080 / TCP

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

NEW QUESTION: 155

Which command is used in a Cisco IOS XE TDM gateway to configure the voice T1/E1 controller to provide clocking?

- A. Clock source network
- B. Clock source line
- C. Clock source internal
- D. Cisco IOS XE TDM gateway T1/E1 controller cannot provide clocking.

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

XE IOS only have line and internal clock source.

<https://www.cisco.com/c/en/us/td/docs/routers/ncs4200/configuration/guide/tdm/16-12-1/b-tdm-t1-e1-xe-16-12-1-ncs4200.html>

NEW QUESTION: 156

Refer to the exhibit.

Refer to the exhibit. Which outgoing m-line SDP is sent to Cisco UCM after matching the appropriate dial peers via Cisco Unified Border Element?

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

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

NEW QUESTION: 157

On which protocol and port combination does Cisco Prime Collaboration receive notifications (Traps and InformRequests) from several network devices in the Collaboration infrastructure for which it has requested notifications?

- A. UDP 162

- B. TCP 80
- C. UDP161
- D. TCP 161

Answer: [\(SHOW ANSWER\)](#)

NEW QUESTION: 158

Which DiffServ marking is the most likely to drop packets?

- A. AF12
- B. AF13
- C. AF11
- D. AF32

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

NEW QUESTION: 159

Where is urgent priority enabled to bypass the T302 timer?

- A. CTI port
- B. route partition
- C. directory number
- D. transformation pattern

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

NEW QUESTION: 160

When a remote office location is set up with limited bandwidth resources, which codec carries the most voice calls?

- A. G.722
- B. G.711
- C. G.723
- D. G.729

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

Cisco recommends using G.729a as the low-bandwidth codec because it is supported on all Cisco Unified IP Phone models as well as most other Cisco Unified Communications devices, therefore it can eliminate the need for transcoding.

NEW QUESTION: 161

Refer to the exhibit. A call is failing to establish between two SIP Devices. The Called device answer with this SDP.

Which SDP parameter causes this issue?

- A. The calling device did not offer aptime value.
- B. The payload for G.711ulaw must be 18.
- C. The media stream is set to sendonly.
- D. The RTP port is set to 0.

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

NEW QUESTION: 162

An engineer configures a Cisco Unified Border Element and must ensure that the codecs negotiated meet the ITSP requirements. The ITSP supports G.711ulaw and G.729 for audio and H.264 for video. The preferred voice codec is G.711. Which configuration meets this requirement?

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

Answer: ([SHOW ANSWER](#))

<https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube/configuration/cube-book/cube-codec-basic.html>

NEW QUESTION: 163

Which settings are needed to configure the SIP route pattern in Cisco Unified Communications Manager?

- A. Pattern usage, IPv6 pattern, and SIP trunk/Route list
- B. Pattern usage, IPv4 pattern, and SIP trunk/Route list
- C. SIP trunk/Route list, description, and IPv4 pattern
- D. Pattern usage, IPv4 pattern, IPv6 pattern, and description

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 164

Refer to the exhibit.

Refer to the exhibit. A customer submits this debug output, captured on a Cisco IOS router.

Assuming that an MGCP gateway is configured with an ISDN BRI interface, which BRI changes resolve the issue?

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

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

NEW QUESTION: 165

Which call flow matches traffic from a Mobile and Remote Access registered endpoint to central call control?

- A. Endpoint>Expressway-C>Expressway-E>Cisco UCM
- B. Endpoint>Expressway-E>Expressway-C> Cisco UCM
- C. Endpoint>Expressway-E> Cisco UCM

D. Endpoint>Expressway-C> Cisco UCM

Answer: A (LEAVE A REPLY)

The call flow for a Mobile and Remote Access registered endpoint to central call control is as follows:

The endpoint registers with the Expressway-C.

The Expressway-C forwards the registration request to the Expressway-E.

The Expressway-E forwards the registration request to the Cisco UCM.

The Cisco UCM registers the endpoint.

When the endpoint places a call, the call flow is as follows:

The endpoint sends the call request to the Expressway-C.

The Expressway-C forwards the call request to the Expressway-E.

The Expressway-E forwards the call request to the Cisco UCM.

The Cisco UCM places the call.

The Expressway-C and Expressway-E are used to provide secure access to the Cisco UCM for endpoints that are not located on the corporate network. The Expressway-C is located on the corporate network, and the Expressway-E is located in the DMZ.

NEW QUESTION: 166

Refer to the exhibit An engineer is troubleshooting codec negotiation between two Cisco phones in different branch offices that connect over a WAN Which codec is used for the call?

A. Option D

B. Option A

C. Option B

D. Option C

Answer: D (LEAVE A REPLY)

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NEW QUESTION: 167

Which settings are needed to configure the SIP route pattern in Cisco Unified Communications Manager?

A. pattern usage, IPv6 pattern, and SIP trunk/Route list

B. pattern usage, IPv4 pattern, IPv6 pattern, and description

C. pattern usage, IPv4 pattern, and SIP trunk/Route list

D. SIP trunk/Route list, description, and IPv4 pattern

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

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmcfg/CUCM_BK_C95ABA82_00_admin-guide-100/CUCM_BK_C95ABA82_00_admin-guide-100_chapter_0100111.pdf

NEW QUESTION: 168

Refer to the exhibit.

Refer to the exhibit There is a call flow between Phone A and Phone B Phone B (holder) places Phone A (holder) on hold Which MRGL and Audio Source are played to Phone A?

- A.** MRGL A and Audio Source 2
- B.** MRGL B and Audio Source 4
- C.** MRGL A and Audio Source 4
- D.** MRGL B and Audio Source 2

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

NEW QUESTION: 169

What is a software-based media resource that is provided by the Cisco IP Voice Streaming Application?

- A.** transcoder
- B.** annunciator
- C.** auto-attendant
- D.** Video conference bridge

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

NEW QUESTION: 170

During the Cisco IP Phone registration process, the TFTP download fails. What are two reasons (or this issue?

(Choose two.)

- A.** Option 150 string was not specified, or an incorrect Option 150 string was specified.
- B.** The Cisco IP Phone does not know the IP address of the TFTP server.
- C.** The Cisco IP Phone does not know the IP address of any of the Cisco UCM Subscriber nodes.
- D.** The DNS server was not specified, which is needed to resolve the DHCP server IP address.
- E.** Option 100 string was not specified, or an incorrect Option 100 string was specified.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 171

Refer to the exhibit.

A Cisco UCM user with directory number 4401 dials 5507. and the call is routed to a Cisco Unified Border Element. Which IP address will the call be sent to?

- A. 10.0.0.2
- B. 10.0.0.3
- C. 10.0.0.4
- D. 10.0.0.1

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

NEW QUESTION: 172

Refer to the exhibit. Endpoint A calls endpoint

B. What is the only audio codec that can be used for the call?

- A. Telephone-event/8000
- B. G7221/16000
- C. PCMA/8000
- D. G722/8000

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

Explanation/Reference:

NEW QUESTION: 173

Which two protocols can be configured for the Cisco Unity Connection and Cisco UCM integration? (Choose two.)

- A. H. 323
- B. SIP
- C. SCCP
- D. MGCP
- E. RTP

Answer: (SHOW ANSWER)

The two protocols that can be configured for the Cisco Unity Connection and Cisco UCM integration are SIP and SCCP.

SIP, or Session Initiation Protocol, is a signaling protocol used for initiating, maintaining, and terminating real-time sessions, including voice, video, and messaging applications.

SCCP, or Skinny Client Control Protocol, is a Cisco proprietary signaling protocol used for controlling Cisco IP phones.

H. 323 is an older signaling protocol that is no longer widely used. MGCP, or Media Gateway Control Protocol, is a protocol used for controlling media gateways. RTP, or Real-time Transport Protocol, is a protocol used for transporting real-time data, such as voice and video

NEW QUESTION: 174

Which statement about Cisco Unified Communications Manager and Cisco IM and Presence backups is true?

- A. Backups are saved as unencrypted.tar files.
- B. Backups should be scheduled during off-peak hours to avoid system performance issues.
- C. Backups are saved as .tar files and encrypted using the web administrator account.

D. Backups are not needed for subscriber Cisco Unified Communications Manager and Cisco IM and Presence servers.

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

NEW QUESTION: 175

Which action prevents toll fraud in Cisco UCM?

- A. Implement route patterns in Cisco UCM.
- B. Configure ad hoc conference restriction.
- C. Allow off-net to off-net transfers.
- D. Implement toll fraud restriction in the Cisco IOS router.

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

NEW QUESTION: 176

A customer wants to conduct B2B video calls with a partner using an on-premises conferencing solution. Which two devices are needed to facilitate this request? (Choose two.)

- A. Expressway-C
- B. MGCP gateway
- C. Cisco TelePresence Management Suite
- D. Cisco Unified Border Element
- E. Expressway-E

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 177

Which type of input is required when configuring a third-party SIP phone?

- A. digest user
- B. serial number
- C. manufacturer
- D. authorization code

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_0_1/sysConfig/CUCM_BK_C733E983_00_cucm-system-configuration-guide-1101/Configure___Third_Party_SIP_Phones.pdf

NEW QUESTION: 178

Refer to the exhibit. An administrator must replace the T1 card with an E1 card. What is the correct configuration if the administrator was asked to configure 12 time slots?

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

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

NEW QUESTION: 179

An administrator configures Cisco UCM to use UDP for SIP signaling and finds that an endpoint cannot make calls.

Which action resolves this issue?

- A. Change the phone security profile.
- B. Change the SIP dial rules.
- C. Change the SIP profile
- D. Change the common phone profile.

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

NEW QUESTION: 180

A Cisco Telepresence SX80 suddenly has issues displaying main video to a display over HDMI. Which command can you use from the SX80 admin CLI to check the video output status to the monitor?

- A. xconfiguration video Output
- B. xStatus HDMI Output
- C. xStatus video Output
- D. xcommand video status

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

NEW QUESTION: 181

Refer to the exhibit. Which two codec permutations should be transcoded by this dspfarm? (Choose two.)

- A. G.729r8 to G.711ulaw
- B. G.729ar8 to G.711alaw
- C. iLBC to G.711ulaw
- D. G.728br8 to G.711alaw
- E. G.722 to G.729r8

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

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NEW QUESTION: 182

Given the H.323 gateway configuration and using Cisco best practices, how must the called party transformation pattern be configured to ensure that a proper ISDN type of number is set?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 183

Which two actions must be taken to provision a new device using self-provisioning?

- A. Enable the self-provisioning IVR in the Cisco UCM
- B. Import the user profile to the corporate LDAP directory.
- C. Link the appropriate service profile to the provisioning template.
- D. Link the appropriate universal device template to the user profile.
- E. Ensure the user has a directory URI and a primary extension.

Answer: A,D ([LEAVE A REPLY](#))

Self-Provisioning of a phone to a user needs:

- * an End user with a primary extension and a Feature Group Template
- * the Feature Group Template needs a User Profile
- * the User Profile needs a Universal Line and Universal Device Template Finally to activate it you need a CTI route point and a configured IVR service to actually register it.

HOWEVER, it looks like you dont actually NEED the IVR to technically fulfill the goal of self-registration, as users can use the URL based registration. but i am not 100% sure if you wont need the IVR anyways, even if you arent using it - the guide is not 100% clear on that

<https://www.cisco.com/c/en/us/support/docs/unified-communications/unified-communications-manager-callmanager/214228-configure-self-provisioning-feature-on-c.html>

NEW QUESTION: 184

An engineer must extend the corporate phone system to mobile users connecting through the internet with their own devices. One requirement is to keep that as simple as possible for end users. Which infrastructure element achieves these goals?

- A. Cisco Express Mobility
- B. Cisco Expressway-C and Expressway-E
- C. Cisco Unified Border Element
- D. Cisco Unified Instant Messaging and Presence

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

Section: Infrastructure and Design

NEW QUESTION: 185

Which recommendation is the best practice for marking video and voice media in a Cisco Unified Communications network?

- A)
- B)
- C)
- D)

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

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

NEW QUESTION: 186

What are two access management mechanisms in Cisco Webex Control Hub? (Choose two.)

- A. multifactor authentication
- B. Active Directory synchronization
- C. attribute-based access control
- D. single sign-on with Google
- E. Client ID/Client Secret

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

The correct answers are A and B.

The two access management mechanisms in Cisco Webex Control Hub are multifactor authentication and Active Directory synchronization.

Multifactor authentication is a security measure that requires users to provide two or more pieces of evidence to verify their identity. This can include something they know, such as a password, and something they have, such as a security token.

Active Directory synchronization is a process that allows Cisco Webex Control Hub to automatically synchronize user accounts from an Active Directory domain. This can simplify user management and provide users with single sign-on access to Cisco Webex Control Hub and other applications.

