

300-075 Dumps

Implementing Cisco IP Telephony and Video, Part 2 (CIPTV2)

<https://www.certleader.com/300-075-dumps.html>



NEW QUESTION 1

You are configuring a SIP trunk with ID333949146 on Cisco Unified Communications Manager to direct calls towards the Cisco VCS C/E. Which two check boxes should you enable in the security profile for the SIP trunk?

- A. Transmit Security Status
- B. Enable Application Level Authorization
- C. Accept Out-of-Dialog REFER
- D. Accept Replaces Header
- E. Accept Unsolicited Notification
- F. Accept Presence Subscription

Answer: DE

NEW QUESTION 2

Which two options are dial-plan challenges in a multisite Cisco Unified Communication Manager deployment?

- A. use of same length directory numbers all over the network
- B. use of +E164 directory numbers and extensions for phones and services
- C. telephony service provider requirements on number format and types
- D. use of MGCP gateways instead of SIP trunks
- E. use of same DID numbers in multiple sites

Answer: AE

NEW QUESTION 3

Which protocol can you use to create a trunk between Cisco Unified Communication Manager and Cisco VCS?

- A. H.323
- B. SIP, H.323
- C. ILS
- D. SIP

Answer: D

NEW QUESTION 4

What is the maximum allowed round-trip between Cisco Unified Communication Manager nodes?

- A. 150 ms
- B. 40 ms
- C. 500 ms
- D. 80 ms

Answer: D

NEW QUESTION 5

Which two gateways or trunks are considered to be call-agent controlled?

- A. SIP trunk registered to a SIP proxy server
- B. H.323 gateway
- C. H.225 trunk (gatekeeper controlled)
- D. Skinny Client Control Protocol gateway
- E. Media Gateway Control Protocol gateway

Answer: CE

NEW QUESTION 6

An engineer opens up a TAC case and wants to implement CCD/SAF service. Which menu navigation sequence must an engineer navigate to when configuring a hosted DN pattern?

- A. Call Routing > Call Control Discovery > Hosted DN Pattern
- B. Advanced Features > SAF > Hosted DN Pattern
- C. Call Routing > SAF > Hosted DN Pattern
- D. Advanced Features > Call Control Discovery > Hosted DN Pattern

Answer: A

NEW QUESTION 7

Which two options are dial plan elements that Global Dial Plan Replication allows independent clusters to share with each other using the Intercluster Lookup service?

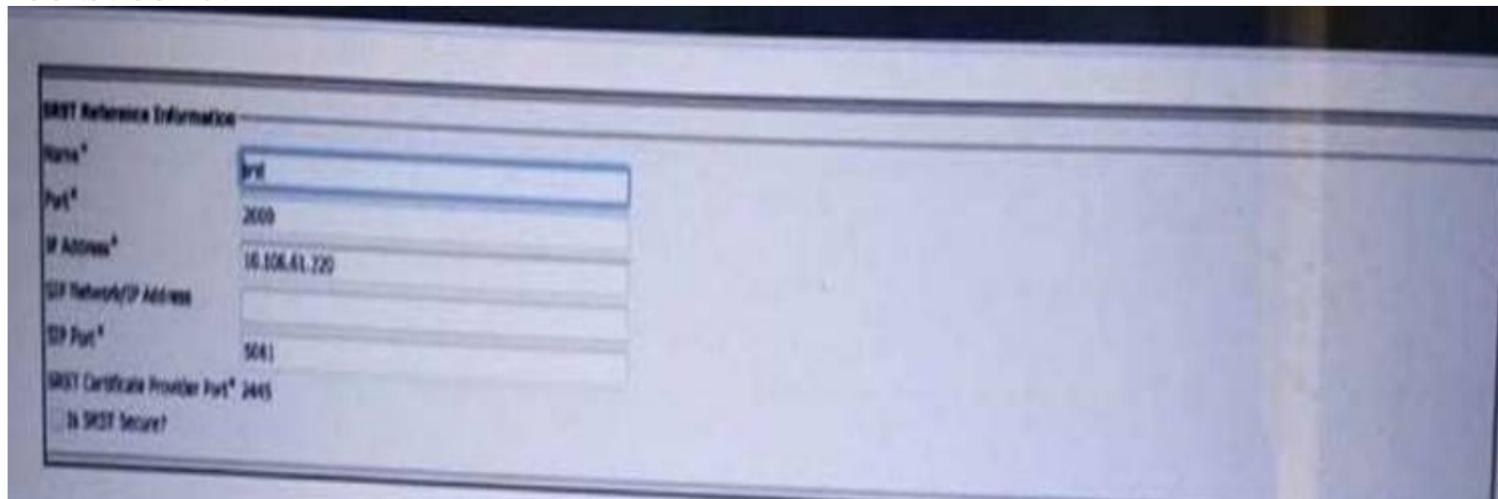
- A. enterprise alternate numbers
- B. voicemail profiles
- C. +W.164 alternate numbers

- D. external phone number masks
- E. actual directory numbers

Answer: BD

NEW QUESTION 8

Refer to the exhibit.



Why are Cisco Unified IP Phones 9971 and 8961 not registering to the SRST gateway?

- A. The SRST reference is not configured on the phone device page.
- B. The SIP network or IP address is not provided.
- C. The SIP port is configured incorrectly in the reference.
- D. Cisco Unified IP Phones 9971 and 8961 do not work in SRST.

Answer: C

NEW QUESTION 9

The network administrator has been investigating bandwidth issues between the central office and remote sites where location-based CAC is implemented. What does the Cisco Unified Communication Manager “Location Out Resources” counter indicate?

- A. This counter represents the total number of times that a call through a location on a particular Cisco Unified Communications Manager failed due to lack of bandwidth.
- B. This counter represents the total number of failed video stream requests (most likely due to lack of bandwidth) in the location where the person who initiated the video conference resides.
- C. This counter represents the total number of times since the last restart of the Cisco IP Voice Streaming Application that a Cisco Unified Communications Manager connection was lost.
- D. This counter represents the total number of times that a call through locations failed due to the lack of bandwidth.

Answer: D

NEW QUESTION 10

Which two roaming sensitive setting in the device pool configuration window override the device level settings when the device roams within or outside a of device mobility group? Choose two

- A. MRGL
- B. codec
- C. device pool
- D. network locale
- E. extension

Answer: AD

NEW QUESTION 10

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

- A. Mobile use of Cisco Jabber requires mobility
- B. Single number reach is a feature of mobility
- C. Mobility is enabled by default
- D. Mobility is used to support mobile phones in Cisco Unified Communications
- E. Mobility just be enabled per user

Answer: BE

NEW QUESTION 11

Which feature can be used to manipulate the calling party numbers of internal calls, post digit analysis?

- A. called party transformation pattern
- B. translation pattern
- C. calling party transformation pattern
- D. external phone mask

Answer: C

NEW QUESTION 14

Which statement about Enhanced Location-based CAC on a Cisco Unified Communications Manager-based voice network

- A. E-LCAC can adjust its knowledge of effective paths due to link failures and reconfigurations.
- B. Every cluster that participates in LBM requires the location be locally configured for location to device association
- C. Each cluster requires a complete view of the modeled topology.
- D. Each cluster should be configured with the immediate neighboring locations such that the topology of each cluster

Answer: C

NEW QUESTION 15

What is the default DHCP/PHB values for Tele Presence streaming video packets in Cisco Unified Communications Manager?

- A. CS6/48
- B. CS3/24
- C. CS4/32
- D. AF41/34
- E. EF/46

Answer: D

NEW QUESTION 20

Which are DSCP values set for video traffic?

- A. DSCP is configured in the device configuration section of Cisco Unified Communications Manager and in the Expressways
- B. The endpoint overrides the DSCP values configured anywhere else
- C. DSCP is configured under service parameters in Cisco Unified Communications Manager and in the Expressways.
- D. DSCP can be configured only in the network with QoS policies
- E. DSCP can be set only on the video endpoint itself to ensure that the correct value is sent.

Answer: D

NEW QUESTION 21

How do you configure the secondary dial tone options for SRST mode?

- A. call-manager failback secondary dialtone O
- B. voice service voip secondary dialtone O
- C. dial-peer voice 1 pots secondary dialtone O
- D. ccm-manager secondary dialtone O

Answer: A

NEW QUESTION 23

To which subzone can you apply bandwidth limitations?

- A. default, traversal, and all manually configured subzones
- B. manually configured subzones
- C. cluster
- D. default

Answer: A

NEW QUESTION 25

Which two statements regarding you configuring a traversal server and traversal client relationship are true? (Choose two.)

- A. VCS supports only the H.460.18/19 protocol for H.323 traversal calls.
- B. VCS supports either the Assent or the H.460.18/19 protocol for H.323 traversal calls.
- C. VCS supports either the Assent or the H.460.18/19 protocol for SIP traversal calls.
- D. If the Assent protocol is configured, a TCP/TLS connection is established from the traversal client to the traversal server for SIP signaling.
- E. A VCS Expressway located in the public network or DMZ acts as the firewall traversal client.

Answer: BD

NEW QUESTION 27

On which two call legs is the media encryption enforced in a Collaboration Edge design? (Choose two)

- A. Expressway-C to internal endpoint
- B. Expressway-E to outside-located endpoint
- C. Expressway-C to Expressway-E
- D. Expressway-E to Cisco Unified Communications Manager
- E. Expressway-C to Cisco Unified Communications Manager

Answer: BC

NEW QUESTION 30

Which option best describes a service that assembles a network model from configured locations and link data in one or more clusters?

- A. LBM Hub
- B. LBM
- C. Shadow
- D. Weight

Answer: B

NEW QUESTION 32

Which two features are part of Cisco Unified Mobility? (Choose Two)

- A. Mobile Voice Access
- B. Device Mobility
- C. Enterprise Feature Access
- D. Shared line
- E. Extension Mobility Cross Cluster

Answer: AC

NEW QUESTION 33

Which three steps are necessary to configure mobile and remote access on Cisco VCS Control? (Choose three)

- A. Register voice gateways with Cisco Unified Communications Manager
- B. Configure Cisco VCS for Cisco Unified Communication
- C. Configure Directory Services for Cisco Unified Communication
- D. Discover Cisco Unified Communicators Manager and IM&P Servers.
- E. After the discovery confirm the connectivity between Cisco VCS Control with Cisco Unified Communication Manager and IM&P
- F. Configure DNS and NTP

Answer: ABF

NEW QUESTION 38

An Enterprise Alternate Number is not reachable in GDPR. What navigation will allow an engineer to configure an automatic switching to a backup number?

- A. PSTN failover numbers
- B. External Phone Number Mask
- C. CFUR
- D. Voicemail

Answer: A

NEW QUESTION 40

In Cisco Unified Communications manager, where can you configure the default codec between two different regions for all endpoints?

- A. Device Pool configuration
- B. Service Parameters
- C. Enterprise Parameters
- D. Endpoint Configuration

Answer: B

NEW QUESTION 44

An administrator is setting up analog phones that connect to a Cisco VG310. Which type of gateway or trunk on Cisco Unified Communication Manager for the Cisco VG310 must the administrator set up to allow the phones to have the call pickup feature?

- A. H.323 gateway
- B. SCCP gateway
- C. H.225 trunk
- D. MGCP gateway
- E. SIP trunk

Answer: B

Explanation: http://www.cisco.com/c/en/us/products/collateral/unified-communications/vg-seriesgateways/product_data_sheet09186a00801d87f6.html

NEW QUESTION 46

Which three steps are required to configure a SIP trunk on Cisco VCS towards cisco unified communications manager? (Choose three)

- A. Configure a neighbor zone that contains the Cisco Unified Communications Manager.
- B. Configure a route pattern zone that contains the Cisco Unified Communications Manager.
- C. Configure a transform that converts numbers to appropriate domains.
- D. Configure a SIP trunk with Cisco Unified Communications Manager destination IP addresses.
- E. Configure a search rule to route calls to the Cisco Unified Communications Manager neighbor zone.
- F. Configure a SIP trunk security profile.

Answer: ACE

NEW QUESTION 49

Which field must be populated to assign a default directory URI partition in Communication Manager?

- A. Directory URI Alias Partition
- B. Alias URI Directory Partition
- C. Alias URI Partition
- D. Directory URI Partition

Answer: A

NEW QUESTION 54

Your company's main number is 408-526-7209, and your employee's directory numbers are 4-digit numbers. Which option should be configured if you want outgoing calls from 4-digit internal directory number to be presented as a 10-digit number?

- A. calling party transformation pattern
- B. AAR group
- C. translation pattern
- D. route pattern

Answer: A

NEW QUESTION 57

How is the peer address configured when Expressway-E has only one NIC enabled and is using static NAT mode?

- A. Expressway-E DHCP
- B. Cisco Unified Communications Manager DHCP
- C. Expressway-E FQDN
- D. Cisco Unified Communications Manager FQDN

Answer: C

NEW QUESTION 60

Which two items must you configure in Cisco Unified Communications Manager to deploy Cisco SAF? (Choose two)

- A. an MWI
- B. voicemail ports
- C. a security profile
- D. a forwarder
- E. a remote destination profile

Answer: CD

NEW QUESTION 62

An engineer must ensure that a device register to Cisco VCS-C. Which protocol accomplish this task?

- A. H.260
- B. H.264
- C. MGCP
- D. SIP

Answer: D

NEW QUESTION 65

A CUCM cluster has been set up with one to one (1:1) call processing redundancy. What two occurrences will be seen when the primary subscriber fails? (Choose two)

- A. The secondary subscriber will start its CallManager service
- B. 40 SCCP phones per second will be able to register to the secondary subscriber.
- C. The secondary subscriber will start its TFTP service
- D. 125 SCCP phones per second will be able to register to the secondary subscriber.
- E. 40 SIP phones per second will be able to register to the secondary subscriber.
- F. 125 SIP phones per second will be able to register to the secondary subscriber

Answer: DE

NEW QUESTION 68

Company X has three locations connected via a low bandwidth WAN. Which two configurations are required in the Cisco Unified Communications Manager region to provide the most suitable use of bandwidth while preserving the call quality? (Choose two)

- A. g722/g711 for interregion calling
- B. g722/g711 codec for all calling
- C. g729 codec for all calling
- D. g729 codec for intraregion calling
- E. g722/g711 codec for interregion calling
- F. g729 codec for interregion calling

Answer: AF

NEW QUESTION 71

In which two locations can you verify that a phone has a standby Cisco Unified communications manager? (Choose two)

- A. phone webpage
- B. RTMT
- C. Cisco Unified Serviceability
- D. phone menu

Answer: AD

NEW QUESTION 75

Select two command. One of which can be used to verify Cisco Ip phone Sip registration and one of which can be used to verify Cisco Ip phone sccp registration on Cisco unified Communications Manager express? (Choose two)

- A. show ccm-manager hosts
- B. show ephone registered
- C. show voice register session-server
- D. show sip-ua status registrar
- E. show telephony-service ephone-dn

Answer: BD

NEW QUESTION 76

Which option configures call preservation for H.323-based SRST mode?

- A. voice service voip h323 call preserve
- B. call preservation not possible with H.323
- C. call-manager-fallback preserve-call
- D. dial-peer voice 1 voip call preserve

Answer: A

NEW QUESTION 77

Which statement about the SAF Client Control is correct?

- A. The SAF Client Control is a configurable inherent component of Cisco Unified Communications Manager.
- B. The SAF Client Control is a non-configurable inherent component of Cisco Unified Communications Manager.
- C. The SAF Client Control is a non-configurable inherent component of the Cisco IOS Routers.
- D. The SAF Client Control is a configurable inherent component of the Cisco IOS Routers.

