## [2017 New 300-075 New Questions For Passing The 300-075 Certification Exam (76-100)

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Roaming-sensitive settings are settings that do not have an impact on call routing. Device Mobility settings, on the other hand, may have an impact on call routing because they modify the device CSS, AAR group, and AAR CSS.Answer: D QUESTION 77When device mobility mode is enabled or disabled for a cluster, to which does the cluster setting apply? A. all phones in the cluster that support device mobilityB. all phones in the cluster that subscribed to device mobilityC. mobile phones in the cluster that support device mobilityD. mobile phones in the cluster that are in default mode Answer: A QUESTION 78Which statement is true when device mobility mode is enabled or disabled in the Phone Configuration window? A. The device mobility mode phone settings take precedence over the service parameter settings.B. The service parameter settings take precedence over the device mobility mode phone settings.C. The combined service parameter settings and the device mobility mode phone settings will be used.D. The default settings will be used due to the conflicts. Answer: A QUESTION 79Which statement about the host portion format in Cisco Unified Communications Manager URI dialing is false? A. The host portion cannot start or end with a hyphen.B. The host portion is not case sensitive.C. The host portion accepts characters a-z, A-Z, 0-9, hyphens, and periods.D. The host portion can have two periods in a row. Answer: D QUESTION 80Which 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 QUESTION 81Which system configuration is used to set a restriction on audio bandwidth? A. regionB. locationC. physical locationD. licensing Answer: B QUESTION 82Which 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 QUESTION 83The 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 QUESTION 84How many active gatekeepers can you can define in a local zone? A. 1B. 2C. 5D. 10E. 15F. unlimited Answer: A QUESTION 85When video endpoints register with Cisco Unified Communications Manager, where are DSCP values configured? A. in Unified CM, under Enterprise Parameters ConfigurationB. in Unified CM, under Device > Device Settings > Device DefaultsC. in Unified CM, under Service Parameters > Cisco CallManager Service > Cluster-wide ParametersD. DSCP parameters are always configured on each individual video endpoint. Answer: C QUESTION 86Which statement best describes globalized call routing in Cisco Unified Communications Manager? A. All incoming calling numbers on the phones are displayed as an E.164 with the + prefix.B. Call routing is based on numbers represented as an E.164 with the + prefix format.C. All called numbers sent out to the PSTN are in E.164 with the + prefix format.D. The CSS of all phones contain partitions assigned to route patterns that are in global format.E. All phone directory numbers are configured as an E.164 with the + prefix. Answer: B QUESTION 87Which device is used to connect to the H.323 gatekeeper? A. H.323 gatewayB. SIP trunkC. H.323 trunkD. MGCP gateway Answer: C QUESTION 88The corporate WAN has been extended to two newly acquired sites and it includes gatekeeper support. Each site has a Cisco CallManager and an H.323 gateway that allows connection to the PSTN. Which connection method is best for these two new customers? A. H.225 trunk (gatekeeper-controlled)B. intercluster trunk (non-gatekeeper controlled)C. SIP trunkD. intercluster trunk (gatekeeper-controlled) Answer: D QUESTION 89Which two are gatekeeper-controlled trunk options that support gatekeeper call administration control? (Choose two.) A. H.323B. H.245C. H.225D. intercluster E. intracluster Answer: CD QUESTION 90Assume 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 subnetB. user settingsC. SRST referenceD. regionE. phone button settings Answer: BEExplanation: Although the phone may have moved from one subnet to another, the physical location and associated services have not changed. QUESTION 91Which three statements about configuring an encrypted trunk between Cisco TelePresence Video Communication Server and Cisco Unified Communications Manager are true? (Choose three.) 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 QUESTION 92Which system configuration is used to set audio codecs? A. regionB. locationC. physical locationD. licensing Answer: A QUESTION 93During device failover to the secondary Cisco Unified Communications Manager server, how does the phone recognize that the primary server is back? A. The secondary server keeps sending keepalive message to the primary server and when it succeeds, it unregisters the phones to force them to register to the primary.B. When the primary server goes online, it sends out an "ALIVE" message via broadcast so that the phones re-register.C. The phones never re-register with the primary server until the active (secondary) one goes down.D. The phone sends keepalive messages to the primary server frequently and when it succeeds, the phone re-registers with it. Answer: D QUESTION 94If the device pool in the phone record does not match the device pools in the matching subnet, what will the system consider the phone to be? A. roamingB. unregisteredC. unknownD. new device Answer: A QUESTION 95Which two Cisco Extension Mobility attributes are available in the user device profile? (Choose two.) A. regionsB. descriptionC. phone button templateD. NTP information Answer: BC QUESTION 96What is the default DSCP/PHB for video conferencing packets in Cisco Unified Communications Manager? A. EF/46B. CS6/48C. AF41/34D. CS3/24 Answer: C QUESTION 97Which feature allows you to specify which endpoints ring when someone calls a user on a specific destination ID? A. FindMEB. Extension MobilityC. Speech ConnectD. Single Number Reach Answer: A QUESTION 98The administrator at Company X is getting user reports of inconsistent quality on video calls between endpoints registered to Cisco Unified Communications Manager. The administrator runs a wire trace while a video call is taking place and sees that the packets are not set to AF41 for desktop video as they should be. Where should the administrator look next to confirm that the correct DSCP markings are being set? A. on the MGCP router at the edge of both networksB. the service parameters in the VCS ControlC. the QoS service parameter in Cisco Unified Communications ManagerD. on the actual Cisco phone itself because the DSCP setting is not part of its configuration file downloaded at registrationE. The setting cannot be changed for video endpoints that are registered to Cisco Unified Communications Manager, but only when they are registered to the VCS Control. Answer: C QUESTION 99Which 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 allB. show dial-peer voice summaryC. show ephone summary D. show voice port summary Answer: A QUESTION 100Which 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 There is no doubt that Lead2pass is the top IT certificate exam material provider. All the braindumps are the latest and tested by senior Cisco lecturers and experts. 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