NEW QUESTION: 187

Refer to the exhibit.

An engineer verifies the configured of an MGCP gateway. The commands are already configured. Which command is necessary to enable MGCP?

- A. Device (config)# mgcp
- B. Device (config) # com-manager active
- C. Device(config)# ccm-manager enable
- D. Device(config)# mgcp enable

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

NEW QUESTION: 188

Refer to the exhibit. When a UC Administrator is troubleshooting DTMF negotiated by this SIP INVITE, which two messages should be examined next to further troubleshoot the issue?

(Choose two.)

- A. REGISTER
- B. UPDATE
- C. PRACK
- D. NOTIFY
- E. SUBSCRIBE

Answer: D,E ([LEAVE A REPLY](#))

Section: Protocols, Codecs, and Endpoints

NEW QUESTION: 189

What is required for Cisco UCM to accept SIP calls with a URI in the format of "sip 2001@cucmpub.cisco.com"?

- A. Set the SIPS URI Handling to True in CallManager Service Parameters.
- B. Change the Destination Address to a Fully Qualified Domain Name on the SIP trunk.
- C. Define Cluster Fully Qualified Domain Name under Servers in Cisco UCM.
- D. Define Cluster Fully Qualified Domain Name in Enterprise Parameters.

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

NEW QUESTION: 190

What is the element of Cisco collaboration infrastructure that allows Jabber clients outside of the network to register in Cisco Unified communications Manager and use its resources?

- A. Cisco Unified Border Element
- B. Cisco prime collaboration provisioning server
- C. Cisco Expressway
- D. Cisco IM and Presence node

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 191

What is set when using COS to mark an Ethernet frame?

- A. Ipp bits
- B. IP ECN bits
- C. DCSP bits
- D. 802.1 p User Priority bits

Answer: ([SHOW ANSWER](#)**)**

When using COS to mark an Ethernet frame, the 802.1 p User Priority bits are set. These bits are used to indicate the priority of the frame. The higher the priority, the more likely the frame is to be transmitted first.

NEW QUESTION: 192

An engineer is configuring IP telephony. The network relies on DHCP to provide TFTP server addresses to the endpoints. Policy requires the endpoints to receive two server addresses. Which DHCP option must be configured?

- A. 66
- B. 143
- C. 150
- D. 166

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

Option 66 provides a single TFTP server address. Option 150 provides a list of TFTP server addresses. Also if both are configured, Cisco phones will target 150 by default before 66.

NEW QUESTION: 193

Which type of input is required when configuring a third-party SIP phone?

- A. serial number350-801 2023-4
- B. manufacturer
- C. digest user
- D. authorization code

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

NEW QUESTION: 194

What are two features of Cisco Expressway that the customer gets if Expressway-C and Expressway-E are deployed?(Choose two.)

- A. session-based access to comprehensive collaboration for remote workers, without the need for a separate VPN client.
- B. highly secure freal-traversal technology to extend organizational reach.
- C. utilization and adoption metrics of all remotely connected devices.
- D. complete endpoint registration and monitoring capabilities for devices that are local and remote.
- E. additional visibility of the edge traffic in an organization.

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

NEW QUESTION: 195

Refer to the exhibit. Endpoint A calls endpoint

B. What is the only audio codec that can be used for the call?

- A. G722/8000
- B. PCMA/8000
- C. Telephone-event/8000
- D. G7221/16000

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

NEW QUESTION: 196

Refer to the exhibit. What is the reason for this call failure?

- A. The available location bandwidth between sites is exceeded
- B. The device pool contains more call processing agents in the CMG group than the endpoint can support
- C. MRGL contains more media groups than the endpoint can support.
- D. The hunt group contains more devices than the endpoint can support

Answer: ([SHOW ANSWER](#))

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NEW QUESTION: 197

Users want their mobile phones to be able to access their Cisco Unity Connection mailboxes with only having to enter their voicemail pin at the login prompt when calling the pilot number. Where should an engineer configure this feature?

- A. alternate extensions
- B. message settings
- C. transfer rules
- D. greetings

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

Section: Collaboration Applications

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/administration/guide/10xcucsagx/10xcucsag170.html

NEW QUESTION: 198

An engineer is deploying the Webex app on a Microsoft Windows computer that has multiple user accounts. Which CLI argument allows the engineer to install the application as a "per machine" installation?

- A. ACCEPT_EULA
- B. INSTALL ROOT
- C. ALLUSERS
- D. FORCELOCKDOWN

Answer: ([SHOW ANSWER](#))

The ALLUSERS property is a command-line argument that can be used to install the Webex app as a "per machine" installation. This means that the app will be installed for all users on the computer, regardless of which user account is currently logged in.

The ACCEPT_EULA property is a command-line argument that can be used to accept the end-user license agreement (EULA) for the Webex app. The INSTALL_ROOT property is a command-line argument that can be used to specify the installation directory for the Webex app. The FORCELOCKDOWN property is a command-line argument that can be used to prevent users from uninstalling the Webex app.

NEW QUESTION: 199

What does average rate limiting allow?

- A. bandwidth up to the Be size
- B. traffic to burst to the Be size when there is available bandwidth
- C. more traffic than the CIR to be sent when there is available bandwidth
- D. transmits traffic bursts up to the Bc size

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

NEW QUESTION: 200

What is a valid class included in the 8-class QoS strategy in a VoIP network?

- A. Broadcast video
- B. Real-Time Interactive
- C. Multimedia conferencing
- D. Assured Forwarding

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 201

Which version is used to provide encryption for SNMP management traffic in collaboration deployments?

- A. SNMPv2c
- B. SNMPv2
- C. SNMPv1
- D. SNMPv3

Answer: ([SHOW ANSWER](#)**)**

Section: Infrastructure and Design

Explanation/Reference:

<https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/snmp/configuration/xe-16/snmp-xe-16-book/nm-snmp-encrypt-snmp-support.html>

NEW QUESTION: 202

After an administrator installs the Cisco Jabber Client, it fails to register with the Cisco UCM server. Which two actions should be taken to troubleshoot this problem? (Choose two.)

- A. Verify the installation of the LMHOST file on the PC
- B. Verify the configuration of the DNS SRV record.
- C. Verify corporate firewall settings for client connections.
- D. Verify Layer 3 connectivity on the gateway.
- E. Reboot the Cisco Unified Border Element Gateway.

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

<https://www.cisco.com/c/en/us/support/docs/unified-communications/jabber/212471-jabber-login.html>

NEW QUESTION: 203

AN administrator needs to help remote employee make a free call to an international destination. The administrator calls the employee, then conferences in the international party. The administrator drops the call, and the employee and the international party continue their conversation. Which action prevents this type of toll fraud in the Cisco USM?

- A. Set service parameter "Drop Ad Hoc Conference" to "When Conference Controller leaves"
- B. Set service parameter "Advanced Ad Hoc Conference" to FALSE
- C. Set service parameter "Drop Ad Hoc Conference" to "Do not allow outside parties "
- D. Set service parameter "Advanced Ad Hoc Conference" to 2.

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

NEW QUESTION: 204

How is bandwidth allocated to traffic flows in a flow-based WFQ solution?

- A. Bandwidth is divided among traffic flows. Voice has priority.
- B. Each type of traffic flow has equal bandwidth.
- C. All the bandwidth is divided based on the QoS marking of the packets.
- D. Voice has priority and the other types of traffic share the remaining bandwidth

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

NEW QUESTION: 205

Which type of greeting in the Call Handler configuration in Cisco Unity Connection overtimes all other greetings?

- A. holidays
- B. supervisory
- C. alternate
- D. priority

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

Alternate

Can be used for a variety of special situations, such as vacations or a leave. An alternate greeting overrides all other greetings.

Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/administration/guide/10xcucsagx/10xcucsag080.html

NEW QUESTION: 206

Which user group is targeted by MRA services?

- A. On-the-go mobile workforce who connect to corporate phone services using their own mobile device
- B. Mobile workers in a host desk environment at HQ who log in every morning at possibility a different phone
- C. Production floor users who need wireless mobility in remote areas of the factory
- D. Call center agents who dial out to remote customers

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

NEW QUESTION: 207

Refer to the exhibit.

The SOP offer/answer has been completed successfully but there is no DTMF when users press keys. What is the cause of the issue?

- A. DTMF was not negotiated on the call.
- B. Payload type 110 was negotiated rather than type 101.
- C. DTMF was negotiated properly in these messages.
- D. G.729 rather than G.711ulaw was negotiated.

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

NEW QUESTION: 208

If a phone needs to register with cucm1.cisco.com, which network service assists with the phone registration process?

- A. SNMP
- B. ICMP
- C. SMTP
- D. DNS

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

According to the Cisco Community website¹, the phone uses DNS to resolve the hostname of the CUCM server (cucm1.cisco.com) to its IP address. DNS is a network service that translates domain names into IP addresses.

NEW QUESTION: 209

An engineer must manually register an analog port with cisco UCM using the MGCP protocol. Which commands are required?

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

D. Option D

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

NEW QUESTION: 210

Which configuration on Cisco Unified Communication Manager is required for SIP MWI to work?

- A. The line partition must be inside the inbound CSS assigned to the CUC SIP trunk.
- B. Assign a MWI extension on the mailbox.
- C. The line partition must be inside the rerouting CSS assigned to the Cisco assigned to the Cisco unity connection SIP trunk.
- D. Set the "Enable message waiting indicator" on the port group.

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/integration/guide/cucm_sip/cucintcucmsip/cucintcucmsip060.html

NEW QUESTION: 211

An engineer is configuring a phone system CISCO UCM and wants to activate TFTP service. The engineer selects the serviceability page for configuration. Which nodes configurable for TFTP?

- A. any two nodes
- B. any node
- C. only nodes that have Cisco UCM service enabled
- D. any subscriber nodes

Answer: ([SHOW ANSWER](#)**)**

TFTP is a network protocol that is used to transfer files between devices. It is often used to transfer firmware and configuration files to network devices. In order to use TFTP, the device must have a TFTP server configured.

In Cisco UCM, the TFTP server is configured on the serviceability page. The TFTP server can be configured on any node that has Cisco UCM service enabled. The TFTP server cannot be configured on nodes that do not have Cisco UCM service enabled.

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NEW QUESTION: 212

Refer to the exhibit. An administrator configured a codec preference list with G.722 and ILBC codecs. Which change must the administrator make in the dial-peer section of the configuration to use this list?

- A. add voice-codecs 20
- B. add session codec 20
- C. add codec preference 20
- D. add voice-class codec 20

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

<https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/cube/configuration/cube-book/voi-negt-aud-code.html>

NEW QUESTION: 213

The IP phones at a customer site do not pick an IP address from the DHCP. An engineer must temporarily disable LLDP on all ports of the switch to leave only CDP. Which two commands accomplish this task? (Choose two.)

- A. Switch# copy running-config startup-config
- B. Switch(config)# no lldp run
- C. Switch(config)# no lldp transmit
- D. Switch# configure terminal
- E. Switch(config)# interface GigabitEthernet1/0/1

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

NEW QUESTION: 214

During the Cisco IP Phone registration process the TFTP download fails. What are two reasons for this issue?

(Choose two)

- A. Option 100 string was not specified, or an incorrect Option 100 string was specified.
- B. The Cisco IP Phone does not know the IP address of the TFTP server.
- C. Option 150 string was not specified, or an incorrect Option 150 string was specified.
- D. The DNS server was not specified, which is needed to resolve the DHCP server IP address.
- E. The Cisco IP Phone does not know the IP address of any of the Cisco UCM Subscriber nodes.

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

NEW QUESTION: 215

A collaboration engineer troubleshoots issues with a Cisco IP Phone 7800 Series. The IPv4 address of the phone is reachable via ICMP and HTTP and the phone is registered to Cisco UCM. However, the engineer cannot reach the CLI of the phone. Which two actions in Cisco UCM resolve the issue? (Choose two)

- A. Enable SSH Access under Product Specific Configuration Layout in Cisco UCM
- B. Enable Settings Access under Product Specific Configuration Layout in Cisco UCM.
- C. Set a username and password under Secure Shell Information in Cisco UCM

- D. Disable Web Access under Product Specific Configuration Layout in Cisco UCM
- E. Enable FIPS Mode under Product Specific Configuration Layout in Cisco UCM

Answer: A,D ([LEAVE A REPLY](#))

NEW QUESTION: 216

Given this H.323 gateway configuration and using cisco best practices, how must the called party transformation pattern to ensure that a proper ISDN type of number is set?

- A)
- B)
- C)
- D)

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 217

Refer to the exhibit.

Refer to the exhibit. A collaboration engineer configures Cisco UCM to act as a DHCP server. What must be done next to configure the DHCP server?