Answer: B

NEW QUESTION 78

The Cisco Unified Communications system of a company has five types of devices:

- Cisco Jabber Desktop
- CP-7965
- DX-650
- EX-60
- MX-200

Which two types of devices are affected when an engineer changes the DSCP for TelePresence Calls parameter? (Choose two.)

- A. DX-650
- B. Cisco Jabber Desktop
- C. CP-7965
- D. EX-80
- E. MX-200

Answer: DE

NEW QUESTION 79

The administrator at company X is trying to set up Extension Mobility and has done these steps:

- Set up end users accounts for the users who need to roam.
- Set up a device profile for the type of phones users will be allowed to log in

Users have reported to the administrators that they are unable to log in to the phones designated for Extension Mobility. Which two options are the two Reasons for this issue? (Choose two)

- A. The user must ensure that their main endpoint is online and registered, otherwise they cannot log in elsewhere
- B. Extension Mobility has not been enabled under Enterprise Parameters.
- C. The username must be numeric only and must match the DN.
- D. The user device profile is not associated to the correct end user.
- E. The Extension Mobility service has not been enabled under the Cisco Unified Serviceability Page.

Answer: DE

NEW QUESTION 81

When you configure QoS on VCS, which settings do you apply if traffic through the VCS should be tagged with DSCP AF41?

- A. Set QoS mode to DiffServ and tag value 32.
- B. Set QoS mode to IntServ and tag value to 34.
- C. Set QoS mode to DiffServ and tag value 34.
- D. Set QoS mode to IntServ and tag value to 32.
- E. Set QoS mode to ToS and tag value to 32.

Answer: C

NEW QUESTION 84

Refer to the exhibit.

<table border="1" style="width: 100%; border-collapse: collapse;"> <thead> <tr> <th style="width: 50%;">RTP Phone Device Configuration</th> <th style="width: 50%;">Partitions</th> </tr> </thead> <tbody> <tr> <td>Device CSS</td> <td>RTP_Emergency ALL_Phones</td> </tr> <tr> <td>AAR CSS</td> <td>RTP_LongDistance</td> </tr> </tbody> </table>	RTP Phone Device Configuration	Partitions	Device CSS	RTP_Emergency ALL_Phones	AAR CSS	RTP_LongDistance	<table border="1" style="width: 100%; border-collapse: collapse;"> <thead> <tr> <th style="width: 50%;">RTP Phone DN Configuration</th> <th style="width: 50%;">Partitions</th> </tr> </thead> <tbody> <tr> <td>Line CSS</td> <td>RTP_Local RTP_LongDistance RTP_International</td> </tr> <tr> <td>AAR Group</td> <td>AAR</td> </tr> </tbody> </table>	RTP Phone DN Configuration	Partitions	Line CSS	RTP_Local RTP_LongDistance RTP_International	AAR Group	AAR								
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U.K_PSTN	9.!																				

Assume a centralized Cisco Unified Communications deployment with the headquarters in the U.K, and remote site in RTP. All route patterns are assigned a route list that points to the local route group. Local route groups have been configured on the U.K and RTP device pools. A U.K. user logs onto an RTP phone using the Cisco Extension Mobility feature and places an emergency call to 0000. Which statement about the emergency call is true?

- A. The call will match the U.K_Emergency route pattern partition and will egress at the RTP gateway.
- B. The call will match the U.K_Emergency route pattern partition and will egress at the U. gateway.
- C. gateway.
- D. The call will match the RTP_Emergency route pattern partition and will egress at the U. gateway.
- E. gateway.
- F. The call will match the RTP_Emergency route pattern partition and will egress at the RTP gateway.
- G. The call will fail.

Answer: A

NEW QUESTION 85

What is the default DSCP value for Cisco Tele Presence calls in Cisco Unified Communications Manager?

- A. 24
- B. 32

- C. 46
- D. 26
- E. 34

Answer: E

NEW QUESTION 87

The VCS Expressway can be configured with security controls to safeguard external calls and endpoints. One such option is the control of trusted endpoints via a whitelist. Where is this option enabled?

- A. on the voice-enabled firewall at the edge of the network
- B. on the VCS under Configuration > registration > configuration
- C. on the TMS server under Registrations > whitelist
- D. on the VCS under System > configuration > Registrations

Answer: B

NEW QUESTION 92

Which two statements about the use of the Intercluster Lookup Service in a multicluster environment are true? (Choose two.)

- A. Cisco Unified Communications Manager uses the ILS to support intercluster URI dialing.
- B. ILS contains an optional directory URI replication feature that allows the clusters in an ILS network to replicate their directory URIs to the other clusters in the ILS network.
- C. Directory URI replication does not need to be enabled individually for each cluster.
- D. To enable URI replication in a cluster, check the Exchange Directory URIs with Remote Clusters check box that appears in the SIP trunk configuration menu.
- E. If the ILS and directory URI replication feature is disabled on a cluster, this cluster still accepts ILS advertisements and directory URIs from other neighbor clusters; it just does not advertise its local directory URIs.

Answer: AB

NEW QUESTION 93

A user reports that when using desk phone A at the office HQ a call can be extended to a cell phone, but when the desk phone B is used at another office branch, the call cannot be executed. Which Cisco Unified Communications Manager configuration item causes this issue?

- A. Extension Mobility
- B. Device Mobility
- C. Mobility soft key
- D. Physical Location

Answer: B

NEW QUESTION 96

When configuring a secure SIP trunk to which Cisco Unified Communications Manager trust store must you upload the Cisco VCS certificate?

- A. CallManager-trust
- B. ipsec-trust
- C. tomcat-trust
- D. TVS-trust

Answer: A

NEW QUESTION 100

In the event that the central CUCM is unreachable, how will an endpoint identify which IP address to connect to for registration and call processing?

- A. DHCP Option 150
- B. SRST Reference
- C. SRST router fallback config
- D. Cisco Discovery Protocol

Answer: B

NEW QUESTION 105

Which three options are supplementary services that are affected by MTP? (Choose three.)

- A. Call Pickup
- B. Call Hold
- C. Call Back
- D. Call Transfer
- E. Speed Dial
- F. Call Park

Answer: BDF

NEW QUESTION 108

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

- A. 34 and AF41
- B. 32 and AF41
- C. 32 and CS4
- D. 46 and EF

Answer: D

NEW QUESTION 112

Which three items must you configure to enable SAF Call Control Discovery? (Choose three.)

- A. a calling search space
- B. hosted DN patterns
- C. translation patterns
- D. route patterns
- E. the SIP or H.323 trunk
- F. hosted DN groups

Answer: BEF

Explanation:

Reference: http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/8_0_2/ccmfeat/fsgd-802cm/fscallcontroldiscovery.pdf

NEW QUESTION 117

For which two reasons should you mark AF41 as the audio and video channels of a video call?

- A. to allow high-definition quality calls over low-speed links
- B. to give video calls a higher priority than AF41 in the QoS policy
- C. to provide separate classes for audio-only calls and video calls
- D. to prioritize video over other high-priority traffic classes
- E. to preserve lip synchronization between the audio and video channels

Answer: DE

NEW QUESTION 121

When you configure a globalized dial plan, in which three ways can you enable ingress gateways to process calls? (Choose three.)

- A. Configure the called-party transformation settings for incoming calls on H.323 gateways.
- B. Configure translation patterns in the partitions used by the gateway calling search space.
- C. Configure SIP trunks between Cisco Unified Communications Manager clusters.
- D. Configure a remote site device pool.
- E. Configure a hunt group.
- F. Configure the gateway with prefix digits to add necessary country and region codes.

Answer: ABF

NEW QUESTION 126

When troubleshooting high CPU utilization within the Command Line Interface (CLI) of a Cisco Unified Communications Manager (CUCM). Which command will show the process CPU usage for all processes?

- A. show network status
- B. show pert query counter Process"% CPU Time"
- C. show pert query counter Partition"% Used"
- D. show pert query counter Process "Process Status"

Answer: B

NEW QUESTION 127

Which option configures the secondary dial tone option for SRST mode to let the users hear the dial tone for PSTN calls?

- A. voice service voip secondary dialtone 0
- B. call-manager-fallback secondary dialtone 0
- C. dial-peer voice 1 pots secondary dialtone 0
- D. ccm-manager secondary dialtone 0

Answer: B

NEW QUESTION 130

Which component provides a list of Cisco Unified Communications Manager node addresses that allow the phone to register to the cluster?

- A. DNS option 150

- B. DHCP option 150
- C. DHCP option 66
- D. configuration file from TFTP server
- E. Primary Cisco Unified Communications Manager node from configuration file additional nodes during registration.

Answer: B

NEW QUESTION 131

Which two statements regarding the Cisco VCS search and transformation process are true? (Choose two.)

- A. The Cisco VCS applies the search rules in priority order (all rules with priority 255 are processed first, then rules with priority 254, and so on).
- B. Transforms do not use priority numbers.
- C. Presearch transforms are applied before call policy is configured and before user policy is applied.
- D. Presearch transforms are applied after call policy is configured but before user policy is applied.
- E. The Cisco VCS applies the search rules in priority order (all rules with priority 1 are processed first, then priority 2, and so on).
- F. You cannot set up search rules according to the protocol SIP or H.323.

Answer: CE

NEW QUESTION 135

An engineer wants to change a dialed URI into another format, in Cisco VCS. Which location does the engineer navigate to accomplish this task?

- A. Translation pattern
- B. Translation profile
- C. Translation rule
- D. Transform

Answer: A

NEW QUESTION 139

An engineer wants to allow more calls on the WAN link from the head office to a branch. The region setting has been changed from G711 to G729. Which other configuration in Cisco Unified communication Manager must be updated to accomplish this goal?

- A. physical location
- B. region
- C. location
- D. AAR group
- E. class of control

Answer: C

NEW QUESTION 140

You are asked to help a customer administrator to set Quality of Service options clusterwide. Which two options are related to QoS settings on Cisco Communications Manager servers? (Choose two)

- A. You can Configure DSCP values for video calls, audio portion of video calls, and immersive video differently.
- B. you can configure clusterwide QoS parameters in "Cisco CallManager" Service Parameter section within the call and the portion on any server within the cluster where "Cisco CallManager" service has been activated.
- C. Audio calls and the audio portion of video calls are tagged with DSCP value 46.
- D. You can configure clusterwide QoS parameters only in the Enterprise Parameter section on the Publisher.
- E. The option "DSCP for Video" sets the DSCP value for all types of video to the same value.

Answer: AD

NEW QUESTION 143

Which type of endpoints support the DSCP for Video Calls parameter in Cisco Unified CM?

- A. only Cisco Telepresence endpoints
- B. any video capable device
- C. only Jabber
- D. only Cisco UC video endpoints

Answer: B

NEW QUESTION 148

How can you configure VCS Expressway to make sure an external caller cannot reach a specific internal address?

- A. You add a search rule to block any call to the specific URI on the internal VCS Control.
- B. VCS Expressway only forwards calls to Cisco Unified Communications Manager who block calls.
- C. You add the specific URI in the firewall section of the VCS Expressway and block it.
- D. You block the call with search rule that has a high priority in the VCS Expressway. You make a transform on the VCS Expressway to direct the call to VCS Control, which can block the call.

Answer: B

NEW QUESTION 151

If delegated credentials checking has been enabled and remote workers can register to the VCS Expressway, which statement is true?
A secure neighbor zone has been configured between the VCS Expressway and the VCS Control.

- A. SIP registration proxy mode is set to Off in the VCS Expressway.
- B. H.323 message credential checks are delegated.
- C. SIP registration proxy mode is set to On in the VCS Expressway.

Answer: B

NEW QUESTION 154

Which configuration for the IP DSCP value in the ToS byte to 5 with Expedited Forwarding is correct?

- A. policy-map policy1 class class1set ip dscp 5 class class2set ip dscp C S3
- B. policy-map policy1 class class1set ip dscp ef class class2 set ip dscp 5
- C. policy-map policy1 class class1set ip dscp 5 class class2 set ip dscp ef
- D. policy-map policy1class class1 set ip dscp 3 class class2 set ip dscp ef

Answer: B

NEW QUESTION 155

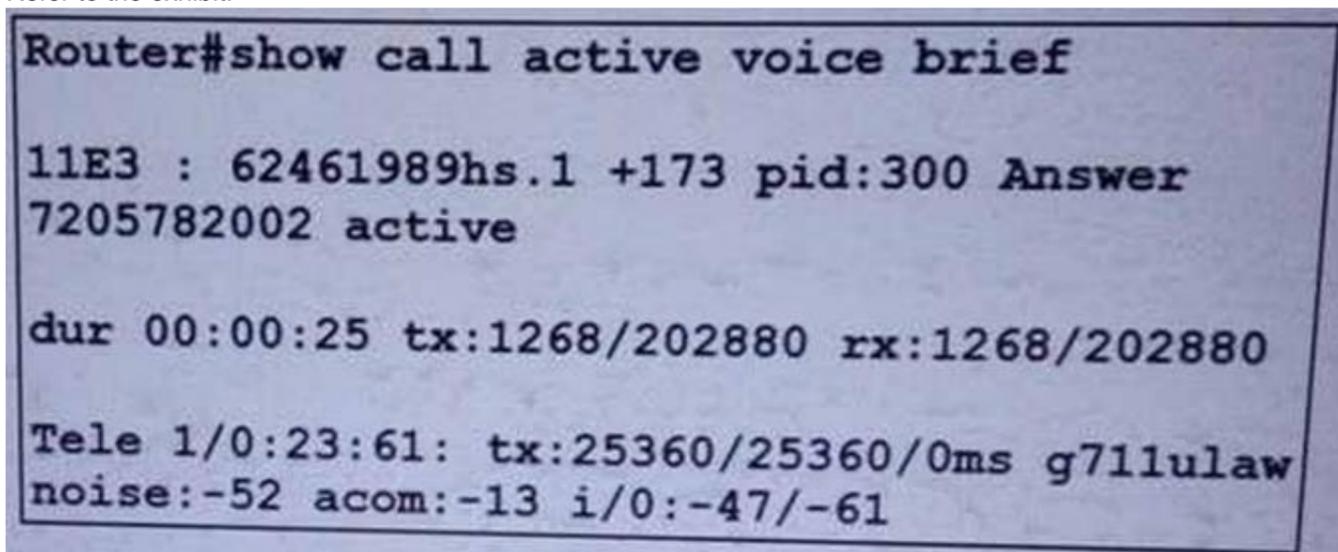
An engineer must ensure that Expressway-E listens and only its legit sender of such traffic is Expressway-C. Which port must be opened to accomplish this task?

- A. 2000
- B. 2111
- C. 222
- D. 233

Answer: A

NEW QUESTION 156

Refer to the exhibit.



An engineer is troubleshooting an active call and sees this output. Wich option does the PID signify/

- A. ISDN callref ID
- B. dial peer
- C. process ID
- D. CCAPI ID

Answer: A

NEW QUESTION 157

Which configuration can you add to Mobile Connect to explicitly allow or block calls to remote destinations based on the caller ID of the caller?

- A. access list
- B. route filter
- C. external call control
- D. call anchoring

Answer: B

NEW QUESTION 158

How does Cisco Extension Mobility work?

- A. A user can log in to any IP phone in the cluster, that phone is dynamically configured with (he user configurations.
- B. A user an define several remote destinations so that a call to the IP phone rings their mobile and home destinations simultaneously.
- C. A user can choose one of several wireless IP phone models for mobile IP communication throughout the WLAN coverage area.
- D. A user can move the phone anywhere within the cluster its calls will be routed out the local gateway.

E. A user can log into any phone in the cluster and receive a DN from a predefined range.

Answer: A

NEW QUESTION 162

Which parameter should be set to prevent H.323 endpoints from registering to Cisco TelePresence Video Communication Server automatically?

