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<http://www.lead2pass.com/300-075.html> QUESTION 126 Which code snippet is required for SAF to be initialized? A. Service FamilyB. External-ClientC. router eigrpD. topology baseAnswer: C QUESTION 127 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 QUESTION 128 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 QUESTION 129 If you want to delete a SAF-enabled trunk from Cisco Unified Communications Manager Administration, what must you do first? A. Disassociate the trunk from the CCD advertising service or CCD requesting service.B. Delete the trunk from the CCD requesting service node.C. Place the Cisco Unified Communications Manager node in standby mode.D. Redirect CCD advertising and requesting services to another Cisco Unified Communications Manager. Answer: A QUESTION 130 Which functionality does ILS use to link all hub clusters in an ILS network? A. FullmeshB. AutomeshC. ILS updatesD. multicast Answer: B QUESTION 131 Which option is known as the location attribute that the global dialplan replication uses to advertise its dial plan information? A. location controllerB. route patternC. route stringD. URI Answer: C QUESTION 132 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. Device Mobility will reconfigure the roaming-sensitive settings of the phone. Answer: BC QUESTION 133 What happens when a user logs in using the Cisco Extension Mobility Service on a device for which the user has no user device profile? A. The Extension Mobility log in fails.B. The device takes on the default device profile for its type.C. The user can log in but does not have access to any features, soft key templates, or button templates.D. The device uses the first device profile assigned to the user in Cisco Unified Communications Manager. Answer: B QUESTION 134 Which three steps are required when configuring extension mobility in Cisco Unified Communications Manager? (Choose three.) A. Create the extension mobility IP Phone Service.B. Check the Home Cluster checkbox on the End User Configuration page.C. Check the Enable Extension Mobility checkbox on the Directory Number Configuration page.D. Unsubscribe all other services from the Cisco IP Phone.E. Create a user Device Profile.F. Subscribe the extension mobility IP Phone Service to the user Device Profile. Answer: AEF QUESTION 135 How many Cisco Unified Mobility destinations can be configured per user? A. 1B. 10C. 4D. 6 Answer: B QUESTION 136 When configuring Cisco Unified Mobility, which parameter defines the access control for a call that reaches out to a remote destination? A. Calling Party Transformation Calling Search Space under Remote Destination Profile InformationB. User Local under Remote Destination Profile InformationC. Rerouting Calling Search Space under Remote Destination Profile InformationD. Rerouting Calling Search Space under Remote Destination informationE. Calling Search Space under Phone Configuration Answer: C QUESTION 137 Which two bandwidth management parameters are available during the configuration of Cisco Unified Communications Manager regions? (Choose two.) A. Default Audio Call RateB. Max Audio Bit RateC. Default Video Call RateD. Max Video Call Bit Rate (Includes Audio)E. Max Number of Video Sessions Answer: BD QUESTION 138 In Cisco Unified Communications Manager, where do you configure the default bit rate for audio and video devices? A. Enterprise ParametersB. Region under Region InformationC. Cisco CallManager service under Service Parameter ConfigurationD. Enterprise Phone Configuration Answer: C QUESTION

139 Which module is the minimum PVDM3 module needed to support video transcoding? A. PVDM3-32B. PVDM3-64C. PVDM3-128D. PVDM3-192 Answer: C QUESTION 140 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 modeB. the size of the cluster that is being designedC. the number of full-duplex media streamsD. the Cisco Unified Communications Manager node setting Answer: A QUESTION 141 Which statement is true when considering a Cisco VoIP environment for regional configuration? A. G.711 requires 128K of bandwidth per call.B. G.729 requires 24K of bandwidth per call.C. The default codec does not matter if you have defined a hardware MTP in your Cisco Unified Communications Manager environment.D. To deploy a Cisco H.323 gatekeeper, you must configure MTP resources on the gatekeeper and only use G.711 between regions. Answer: B QUESTION 142 When you use the Query wizard to configure the trace and log central feature to collect install logs, if you have servers in a cluster in a different time zone, which time is used? A. TLC adjusts the time change appropriately.B. TLC uses its local time for all systems.C. TLC queries