- A. Add a DHCP subnet to the DHCP server under Cisco UCM Administration.
- B. Restart the TFTP service under Cisco Unified Serviceability.
- C. Add the new DHCP server to the primary DNS server
- D. Restart the Cisco DHCP Monitor Service under Cisco Unified Serviceability

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

NEW QUESTION: 218

In an eleven-class queueing model following Realtime, Best effort, and scavenger queueing rules. What recommended percentage of bandwidth must be allocated for interactive video?

- A. 20%
- B. 10%
- C. 15%
- D. 12%

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

NEW QUESTION: 219

Which Cisco Unified Communications Manager service parameter should be enabled disconnect a multiparty call when the call initiator hangs up?

- A. Drop Ad Hoc Conference

- B. H.225 Block Setup Destination
- C. Block OffNet To OffNet Transfer
- D. Enterprise Feature Access Code for Conference

Answer: (SHOW ANSWER)

Section: Infrastructure and Design

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmsys/CUCM_BK_SE5FCFB6_00_cucm-system-guide-100/CUCM_BK_SE5FCFB6_00_cucm-system-guide-100_chapter_011000.html#CUCM_TK_DFC66444_00

NEW QUESTION: 220

An engineer encounters third-party devices that do not support Cisco Discovery Protocol. What must be configured on the network to allow device discovery?

- A. LLDP
- B. TFTP
- C. LACP
- D. SNMP

Answer: A (LEAVE A REPLY)

LLDP (Link Layer Discovery Protocol) is a vendor-neutral network discovery protocol that is used to discover the topology of a network. LLDP is similar to CDP (Cisco Discovery Protocol), but it is not proprietary to Cisco. LLDP is supported by a wide range of network devices, including switches, routers, and firewalls.

To configure LLDP on a network, you must enable LLDP on the devices that you want to discover. You can then use a network management tool, such as Cisco Network Assistant, to view the topology of the network.

The other options are incorrect. TFTP (Trivial File Transfer Protocol) is a network protocol that is used to transfer files between devices. LACP (Link Aggregation Control Protocol) is a network protocol that is used to aggregate multiple network links into a single logical link. SNMP (Simple Network Management Protocol) is a network protocol that is used to manage network devices.

NEW QUESTION: 221

A customer has Cisco Unity Connection that is integrated with LDAP. As a unity connection administrator, you have received a request to change the first for VM user. Where must the change be performed?

- A. Cisco IM and presence
- B. Cisco Unified Communications Manager end user
- C. Cisco unity connection
- D. Active directory

Answer: D (LEAVE A REPLY)

NEW QUESTION: 222

Refer to the exhibit.

An engineer must modify the existing QoS policy-map statement to implement LLQ for voice traffic. Which change must the engineer make in the configuration?

- A. bandwidth 170 to reserve 170
- B. bandwidth 170 to percent 170
- C. bandwidth 170 to priority 170
- D. bandwidth 170 to LLQ 170

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 223

An engineer must configure switch port 5/1 to send CDP packets to configure an attached Cisco IP phone to trust tagged traffic on it's access port. Which command is required to complete the configuration?

- A. platform qos trust extend cos 5
- B. platform qos trust extend cos 3
- C. platform qos trust extend
- D. platform qos extend trust

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

NEW QUESTION: 224

Which access control group is required on an end user to allow Jabber to do deskphone mode?

- A. Standard CTI Enabled
- B. Allow control of device from CTI
- C. Standard CTI Allow Reception of SRTP key Material
- D. Standard CTI Secure Connection

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

NEW QUESTION: 225

Users dial a 9 before a 10-digit phone number to make an off-net call All 11 digits are sent to the Cisco Unified Border Element before going out to the PSTN The PSTN provider accepts only 10 digits. Which configuration is needed on the Cisco Unified Border Element for calls to be successful?

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

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

NEW QUESTION: 226

Which Cisco Unified Communications Manager configuration is required for SIP MWI integrations?

- A. Enable "Accept unsolicited notification" on the SIP Trunk Security Profile.
- B. Enable "Accept presence subscription" on the SIP Trunk Security Profile.
- C. Select "Redirecting Diversion Header Delivery - Inbound" on the SIP trunk.
- D. Select "Redirecting Diversion Header Delivery - Outbound" on the SIP trunk.

Answer: ([SHOW ANSWER](#))

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NEW QUESTION: 227

Refer to exhibit. An engineer configured a codec preference list on a Cisco Unified Border Element. Which command is required on the dial peer to ensure that all codecs are offered during an early offer call?

- A. voice-class codec 10 early-offer
- B. voice class codec 10
- C. voice-class codec 10 offer-all
- D. voice-class codec 10

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 228

Regarding SIP integrations with Cisco Unified Communications Manager, if the Cisco Unity Connection is configured to listen for incoming IPv4 and IPv6 traffic, how should the addressing mode be set up in the Cisco Unity Connection?

- A. Set up media ports for each port group to use IPv4.
- B. Set up for each group to use IPv4 and IPv6.
- C. Set up IPv4 and IPv6 in cisco Unified CM.
- D. Set up is not required.

Answer: ([SHOW ANSWER](#))

For SIP integrations with Cisco Unified CM, if Unity Connection is configured to listen for incoming IPv4 and IPv6 traffic, you can configure the addressing mode that Unity Connection uses for call control signaling for each port group to use either IPv4 or IPv6. (This mode is also used when connecting to a TFTP server.) In addition, you can configure the addressing mode that Unity Connection uses for media for each port group to use either IPv4 or IPv6.

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/11x/design/guide/b_11xcucdg/b_11xcucdg_chapter_00.html?bookSearch=true

NEW QUESTION: 229

Refer to the exhibit. An engineer configures ISDN on a voice gateway. The provider confirms that the PRI is configured with 10 channels the engineer ordered and is working from the provider side, but the engineer cannot get a B-channel to carry voice. The rest of the configuration for the serial interface and voice network is functioning correctly. Which actions must be taken to carry voice?

- A.** The engineer used a T1 interface but must use an E1 interface.
- B.** The pri-group timeslots command must be 0-9 for the 10 channels because all values on a router start with 0.
- C.** The engineer must activate the controller card on the voice gateway before configuring the device.
- D.** The engineer must manually revert the order of using the channels.

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

NEW QUESTION: 230

Refer to the exhibit. An engineer must implement toll fraud prevention on a Cisco UCM cluster. Only the SIP protocol must be allowed for connections passing through Cisco Unified Border Element. What must be configured?

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

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 231

An administrator has been asked to implement toll fraud prevention in Cisco UCM. Which tool is used to begin this process?

- A.** Cisco UCM class of service
- B.** Cisco Unified Real-Time Monitoring Tool
- C.** Cisco Unified Mobility
- D.** Cisco UCM Access Control Group restrictions

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

NEW QUESTION: 232

A network administrator deleted a user from the LDAP directory of a company. The end user shows as Inactive LDAP Synchronize User in Cisco Unified Communications Manager. Which step is next to remove this user from Cisco Unified Communications Manager?

- A.** Restart the Dirsync service after the user is deleted from LDAP directory.

- B. Delete the user directly from Cisco Unified Communications Manager.
- C. Wait 24 hours for the garbage collector to remove the user.
- D. Execute a manual sync to refresh the local databased and delete the end user.

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab11/collab11/directry.html#pgfld-1045229

NEW QUESTION: 233

Refer to the exhibit.

Refer to the exhibit. How must the +E.164 translation pattern be configured to reach international number 496929810?

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

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

NEW QUESTION: 234

A remote office has a less-than-optimal WAN connection end experiences packet loss, delay, and jitter.

Which VoIP codec should be used in this situation?

- A. G.729A
- B. G.711alaw
- C. G.722.1
- D. iLBC

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

NEW QUESTION: 235

Refer to the exhibit. A call to an international number has failed.

Which action corrects this problem?

- A. Add the bearer-cap speech command to the voice port.
- B. Strip the leading 011 from the called party number
- C. Add the isdn switch-type primart-dms100 command to the serial interface.
- D. Assign a transcoder to the MRGL of the gateway.

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

NEW QUESTION: 236

Refer to the exhibit.

An engineer verifies the configuration of an MGCP gateway. The commands are already configured. Which command is necessary to enable MGCP?

- A. Device(config)# mgcp enable

- B. Device(config)# ccm-manager enable
- C. Device (config) # com-manager active
- D. Device (config)# mgcp

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

NEW QUESTION: 237

Refer to the exhibit.

Refer to the exhibit. An engineer is troubleshooting why PSTN phones are not receiving the caller's name when called from a remote Cisco UCM site. An ISDN PRI connection is being used to reach the PSTN. What must the administrator select to resolve the issue?

- A. isdn enable did
- B. isdn outgoing display-ie
- C. isdn send display le
- D. isdn supp-service name calling

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

NEW QUESTION: 238

Which configuration on Cisco UCM is required for SIP MWI to work?

- A. Assign an MWI extension on the mailbox.
- B. The line partition must be inside the inbound CSS assigned to the CUC SIP trunk.
- C. The line partition must be inside the rerouting CSS assigned to the Cisco Unity Connection SIP trunk.
- D. Set the "Enable message waiting indicator" on the port group.

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

The line partition must be inside the inbound CSS assigned to the CUC SIP trunk. This ensures that the SIP MWI messages are sent to the correct destination.

The other options are incorrect because:

Assigning an MWI extension on the mailbox is not required for SIP MWI to work.

The line partition does not need to be inside the rerouting CSS assigned to the Cisco Unity Connection SIP trunk.

Setting the "Enable message waiting indicator" on the port group is not required for SIP MWI to work.

NEW QUESTION: 239

How can an administrator stop Cisco Unified Communications Manager from advertising the OPUS codec for recording enabled devices?

- A. Router recorded calls through Cisco Unified Border Element because it does not support OPUS.
- B. Go to the phone's configuration page and set "Advertise OPUS Codec" to be "false"
- C. In CUCM Service Parameters set "Opus codec Enables" to "Enabled for all devices Except Recording-Enabled devices."

D. Integrate the Cisco Unified CM with a recording solution that does not support OPUS.

Answer: (SHOW ANSWER)

NEW QUESTION: 240

What are two key features of the Expressway series? (Choose two.)

- A. device registration over the Internet
- B. VPN connection toward the internal UC resources
- C. IP to PSTN call connectivity
- D. SIP header modification
- E. B2B calls

Answer: A,E (LEAVE A REPLY)

NEW QUESTION: 241

Refer to the exhibit There is a call flow between Phone A and Phone B Phone B (holder) places Phone A (holder) on hold Which MRGL and Audio Source are played to Phone A?

- A. MRGL B and Audio Source 4
- B. MRGL A and Audio Source 4
- C. MRGL B and Audio Source 2
- D. MRGL A and Audio Source 2

Answer: D (LEAVE A REPLY)

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NEW QUESTION: 242

How does an administrator make Cisco IP phone display the last 10 digits of the calling number when the call is in the connected state, and also display the calling number in the E.164 format within call history on the phone?

- A. Configure a calling party transformation pattern that keeps only last 10 digits.
- B. Configure a translation pattern that has a calling Party Transform Mask of XXXXXXXXXXXX.
- C. On the inbound SIP trunk, change Significant Digits to 10.
- D. Change the service parameter Apply Transformations On Remote Number to True.

Answer: D (LEAVE A REPLY)

The outbound call's calling party transformation CSS (also referred to as localization or remote number calling party transformation CSS) can also be used to localize remote connected party information.

To enable this, the advanced service parameter Apply Transformations On Remote Number must be enabled.

NEW QUESTION: 243

Why would we not include an end user's PC device in a QoS trust boundary?

- A. The end user may incorrectly tag their traffic to be prioritized over other network traffic
- B. The end user could incorrectly tag their traffic to bypass firewalls.
- C. There is no reason not to include an end user's PC device in a QoS trust boundary
- D. The end user could incorrectly tag their traffic to advertise their PC as a default gateway.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 244

An administrator is asked to implement toll fraud prevention in Cisco UCM, specifically to restrict off-net to off-net call transfers How is this implemented?

- A. Enforce ad-hoc conference restrictions
- B. Implement time-of-day routing.
- C. Use the correct route filters.
- D. Set the appropriate service parameter

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

NEW QUESTION: 245

A collaboration engineer must optimize the dial plan within Cisco UCM. There are multiple remote sites. And each site has its own route patterns and local gateways.

What should the engineer do on the cisco UCM to optimize the dial plan?

- A. Configure a Standard Local Route Group to use a single route pattern for all calls within the cluster.
- B. Implement ILS with GDPR so the dial plan can dynamically replicate across clusters.
- C. Leverage Cisco UCM Express as SRST so the phones can have more features of each remote site
- D. Create a centralized dial plan with a Cisco UCM Session Management Edition cluster or a Cisco gatekeeper.

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

NEW QUESTION: 246

How many DNS SRV entries can be defined in the SIP trunk destination address field in Cisco Unified Communications Manager?

