[New Exam Dumps 100% Valid 300-080 Exam Questions and Answers Everyone Want To Download

New Updated 300-080 Exam Questions from PassLeader 300-080 PDF dumps! Welcome to download the newest PassLeader 300-080 VCE dumps: **http://www.passleader.com/300-080.html** (141 Q&As) Keywords: 300-080 exam dumps, 300-080 exam questions, 300-080 VCE dumps, 300-080 PDF dumps, 300-080 practice tests, 300-080 study guide, 300-080 braindumps, Troubleshooting Cisco IP Telephony & Video v1.0 Exam NEW QUESTION 76   Phone A is able to dial the directory number of Phone B and complete a call. However, when Phone B dials the directory number of Phone A, Phone B receives a fast-busy tone. What is causing this issue? A.   Phone A and Phone B are in different partitions. B.   Phone B does not have Phone A in its partition.   C.   Phone B is not in the CSS of Phone A. D.   Phone B does not have the CSS of Phone A in its partition.   E.   Phone B is not registered.   F.   Phone B has the incorrect CSS to dial Phone A.   Answer: F NEW QUESTION 77   An inbound call from the PSTN is not reaching the directory number that it is calling. When the PSTN phone calls the correct DID, only a dial tone is heard. Which command resolves this issue? A.   (config-dial-peer)#direct-inward-dial B.   (config-controller)# no provide-outside-dialtone   C.   (config-if)#no dial-tone D.   (config-dial-peer)# no dial-tone   E.   (config-if)#direct-inward-dial F.   (config) allow inbound dial-peer 1 Answer: A NEW QUESTION 78   When a caller dials 9 plus an external seven-digit number, the caller hears a fast-busy tone after a period of silence. What is causing the silence? A.   There is no dial route for 9XXXXXXX on Cisco Unified Communications Manager. B.   The gateway is not dropping the leading 9, and the PSTN fails.   C.   The T302 timer is waiting to expire.   D.   The caller does not have the PSTN partition in the CSS.   E.   The caller dialed the wrong number.   F.   To dial successfully, the caller must enter a Forced Authorization Code. Answer: C NEW QUESTION 79   In a single-site deployment model, the internal endpoints are unable to dial from one to the other. What are two possible causes? (Choose two.) A.   The PSTN gateway is not configured. B.   The called endpoint does not have the SIP trunk enabled.   C.   The called endpoint is not registered.   D.   The calling endpoint is not in the CSS of the called endpoint. E.   The called endpoint is not in the partition of the calling endpoint.   F.   The calling endpoint is not configured for the correct CoS. Answer: CF NEW QUESTION 80   What is the default interval for SAF hello packets? A.   15 seconds   B.   15 seconds on links with speeds that are slower than T1 speeds   C.   40 seconds   D.   40 seconds onlinks with speeds that are slower than T1 speeds   E.   60 seconds   F.   60 seconds onlinks with speeds that are slower than T1 speeds Answer: F NEW QUESTION 81   Which configuration can be dynamically set using the Cisco Unified Communications Manager Device Mobility feature? A.   phone model and protocol   B.   SRST reference and directory number   C.   CSS and local gateway   D.   partition and CSS E.   media resources and permanent bridges Answer: C NEW QUESTION 82   Which configuration is required on Cisco TelePresence Server, in order to support 1080p resolution? A.   Screen licenses must be configured.   B.   Cisco TelePresence Server must be in remotely managed mode.   C.   Cisco TelePresence Server must be in HD mode.   D.   Cisco TelePresence Server must be configured with Cisco TelePresence Conductor.   E.   Cisco TelePresence Server must be in Full HD mode. Answer: E NEW QUESTION 83   A user is dialing an external PSTN number with a prefix of 01 from a Cisco TelePresence SX10 Quick Set in a Cisco VCS environment. In the past, the Cisco VCS and the ISDN gateway were correctly configured with a prefix of 01, but the calls are now failing. What are three possible causes? (Choose three.) A.   The Cisco VCS Control is down. B.   The interworking setting is turned off.   C.   The audio feature in the Cisco TelePresenceSX10 is turned off.   D.   The SIP trunk is not configured on the gateway. E.   01 is not a valid prefix.   F.   ISDN is not enabled on the Cisco TelePresenceSX10. G.   The Cisco TelePresenceSX10 is not registered to the Cisco VCS Control.   H.   The Cisco TelePresenceSX10 is not registered to the Cisco Express C. Answer: ABG NEW QUESTION 84   When the command utils dbreplication status is executed on the Cisco Unified Communications Manager CLI, which step should be taken next to check the database replication status? A.   View the activelog file.   B.   Run the same command on all nodes of the cluster.   C.   Restart the Cisco CallManager service.   D.   The

command utils dbreplication runtimestate must be run on the publisher.   E.   The command utils dbreplication runtimestate must be run on the subscriber. Answer: A NEW QUESTION 85   Which tool can you use to see SAF advertisements in Cisco Unified Communications Manager? A.   Cisco Unified Real-Time Monitoring Tool B.   show eigrp neighbors command   C.   debug eigrp commands D.   Terminal Monitor Answer: A NEW QUESTION 86   URI dialing is enabled between two clusters with the default options. The engineer that set up the URI dialing verified that all was working properly. However, a user from one of the clusters cannot dial using URI to a user in the same cluster. What do you do to resolve this issue? A.   Verify the password that is used by the authentication under Intercluster Lookup Service configuration.   B.   Find out if the URI address of the called user has a capital letter in the URI string.   C.   Verify that Intercluster Lookup Service is set up correctly.   D.   Verify USN Data Synchronization Status. Answer: B NEW QUESTION 87   ......   Download the newest PassLeader 300-080 dumps from passleader.com now! 100% Pass Guarantee! 300-080 PDF dumps & 300-080 VCE dumps: **http://www.passleader.com/300-080.html** (141 Q&As)