- A. On the VCS, navigate to Configuration, Protocols, H.323, and set Auto Discover to off.
- B. On the VCS, navigate to Configuration, Protocols, H.323, and set Auto Registration to off.
- C. On the VCS, navigate to Configuration, Registration, Allow List, and set Auto Registration to off.
- D. On the VCS, navigate to Configuration, Registration, Configuration, and set Auto Registration to off.

Answer: A

NEW QUESTION 165

Widgets.com's Cisco TelePresence Video Communication Server allows SIP and H.323 registrations.

Which local zone search rule configuration allows SIP registered endpoints to connect to H.323 endpoints that register with an H.323 E.164 number only?

A)

Mode: Alias pattern match
 Pattern type: Regex
 Pattern string: (*)@widgets.com
 Pattern behavior: Replace
 Replace string: \d+
 On successful match: Continue
 Target: Local Zone
 State: Enabled

B)

Mode: Alias pattern match
 Pattern type: Regex
 Pattern string: (+)@widgets.com
 Pattern behavior: Replace
 Replace string: *
 On successful match: Continue
 Target: Local Zone
 State: Enabled

C)

Mode: Alias pattern match
 Pattern type: Regex
 Pattern string: (+)@widgets.com
 Pattern behavior: Replace
 Replace string: \\$
 On successful match: Continue
 Target: Local Zone
 State: Enabled

D)

Mode: Alias pattern match
 Pattern type: Regex
 Pattern string: (+)@widgets.com
 Pattern behavior: Replace
 Replace string: \d
 On successful match: Continue
 Target: Local Zone
 State: Enabled

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

Answer: D

NEW QUESTION 168

You want to avoid unnecessary interworking in Cisco TelePresence Video Communication Server, such as where a call between two H.323 endpoints is made over SIP, or vice versa

- A. Which setting is recommended?
- B. H.323 - SIP interworking mod
- C. Reject
- D. H.323 - SIP interworking mod
- E. On
- F. H.323 - SIP interworking mod
- G. Registered only
- H. H.323 - SIP interworking mod
- I. Off
- J. H.323 - SIP interworking mod
- K. Variable

Answer: C

NEW QUESTION 173

Which three statements about configuring an encrypted trunk between Cisco TelePresence Video Communication Server and Cisco Unified Communications Manager are true? (Choose three.)

- A. The root CA of the VCS server certificate must be loaded in Cisco Unified Communications Manager.
- B. A SIP trunk security profile must be configured with Incoming Transport Type set to TCP+UDP.
- C. The Cisco Unified Communications Manager trunk configuration must have the destination port set to 5061.
- D. A SIP trunk security profile must be configured with Device Security Mode set to TLS.
- E. A SIP trunk security profile must be configured with the X.509 Subject Name from the VCS certificate.
- F. The Cisco Unified Communications Manager zone configured in VCS must have SIP authentication trust mode set to On.
- G. The Cisco Unified Communications Manager zone configured in VCS must have TLS verify mode set to Off.

Answer: ACE

NEW QUESTION 176

Which two statements about configuring mobile and remote access on Cisco TelePresence Video Communication Server Expressway are true? (Choose two.)

- A. The traversal server zone on Expressway-C must have a TLS verify subject name configured.
- B. The traversal client zone and the traversal server zone Media encryption mode must be set to Force encrypted.
- C. The traversal client zone and the traversal server zone Media encryption mode must be set to Auto.
- D. The traversal client zone on Expressway-C Media encryption mode must be set to Auto.
- E. The traversal client zone and the traversal server zone must be set to SIP TLS with TLS verify mode set to On.

Answer: BE

NEW QUESTION 178

Which two statements regarding IPv4 Static NAT address 209.165.200.230 has been configured on a VCS Expressway are true? (Choose two.)

- A. The Advanced Networking or Dual Network Interfaces option key has been installed.
- B. VCS rewrites the Layer 3 source address of outbound SIP and H.323 packets to 209.165.200.230.
- C. VCS applies 209.165.200.230 to outbound SIP and H.323 payload messages.
- D. With static NAT enabled on the LAN2 interface, VCS applies 209.165.200.230 to outbound H.323 and SIP payload traffic exiting the LAN1 interface.

Answer: AC

NEW QUESTION 180

Which configuration does Cisco recommend for the peer address on the Expressway-C secure traversal zone when the Expressway-E has one NIC enabled?

- A. Expressway-E internal IP address
- B. Expressway-E external IP address
- C. Expressway-E internal FQDN
- D. Expressway-E external FQDN

Answer: B

NEW QUESTION 184

Which two options should be used to create a secure traversal zone between the Expressway-C and Expressway-E? (Choose two.)

- A. Expressway-C and Expressway-E must trust each other's server certificate.
- B. One Cisco Unified Communications traversal zone for H.323 and SIP connections.
- C. A separate pair of traversal zones must be configured if an H.323 connection is required and Interworking is disabled.
- D. Enable username and password authentication verification on Expressway-E.
- E. Create a set of username and password on each of the Expressway-C and Expressway-E to authenticate the neighboring peer.

Answer: AC

NEW QUESTION 188

Which three commands are necessary to override the default CoS to DSCP mapping on interface FastEthernet0/1? (Choose three.)

- A. mls qos map cos-dscp 0 10 18 26 34 46 48 56
- B. mls qos map dscp-cos 8 10 to 2
- C. mls qos
- D. interface FastEthernet0/1 mls qos trust cos
- E. interface FastEthernet0/1 mls qos cos 1
- F. interface FastEthernet0/2mls qos cos 1

Answer: ACD

NEW QUESTION 193

Refer to the exhibit.

```
dial-peer voice 901 pots
destination-pattern 9011T
port 1/0:23
```

Which configuration change is needed to enable NANP international dialing during MGCP fallback?

- A. Change the dial peer to dial-peer voice 901 voip.
- B. Change the dial peer to dial-peer voice 9011 pots.
- C. Add the command prefix 011 to the dial peer.
- D. Add the command prefix 9011 to the dial peer.

Answer: C

NEW QUESTION 196

Which command displays the detailed configuration of all the Cisco Unified IP phones, voice ports, and dial peers of the Cisco Unified SRST router?

- A. show call-manager-fallback all
- B. show dial-peer voice summary
- C. show ephone summary
- D. show voice port summary

Answer: A

NEW QUESTION 200

Which two options should be selected in the SIP trunk security profile that affect the SIP trunk pointing to the VCS? (Choose two.)

- A. Accept Unsolicited Notification
- B. Enable Application Level Authorization
- C. Accept Out-of-Dialog REFER
- D. Accept Replaces Header
- E. Accept Presence Subscription

Answer: AD

NEW QUESTION 202

Which sign is prefixed to the number in global call routing?

- A. -
- B. +
- C. #
- D. @
- E. &
- F. *

Answer: B

NEW QUESTION 206

Which statement about the function of the "+" symbol in the E.164 format is true?

- A. The "+" symbol represents the international country code.
- B. The "+" symbol represents the international call prefix.
- C. The "+" symbol matches the preceding element one or more times.
- D. The "+" symbol matches the preceding element zero or one time.

Answer: B

NEW QUESTION 209

Which statement is true regarding the configuration of SAF Forwarder?

- A. In a multisite dial plan, SAF Forwarders may exist in multiple autonomous systems.

- B. The client label that is configured in Cisco Unified Communications Manager must match the configuration on the SAF Forwarder router.
- C. There should not be multiple nodes of Cisco Unified Communications Manager clusters acting as SAF clients.
- D. The destination IP address must match the loopback address of the SAF router.

Answer: A

NEW QUESTION 212

Assume that local route groups are configured. When an IP phone moves from one device mobility group to another, which two configuration components are not changed? (Choose two.)

- A. IP subnet
- B. user settings
- C. SRST reference
- D. region
- E. phone button settings

Answer: BE

NEW QUESTION 217

Which statement about setting up FindMe in Cisco TelePresence Video Communication Server is true?

- A. Users are allowed to delete or change the address of their principal devices.
- B. Endpoints should register with an alias that is the same as an existing FindMe ID.
- C. If VCS is using Cisco TMS provisioning, users manage their FindMe accounts via VCS.
- D. A VCS cluster name must be configured.

Answer: D

NEW QUESTION 222

Which action configures phones in site A to use G.711 to site B, but uses G.729 to site C?

- A. Configure Cisco Unified Communications Manager regions.
- B. Configure Cisco Unified Communications Manager locations.
- C. Configure transcoder resources in Cisco Unified Communications Manager.
- D. Configure a gatekeeper.

Answer: A

NEW QUESTION 224

Which system configuration is used to set audio codecs?

- A. region
- B. location
- C. physical location
- D. licensing

Answer: A

NEW QUESTION 228

A voicemail product that supports only the G.711 codec is installed in headquarters. Which action allows branch Cisco IP phones to function with voicemail while using only the G.729 codec over the WAN link to headquarters?

- A. Configure Cisco Unified Communications Manager regions.
- B. Configure transcoding within Cisco Unified Communications Manager.
- C. Configure transcoding resources in Cisco IOS and assign to the MRGL of Cisco IP phones.
- D. Configure transcoder resources in the branch Cisco IP phones.

Answer: C

NEW QUESTION 232

Which system configuration is used to set a restriction on audio bandwidth?

- A. region
- B. location
- C. physical location
- D. licensing

Answer: B

NEW QUESTION 234

The network administrator has been investigating bandwidth issues between the central office and remote sites where location-based CAC is implemented. What does the Cisco Unified Communications Manager "LocationOutOfResources" counter indicate?

- A. This counter represents the total number of times that a call on a particular Cisco Unified Communications Manager through the location failed due to lack of

bandwidth.

B. This counter represents the total number of times that a call through locations failed due to the lack of bandwidth.

C. This counter represents the total number of failed video-stream requests (most likely due to lack of bandwidth) in the location where the person who initiated the video conference resides.

D. This counter represents the total number of times since the last restart of the Cisco IP Voice Streaming Application that a Cisco Unified Communications Manager connection was lost.

Answer: B

NEW QUESTION 236

Scenario:

There are two call control systems in this item.

The Cisco UCM is controlling the DX650, the Cisco Jabber for Windows Client, and the 9971 Video IP Phone.

The Cisco VCS is controlling the SX20, the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows.

SX20 System information:

System Information

General

Product:	Cisco TelePresence SX20
Last boot:	Last Wednesday at 21:43
Serial number:	ABCD12345678
Software version:	TC7.3.0
Installed options:	PremiumResolution
System name:	MySystem
IPv4:	192.168.1.128
IPv6:	2001:DB8:1001:2002:3003:4004:5005:F00F
MAC address:	01:23:45:67:89:AB
Temperature:	58.5°C / 137.3°F

H323

Status:	Registered
Gatekeeper:	192.168.1.1
Number:	123456
ID:	firstname.lastname@company.com

SIP Proxy 1

Status:	Registered
Proxy:	192.168.1.2
URI:	firstname.lastname@company.com

DX650 Configuration:

The screenshot shows the Cisco DX650 Configuration interface. On the left, there is a list of 16 lines. Line 1 is selected and shows 'Line [1] - 3304 in Devices'. Line 2 is 'Line [2] - Add a new DN'. Lines 3-15 are 'Add a new SD'. Line 16 is 'Add a new SD'. On the right, the configuration details for the selected line are shown:

- Product Type:** Cisco DX650
- Device Protocol:** SIP
- Real-time Device Status:**
 - Registration: Unregistered
 - IPv4 Address: 172.18.32.119
 - Active Load ID: sipdx650.10-2-3-26
 - Inactive Load ID: sipdx650.10-1-1-78
 - Download Status: None
- Device Information:**
 - Device is Active
 - Device is trusted
 - MAC Address*: DOC78914131D
 - Description: DX650 Pod 3
 - Device Pool*: Default
 - Common Device Configuration: < None >
 - Phone Button Template*: Cisco DX650 SIP
 - Softkey Template: < None >
 - Common Phone Profile*: Standard Common Phone Profile
 - Calling Search Space: All-Devices
 - AAR Calling Search Space: < None >

MRGL:

MRGL

Status
i 1 records found

Media Resource Group List (1 - 1 of 1)

Find Media Resource Group List where Name begins with **Find** **Clear Filter** **+** **-**

<input type="checkbox"/>	Name ^
<input type="checkbox"/>	MRGL-All

Add New **Select All** **Clear All** **Delete Selected**

DP:

DP

Status
i 3 records found

Device Pool (1 - 3 of 3)

Find Device Pool where Device Pool Name begins with **Find** **Clear Filter** **+** **-**

<input type="checkbox"/>	Name ^	Cisco Unified CM Group	Region	Date/Ti
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	GSM	Default	GSM	CMLocal

Add New **Select All** **Clear All** **Delete Selected**

Locations:

Locations

Locations (1 - 3 of 3)

Find Locations where Location begins with **Find** **Clear Filter** **+** **-**

<input type="checkbox"/>	
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

Add New **Select All** **Clear All** **Delete Selected**

AARG:

AARG

Automated Alternate Routing Group

Find Automated Alternate Routing Group where Name begins with **Find** **Clear Filter** **+** **-**

No active query. Please enter your search criteria using the options above.

CSS:

CSS

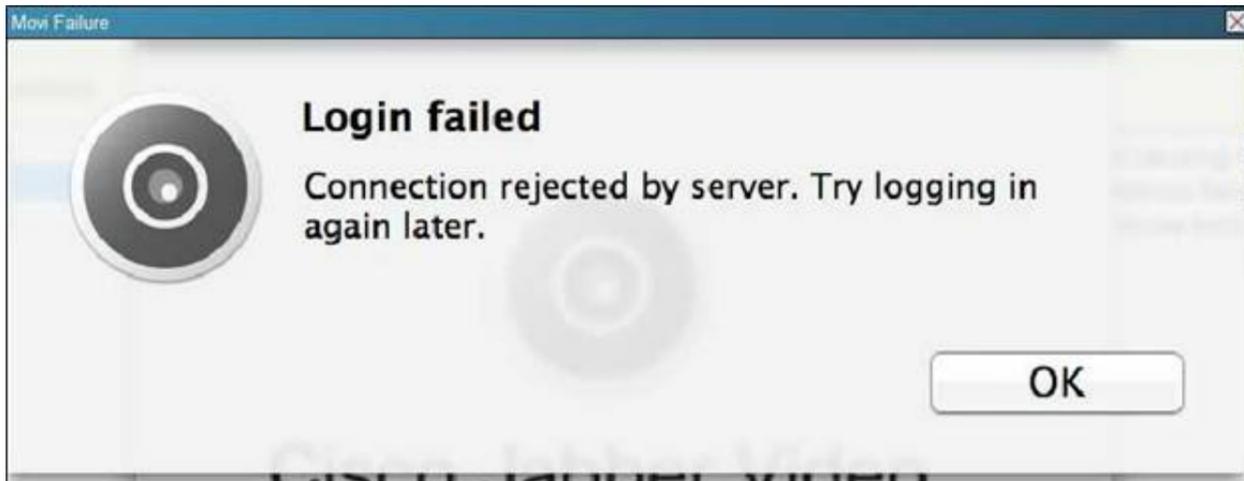
Calling Search Space (1 - 2 of 2)

Find Calling Search Space where CSS Name begins with **Find** **Clear Filter** **+** **-**