- A. 1
- B. 8

C. 16

D. 4

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

Section: Infrastructure and Design

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1/sysConfig/CUCM_BK_SE5DAF88_00_cucm-system-configuration-guide-1151/CUCM_BK_SE5DAF88_00_cucm-system-configuration-guide-1151_chapter_01110.html

NEW QUESTION: 247

Which task is required when configuring self-provisioning for an end user in Cisco UCM?

A. Associate the end user to a SIP Profile.

B. Disable Auto-Registration.

C. Enable Auto-Registration.

D. Associate the end user to the Standard CCM Super Users group.

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

NEW QUESTION: 248

Refer to the exhibit.

Refer to the exhibit. The SDP offer/answer has been completed successfully but there is no DTMF when users press keys. What is the cause of the issue?

A. G.729 rather than G.711ulaw was negotiated.

B. Payload type 110 was negotiated rather than type 101.

C. DTMF was not negotiated on the call.

D. DTMF was negotiated properly in these messages.

Answer: (SHOW ANSWER)

NEW QUESTION: 249

What is set when using CoS to mark an Ethernet frame?

A. 802.1p User Priority bits

B. IPP bits

C. IP ECN bits

D. DCSP bits

Answer: (SHOW ANSWER)

NEW QUESTION: 250

How does an administrator ensure compliance with ITU-T G.114 latency requirements when deploying voice and video applications over routes that include a geostationary satellite?

A. The application will work because it will be below the recommended latency requirement. If an engineer uses compressed headers to speed up the voice packets.

- B.** Using G.729 voice codec Instead of G.711 voice codec will reduce the packet size and the latency to stay within the recommended threshold.
- C.** The satellite is 35,786 km (22,236 miles) above sea level (from the equator), so an engineer will not get below 150 ms latency in one direction and will need to manage the expectation of the user.
- D.** For voice and video applications over satellite, Cisco recommends using WAN optimization compression, which will save space or time of the transmission. The latency will then be below the recommended latency threshold for voice.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 251

Which attribute contains an XMPP stanza?

- A.** type
- B.** iq
- C.** message
- D.** presence

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

Section: Collaboration Applications

Explanation/Reference:

<https://www.ciscolive.com/c/dam/r/ciscolive/apjc/docs/2011/pdf/TECUCC-2611.pdf> slide 18

NEW QUESTION: 252

Which dial plan function restricts calls that are made by a lobby phone to internal extensions only?

- A.** endpoint addressing
- B.** calling privileges
- C.** manipulation of dialed destination
- D.** path selection

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 253

Refer to the exhibit. A collaboration engineer adds an analog gateway to a Cisco UCM cluster. The engineer chooses MGCP over SCCP as the gateway protocol.

Which two actions ensure that the gateway registers? (Choose two.)

- A.** Delete and re-add the gateway configuration in Cisco UCM.
- B.** Enter "mgcp" on the gateway in configuration mode.
- C.** Enter "no sccp" on the gateway in configuration mode.
- D.** Enter "ccm-manager config" on the gateway in configuration mode.
- E.** Enter "ccm-manager mgcp" on the gateway in configuration mode.

Answer: ([SHOW ANSWER](#))

To enable the gateway to communicate with Cisco CallManager through the Media Gateway Control Protocol (MGCP) and to supply redundant control agent services, use the "ccm-manager mgcp" command in global configuration mode.

http://www.cisco.com/en/US/docs/ios/12_3t/voice/command/reference/vrht_c4_ps5207_TSD_Products_Command_Reference_Chapter.html#wp1072910 Do a mgcp / no mgcp once its added.

Make sure that the domain name on cucm is the same as it appears in the 'show ccm-manager' output on the gateway.

NEW QUESTION: 254

Refer to the exhibit.

You deploy Mobile and Remote Access for Jabber and discover that Jabber for Windows does not register to cisco Unified Communications Manager while outside of the office. What is a cause of this issue?

- A. The DNS record should be changed from _collab-edge._tls example.com.
- B. Server 4.2.2.2 is not a valid DNS server.
- C. The DNS record type should be changed from SRV to A.
- D. The DNS record should be created for _cisco-uds._tcp example.com.

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

NEW QUESTION: 255

Refer to the exhibit. An administrator must fix the SRV records to ensure that server1.sample.com is always contacted first from the three servers. Which solution should the engineer apply to resolve this issue?

- A. Priority = 10, Weight = 10
- B. Priority = 100, Weight = 90
- C. Priority = 10, Weight = 5
- D. Priority = 5, Weight = 70

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 256

Refer to the exhibit. This INVITE is sent to an endpoint that only supports G.729.

What must be done for this call to succeed?

- A. ADD a media termination point that supports G.711ulaw and G.729.
- B. Noting: both sides support G.729.
- C. Add a transcoder that supports G.711ualw and G.729.
- D. Nothing: both sides support payload type 101.

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

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NEW QUESTION: 257

What are two characteristics of priority queueing? (Choose two.)

- A. It ensures that critical traffic queues are serviced first.
- B. It guarantees bandwidth for user-defined traffic classes.
- C. It places traffic into one of four queues.
- D. It applies strict priority to user-defined traffic classes.
- E. It allocates an equal share of bandwidth to each flow.

Answer: A,C ([LEAVE A REPLY](#))

NEW QUESTION: 258

Refer to the exhibit.

You deploy Mobile and Remote Access for Jabber and discover that Jabber for Windows does not register to Cisco Unified Communications Manager while outside of the office. What is a cause of this issue?

- A. The DNS record should be changed from _collab-edge_tcp example.com.
- B. The DNS record should be created for _cisco-uds_tcp example.com.
- C. The DNS record type should be changed from SRV to A.
- D. Server 4.2.2.2 is not a valid DNS server.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 259

A Cisco TelePresence SX80 suddenly has issues displaying main video to a display over HDMI. Which command can you use from the SX80 admin CLI to check the video output status to the monitor?

- A. xCommand Video Status
- B. xConfiguration Video Output
- C. xStatus HDMI Output
- D. xStatus Video Output

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

NEW QUESTION: 260

Refer to the exhibit.

Refer to the exhibit. A collaboration engineer configures Cisco UCM to act as a DHCP server. What must be done next to configure the DHCP server?

- A. Restart the TFTP service under Cisco Unified Serviceability.
- B. Restart the Cisco DHCP Monitor Service under Cisco Unified Serviceability
- C. Add a DHCP subnet to the DHCP server under Cisco UCM Administration.
- D. Add the new DHCP server to the primary DNS server

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

NEW QUESTION: 261

In an eleven-class queueing model following realtime, best effort, and scavenger queueing rules, what recommended percentage of bandwidth must be allocated for interactive video?

- A. 10%
- B. 12%
- C. 15%
- D. 20%

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

<https://www.ciscopress.com/articles/article.asp?p=357102&seqNum=7#:~:text=If%20Scavenger%20and%20Bulk%20traffic,illustrated%20in%20Figure%2012%2D10>

NEW QUESTION: 262

Refer to the exhibit. An engineer is troubleshooting a codec negotiation issue where both endpoints that are involved in the call support the codecs listed in the exhibit.

Which audio codec is selected if a call between two endpoints in Region 1 is placed?

- A. G.711u
- B. G.729
- C. G.722
- D. G.711a

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

NEW QUESTION: 263

An administrator works with an ISDN PRI that is connected to a third-party PBX.

The ISDN link does not come up; and the administrator finds that the third-party PBX uses the QSIG signalling method.

Which command enables the Cisco IOS Gateway to use QSIG signalling on the ISDN link?

- A. isdn switch-type basic-ni
- B. isdn switch-type primary-qsig
- C. isdn incoming-voice voice
- D. isdn switch-type basic-qsig

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

NEW QUESTION: 264

Refer to the Exhibit.

Which command is required to allow this media resource to handle Video Media streams?

- A. maximum sessions hardware 50
- B. codec pass-through
- C. video codec h264
- D. associate application Cisco Unified Border Element

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

NEW QUESTION: 265

Which two functions are provided by Cisco Expressway Series? (Choose two.)

- A. voice and video transcoding
- B. voice and video conferencing
- C. interworking of SIP and H.323
- D. intercluster extension mobility
- E. endpoint registration

Answer: ([SHOW ANSWER](#)**)**

The Cisco Expressway Series provides the following functions:

Voice and video transcoding

Interworking of SIP and H.323

Firewall traversal

Session border controller (SBC) functionality

Endpoint registration

Call admission control (CAC)

Quality of service (QoS)

Security

The Cisco Expressway Series does not provide voice and video conferencing or intercluster extension mobility.

NEW QUESTION: 266

According to the QoS Baseline Model, drag and drop the applications from the left onto the correct per-Hop Behaviour values on the right.

Answer:

NEW QUESTION: 267

How are E.164 called-party numbers normalized on a globalized call-routing environment in Cisco Unified Communications Manager?

- A. Normalization is achieved by setting up calling search and partitions at the SIP trunk for PSTN connection.
- B. Call ingress must be normalized before the call being routed.
- C. Normalization is achieved by stripping or translating the called numbers.
- D. Normalization is not required

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

NEW QUESTION: 268

Which field is configured to change the caller ID information on a SIP route pattern?

- A. Calling Party Transformation Mask
- B. Connected Line ID Presentation
- C. Route Partition
- D. Called Party Transformation Mask

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

NEW QUESTION: 269

A network administrator with ID392116981 has determined that a WAN link between two Cisco UCM clusters supports only 1 Mbps of bandwidth for voice traffic. How many calls does this link support if G.711 as the audio codec is used?

- A. 15
- B. 16
- C. 13
- D. 12

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

NEW QUESTION: 270

Which endpoint feature is supported using Mobile and Remote Access through Cisco Expressway?

- A. SRST
- B. SSO
- C. H.323 registration proxy to Cisco Unified Communications Manager
- D. MGCP gateway registration

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

Explanation/Reference:

NEW QUESTION: 271

A collaboration engineer adds a voice gateway to Cisco UCM. The engineer creates a new gateway device in Cisco UCM, selects VG320 as the device type and selects MGCP as the protocol. What must be done next to add the gateway to the Cisco UCM database?

- A. Select a device pool for the new gateway
- B. Select the DTMF relay type for the gateway
- C. Add the FQDN or hostname of the device
- D. Configure the module in slot 0 of the new gateway

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

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<https://www.trainingdump.com/Cisco/350-801-practice-exam-dumps.html> (364 Q&As Dumps, **40%OFF Special Discount: Exam-Tests**)

NEW QUESTION: 272

According to the QoS Baseline Model, drag and drop the applications from the left onto the correct per-Hop Behaviour values on the right.

Answer:

NEW QUESTION: 273

An engineer configures a Cisco Unified Border Element and must ensure that the codecs negotiated meet the ITSP requirements. The ITSP supports G.711ulaw and G.729 for audio and H.264 for video. The preferred voice codec is G.711. Which configuration meets this requirement?

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

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

NEW QUESTION: 274

A Cisco UCM administrator sets up new route patterns to support phones in four different locations, all with local gateways. The administrator wants to use the same route pattern for all four locations.

How must the system be configured to achieve this goal?

- A. Use standard local route groups.
- B. Add a CSS to each local gateway.
- C. Use CSS alternate routing rules.
- D. Use transforms in the route groups.

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

NEW QUESTION: 275

An engineer must configure an MGCP gateway and register it to Cisco Unified Communications Manager.

Which prerequisite must be met before applying the gateway commands to enable MGCP?

- A. The MGCP gateway and the Cisco Unified CM must be able to communicate over ports 5060 and 5061.

- B.** Cisco Unified CM and the MGCP gateway must utilize the SIP OPTIONS ping feature to monitor status.
- C.** The MGCP gateway must have voice service VoIP configured.
- D.** The MGCP gateway and the Cisco Unified CM must be able to communicate over ports 2427, 2428, and 2727.

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

Section: Cisco IOS XE Gateway and Media Resources

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/ios/voice/cminterop/configuration/guide/12_4t/vc_12_4t_book/vc_ucm_mgcp_gw.html

NEW QUESTION: 276

An administrator installs a new Cisco Telepresence video endpoint and receives this error: "AOR is not permitted by Allow/Deny list." Which action should be taken to resolve this problem?

- A.** Correct the restriction policy settings.
- B.** Change the SIP trunk configuration.
- C.** Upload a new policy in VCS.
- D.** Reboot the VCS server and attempt reregistration.

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

https://www.cisco.com/c/en/us/td/docs/telepresence/infrastructure/articles/vcs_endpoint_registration_problems_kb_460.html

NEW QUESTION: 277

Refer to the exhibit. An administrator configures a secure SIP trunk on Cisco UCM. Which value is needed in the Secure Certificate Subject or Subject Alternate Name field to accomplish this task?