for the time zone as part of configuration.D. TLC produces an error and must be run remotely. Answer: A QUESTION 143 When a call is made from a video endpoint to a Cisco TelePresence EX90 that is registered to a Cisco VCS Control, which portion of the destination URI is the first match that is attempted? A. the full URI, including the domain portionB. the destination alias, without the domain portionC. the E.164 number that is assigned to the Cisco TelePresence EX90D. the directory number that is assigned to the Cisco TelePresence EX90 Answer: B QUESTION 144 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 controlB. intercluster trunk without gatekeeper controlC. SIP trunkD. h225 trunk Answer: B QUESTION 145 Refer to the exhibit. 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. cs4B. efC. af23D. af41 Answer: D QUESTION 146 What is the default DSCP/PHB for TelePresence video conferencing packets in Cisco Unified Communications Manager? A. EF/46B. CS6/48C. AF41/34D. CS3/24E. CS4/32 Answer: E QUESTION 147 How is a SIP trunk in Cisco Unified Communications Manager configured for SIP precondition? A. The configuration is done by selecting a SIP precondition trunk for trunk type.B. The configuration is automatically selected when RSVP is enabled for the location assigned to the trunk.C. SIP precondition is configured by selecting E2E for RSVP over SIP on the default SIP profile assigned to the SIP trunk.D. SIP precondition is configured by configuring a new SIP profile and selecting E2E for RSVP over SIP. The new SIP profile must then be assigned to the SIP trunk. Answer: D QUESTION 148 Which statement about SIP precondition is most correct? A. When configuring SIP precondition, the SIP trunk must have access to an RSVP agent.B. When configuring SIP precondition, the IP phones must have access to an RSVP agent.C. When configuring SIP precondition, the IP phones and SIP trunk must have access to an RSVP agent.D. RSVP agents are only required for the IP phones. SIP trunks require RSVP agents only when fall back to local RSVP is configured.E. SIP trunk will always require RSVP agents regardless of what RSVP type is configured. Answer: D QUESTION 149 Refer to the exhibit. Assume that NANP is being used and 9 is used for PSTN access code Long distance national calls are preceded with 1. How should the HQ Cisco Unified Communications Manager be configured for calls to 3XXX to be sent to the gatekeeper at 1 0 1 6 1 with PSTN backups? A. Configure a route pattern for 3XXX Assign this route pattern to a route list that points to two route groups. The first route group contains the H 225 trunk. The second route group contains the MGCP gateway with prefix digits 1 408555 for the outgoing called number.B. Configure a route pattern for 1#3XXX Assign this route pattern to a route list that points to a route group that lists the H 225 trunk as first choice and the MGCP gateway as a second choice.C. Configure a route pattern for 4085543XXX. Assign this route pattern to a route list that points to two route groups. The first route group contains the H 226 trunk. The second route group contains MGCP gateway.D. Configure a route pattern for 3XXX Assign this route pattern to a route list that points to two route groups. The first route group contains the H 225 trunk. The second route group contains MGCP gateway with prefix digits 91 408554 for the called number. Answer: A QUESTION 150 Refer to the exhibit. IT shows an H.323 gateway configuration in a Cisco Unified Communications Manager environment. An inbound PSTN call to this H.323 gateway fails to connect to IP phone extension 2001. The PSTN user hears a reorder tone. Debug isdn q931 on the H.323 gateway shows the correct called-party number as 5015552001. Which two configuration changes can correct this issue? (Choose two.) A. Add port 1/0:23 to dial-peer voice 123 pots.B. Ensure that the Significant Digits for inbound calls on the H.323 gateway configuration is 4.C. Add a voice translation profile to truncate the number from 10 digits to 4 digits. Apply the voice translation profile to the Voice-port. The configuration field "Significant Digits for inbound calls" is left at default (All).D. Add the command h323-gateway voip id on interface vlan120.E. Change the destination-pattern on the dial-peer voice 23000 VoIP to 501501? and change the Significant Digits for inbound calls to 4. Answer: BEE Explanation: Choose the number of significant digits to collect, from 0 to 32. Cisco Unified Communications Manager counts significant digits from the right (last digit) of the number that is called. Lead2pass offers the latest 300-075 PDF and VCE dumps

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