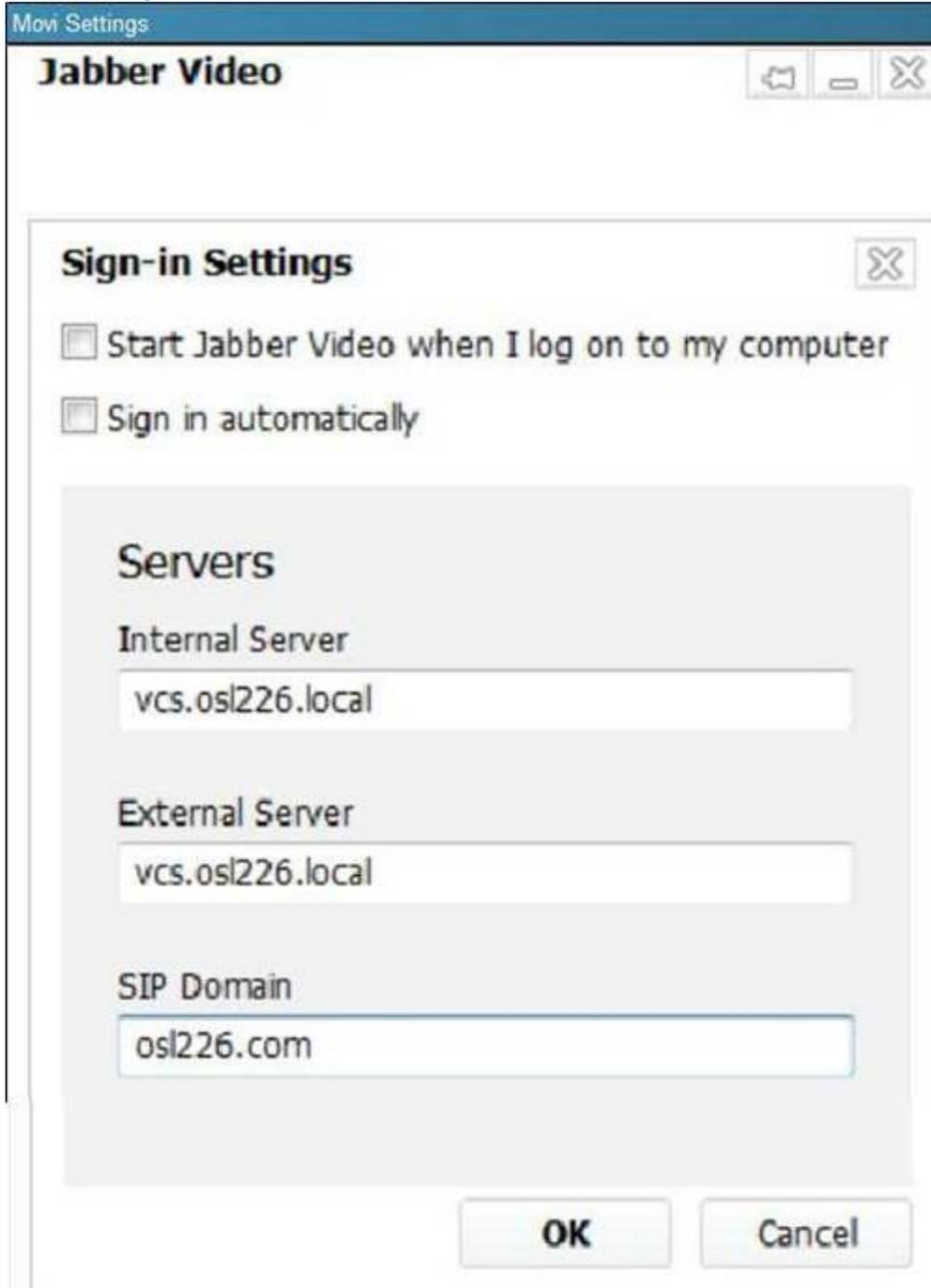
<input type="checkbox"/>	CSS Name ^
<input type="checkbox"/>	A11-Devices
<input type="checkbox"/>	All-Devices

Add New **Select All** **Clear All** **Delete Selected**

Movi Failure:



Movie Settings:



Which device configuration option will allow an administrator to assign a device to specific rights for making calls to specific DNSs?

- A. Media Resource Group List
- B. Device Pool
- C. Location
- D. AAR Group
- E. Calling Search Space

Answer: E

NEW QUESTION 240

Scenario:

There are two call control systems in this item.

The Cisco UCM is controlling the Cisco Jabber for Windows Client, and the 7965 and 9971 Video IP Phone.

The Cisco VCS is controlling the SX20, the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows DP:

DP

Status

3 records found

Device Pool (1 - 3 of 3)

Find Device Pool where Device Pool Name begins with

<input type="checkbox"/>	Name ^	Osco Unified CM Group	Region	Date/Ti
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	GSM	Default	GSM	CMLocal

Locations:

Locations

Locations (1 - 3 of 3)

Find Locations where Location begins with

<input type="checkbox"/>	Location
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

CSS:

CSS

Calling Search Space (1 - 2 of 2)

Find Calling Search Space where CSS Name begins with

<input type="checkbox"/>	CSS Name ^
<input type="checkbox"/>	A11-Devices
<input type="checkbox"/>	All-Devices

Movie Failure:

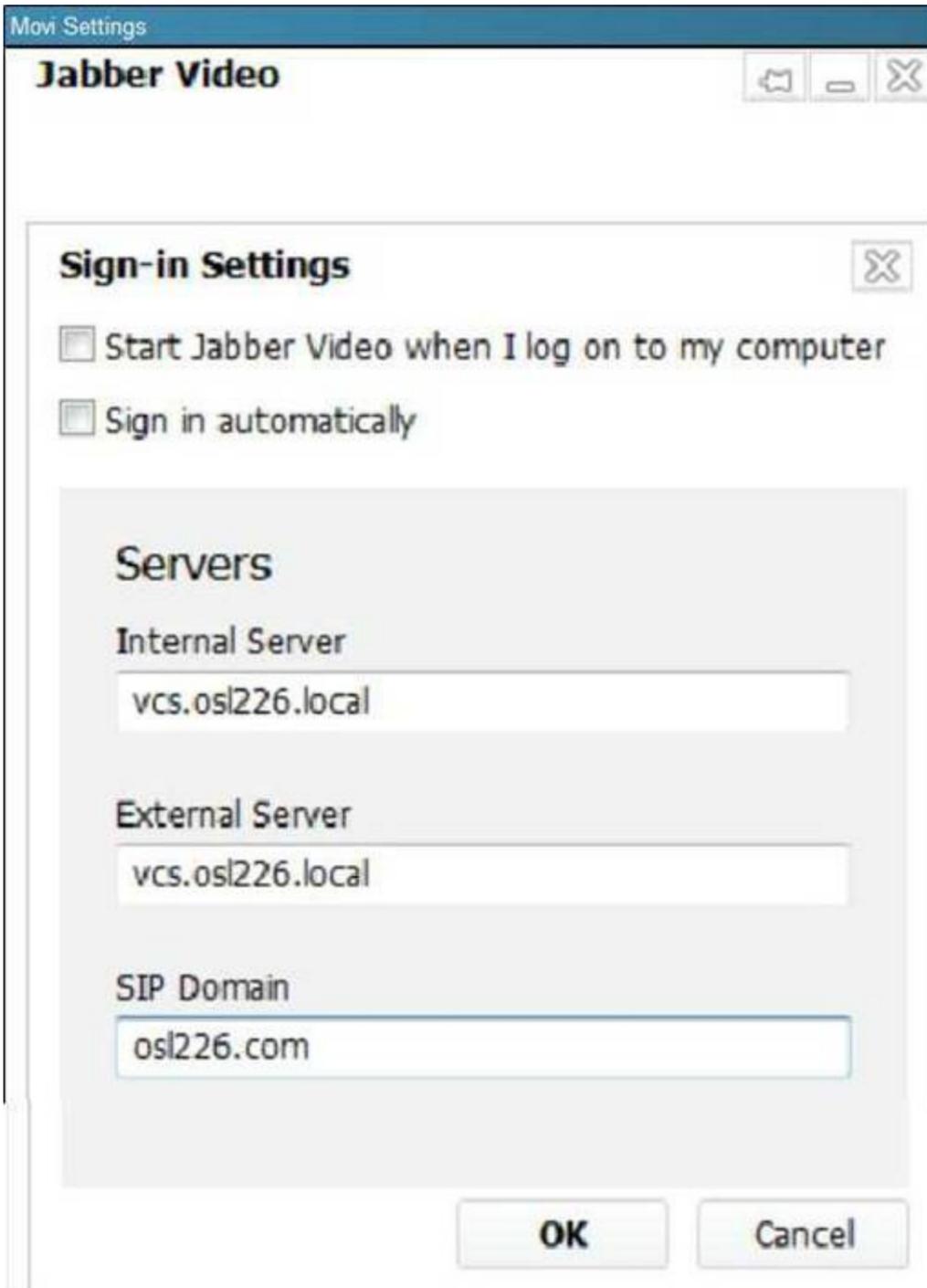
Movie Failure



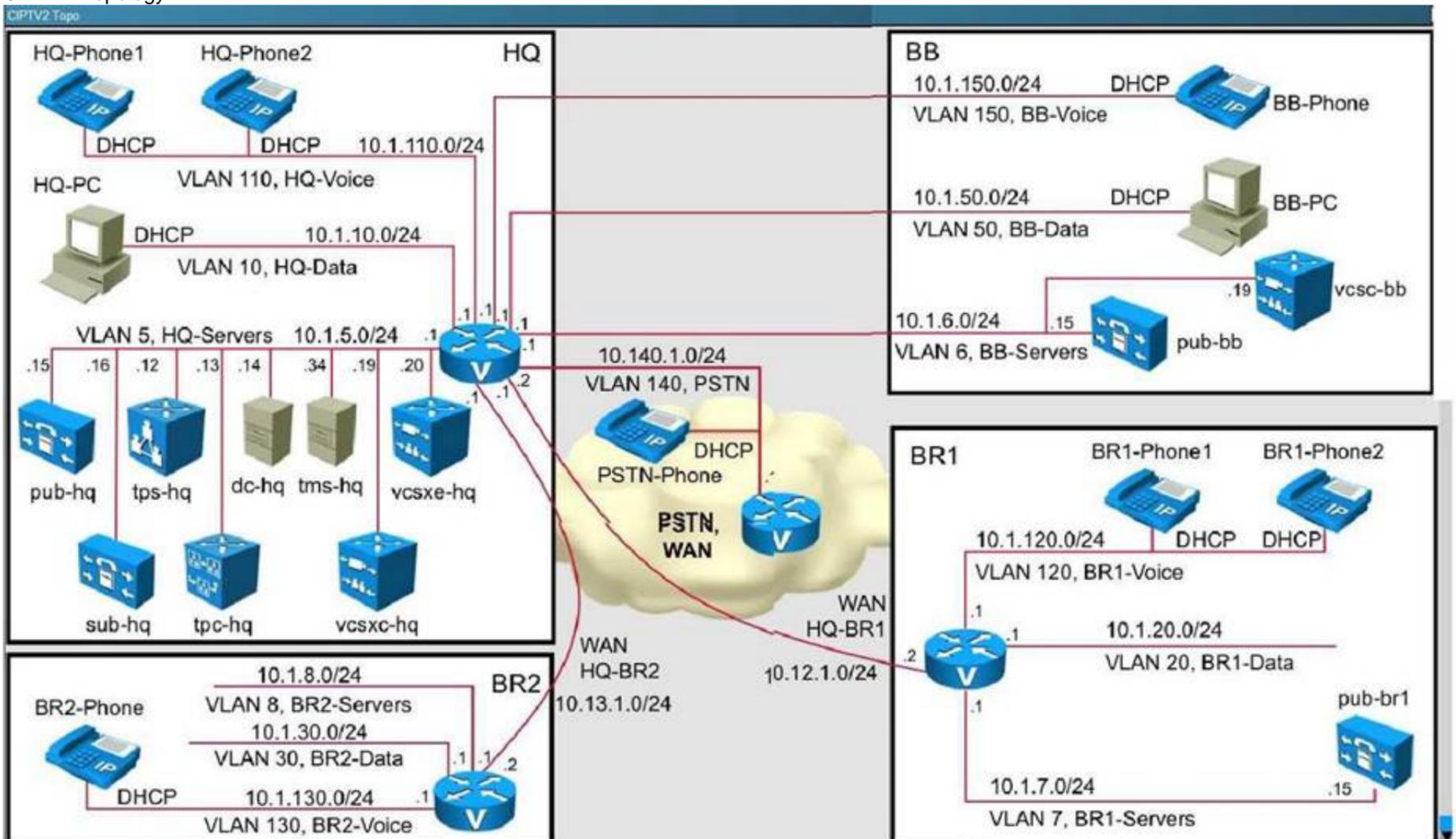
Login failed

Connection rejected by server. Try logging in again later.

Movie Settings:



CIPTV2 Topology:



Subzone:

- **Name:** HQ
- **Authentication policy:** Treat as authenticated
- **Total bandwidth available - Bandwidth restriction:** Limited
- **Total bandwidth available - Total bandwidth limit (kbps):** 512
- **Calls into or out of this subzone - Bandwidth restriction:** Limited
- **Calls into or out of this subone - Total bandwidth limit (kbps):** 0
- **Calls entirely within this subzone Bandwidth restriction:** Limited
- **Calls entirely within this subzone Total bandwidth limit (kbps):** 200

Links:

Links		You are here: Configuration > Bandwidth > Links					
Name	Node 1	Node 2	Pipe 1	Pipe 2	Calls	Bandwidth used	Actions
<input type="checkbox"/> DefaultSZtoClusterSZ	DefaultSubZone	ClusterSubZone			0	0 kbps	View/Edit
<input type="checkbox"/> DefaultSZtoDefaultZ	DefaultSubZone	DefaultZone			0	0 kbps	View/Edit
<input type="checkbox"/> DefaultSZtoTraversalSZ	DefaultSubZone	TraversalSubZone			0	0 kbps	View/Edit
<input type="checkbox"/> SubZone001ToDefaultSZ	HQ	DefaultSubZone			0	0 kbps	View/Edit
<input type="checkbox"/> SubZone001ToTraversalSZ	HQ	TraversalSubZone			0	0 kbps	View/Edit
<input type="checkbox"/> TraversalSZtoDefaultZ	TraversalSubZone	DefaultZone			0	0 kbps	View/Edit
<input type="checkbox"/> VCS HQ - toHQ	HQ	to HQ	to HQ pipe		1	128 kbps	View/Edit

Pipe:

Name: to HQ pipe

Total Bandwidth available – Bandwidth restriction: Limited

Total Bandwidth available – Total bandwidth limit (kbps): 256

Calls through this pipe – Bandwidth restriction: Limited

Calls through this pipe – Per call bandwidth limit (kbps): 128

What two issues could be causing the Cisco Jabber Video for TelePresence failure shown in the exhibit? (Choose two)

- A. Incorrect username and password.
- B. Wrong SIP domain configured.
- C. User is not associated with the device.
- D. IP or DNS name resolution issue.
- E. CSF Device is not registered.
- F. IP Phone DN not associated with the user.

Answer: BD

NEW QUESTION 242

Scenario:

There are two call control systems in this item. The Cisco UCM is controlling the DX650, the Cisco Jabber for Windows Client, and the 9971 Video IP Phone. The Cisco VCS and TMS control the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows.