- A.** the common name of the remote device certificates
- B.** the common name of the Cisco UCM CallManager certificate
- C.** the fully qualified domain name of all Cisco UCM nodes that run the CallManager service
- D.** the fully qualified domain name of the remote device that is configured on the SIP trunk

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 278

What is the major difference between the two possible Cisco IM and Presence high-availability modes?

- A.** Balanced mode does not provide user load balancing, but it provides in the event of an outage. Active/ standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

B. Balanced mode provides user load balancing and user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, and it also provides load balancing.

C. Balanced mode provides user load balancing and user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

D. Balanced mode provides user load balancing and user failover only for manually generated failovers.

Active/standby mode provides an unconfigured standby node in the event of an outage, but it does not provide load balancing.

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

NEW QUESTION: 279

Which description of the Mobile and Remote Access feature is true?

A. Collaboration Edge feature that enables remote individuals to access all enterprise collaboration services using a PC within the corporate environment

B. Collaboration Edge feature that enables remote individuals to perform international calls from Jabber with a VPN connection

C. Collaboration Edge feature that enables remote individual to access enterprise collaboration services via Jabber with the use of a VPN connection

D. Collaboration Edge feature that enables remote individuals to access enterprise collaboration services via Jabber without the use of a VPN connection

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

NEW QUESTION: 280

A customer wants a video conference with five Cisco TelePresence IX5000 Series systems.

Which media resource is necessary in the design to fully utilize the immersive functions?

A. Cisco Meeting Server

B. software conference bridge on Cisco Unified Communications Manager

C. Cisco PVD4-128

D. Cisco Webex Meetings Server

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

NEW QUESTION: 281

Refer to the exhibit.

Refer to the exhibit. An administrator must fix the SRV records to ensure that server1.

sample.com is always contacted first from the three servers. Which solution should the engineer apply to resolve this issue?

A. Priority = 10, Weight = 5

B. Priority = 10, Weight = 10

C. Priority = 5, Weight = 70

D. Priority = 100, Weight = 90

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

NEW QUESTION: 282

Refer to the exhibit.

Which two codec permutations should be transcoded by this dspfarm? (Choose two)

A. G.729ar8 to G.711alaw

B. G.722 to G.729r8

C. iLBC to G.711ulaw

D. G.729r8 to G.711ulaw

E. G.729br8toG.711alaw

Answer: C,E ([LEAVE A REPLY](#))

NEW QUESTION: 283

Which data security feature is included with the Pro Pack for Cisco Webex Control Hub?

A. encryption in transit

B. encryption of content

C. Hybrid Data Security end-to-end

D. end-to-end encryption for meetings

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

NEW QUESTION: 284

What is the importance of locations when using location-based CAC within Cisco UCM?

A. They define the geographic regions to which resources are assigned.

B. They define the maximum bandwidth of the codec that is used for a call.

C. They define the minimum bandwidth of the codec that is used for a call.

D. They define the amount of available bandwidth for the link.

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

NEW QUESTION: 285

Which two DNS records must be created to configure Service Discovery for or premises Jabber? (Choose two.)

A. _cisco-uds._tls.cisco.com pointing to the IP address of Cisco Unified Communications Manager

B. _cuplogin._tcp.cisco.com pointing to a record of IM&P

C. _cuplogin._tls.cisco.com pointing to the IP address of IM&P

D. _cisco-uds._tcp.cisco.com pointing to a record of Cisco Unified CM

E. _xmpp._tls.cisco.com pointing to a record of IM&P

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

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/jabber/Windows/9_7/

CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide/CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide_chapter_010.html

NEW QUESTION: 286

Which Cisco Unified Communications Manager configuration is required for SIP MWI integrations?

- A. Select "Redirecting Diversion Header Delivery - Inbound" on the SIP trunk.
- B. Enable "Accept presence subscription" on the SIP Trunk Security Profile.
- C. Enable "Accept unsolicited notification" on the SIP Trunk Security Profile.
- D. Select "Redirecting Diversion Header Delivery - Outbound" on the SIP trunk.

Answer: (SHOW ANSWER)

Section: Collaboration Applications

Explanation

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<https://www.trainingdump.com/Cisco/350-801-practice-exam-dumps.html> (364 Q&As Dumps, **40%OFF Special Discount: Exam-Tests**)

NEW QUESTION: 287

what causes poor voice quality and video pixelization in a video call?

- A. Cisco Unified Communications Manager is configured to use G.711 instead G.729.
- B. 1 Gbps network ports are used instead of 100 Mbps network ports.
- C. The QoS is configured incorrectly.
- D. A firewall is blocking the RTP ports.

Answer: C (LEAVE A REPLY)

NEW QUESTION: 288

Which DTMF relay method configured on a SIP dial-peer will ensure that a media resource is not invoked by Unified CM for calls to UCCX IVRs?

- A. dtmf-relay slp-kpml
- B. dtmf-relay rtp-nte
- C. sdtmf-relay h245-signal
- D. dtmf-relay sip-notify

Answer: A (LEAVE A REPLY)

UCCX only supports out-of-band DTMF. This can be applied at the CUBE to avoid a media resource being needed to address as it can directly handle the DTMF relay from RTP-NTE to SIP-KPML

https://www.cisco.com/c/dam/en/us/td/docs/ios-xml/ios/voice/cube_uccx/cube_uccx_interop_bestprac_guide.pdf

NEW QUESTION: 289

Which settings are needed to configure the SIP route pattern in Cisco Unified Communications Manager?

- A. Pattern usage, IPv4 pattern, IPv6 pattern, and description
- B. Pattern usage, IPv4 pattern, and SIP trunk/Route list
- C. Pattern usage, IPv6 pattern, and SIP trunk/Route list
- D. SIP trunk/Route list, description, and IPv4 pattern

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmcfg/CUCM_BK_C95ABA82_00_admin-guide-100/CUCM_BK_C95ABA82_00_admin-guide-100_chapter_0100111.pdf

NEW QUESTION: 290

When designing location-based CAC, how much overprovisioning is calculated for video bandwidth?

- A. 15%
- B. 10%
- C. 20%
- D. 25%

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

NEW QUESTION: 291

Given this H.323 gateway configuration and using cisco best practices, how must the called party transformation pattern to ensure that a proper ISDN type of number is set?

- A)
- B)
- C)
- D)

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

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

NEW QUESTION: 292

A company deploys centralized cisco ucm architecture for a hub location and two remote sites.

- The company has only one ITSP connection at the hub connection, and ITSP supports only G.711 calls
- Remote site A has a 1-Gbps fiber connection to the hub connection and calls to and from remote side A use G.711 codec
- Remote site B has a 1 T1 connection to the hub location and calls to and from remote site B use G.729 codec

Based on the provided guidance, a Cisco voice engineer must design media resource management for the customer. What is the method that needs to be followed?

- A.** configure the software transcoder on Cisco UCM to support voice calls to and from both remote sites.
- B.** configure the hardware transcoder on the hub location router.
- C.** configure the hardware transcoder on the site A router.
- D.** configure the hardware transcoder on the site B router.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 293

A user dials 9011841234567 to reach Vietnam. Which steps send the call to the PSTN provider as 011841234567?

- A)
- B)
- C)
- D)
- A.** Option B
- B.** Option A
- C.** Option D
- D.** Option C

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

NEW QUESTION: 294

Which DTMF relay method configured on a SIP dial-peer will ensure that a media resource is not invoked by Unified CM for calls to UCCX IVRs?

- A.** dtmf-relay rtp-nte
- B.** dtmf-relay sip-notify
- C.** dtmf-relay slp-kpml
- D.** sdtmf-relay h245-signal

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

NEW QUESTION: 295

Refer to the exhibit.

Refer to the exhibit. What is the result of applying these two rules to a voice translation profile for use with an ISDN T1 PRI on a Cisco Voice Gateway?

- A. The ISDN Type is modified to the administrator's defined value.
- B. Any zero is stripped from the numeric phone number.
- C. The ISDN Plan is modified to the administrator's defined value.
- D. The leading Plus is stripped from the numeric phone number.

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

NEW QUESTION: 296

Which statement describes the outcome when the trust boundary is defined at the Cisco IP phone?

- A. Packets or Ethernet frames are not remarked by the IP phone.
- B. Packets or Ethernet frames are remarked at the distribution layer switch.
- C. Packets of Ethernet frames are remarked at the access layer switch.
- D. Packets of Ethernet frames are not remarked at the access layer switch.

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

In the picture above the trust boundary is at the Cisco IP phone, this means that we won't remark any packets or Ethernet frames anymore at the access layer switch.

<https://networklessons.com/quality-of-service/how-to-configure-qos-trust-boundary-on-cisco-switches>

NEW QUESTION: 297

Refer to the exhibit. What is the result of applying these two rules to a voice translation-profile for use with an ISDT1 PRI on a Cisco Voice gateway?

- A. The ISDN Type is modified to the administrators defined value.
- B. Any Zero is stripped from the numeric phone number.
- C. The ISDN Plan is modified to the administrators defined value.
- D. The leading Plus is stripped from the numeric phone number.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 298

An engineer is integrating Unity Connection with Cisco UCM. Which two actions must be configured so that recording and playback from the IP phones works at all times, including peak traffic hours? (Choose two.)

- A. The phone system to which the phones are registered in Unity Connection has the Default Trap Switch check box enabled.
- B. If it's a Unity Connection Cluster, ensure that replication is fine and not in split-brain mode.
- C. Add dedicated dial-out ports with the allow trap connections setting selected.
- D. Increase the number of voice ports.
- E. Ensure that you have set up SIP Digest Authentication on the SIP trunk security profile.

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

NEW QUESTION: 299

An engineer is trying to implement a new IP phone system in a company. The phones connected to the internal network are reporting a status of "unknown." There is no firewall between the Cisco UCM and the phones. What is a possible reason for the unknown status?

- A. DNS is not enabled for the phones
- B. Cisco Discovery Protocol is not enabled, and ping sweep is enabled
- C. DHCP option 150 is not set up correctly
- D. The IP phone management service is not enabled under Services

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

NEW QUESTION: 300

An administrator is integrating a Cisco Unity Express module to a Cisco UCME system. A test call is placed to the Cisco Unity Express pilot number, but the administrator receives a busy signal.

The dial-peer is configured as follows:

How is this issue resolved?

- A. The dial-peer needs to be reconfigured to support the G.711 voice codec.
- B. The dial-peer needs to be reconfigured to support vad.
- C. The destination pattern needs to be changed to match the dialed number of 4100.
- D. The dial-peer needs to be reconfigured to support H.323 instead of SIP.

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

"A voice activity detection (VAD) must be switched off and a codec must be designated."

<https://www.cisco.com/c/en/us/support/docs/voice-unified-communications/unity-express/62609-tdcmecue.html#anc17>

NEW QUESTION: 301

An administrator troubleshoots a DNS issue with Cisco UCM and observes that the hostname resolution returns the wrong IP address. The DNS team informs the administrator that the A/AAAA record was recently updated on the DNS server. Which action must be taken on Cisco UCM to reflect the correct entries?

- A. Flush the current cache using this CLI command: `admin:utils network name-service hosts cache validate`
- B. Restart the DNS server to flush any stale entries
- C. Delete the ARP entry of the host: `utils network arp delete`
- D. Flush the current cache using this CLI command: `admin:utils network name-service hosts cache invalidate`

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

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NEW QUESTION: 302

Refer to the exhibit. An engineer must deploy a Cisco Unified Border Element that will allow the last four digits of a dialed number to be sent from the Cisco Unified Border Element to Cisco UCM. The configuration in the exhibit already exists on the Cisco Unified Border Element. Which additional configuration must be performed on the Cisco Unified Border Element?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 303

How can an engineer determine location-based CAC bandwidth requirements for Cisco Unified communication Manager?

- A. Execute the Resource Reservation protocol to return location-based requirements.
- B. Calculate the number of calls against the license for Cisco Unified Border Element to determine calls per location.
- C. Set the requirements in the service parameters.
- D. Add the requirements for each audio and video codec and how many calls must be supported.

Answer: ([SHOW ANSWER](#))

CAC is configured in location, and is the total bandwidth for all calls.

NEW QUESTION: 304

In Cisco UCM, an engineer must create a new SIP route pattern to allow ccnp.cisco.com for all internal callers. The internal callers have a calling search space named CSSJINTERNAL that includes the partition named PTJINTERNAL. If allowed, the calls must be routed to a SIP trunk named SIP_CISCO_TRK. Drag and drop the values from the left onto the corresponding fields on the right.

Answer:

NEW QUESTION: 305

Refer to the exhibit.

Refer to the exhibit. A collaboration engineer is configuring the Cisco UCM IM and Presence Service. Which two steps complete the configuration of the SIP trunk security profile? (Choose two.)

- A. Check the box to accept replaces header.
- B. Check the box to enable application-level authorization.
- C. Check the box to allow charging header.
- D. Check the box to transmit security status.
- E. Check the box to accept unsolicited notification.