DP:

DP

Status

3 records found

Device Pool (1 - 3 of 3)

Find Device Pool where Device Pool Name begins with

<input type="checkbox"/>	Name ^	Cisco Unified CM Group	Region	Date/TI
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	GSM	Default	GSM	CMLocal

Locations:

Locations

Locations (1 - 3 of 3)

Find Locations where Location begins with

<input type="checkbox"/>	Location
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

CSS:

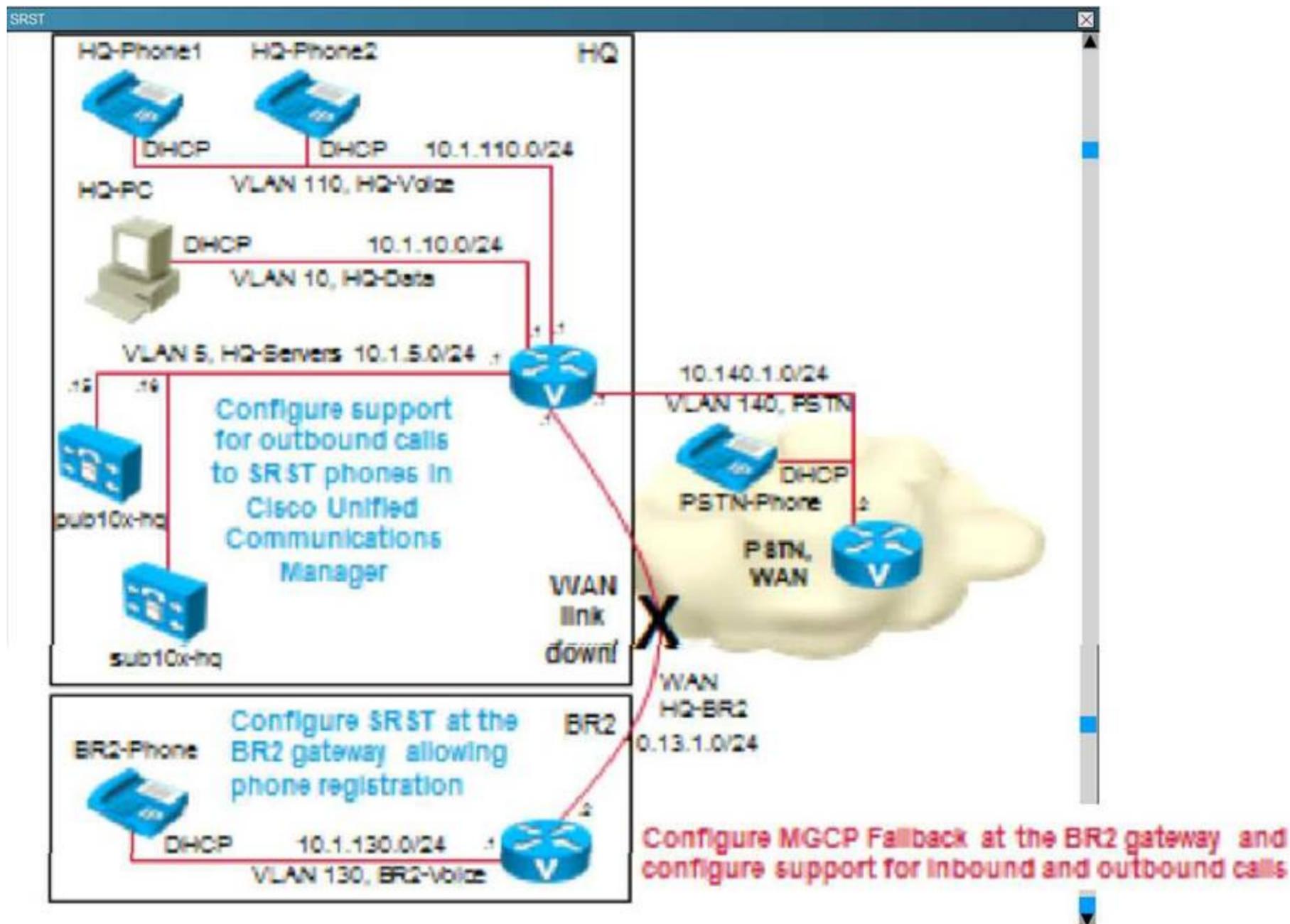
CSS

Calling Search Space (1 - 2 of 2)

Find Calling Search Space where CSS Name begins with

<input type="checkbox"/>	CSS Name ^
<input type="checkbox"/>	All-Devices
<input type="checkbox"/>	All-Devices

SRST:



SRST-BR2-Config:

```

SRST-BR2 Config
  ■ Name: BR2
  ■ Port: 2000
  ■ IP Address: 10.1.5.15
  ■ SIP Network/IP Address: 10.1.5.15
  
```

BR2 Config:

```
BR2 Config
voice service voip
  sip
    bind control source-interface
GigabitEthernet0/0/0.130
    bind media source-interface
GigabitEthernet0/0/0.130
  registrar server
  !
voice register global
  max-dn 1
  max-pool 1
  !
voice register pool 1
  id network 10.1.130.0 mask 255.255.25
call-manager-fallback
  ip source-address 10.1.130.1
  max-dn 1 dual-line
  max-ephones 1
```

SRSTPSTNCall:

At the HQ cluster, the CFUR for the directory number that is applied to BR2 phone (+442288224001) has been configured:

- Forward Unregistered Internal Destination: **+442288224001**
- Forward Unregistered Internal Calling Search Space: System css
- Forward Unregistered External Destination: **+442288224001**
- Forward Unregistered External Calling Search Space: System css

After configuring the CFUR for the directory number that is applied to BR2 phone (+442288224001), the calls fail from the PSTN. Which two of the following configurations if applied to the router, would remedy this situation? (Choose two.)

- A. dial-peer voice 1 potsincoming called-number 228822... direct-inward-dialport 0/0/0:15
- B. dial-peer voice 1 potsincoming called-number 228822... direct-inward-dialport 0/0/0:13
- C. voice translation-rule 1 rule 1/?228821....S//+44&/ exit!voice translation-profile pstn-in translate called 1!voice-port 0/0/0:15translation-profile incoming pstn-in
- D. voice translation-rule 1 rule 1/?228822....S//+44&/ exit!voice translation-profile pstn-in translate called 1!voice-port 0/0/0:15translation-profile incoming pstn-in
- E. The router does not need to be configured.

Answer: AD

NEW QUESTION 245

Scenario:

There are two call control systems in this item.

The Cisco UCM is controlling the DX650, the Cisco Jabber for Windows Client, and the 9971 Video IP Phone.

The Cisco VCS and TMS control the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows

DP:

DP

Status

3 records found

Device Pool (1 - 3 of 3)

Find Device Pool where Device Pool Name begins with

<input type="checkbox"/>	Name ^	Cisco Unified CM Group	Region	Date/TI
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	GSM	Default	GSM	CMLocal

Locations:

Locations

Locations (1 - 3 of 3)

Find Locations where Location begins with

<input type="checkbox"/>	Location
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

CSS:

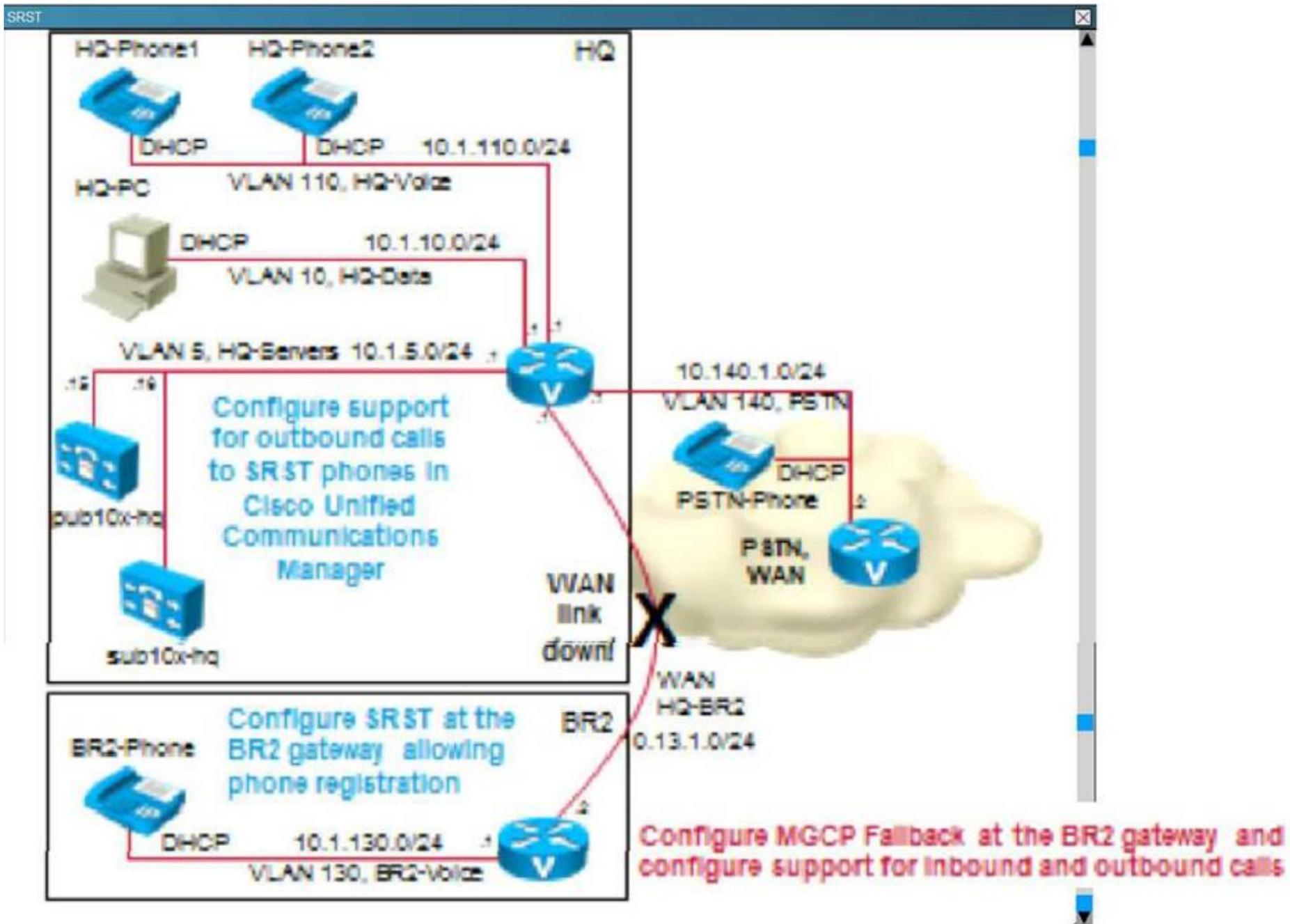
CSS

Calling Search Space (1 - 2 of 2)

Find Calling Search Space where CSS Name begins with

<input type="checkbox"/>	CSS Name ^
<input type="checkbox"/>	All-Devices
<input type="checkbox"/>	All-Devices

SRST:



SRST-BR2-Config:

```

SRST-BR2-Config
■ Name: BR2
■ Port: 2000
■ IP Address: 10.1.5.15
■ SIP Network/IP Address: 10.1.5.15
    
```

BR2 Config:

```

BR2 Config
voice service voip
  sip
    bind control source-interface
GigabitEthernet0/0/0.130
    bind media source-interface
GigabitEthernet0/0/0.130
  registrar server
  !
voice register global
  max-dn 1
  max-pool 1
  !
voice register pool 1
  id network 10.1.130.0 mask 255.255.25
call-manager-fallback
  ip source-address 10.1.130.1
  max-dn 1 dual-line
  max-ephones 1
  
```

SRSTPSTNCall:

At the HQ cluster, the CFUR for the directory number that is applied to BR2 phone (+442288224001) has been configured:

- Forward Unregistered Internal Destination: **+442288224001**
- Forward Unregistered Internal Calling Search Space: System css
- Forward Unregistered External Destination: **+442288224001**
- Forward Unregistered External Calling Search Space: System css

Which device configuration option will allow an administrator to control bandwidth between calls placed between branches?

- A. Media Resource Group List
- B. Device Pool
- C. Location
- D. AAR Group
- E. Regions

Answer: C

NEW QUESTION 246

Scenario:

There are two call control systems in this item.

The Cisco UCM is controlling the DX650, the Cisco Jabber for Windows Client, and the 7965 and 9971 Video IP Phones.

The Cisco VCS and TMS control the Cisco TelePresence Conductor, the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows.

DNS Server:

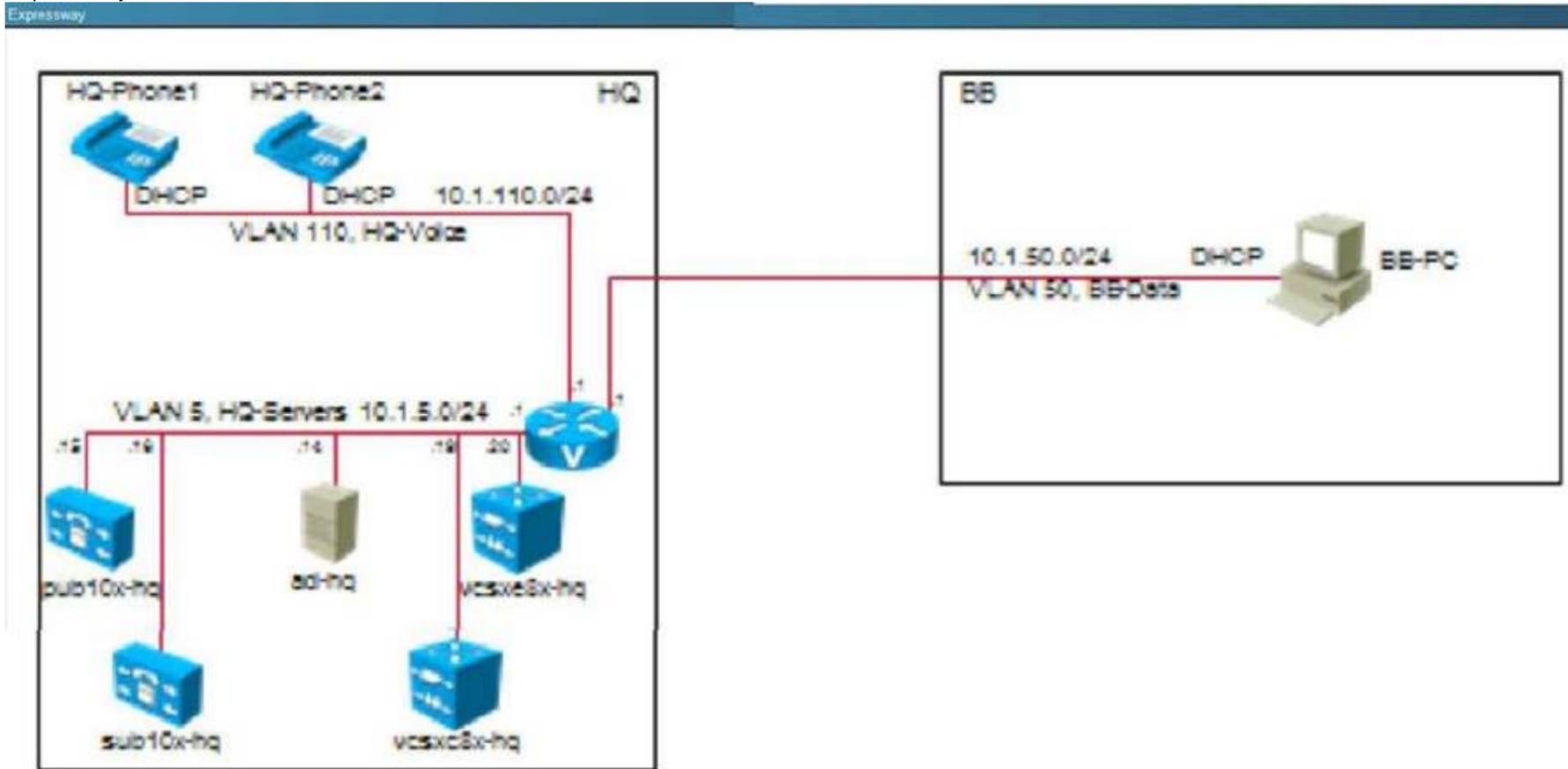
```

ip dns server
ip host _cisco-uds._tcp.hq.cisco.com srv 1 1 8443 10.1.5.15
ip host _cisco-uds._tcp.hq.cisco.com srv 1 1 8443 10.1.5.16
ip host pub10x-hq.collab10x.cisco.com 10.1.5.15
ip host sub10x-hq.collab10x.cisco.com 10.1.5.16
ip host pub10x-hq.hq.collab10x.cisco.com 10.1.5.15
ip host sub10x-hq.hq.collab10x.cisco.com 10.1.5.16
ip host hq.cisco.com 10.1.5.1
    
```

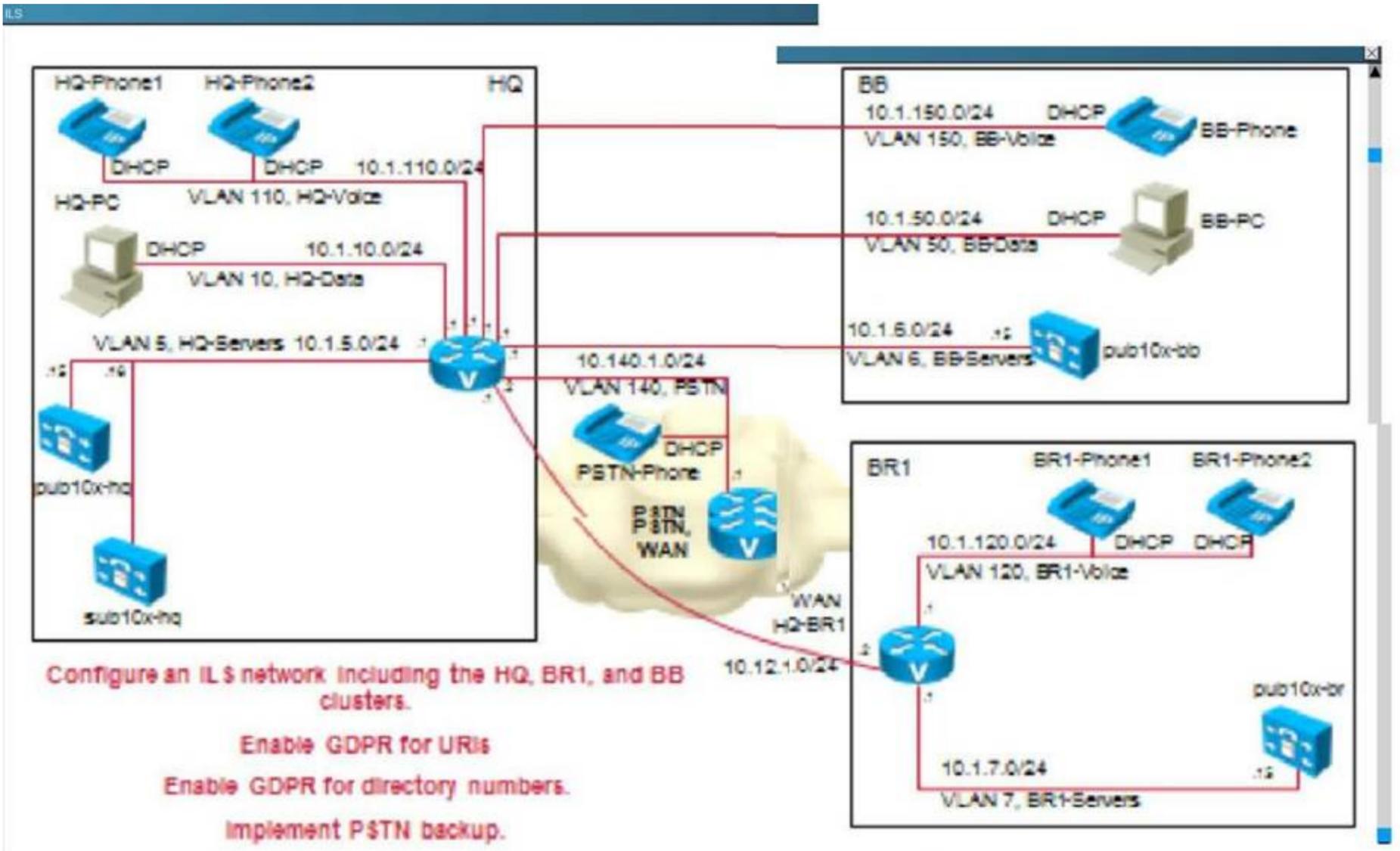
Device Pool:

Device Pool (1 - 3 of 3)				
Name	Osco Unified CM Group	Region	Date/Time	
<input type="checkbox"/> Default	Default	Default	CMLocal	
<input type="checkbox"/> Default	Default	Default	CMLocal	
<input type="checkbox"/> GSM	Default	GSM	CMLocal	

Expressway:



ILS:



Locations:

Locations (1 - 3 of 3)

Find Locations where Location begins with Find Clear Filter + -

<input type="checkbox"/>	
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

Add New Select All Clear All Delete Selected

MRA:

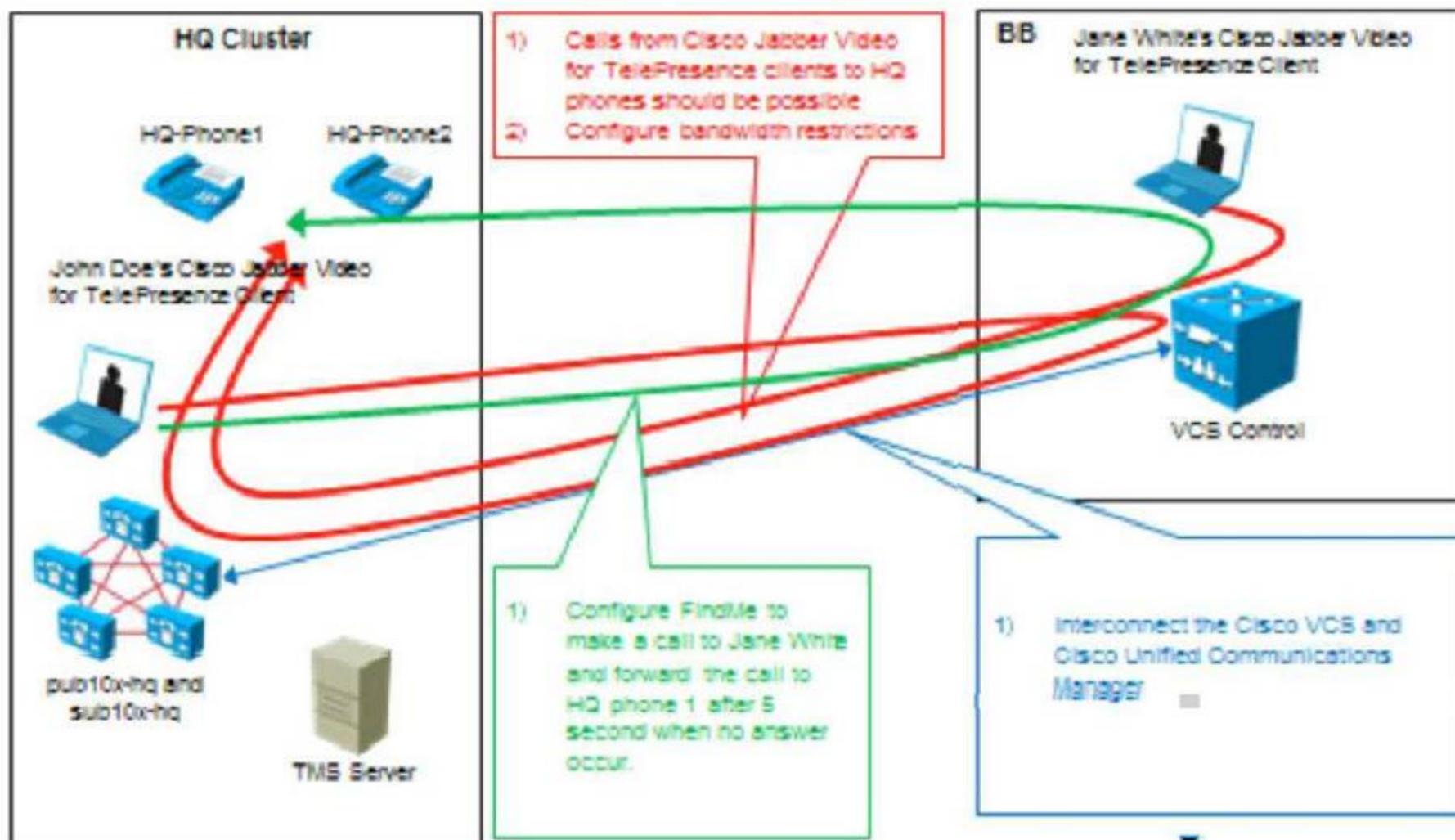
```
ip host _collab-edge._tls.hq.cisco.com SRV 1 1 8443 vcsxe8x-
hq.hq.collab10x.cisco.com
ip host vcsxe8x-hq.hq.collab10x.cisco.com 10.1.5.20
```

Speed Dial:

Phone	Speed Dial Button	Speed Dial Destination
HQ phone 1	1	hq2@cisco.lab
HQ phone 1	2	br1@cisco.lab
HQ phone 1	3	bb@cisco.lab
HQ phone 2	1	hq1@cisco.lab
HQ phone 2	2	br1@cisco.lab
HQ phone 2	2	br1@cisco.lab
HQ phone 2	3	bb@cisco.lab
BR1 phone 1	1	hq1@cisco.lab
BR1 phone 1	2	hq2@cisco.lab
BR1 phone 1	3	bb@cisco.lab
BB phone	1	hq1@cisco.lab
BB phone	2	hq2@cisco.lab

SIP Trunk:

SIP Trunk



Which three configuration tasks need to be completed on the router to support the registration of Cisco Jabber clients? (Choose three.)

- A. The DNS server has the wrong IP address.
- B. The internal DNS Service (SRV) records need to be updated on the DNS Server.
- C. Flush the DNS Cache on the client.
- D. The DNS AOR records are wrong.

E. Add the appropriate DNS SRV for the Internet entries on the DNS Server.

Answer: BCE

NEW QUESTION 249

Scenario:

There are two call control systems in this item.

The Cisco UCM is controlling the DX650, the Cisco Jabber for Windows Client, and the 7965 and 9971 Video IP Phones.

The Cisco VCS and TMS control the Cisco TelePresence Conductor, the Cisco TelePresence MCU, and the Cisco Jabber TelePresence for Windows.

DNS Server:

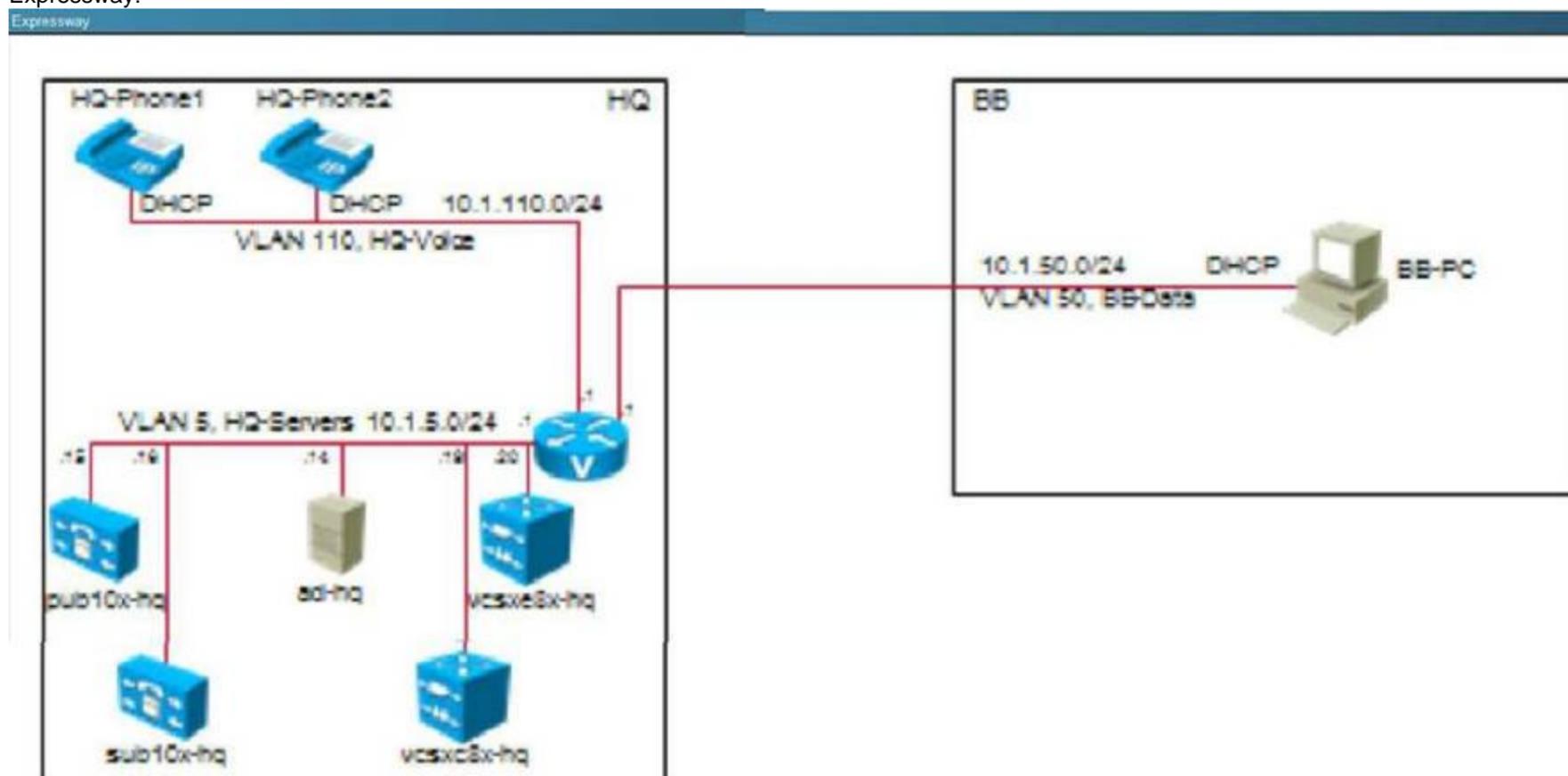
```
ip dns server
ip host _cisco-uds._tcp.hq.cisco.com srv 1 1 8443 10.1.5.15
ip host _cisco-uds._tcp.hq.cisco.com srv 1 1 8443 10.1.5.16
ip host pub10x-hq.collab10x.cisco.com 10.1.5.15
ip host sub10x-hq.collab10x.cisco.com 10.1.5.16
ip host pub10x-hq.hq.collab10x.cisco.com 10.1.5.15
ip host sub10x-hq.hq.collab10x.cisco.com 10.1.5.16
ip host hq.cisco.com 10.1.5.1
```

Device Pool:

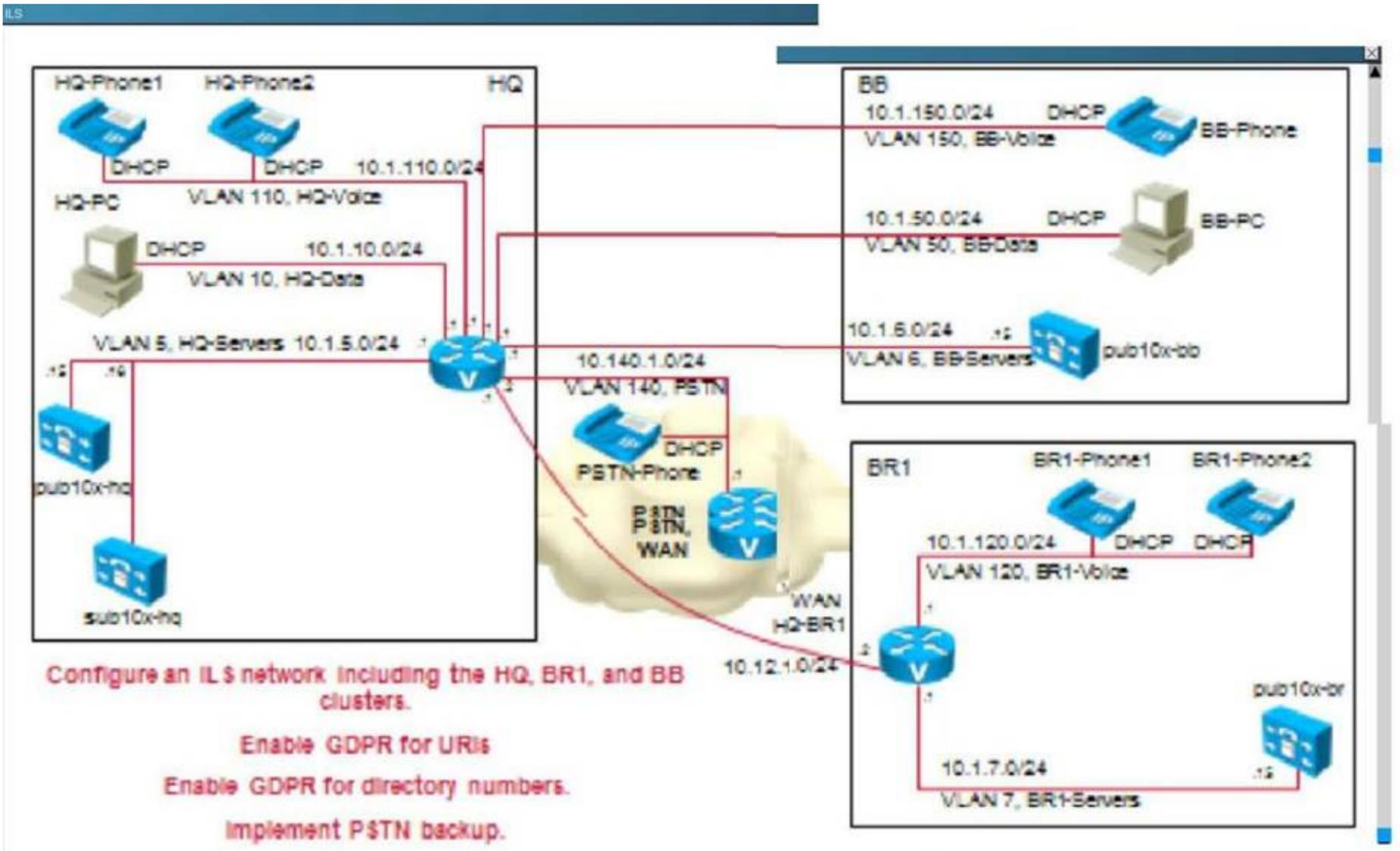
Device Pool				
Status				
3 records found				
Device Pool (1 - 3 of 3)				
Find Device Pool where		Device Pool Name	begins with	Find Clear Filter
<input type="checkbox"/>	Name ^	Cisco Unified CM Group	Region	Date/Ti
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	Default	Default	Default	CMLocal
<input type="checkbox"/>	GSM	Default	GSM	CMLocal

Buttons: Add New, Select All, Clear All, Delete Selected

Expressway:



ILS:



Locations:

Locations

Locations (1 - 3 of 3)

Find Locations where Location begins with Find Clear Filter

<input type="checkbox"/>	
<input type="checkbox"/>	Hub_None
<input type="checkbox"/>	Phantom
<input type="checkbox"/>	Shadow

MRA:

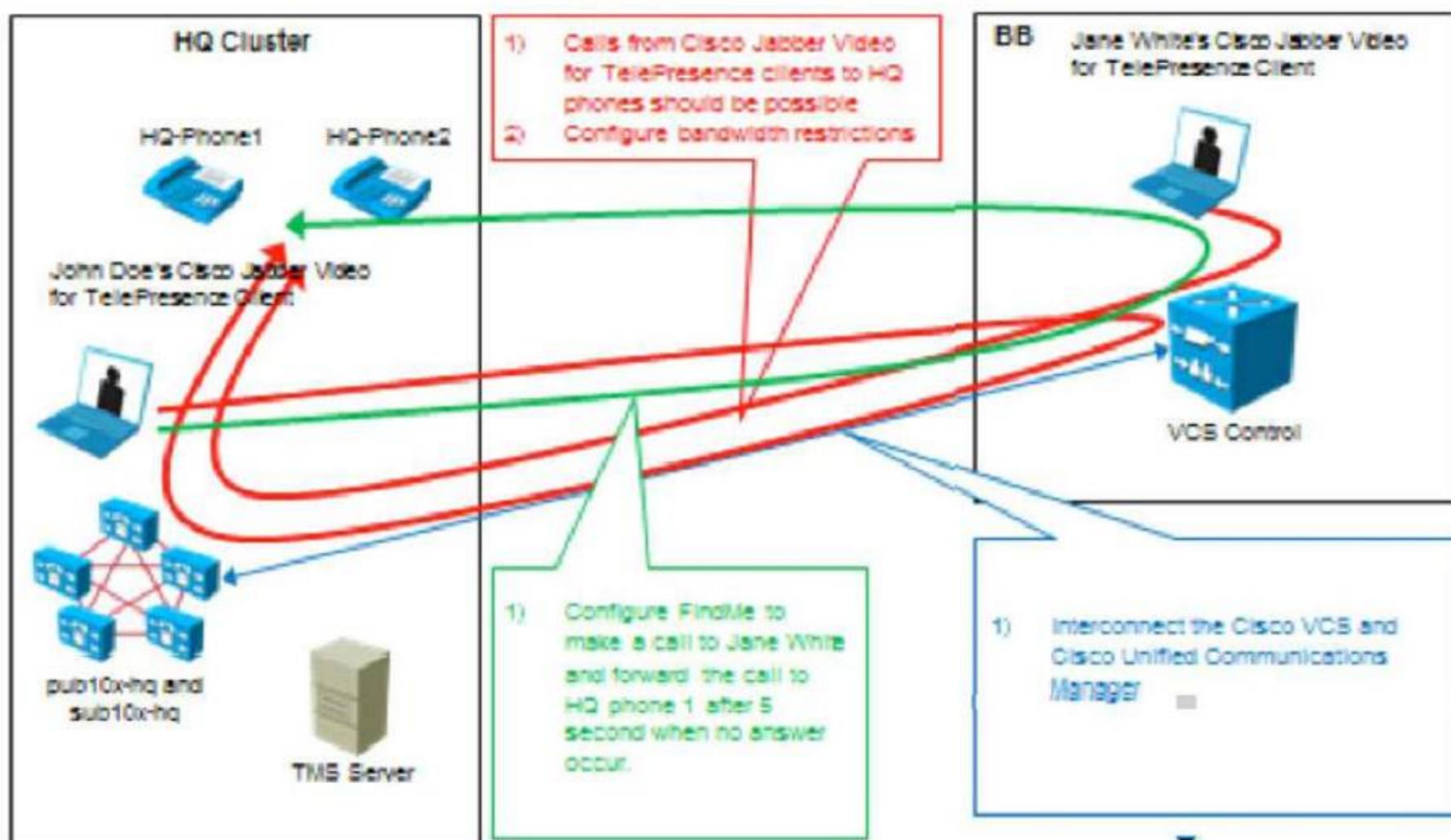
```
ip host _collab-edge._tls.hq.cisco.com SRV 1 1 8443 vcsxe8x-
hq.hq.collab10x.cisco.com
ip host vcsxe8x-hq.hq.collab10x.cisco.com 10.1.5.20
```

Speed Dial:

Phone	Speed Dial Button	Speed Dial Destination
HQ phone 1	1	hq2@cisco.lab
HQ phone 1	2	br1@cisco.lab
HQ phone 1	3	bb@cisco.lab
HQ phone 2	1	hq1@cisco.lab
HQ phone 2	2	br1@cisco.lab
HQ phone 2	2	br1@cisco.lab
HQ phone 2	3	bb@cisco.lab
BR1 phone 1	1	hq1@cisco.lab
BR1 phone 1	2	hq2@cisco.lab
BR1 phone 1	3	bb@cisco.lab
BB phone	1	hq1@cisco.lab
BB phone	2	hq2@cisco.lab

SIP Trunk:

SIP Trunk



The intercluster URI call routing no longer allows calls between sites. What is the reason why this would happen?

- A. Wrong SIP domain configured.
- B. User is not associated with the device.
- C. IP or DNS name resolution issue.
- D. No SIP route patterns for cisco.lab exist.

Answer: CD

NEW QUESTION 254

Which zone will the VCS Control use to route calls to the VCS Expressway?

- A. neighbor zone
- B. DNS zone
- C. traversal client zone
- D. ENUM zone

Answer: C

NEW QUESTION 256

You want to configure Cisco VCS SIP endpoints and H.323 endpoints so that they communicate with one another. To do this, which format must you use in the Search Rule?

- A. name@domain
- B. IP Address (192.168.100.0)
- C. name@hostname
- D. name@IP Address (192.168.100.0)

Answer: A

NEW QUESTION 258

A new administrator at Company X has deployed a VCS Control on the LAN and VCS Expressway in the DMZ to facilitate VPN-less SIP calls with users outside of the network. However, the users report that calls via the VCS are erratic and not very consistent. What must the administrator configure on the firewall to stabilize this deployment?

- A. The VCS Control should not be on the LAN, but it must be located in the DMZ with the Expressway.
- B. The firewall at Company X must have all SIP ALG functions disabled.
- C. The firewall at Company X requires a rule to allow all traffic from the DMZ to pass to the same network that the VCS Control is on.
- D. A TMS server is needed to allow the firewall traversal to occur between the VCS Expressway and the VCS Control servers.

Answer: B

NEW QUESTION 263

Company X wants to implement RSVP-based Call Admission Control and move away from the current location-based configuration. Where does the administrator go to create a default profile?

- A. System > Call Manager > Clusterwide > Service Parameters > RSVP
- B. System > Service Parameters > RSVP
- C. System > Service Parameters > Call Manager > Clusterwide parameters > RSVP
- D. on each MGCP gateway at all remote locations

Answer: C

NEW QUESTION 266

Which two statements about remote survivability are true? (Choose two.)

- A. SRST supports more Cisco IP Phones than Cisco Unified Communications Manager Express in SRST mode.
- B. Cisco Unified Communications Manager Express in SRST mode supports more Cisco IP Phones than SRST.
- C. MGCP fallback is required for ISDN call preservation.
- D. MGCP fallback functions with SRST.

Answer: AD

NEW QUESTION 271

Which two options enable routers to provide basic call handling support for Cisco Unified IP Phones if they lose connection to all Cisco Unified Communications Manager systems? (Choose two.)

- A. SCCP fallback
- B. Cisco Unified Survivable Remote Site Telephony
- C. Cisco Unified Communications Manager Express
- D. MGCP fallback
- E. Cisco Unified Communications Manager Express in SRST mode

Answer: BE

NEW QUESTION 273

How long is the default keepalive period for SRST in Cisco IOS?

- A. 45 sec
- B. 30 sec
- C. 60 sec
- D. 120 sec

Answer: B

NEW QUESTION 278

Which three commands can be used to verify SRST fallback mode? (Choose three.)

- A. show telephony-service all
- B. show telephony-service ephone-dn
- C. show telephony-service ephone
- D. show telephony-service voice-port
- E. show telephony-service tftp-bindings

Answer: ABC

NEW QUESTION 280

Company X has a Cisco Unified Communications Manager cluster and a Cisco Unity Connection cluster at its head office and implemented SRST for its branch offices. One Monday at 2:00 pm, the WAN connection to a branch office failed and stayed down for 45 minutes. That day the help desk received several calls from the branch saying their voicemail was not working but they were able to make and receive calls. Why did the users not realize the WAN was down and prevented access to their voicemail?

- A. All the phones should have started ringing the instant the WAN connection failed to signal the start of SRST mode.
- B. All calls should have dropped when the WAN failed so users would be instantly aware.
- C. Although the phones were still working, the users should have noticed that the phone displays said "SRST Fallback Active".
- D. The voice administrators at the head office did not call the users to notify them.

Answer: C

NEW QUESTION 282

When using SAF, how do you prevent multiple nodes in a cluster from showing up in the Show Advance section of the SAF Forwarder configuration?

- A. Configure the publisher node only in the SAF Forwarder configuration page.
- B. Append an @ symbol at the end of the client label value in the SAF Forwarder configuration page.
- C. Configure the correct node in the EIGRP configuration of the gateway router that is associated with the Cisco Unified Communications Manager node.
- D. Configure the SAF Security Profile Configuration to support only a single node.

Answer: B

NEW QUESTION 283

Which functionality does ILS use to link all hub clusters in an ILS network?

- A. Fullmesh
- B. Automesh
- C. ILS updates
- D. multicast

Answer: B

NEW QUESTION 284

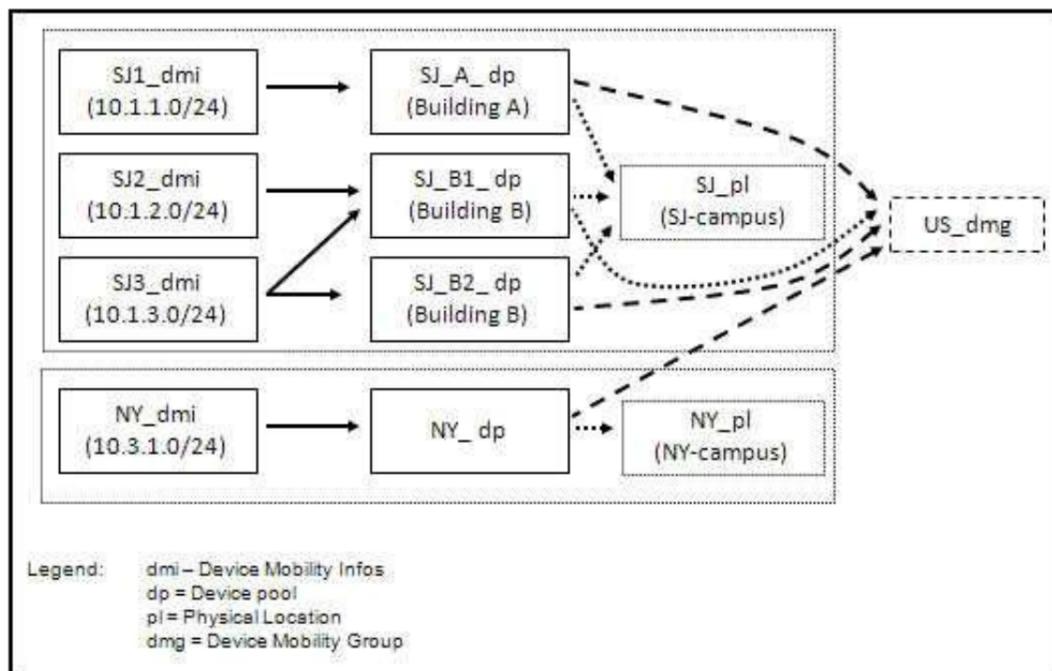
Which option is known as the location attribute that the global dialplan replication uses to advertise its dial plan information?

- A. location controller
- B. route pattern
- C. route string
- D. URI

Answer: C

NEW QUESTION 287

Refer to the exhibit.



If an IP phone in San Jose roams to New York, which two IP phone settings will be modified by Device Mobility so that the phone can place and receive calls in New York? (Choose two.)