Answer: A,E ([LEAVE A REPLY](#))

NEW QUESTION: 306

Drag and Drop Question

Drag and drop the SNMPv3 message types from the left onto the corresponding definitions on the right.

Answer:

NEW QUESTION: 307

Which SNMP service must be activated manually on the Cisco Unified Communications Manager after installation?

- A. Connection SNMP Agent
- B. Host Resources Agent
- C. SNMP Master Agent
- D. Cisco CallManager SNMP

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 308

What is the major difference between the two possible Cisco IM and presence high-availability modes?

A. Balanced mode provides user load balancing and user failover only for manually generated failovers.

Active/standby mode provides an unconfigured standby node in the event of an outage, but it does not provide load balancing.

B. Balanced mode provides user load balancing and user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

C. Balanced mode does not provide user load balancing, but it provides user failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

D. Balanced mode provides user load failover in the event of an outage.

Active/standby mode provides an always on standby node in the event of an outage, and is also provides load balancing.

Answer: B (LEAVE A REPLY)

Balanced mode: This mode provides redundant high availability with automatic user load balancing and user failover in the event that one nodes fails because of component failure or power outage.

Active/standby mode: The standby node automatically takes over for the active node if the active node fails. It does not provide automatic load balancing.

NEW QUESTION: 309

What is a description of the DiffServ model used for implementing QoS?

- A. AF41 has higher drop precedence than AF42, which has higher drop precedence than AF43.
- B. Voice and video calls are marked with different DSCP values and placed in different queues.
- C. AF43 has higher drop precedence than AF42 but lower drop precedence than AF41.
- D. RTP traffic from voice and video calls is marked EF and placed in the same queue.

Answer: B (LEAVE A REPLY)

<https://www.cisco.com/c/en/us/support/docs/quality-of-service-qos/qos-packet-marking/10103-dscpvalues.html>

NEW QUESTION: 310

Which statement about Cisco Unified Communications Manager and Cisco IM and Presence backups is true?

- A. Backups should be scheduled during off-peak hours to avoid system performance issues.
- B. Backups are saved as .tar files and encrypted using the web administrator account.
- C. Backups are saved as unencrypted .tar files.
- D. Backups are not needed for subscriber Cisco Unified Communications Manager and Cisco IM and Presence servers.

Answer: (SHOW ANSWER)

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/11_5_1_SU1/Administration/cucm_b_administration-guide-1151su1/cucm_b_administration-guide-1151su1_chapter_01010.html#CUCM_TK_S7FC26D5_00

NEW QUESTION: 311

Refer to the exhibit. An administrator must fix the SRV records to ensure that server1.sample.com is always contacted first from the three servers. Which solution should the engineer apply to resolve this issue?

- A. Priority = 5, Weight = 70
- B. Priority = 10, Weight = 5
- C. Priority = 10, Weight = 10
- D. Priority = 100, Weight = 90

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 312

Refer the exhibit.

Given this "debug isdn q921" output, what is the problem with the PRI?

- A. PRI does not have an IP address configured on the interface.
- B. Noting, the PRI is sending keepalives.
- C. Layer 1 is down on the controller.
- D. Layer 2 is down on the controller.

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

NEW QUESTION: 313

An administrator needs to stop the leading 9 from outbound calls on an IOS Voice Gateway and ensure that the system handles 911 emergency calls Which configuration is needed to accomplish this task?

- A)
- B)
- C)
- D)
- A. Exhibit A
- B. Exhibit B
- C. Exhibit C
- D. Exhibit D

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

NEW QUESTION: 314

Which Webex Calling construct is used to organize calling features within a physical site?

- A. client settings
- B. locations
- C. service settings
- D. call routing

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

A location is a physical site that contains users, devices, and resources. Locations are used to organize calling features within a physical site. For example, you can create a location for each of your offices and then assign users, devices, and resources to that location. This will allow you to manage calling features for each office separately.

NEW QUESTION: 315

Which command is used in Cisco IOS XE TDM gateway to configure the voice T1/E1 controller to provide clocking?

- A. clock source line

- B. Cisco IOS XE TDM gateway T1/E1 controller cannot provide clocking.
C. clocking source internal
D. clocking source network

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

Explanation/Reference:

<https://www.cisco.com/c/en/us/td/docs/routers/access/interfaces/NIM/software/configuration/guide/4gen-t1-e1-nim-guide.html>

NEW QUESTION: 316

An engineer implements a new Cisco UCM based telephony system per these requirements:

- The local Ethernet bandwidth is sized based on the total bandwidth per call
- A G.736 codec is used.
- The bit rate is 64 kbps
- The codec sample interval is 10 ms.
- The voice payload size is 160 bytes per 20 ms.

What should the size of the Ethernet bandwidth be per call?

- A. 31.2 kbps
B. 38.4 kbps
C. 55.2 kbps
D. 87.2 kbps

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

All details match with G711 protocol, for which the calculation is :

$(160 + (40 + 18)) / 160 \times 64 = 87.2 \text{ kbps}$

<https://www.cisco.com/c/en/us/support/docs/voice/voice-quality/7934-bwidth-consume.html>

Codec Information Bandwidth Calculations Codec & Bit Rate (Kbps) Codec Sample Size (Bytes) Codec Sample Interval (ms) Mean Opinion Score (MOS) Voice Payload Size (Bytes) Voice Payload Size (ms) Packets Per Second (PPS) Bandwidth MP or FRF.12 (Kbps) Bandwidth w/cRTP MP or FRF.12 (Kbps) Bandwidth Ethernet (Kbps)

G.711 (64 Kbps) 80 Bytes 10 ms 4.1 160 Bytes 20 ms 50 82.8 Kbps 67.6 Kbps 87.2 Kbp

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NEW QUESTION: 317

Due to provider requirements, outgoing calls from the Enterprise to the PSTN must start with channel 1.

Which ISDN command changes the channel selection an IOS to meet this requirement?

- A. isdn protocol-emulate network
- B. isdn bchan-number-order ascending
- C. isdn incoming-voice voice
- D. isdn bchan-number-order decending

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

NEW QUESTION: 318

An engineer with troubleshoots poor voice quality on multiple calls. After looking at packet captures, the engineer notices high levels of jitter. Which two areas does the engineer check to prevent jitter? (Choose two.)

- A. The network meets bandwidth requirements
- B. Voice packets are classified and marked
- C. MTP is enable on the sip trunk to cisco Unified Border Elemnet
- D. Cisco UBE manages voice traffic, not data traffic
- E. All devices use wired connections instead of wireless connections

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 319

Refer to the exhibit. A collaboration engineer is configuring the Cisco UCM IM and Presence Service. Which two steps complete the configuration of the SIP trunk security profile? (Choose two.)

- A. Check the box to enable application-level authorization.
- B. Check the box to transmit security status.
- C. Check the box to accept replaces header.
- D. Check the box to accept unsolicited notification.
- E. Check the box to allow charging header.

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 320

Which two configuration elements are part of the Cisco Unified Communications Manager toll-fraud prevention? (Choose two.)

- A. SIP trunk security profile
- B. Calling Search Space
- C. SUBSCRIBE Calling Search Space
- D. feature control policy
- E. partition

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

Section: Call Control

Explanation/Reference:

NEW QUESTION: 321

Users dial a 9 before a 10-digit phone number to make an off-net call. All 11 digits are sent to the Cisco Unified Border Element before going out to the PSTN. The PSTN provider accepts only 10 digits. Which configuration is needed on the Cisco Unified Border Element for calls to be successful?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 322

Which two types of device are supported by the Bulk Administration Tool? (Choose two.)

- A. H.322
- B. Cisco Unified IP phones (all models)
- C. SIP trunks
- D. H.225 trunks
- E. music on hold servers

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

Section: Protocols, Codecs, and Endpoints

NEW QUESTION: 323

Refer to the exhibit.

An administrator must fix the SRV records to ensure that server1.sample.com is always contacted first from the three servers. Which solution should apply to resolve this issue?

- A. Priority = 10, Weight = 5
- B. Priority = 100, Weight = 90
- C. Priority = 10, Weight = 10
- D. Priority = 5, Weight = 70

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 324

An engineer deploys a Cisco Expressway-E server for a customer who wants to utilize all features on the server. Which feature does the engineer configure on the Expressway-E?

- A. SIP gateway for PSTN providers
- B. H.323 endpoint registrations
- C. Mobile and Remote Access
- D. VTC bridge

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 325

Which action prevents toll fraud in Cisco Unified Communications Manager?

- A. Implement toll fraud restriction in the Cisco IOS router.
- B. Implement route patterns in Cisco Unified CM.
- C. Allow off-net to off-net transfers.
- D. Configure ad hoc conference restriction.

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

Explanation/Reference: <https://www.cisco.com/c/en/us/support/docs/voice-unified-communications/unified-communications-manager-express/107626-cme-toll-fraud.html>

NEW QUESTION: 326

Refer to the exhibit.

This INVITE is sent to an endpoint that only supports G.729. what must be done for this call to succeed?

- A. Noting; both sides support G.729.
- B. ADD a media termination point that supports G.711ulaw and G.729.
- C. Nothing; both sides support payload type.101.
- D. Add a transcoder that supports G.711ualw and G.729.

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

NEW QUESTION: 327

Which version is used to provide encryption for SNMP management traffic in collaboration deployments?

- A. SNMPv1
- B. SNMPv3
- C. SNMPv2
- D. SNMPv2c

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

<https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/snmp/configuration/xe-16/snmp-xe-16-book/nm-snmp-encrypt-snmp-support.html>

NEW QUESTION: 328

Which wildcard must an engineer configure to match a whole domain in SIP route patterns?

- A. *
- B. @
- C. .
- D. !

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/10_0_1/ccmcfg/CUCM_BK_C95ABA82_00_admin-guide-100/CUCM_BK_C95ABA82_00_admin-guide-

NEW QUESTION: 329

Refer to the exhibit.

There is a call flow between Phone A and Phone B Phone B (holder) places Phone A (holder) on hold Which MRGL and Audio Source are played to Phone A?

- A. MRGL A and Audio Source 2
- B. MRGL A and Audio Source 4
- C. MRGL B and Audio Source 4
- D. MRGL B and Audio Source 2

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

NEW QUESTION: 330

An engineer wants to manually deploy a Cisco Webex DX80 video endpoint to a remote user.

Which type of provisioning is configured on the endpoint?

- A. CUBE
- B. CMS
- C. CUCM
- D. Edge

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

The provisioning modes/types are Off/Auto/CUCM/Edge/Spark/TMS/VCS.

https://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/ce95/dx70-dx80-administrator-guide-ce95.pdf#D1536209_DX70-DX80_Administrator_Guide_CE95.indd%3A.859768%3A307505

NEW QUESTION: 331

Which two DNS records must be created to configure Service Discovery for on-premises Jabber? (Choose two.)

- A. _cuplogin._tcp.cisco.com pointing to a record of IM&P
- B. _cisco-uds._tcp.cisco.com pointing to a record of Cisco Unified CM
- C. _cuplogin._tls.cisco.com pointing to the IP address of IM&P
- D. _cisco-uds._tls.cisco.com pointing to the IP address of Cisco Unified Communications manager
- E. _xmpp._tls.cisco.com pointing to a record of IM&P

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/jabber/Windows/9_7/CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide/CJAB_BK_C606D8A9_00_cisco-jabber-dns-configuration-guide_chapter_010.html

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NEW QUESTION: 332

Refer to the exhibit.

When making a call to a MRA client, what are the combinations of protocol on each of the different sections A-B-C?

- A. SIP TLS (A) +SIP TLS (B) +SIP TLS (C)
- B. SIP TCP/TLS (A) + SIP TLS (B) + SIP TLS (C)
- C. SIP TCP/TLS +SIP TCP/TLS (B) + SIP TCP/TLS (C)
- D. IP TCP/TLS(A) +SIP TCP/TLS (B) +TLS (C)

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 333

An engineer must deploy the Cisco Webex app to a Virtual Desktop Infrastructure (VDI) environment and support full-featured meetings. Which action must the engineer take to meet these requirements?

- A. Configure the VDI thin client and install the Webex app
- B. Install three separate VDI plugins on the VDI thin client
- C. From Control Hub, configure VDI optimization for the Webex app
- D. Install two separate VDI plugins on the VDI thin client

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 334

Which command is used in Cisco IOS XE TDM gateway to configure the voice T1/E1 controller to provide clocking?

- A. clock source line
- B. Cisco IOS XE TDM gateway T1/E1 controller cannot provide clocking.
- C. clock source internal
- D. clock source network

Answer: ([SHOW ANSWER](#))

Section: Cisco IOS XE Gateway and Media Resources

Explanation/Reference:

<https://www.cisco.com/c/en/us/td/docs/routers/access/interfaces/NIM/software/configuration/guide/4gen-t1-e1-nim-guide.html>

NEW QUESTION: 335

An administrator is configuring a new Cisco UCM with PSTN capabilities. Due to bandwidth constraints, audio compression is used on the codec. DTMF must work as expected because the customer is calling many call centers where the users must select options in the call. Where is DTMF out-of-band in a CCM 12.5 with SIP-based gateway configured?

- A. in the DTMF setting under SIP profile on the Cisco Unified Border Element
- B. in DTMF settings in the audio codec preference list under regions in the Cisco UCM
- C. in the dial peer on the Cisco IOS router
- D. in regions on the Cisco UCM where the appropriate codec to use is set

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

NEW QUESTION: 336

Refer to the exhibit. A call from +1 613 555 1234 that is sent out through the Nice Gateways should result in a calling party of 001 613 555 1234 with the numbering type "international". Which configuration accomplishes this goal?

- A. \+.! pre-dot 00 international
- B. \+1 ! none pre-dot 001 international
- C. 613XXXXXXX none +011 international
- D. \+ 0011 pre-dot 1 international

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

NEW QUESTION: 337

An engineer must manually provision a Cisco IP Phone 8845 using SIP. Which two fields must be configured for a successful provision? (Choose two.)

- A. Media resources group list
- B. CSS
- C. SIP profile
- D. Location
- E. Device security profile

Answer: C,E ([LEAVE A REPLY](#))

NEW QUESTION: 338

According to the QoS Baseline Model, drag and drop the applications from the left onto the correct per-Hop Behaviour values on the right.

Answer:

Explanation

Voice = EF

Interactive Video = AF41

Bulk Data = AF11

Call-Signaling = AF31/CS3

Network Management = CS2

https://www.cisco.com/c/en/us/td/docs/solutions/Enterprise/WAN_and_MAN/QoS_SRND/QoS-SRND-Book/Qo

NEW QUESTION: 339

Which Cisco Unity Connection call management element can be used to answer calls, provide callers with a list of menu options, and transfer calls?

- A. call handler
- B. call routing table
- C. interview handler
- D. directory handler

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

NEW QUESTION: 340

An engineer must deploy the Cisco Web*ex app to a Windows Virtual Desktop Infrastructure environment that has a roaming database named spark roaming_store stored in a user's AppData\Roaming directory. Which two command line arguments must be used when running the installer? (Choose two.)

- A. ALLUSERS=0
- B. ENABLEVDI=1
- C. ALLUSERS=1
- D. ENABLEVDI=2
- E. ROAMINGENABLED=1

Answer: ([SHOW ANSWER](#)**)**

The Cisco Webex app can be installed on a Windows Virtual Desktop Infrastructure (VDI) environment by using the following command-line arguments:

ENABLEVDI=1 - This argument enables VDI mode for the Webex app.

ROAMINGENABLED=1 - This argument enables roaming for the Webex app.

The ALLUSERS argument is not required when installing the Webex app on a VDI environment.

The ENABLEVDI argument must be set to 1, and the ROAMINGENABLED argument must be set to 1.

The following is an example of the command that can be used to install the Webex app on a VDI environment:

Code snippet

```
msiexec /i WebexApp.msi ENABLEVDI=1 ROAMINGENABLED=1
```

NEW QUESTION: 341

A collaboration engineer must configure Cisco Unified Border Element to support up to five concurrent outbound calls across an Ethernet Link with a bandwidth of 160 kb to the Internet Telephony service provider.

Which set of commands allows the engineer to complete the task without compromising voice quality?

- A)
- B)
- C)
- D)

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

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

NEW QUESTION: 342

An incoming off-net call to a user fails. An engineer notices that the off-net call is G.711, but the phone accepts only G.729. Which media resource on a Cisco Unified Border Element and Cisco Unified Communications Manager must the engineer configure to manage the codec negotiation?

- A. transcoder
- B. MTP
- C. MOH
- D. CFB

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

NEW QUESTION: 343

In the cisco expressway solution, which two features does mobile and Remote access provide? (Choose two)

- A. VPN-based enterprise access for a subset of Cisco Unified IP Phone models
- B. Secure reverse proxy firewall travelsal connectivity
- C. the ability to register third-party SIP or H.323 devices on Cisco UCM without requiring VPN
- D. the ability of Cisco IP Phones to access the enterprise through VPN connection
- E. the ability for remote users and their devices to access and consume enterprise collaboration applications and services

Answer: ([SHOW ANSWER](#)**)**

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/expressway/config_guide/X12-5/exwy_b_mra-expressway-deployment-guide/exwy_b_mra-expressway-deployment-guide_chapter_00.html

NEW QUESTION: 344

Location-based call admission control is used in Cisco UCM to control the amount of bandwidth allowed between predefined locations. What must be considered when calculating the number of calls that can be made?

- A. A limit of 900 kb is the total amount of actual used RTP traffic so the oversubscription can be done for multiple calls adding its maximum RTP stream requirements.

- B.** A limit of 900 kb is the total amount of bandwidth used for the sum of the media and signaling in the calls. The system allows for an actual of 20% overhead of IP packet headers. This is to compare the use of ISDN lines with no IP overhead.
- C.** A limit of 900 kb is the total amount of bandwidth used for the sum of calls. The use of compression does not impact the amount of calls because only the codec definition is considered when calculating the CAC in Cisco UCM.
- D.** A limit of 900 kb is the total amount of bandwidth used for the sum of calls of the defined codec including headers and media traffic.

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

NEW QUESTION: 345

Refer to the exhibit.

Refer to the exhibit. Given this "debug isdn q921" output, what is the problem with the PRI?

- A.** Layer 1 is down on the controller.
- B.** PRI does not have an IP address configured on the interface.
- C.** Nothing, the PRI is sending keepalives.
- D.** Layer 2 is down on the controller.

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

NEW QUESTION: 346

What is the major difference between the two possible Cisco IM and presence high-availability modes?

- A.** Balanced mode does not provide user load balancing, but it provides user failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.
- B.** Balanced mode provides user load balancing and user failover only for manually generated failovers.

Active/standby mode provides an unconfigured standby node in the event of an outage, but it does not provide load balancing.

- C.** Balanced mode provides user load failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, and it also provides load balancing.
- D.** Balanced mode provides user load balancing and user failover in the event of an outage. Active/standby mode provides an always on standby node in the event of an outage, but it does not provide load balancing.

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

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NEW QUESTION: 347

How can an administrator stop Cisco Unified Communications Manager from advertising the OPUS codec for recording enabled devices?

- A. Go to the phone's configuration page and set "Advertise OPUS Codec" to be "false"
- B. Router recorded calls through Cisco Unified Border Element because it does not support OPUS.
- C. In CUCM Service Parameters set "Opus codec Enables" to "Enabled for all devices Except Recording-Enabled devices."
- D. Integrate the Cisco Unified CM with a recording solution that does not support OPUS.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 348

Which two elements of a dial plan define the domains that are accessible and are assigned to an endpoint?

(Choose two.)

- A. Call Admissions Control
- B. partitions
- C. Route patterns
- D. Translation patterns
- E. Calling Search Spaces

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

NEW QUESTION: 349

Which IP Precedence value is used to classify a call signalling packet?

- A. 6
- B. 5
- C. 3
- D. 4

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

NEW QUESTION: 350

A company deploys centralized cisco ucm architecture for a hub location and two remote sites. The company has only one ITSP connection at the hub connection, and ITSP supports only G.711 calls Remote site A has a 1-Gbps fiber connection to the hub connection and calls to and from remote side A use G.711 codec Remote site B has a 1 T1 connection to the hub location and calls to and from remote site B use G.729 codec Based on the provided guidance, a Cisco

voice engineer must design media resource management for the customer. What is the method that needs to be followed?

- A. configure the hardware transcoder on the site B router.
- B. configure the hardware transcoder on the site A router.
- C. configure the hardware transcoder on the hub location router.
- D. configure the software transcoder on Cisco UCM to support voice calls to and from both remote sites.

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

NEW QUESTION: 351

An engineer is integrating Unity Connection with Cisco UCM. Which two actions must be configured so that recording and playback from the IP phones works at all times, including peak traffic hours? (Choose two.)

- A. If it's a Unity Connection Cluster, ensure that replication is fine and not in split-brain mode.
- B. Increase the number of voice ports.
- C. Add dedicated dial-out ports with the allow trap connections setting selected.
- D. Ensure that you have set up SIP Digest Authentication on the SIP trunk security profile.
- E. The phone system to which the phones are registered in Unity Connection has the Default Trap Switch check box enabled.

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 352

An engineer is configuring Cisco Jabber for Windows and must implement desk phone control mode for some of the users.

Which access control group must be configured for those users to enable this functionality?

- A. Standard CTI Enabled
- B. Standard CTI Allow Reception of SRTP Key Material
- C. Allow Control of Device from CTI
- D. Standard CTI Secure Connection

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

NEW QUESTION: 353

Refer to the exhibit.

A collaboration engineer needs to replace the original, single NTP server that was configured during the initial install of a Cisco UCM server. What is the first step to accomplish this task?

- A. Restart the NTP service on Cisco UCM.
- B. Stop the NTP service on Cisco UCM.
- C. Enable NTP authentication for the new NTP server on Cisco UCM.
- D. Delete the original NTP server from Cisco UCM

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 354

An engineer is designing a load balancing solution for two Cisco Unified Border routers. The first router (cube1.abc.com) takes 60% of the calls and the second router (cube2.abc.com) takes 40% of the calls. Assume all DNS A records have been created. Which two SRV records are needed for a load balanced solution? (Choose two.)

- A. sip_udp.abc.com 60 IN SRV 3 60 5060 cube2.abc.com
- B. sip_udp.abc.com 60 IN SRV 2 60 5060 cube1.abc.com
- C. sip_udp.abc.com 60 IN SRV 1 40 5060 cube2.abc.com
- D. sip_udp.abc.com 60 IN SRV 1 60 5060 cube1.abc.com
- E. sip_udp.abc.com 60 IN SRV 60 1 5060 cube1.abc.com

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 355

Which action enables Cisco MRA?

- A. Clients such as Cisco Jabber can use call control on Cisco UCM.
- B. Internal SIP clients registered to Cisco UCM can call external companies
- C. VPN connectivity can be established to Cisco UCM.
- D. Cisco UCC Express clients can obtain VPN connectivity to Cisco UCC Enterprise.

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

NEW QUESTION: 356

Refer to the exhibit.

A call is failing to establish between two SIP Devices. The Called device answer with this SDP. Which SDP parameter causes this issue?

- A. The media stream is set to sendonly.
- B. The calling device did not offer aptime value.
- C. The payload for G.711ulaw must be 18.
- D. The RTP port is set to 0.

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

NEW QUESTION: 357

Which packet delay is the maximum supported between Cisco Unified Communications Manager nodes for clustering over WAN deployments?

- A. 150 ms round trip
- B. 510 ms round trip
- C. 40 ms round trip
- D. 80 ms round trip

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

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab11/collab11/callpros.html

NEW QUESTION: 358

An engineer troubleshoots a Cisco Jabber login problem on a Windows PC in the corporate network. The login fails with the error message "Cannot find your services automatically. Click advanced settings to set up manually." Which action should the engineer take first?

- A. Verify whether the cup-xmpp certificates are valid.
- B. Perform a manual DNS lookup of SRV record _collab-edge._tls.domain.com.
- C. Perform a manual DNS lookup of SRV record _cisco-uds.tcp.domain.com.
- D. Verify the username and password and try again

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

NEW QUESTION: 359

Endpoint A is attempting to call endpoint

B. Endpoint A only supports G.711ulaw with a packetization rate of 20 ms, and endpoint B supports packetization rate of 30 ms for G.711ulaw. Which two media resources are allocated to normalize packetization rates through transrating? (Choose two.)

- A. software transcoder on Cisco UCM
- B. hardware transcoder on Cisco IOS Software
- C. software MTP on Cisco IOS Software
- D. software MTP on Cisco UCM
- E. hardware MTP on Cisco IOS Software

Answer: ([SHOW ANSWER](#)**)**

Section: Protocols, Codecs, and Endpoints

NEW QUESTION: 360

Which Cisco Collaboration Edge architecture product allows remote endpoints to leverage corporate on-premises Cisco Unified Communications infrastructure?

- A. Cisco Umbrella
- B. Cisco Webex
- C. Cisco Unified Communications Mobile and Remote Access
- D. Cisco VPN Client

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

NEW QUESTION: 361

Refer to the exhibit. How must the +E.164 translation pattern be configured to reach international number

496929810?

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

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

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NEW QUESTION: 362

In a Cisco Unified Communications Manager design where +E.164 destinations are populated in directory entries, which call routing practice is critical to prevent unnecessary toll charges caused by internal calls routed through the PSTN?