- A. The physical locations are not different, so the configuration of the phone is not modified.
- B. The physical locations are different, so the roaming-sensitive parameters of the roaming device pool are applied.
- C. The device mobility groups are the same, so the Device Mobility-related settings are applied in addition to the roaming-sensitive parameters.
- D. The Device Mobility information is associated with one or more device pools other than the home device pool of the phone, so one of the associated device pools is chosen based on a round-robin load-sharing algorithm.
- E. The Device Mobility information is associated with the home device pool of the phone, so the phone is considered to be in its home location.
- F. Device Mobility will reconfigure the roaming-sensitive settings of the phone.

Answer: BC

NEW QUESTION 290

How many Cisco Unified Mobility destinations can be configured per user?

- A. 1
- B. 10
- C. 4
- D. 6

Answer: B

NEW QUESTION 295

Which two bandwidth management parameters are available during the configuration of Cisco Unified Communications Manager regions? (Choose two.)

- A. Default Audio Call Rate
- B. Max Audio Bit Rate
- C. Default Video Call Rate
- D. Max Video Call Bit Rate (Includes Audio)
- E. Max Number of Video Sessions

Answer: BD

NEW QUESTION 296

In Cisco Unified Communications Manager, where do you configure the default bit rate for audio and video devices?

- A. Enterprise Parameters
- B. Region under Region Information
- C. Cisco CallManager service under Service Parameter Configuration
- D. Enterprise Phone Configuration

Answer: C

NEW QUESTION 297

Which module is the minimum PVDM3 module needed to support video transcoding?

- A. PVDM3-32
- B. PVDM3-64
- C. PVDM3-128
- D. PVDM3-192

Answer: C

NEW QUESTION 301

When implementing a Media Termination Point, what determines the number of sessions that is supported on each DSP?

- A. the codecs that are used in universal transcoding mode
- B. the size of the cluster that is being designed
- C. the number of full-duplex media streams
- D. the Cisco Unified Communications Manager node setting

Answer: A

NEW QUESTION 302

In a distributed call processing network with locations-based CAC, calls are routed to and from intercluster trunks. Which trunk type is implemented in this network?

- A. intercluster trunk with gatekeeper control
- B. intercluster trunk without gatekeeper control
- C. SIP trunk
- D. h225 trunk

Answer: B

NEW QUESTION 303

Refer to the exhibit.

```
Router(config)#class-map VIDEO
Router(config-cmap)#match dscp ?
<0-63> Differentiated services codepoint value
af11 Match packets with AF11 dscp (001010)
af12 Match packets with AF12 dscp (001100)
af13 Match packets with AF13 dscp (001110)
af21 Match packets with AF21 dscp (010010)
af22 Match packets with AF22 dscp (010100)
af23 Match packets with AF23 dscp (010110)
af31 Match packets with AF31 dscp (011010)
af32 Match packets with AF32 dscp (011100)
af33 Match packets with AF33 dscp (011110)
af41 Match packets with AF41 dscp (100010)
af42 Match packets with AF42 dscp (100100)
af43 Match packets with AF43 dscp (100110)
cs1 Match packets with CS1(precedence 1) dscp (001000)
cs2 Match packets with CS2(precedence 2) dscp (010000)
cs3 Match packets with CS3(precedence 3) dscp (011000)
cs4 Match packets with CS4(precedence 4) dscp (100000)
cs5 Match packets with CS5(precedence 5) dscp (101000)
cs6 Match packets with CS6(precedence 6) dscp (110000)
cs7 Match packets with CS7(precedence 7) dscp (111000)
default Match packets with default dscp (000000)
ef Match packets with EF dscp (101110)

Router(config-cmap)#match dscp █
```

The "DSCP for Video Calls" Cisco CallManager service parameter is set to 34. What is the correct DSCP value to use when configuring a class map in a Cisco IOS router?

- A. cs4
- B. ef
- C. af23
- D. af41

Answer: D

NEW QUESTION 304

A voice engineer is enabling video capabilities between H.323 and SIP endpoints. Which component allows for standardized caller addresses between the endpoints?

- A. search rules
- B. SIP route pattern
- C. policy service
- D. transform

Answer: B

Explanation:

Reference: http://www.cisco.com/en/US/tech/tk652/tk701/technologies_configuration_guide_chapter09186a00800eadee.html

NEW QUESTION 305

How many nodes can a phone establish a connection to at the same time?

- A. 4
- B. 3
- C. 1
- D. 2

Answer: D

NEW QUESTION 306

Which three steps configure Cisco Unified Survivable Remote Site Telephony for SIP phones? (Choose three.)

- A. Configure voice register pool.
- B. Configure voice register global dn.
- C. Configure an SRST reference.
- D. Configure a phone NTP reference.
- E. Configure the SIP registrar.
- F. Configure telephony service.

Answer: ACE

NEW QUESTION 311

An engineer is working on a Cisco VCS Control routing configuration and wants users to be able to dial ccnpcollab and have calls routed to ccnpcollab@cisco.com. Which option achieves this aim?

- A. search rules
- B. transforms
- C. access rules
- D. call policy

Answer: B

NEW QUESTION 314

An engineer must resolve a call failure issue. When using RTMT, the engineer notices that the Location Bandwidth Manager-OutOfResources counter is showing a positive value. Which option is the cause of the call failure?

- A. lack of audio bandwidth
- B. lack of video bandwidth
- C. lack of transcoding resources
- D. lack of audio or video bandwidth
- E. lack of conferencing resources

Answer: A

NEW QUESTION 318

While troubleshooting a connectivity issue between Cisco Unified Communications Manager, Expressway-C, and Expressway-E, an engineer sees this output in the Expressway-E logs. Event="Authentication Failed" Service="SIP" Src-ip="10.50.2.1" Src-port="25723" Detail="Incorrect authentication credential for user" Protocol "TLS" Method="OPTIONS" Level="1"
What is the cause of this issue?

- A. The Expressway-C Traversal Server username/password do not match the Expressway-E Traversal Zone username/password.
- B. The Expressway-C Traversal Client username/password do not match the Expressway-E Traversal Server username/password.
- C. The Expressway-C Traversal Zone username/password do not match the Expressway-E Traversal Zone username/password.
- D. The Expressway-C Traversal Zone username/password do not match the Expressway-E Traversal Client username/password.
- E. The Expressway-C Traversal Server username/password do not match the Expressway-E Traversal Client username/password.

Answer: B

NEW QUESTION 320

An engineer is performing an international multisite deployment and wants to create an effective backup method to access TEHO destinations in case the call limit triggers. Which technology provides this ability?

- A. AAR
- B. CFUR
- C. LRG
- D. SRST

Answer: A

NEW QUESTION 322

An engineer is configuring a SIP profile for Cisco VCS SIP trunk on Cisco Unified Communications Manager and enables the option "Use Fully Qualified Domain Name" in SIP Requests. Which result is achieved by enabling this option?

- A. Resolve FQDN using DNS type SRV record.
- B. Resolve FQDN using DNS type A record.
- C. Ensure FQDN is used in SIP Identity header.
- D. Ensure FQDN is used in SIP Request header.

Answer: C

Explanation:

Reference: http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/8_6_1/ccmcfg/bccm-861cm/b06siprf.html

NEW QUESTION 327

Which option indicates the best QoS parameters for interactive video?

- A. 0% Max Loss, 100 ms One-way Latency, 30 ms Jitter, 20% Overprovisioning
- B. 1% Max Loss, 160 ms One-way Latency, 60 ms Jitter, 10% Overprovisioning
- C. 1% Max Loss, 150 ms One-way Latency, 30 ms Jitter, 20% Overprovisioning
- D. 5% Max Loss, 5 s One-way Latency, 30 ms Jitter, 20% Overprovisioning

Answer: C

Explanation:

Reference: http://www.cisco.com/c/en/us/td/docs/solutions/Enterprise/WAN_and_MAN/QoS_SRND/QoS-SRND-Book/QoSIntro.html (interactive video)

NEW QUESTION 330

An engineer is configuring Global Dial Plan Replication and wants to prevent the local cluster from routing the Vice President number 5555555555 to the remote cluster. Which action accomplishes this task?

- A. Create a block route pattern.
- B. Create a block learned pattern.
- C. Create a block transformation pattern.
- D. Create a block translation pattern.

Answer: C

NEW QUESTION 331

Which two steps must you take when implementing TEHO in your environment? (Choose two.)

- A. Implement local failover.
- B. Implement SIP to POTS.
- C. Load-balance PRI connections.
- D. Load-balance route lists within the cluster.
- E. Implement ICT trunks to remote locations.
- F. Implement centralized failover.

Answer: AD

NEW QUESTION 334

You have deployed a Cisco 2821 ISR to perform as an SRST voice gateway at a remote site. During a network failure between the remote site and the central office, some of the phones located at the remote site are unable to make phone calls. Which two options are potential causes of the problem? (Choose two.)

- A. The site has exceeded the number of SRST endpoints supported by the voice gateway.
- B. The ccm-manager fallback command is configured incorrectly on the voice gateway.
- C. Phones at the remote site are assigned to the incorrect device pool.
- D. The ccm-manager fallback-mgcp command is configured incorrectly on the voice gateway.
- E. The site has exceeded the number of simultaneous calls allowed in SRST mode.

Answer: AC

NEW QUESTION 335

Refer to the exhibit.



An engineer is deploying a new Cisco VCS Expressway for a company and has configured the IP address and the system name. After logging into the Cisco VCS Expressway admin page, the engineer sees this output. Which four options must be configured to complete the Cisco VCS Expressway system configuration? (Choose four.)

- A. NTP server
- B. SIP server
- C. LDAP server
- D. security certificate
- E. DHCP server
- F. DNS server
- G. SIP URI
- H. Cisco Unified Communications Manager IP address

Answer: ACDF

NEW QUESTION 339

After forgetting to log out of his IP phone in the main office, an Extension Mobility user is unable to log in to a different IP phone at a remote office. Which option is a possible reason for the problem?

- A. The phone at the remote location is a different model than the phone in the user's main office.
- B. The user's Extension Mobility profile is misconfigured.
- C. The user can log in to only one device at a time.
- D. The device pool is misconfigured.

Answer: C

NEW QUESTION 342

An engineer has configured a Cisco EX60 to register with a Cisco VCS-C, but the device is not showing up as registered. During troubleshooting, which component will the engineer likely find missing in the configuration?

- A. gatekeeper
- B. MCU
- C. default gateway
- D. TMS
- E. DNS

Answer: A

NEW QUESTION 343

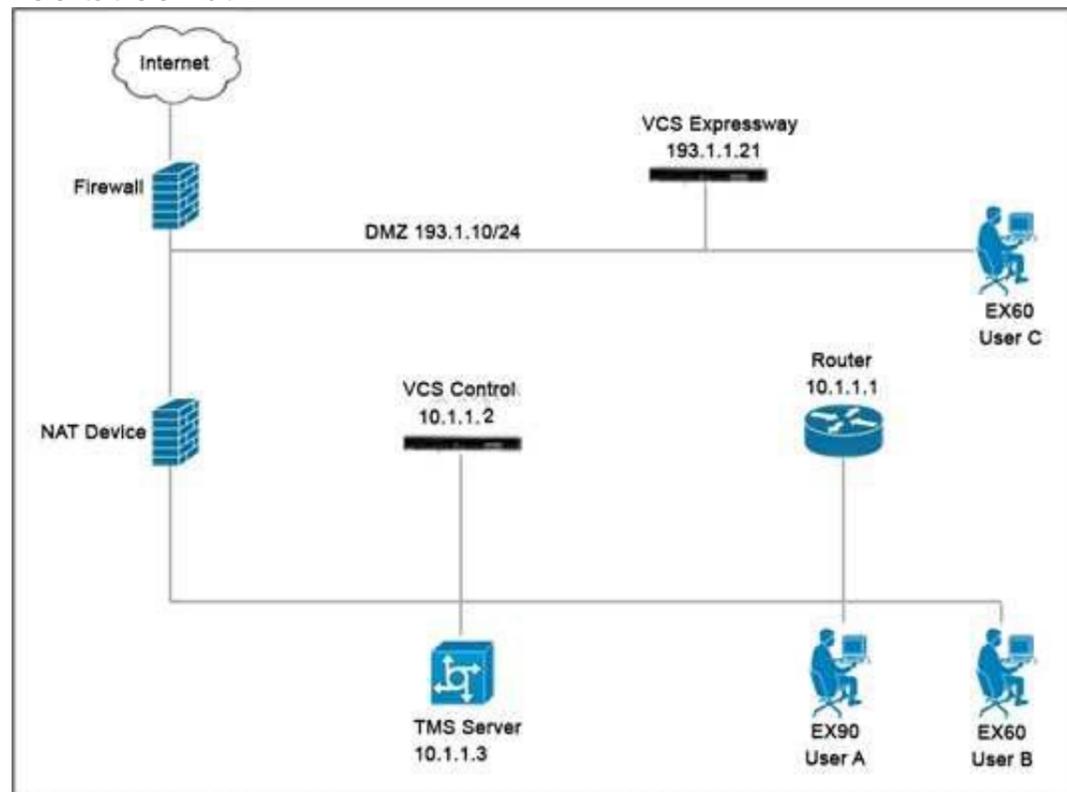
An engineer is configuring a new DX-80 in Cisco Unified Communications Manager. Where can an engineer verify the default DSCP value of AF41?

- A. enterprise phone configuration
- B. common phone profile
- C. service parameters
- D. enterprise parameters

Answer: C

NEW QUESTION 347

Refer to the exhibit.



Which three statements about when user A calls user using SIP are true? (Choose three.)

- A. SIP TCP/TLS ports must be opened from internal to DMZ and vice versa.
- B. Deploying a Cisco VCS Expressway behind a NAT mandates the use of the Advanced Networking option key.
- C. Cisco VCS Control and Cisco VCS Expressway support static NAT.
- D. Deploying a Cisco VCS Control inside a NAT mandates the use of the Advanced Networking option key.
- E. RT and RTCP ports must be opened at the firewall from internal to DMZ and vice versa
- F. The NAT device must translate from 10.X.X X to 193.1.1.X and vice versa.

Answer: ABE

NEW QUESTION 349

When configuring intercluster URI dialing, an engineer gets the error message "Local cluster cannot connect to the ILS network". Which three reasons for this error are true? (Choose three.)

- A. The SIP route patterns have not been properly configured.
- B. The Tomcat certificates do not match.
- C. The Cisco Unified Resource Identifier service needs a restart.
- D. The ILS authentication password does not match.

- E. The cluster ID does not match.
- F. One cluster is using TLS certificate, and the other is using Password.

Answer: BDF

NEW QUESTION 350

When you configure TEHO for long-distance calls and use the local PSTN gateways as fallback, how many route patterns do you require for a cluster with five sites that are located in different area codes?

- A. 6 when using a local route group
- B. 10 when not using a local route group
- C. 5 when using a local route group
- D. 15 when not using a local route group

Answer: A

NEW QUESTION 353

Which two configurable options are available to enable Early Offer for calls over a Cisco Unified Communications Manager SIP trunk? (Choose two.)

- A. Early Offer support for voice and video calls Mandatory (insert MTP if needed)
- B. Media Termination Point Required
- C. Use Trusted Relay Point
- D. Accept Audio Codec Preferences in Received Offer
- E. No Media Termination Point Required

Answer: AB

NEW QUESTION 357

Where do you go to configure restrictions on audio bandwidth?

- A. location
- B. partition
- C. region
- D. enterprise parameters
- E. serviceability

Answer: A

NEW QUESTION 361

On which Cisco Unified Communications Manager configuration parameter does the CODEC that a Cisco IP Phone uses for a call depend?

- A. media resources
- B. region
- C. physical location
- D. enterprise parameters
- E. location

Answer: B

NEW QUESTION 365

.....

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