- A. tail-end hop-off
- B. forced on-net routing
- C. forced authorization codes
- D. automated alternate routing
- E. client matter codes

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 363

Which two actions must be taken to provision a new device using self-provisioning?

- A. Link the appropriate service profile to the provisioning template.
- B. Link the appropriate universal device template to the user profile.
- C. Enable the self-provisioning IVR in the Cisco UCM
- D. Import the user profile to the corporate LDAP directory.
- E. Ensure the user has a directory URI and a primary extension.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 364

Which method is used to avoid toll fraud with Cisco Unified Communications Manager calls?

- A. call policy service
- B. TOLLFRAUD_APP
- C. default zone access rules
- D. class of service

Answer: ([SHOW ANSWER](#))

Section: Call Control

NEW QUESTION: 365

Refer the exhibit. Given this "debug isdn q921" output, what is the problem with the PRI?

- A. Layer 1 is down on the controller.
- B. PRI does not have an IP address configured on the interface.
- C. Noting, the PRI is sending keepalives.
- D. Layer 2 is down on the controller.

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

If Layer 2 is stable, the router and switch must begin to synchronize with each other. The Set Asynchronous Balanced Mode Extended (SABME) message appears on the display. This message indicates that Layer 2 tries to initialize with the other side. Either side can send the message and try to initialize with the other side. If the router receives the SABME message, it must send back an Unnumbered Acknowledge frame (UAf).

<https://www.cisco.com/c/en/us/support/docs/wan/t1-e1-t3-e3/8131-T1-pri.html>

NEW QUESTION: 366

An administrator executes the debug isdn q931 command while debugging a failed call.

After a test call is placed, the logs return a disconnect cause code of 1.

What is the cause of this problem?

- A. The destination number rejects the call
- B. The destination number is busy
- C. The media resource is unavailable
- D. The dialed number is not assigned to an endpoint

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

<https://www.cisco.com/c/en/us/support/docs/dial-access/integrated-services-digital-networks-isdn-channel-associated-signaling-cas/8607-isdn-disc-code.html>

NEW QUESTION: 367

Endpoint A is attempting to call endpoint B. Endpoint A only supports G.711 ulaw with a packetization rate of

20 ms, and endpoint B supports packetization rate of 30 ms for G.711ulaw. Which two media resources are allocated to normalize packetization rates through transrating?

- A. Software transcoder on Cisco unified Communications manager
- B. Software MTP on Cisco IOS Software
- C. Software MTP on Cisco Unified Communication Manager
- D. Hardware MTP on Cisco IOS Software
- E. Hardware transcoder on Cisco IOS Software

Answer: A,C ([LEAVE A REPLY](#))

NEW QUESTION: 368

Refer to the exhibit.

Refer to the exhibit. An administrator accesses the terminal of a Cisco UCM and starts a packet capture. Which two commands must the administrator use on Cisco UCM to generate DNS traffic? (Choose two.)

- A. show cdp neighbor
- B. utils diagnose module validate Network
- C. utils ntp status
- D. utils diagnose test
- E. show version active

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

NEW QUESTION: 369

An administrator works with an ISDN PRI that is connected to a third-party PBX. The ISDN link does not come up. and the administrator finds that the third-party PBX uses the OSIG signaling method. Which command enables the Cisco IOS Gateway to use QSIG signaling on the ISDN link?

- A. isdn switch-type basic-ni
- B. isdn switch-type basic-qsig
- C. isdn switch-type primary-qsig
- D. isdn incoming-voice voice

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

NEW QUESTION: 370

An employee of company ABC just quit. The IT administrator deleted the employee's user id from the active directory at 10 a.m. on March 4th. The nightly sync occurs at 10 p.m. daily. The IT administrator wants to troubleshoot and find a way to delete the user id as soon as possible. How is this issue resolved?

- A. Wait until 10 pm on March 4th when the user is automatically removed from Cisco UCM
- B. Wait until 10 pm on March 5th when the user is automatically removed from Cisco UCM.
- C. Wait until 3:15 a.m. on March 6th for garbage collection to remove the user from Cisco UCM.
- D. Wait until 3:15 a.m. on March 5th for garbage collection to remove the user from Cisco UCM

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

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab11/collab11/directry.html#pgfId-1045229

NEW QUESTION: 371

Refer to the exhibit. A Cisco UCM user with directory number 4401 dials 5507 and the call is routed to a Cisco Unified Border Element. Which IP address will the call be sent to?

- A. 10.0.0.3
- B. 10.0.0.1
- C. 10.0.0.2
- D. 10.0.0.4

Answer: ([SHOW ANSWER](#))

Once an incoming call is matched by an inbound dial peer with an active destination dial-peer group, dial peers from this group are used to route the incoming call.

No other outbound dial-peer provisioning to select outbound dial peers is used.
A preference can be defined for each dial peer in a dial-peer group.

NEW QUESTION: 372

What is set when using CoS to mark an Ethernet frame?

- A. IPP bits
- B. DCSP bits
- C. IP ECN bits
- D. 802.1p User Priority bits

Answer: ([SHOW ANSWER](#))

Section: QoS

NEW QUESTION: 373

User A calls user B.

The call gets connected, but the quality is bad. What are two reasons for this issue? (Choose two.)

- A. Incorrect partition
- B. Incompatible codec
- C. Incorrect QoS
- D. Network congestion
- E. No region relationship

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 374

Refer to the exhibit. Calls to Cisco Unity Connection are failing across Cisco Unified Border Element when callers try to select a menu prompt. Why is this happening and how is it fixed?

- A. Cisco Unified Border Element is not sending any support for DTMF. DTMF configuration must be added to the appropriate dial peer.
- B. Cisco Unified Border Element is sending the incorrect media IP address. The IP address of the loopback interface must be advertised in the SDP.
- C. Cisco Unity Connection is configured on G.729 only. Cisco Unity Connection must be reconfigured to support iLBC.
- D. The Cisco Unity Connection Call Handler is configured for a -Release to Switch" transfer type. Transfers must be disabled for the Cisco Unity Connection Call Handler.

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 375

A remote office has a less-than-optimal WAN connection and experiences packet loss, delay and jitter. Which VoIP codec is used in this situation?

- A. G.711alaw
- B. G.729A

C. G722.1

D. iLBC

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

NEW QUESTION: 376

Where is the default for Maximum Session Bit Rate for a region configured?

A. Service Parameter Configuration

B. Enterprise Phone Configuration

C. Enterprise Parameters Configuration

D. Region Configuration

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

Explanation/Reference:

https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/9_1_1/ccmcfg/CUCM_BK_A34970C5_00_admin-guide-91/CUCM_BK_A34970C5_00_admin-guide-91_chapter_0111.html

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<https://www.trainingdump.com/Cisco/350-801-practice-exam-dumps.html> (364 Q&As Dumps, **40%OFF Special Discount: Exam-Tests**)

NEW QUESTION: 377

Refer to the exhibit.

Refer to the exhibit. Which action must an engineer take to implement self-provisioning on a primary communications manager server?

A. Change the SIP Phone Secure Port.

B. Uncheck the auto-registration Disabled checkbox.

C. Select a different Universal Line Template.

D. Select a different Universal Device Template.

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

NEW QUESTION: 378

An engineer configures Cisco Unified Communications Manager to prevent toll fraud. At which two points does the engineer block the pattern in Cisco Unified CM to complete this task?

(Choose two.)

A. route pattern

B. route group

- C. translation pattern
- D. partition
- E. CSS

Answer: C,E ([LEAVE A REPLY](#))

Section: Call Control

NEW QUESTION: 379

Which configuration tells a switch port to send Cisco Discovery protocol packets that configure an attached Cisco IP phone to trust tagged traffic that is received from a device that is connected to the access port on the Cisco IP phone?

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

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

NEW QUESTION: 380

An administrator configured SIP route patterns cisco com and uc(5-7] cisco 1 com in Cisco UCM. A route pattern must be added for ucce2.cisco1.com. Which two patterns complete the configuration*? (Choose two.)

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

Answer: C,D ([LEAVE A REPLY](#))

NEW QUESTION: 381

Refer to the exhibit A user takes a phone from San Francisco to New York for a short reassignment. The phone was set up to use the San Francisco device pool, and device mobility is enabled on the Cisco UCM. The user makes a call that matches a route pattern in a route list that contains the Standard Local Route Group. To where does the call retreat?

- A. The call fails because the Standard Local Route Group is being used only if no configuration is set for the device pools
- B. The call egresses in San Francisco because the user uses device mobility and is allowed to roam while still keeping the number and resources assigned in San Francisco

C. The call fails because device mobility is turned on. and the phone is not configured in New York The engineer must configure which sites the device should be roaming to

D. The call egresses in New York because the device automatically is assigned a New York device pool and uses the local gateway

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

NEW QUESTION: 382

Which transport protocol does the application layer protocol SNMP use?

A. UDP

B. SIP

C. HTTP

D. XML

Answer: ([SHOW ANSWER](#)**)**

NEW QUESTION: 383

An engineer is setting up a system with voice and video endpoints using auto-QoS policy on the switches.

Which DSCP values are expected for interactive voice and video?

A. EF AND AF21

B. EF AND AF41

C. EF AND CS3

D. EF AND CS6

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

NEW QUESTION: 384

Users dial a 9 before a 10-digit phone number to make an off-net call. All 11 digits are sent to the Cisco Unified Border Element before going out to the PSTN The PSTN provider accepts only 10 digits. Which configuration is needed on the Cisco Unified Border Element for calls to be successful?

A. Option C

B. Option B

C. Option A

D. Option D

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

NEW QUESTION: 385

Based on the provided guidance a cisco voice engineer must design media resource management for the customer what is the method that needs to be followed?

A. configure the hardware transcoder on the site B router.

B. configure the hardware transcoder on the site A router

C. configure the software transcoder on Cisco UCM to support voice calls to and from both remote sites

D. configure the hardware transcoder on the hub location router

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

NEW QUESTION: 386

Which configuration tells a switch port to send Cisco Discovery protocol packets that configure an attached Cisco IP phone to trust tagged traffic that is received from a device that is connected to the access port on the Cisco IP phone?

A)

B)

C)

D)

A. Option D

B. Option C

C. Option A

D. Option B

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

NEW QUESTION: 387

What are two functions of Cisco Expressway in the Collaboration Edge? (Choose two.)

A. The Expressway-C and Expressway-E pair can interconnect H.323-to-SIP calls for voice.

B. Expressway-E provides a perimeter network that separates the enterprise network from the Internet.

C. The Expressway-C and Expressway-E pair can enable connectivity from the corporate network to the PSTN via a T1/E1 trunk.

D. Expressway-E provides a VPN entry point for Cisco IP phones with a Cisco AnyConnect client using authentication based on certificates.

E. Expressway-C provides encryption (or Mobile and Remote Access but not (or business-to-business communications.

Answer: A,E ([LEAVE A REPLY](#))

NEW QUESTION: 388

Why would we not include an end user's PC device in a QoS trust boundary?

A. The end user could incorrectly tag their traffic to bypass firewalls.

B. The end user may incorrectly tag their traffic to be prioritized over other network traffic.

C. There is no reason not to include an end user's PC device in a QoS trust boundary.

D. The end user could incorrectly tag their traffic to advertise their PC as a default gateway.

Answer: ([SHOW ANSWER](#)**)**

Section: QoS

NEW QUESTION: 389

An engineer must create a route pattern for a SIP trunk with the pattern named Unity_Con_SIP_Trunk and a voicemail pilot number 8000/< None > in Cisco Unity Connection. Which value must the engineer enter for the Gateway/Route List field?

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

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 390

How many DNS SRV entries can be defined in the SIP trunk destination address field in Cisco Unified Communications Manager?

- A. 1
- B. 16
- C. 8
- D. 4

Answer: ([SHOW ANSWER](#))

NEW QUESTION: 391

Refer to the exhibit. What is the registration state of the analog port in this debug output?

- A. The analog port is registered to Cisco UCM.
- B. The analog port failed to register to Cisco UCM with an error code 200.
- C. The analog port is currently shut down.
- D. The MGCP Gateway is not communicating with the Cisco UCM.

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

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NEW QUESTION: 392

Which service on the Presence Server is responsible for maintaining the point-to-point chat connections between Jabber clients?

- A. Cisco XCP Router
- B. Cisco XCP Text Conference Manager

C. Cisco XCP XMPP Federation Manager

D. Cisco SIP Proxy

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

NEW QUESTION: 393

Refer to the exhibit.

A call is failing to establish between two SIP Devices. The Called device answer with this SDP. Which SDP parameter causes this issue?

A. The payload for G.711ulaw must be 18.

B. The calling device did not offer aptime value.

C. The media stream is set to sendonly.

D. The RTP port is set to 0.

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

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