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2016/07 Cisco Official News!!! CCNP Collaboration 300-080 CTCOLLAB: Troubleshooting Cisco IP Telephony & Video v1.0 Exam Questions and Answers New Updated Today! Instant Download Free 300-080 PDF & 300-080 VCE from Braindump2go.com. **100% Exam Pass Guaranteed!** NEW QUESTION 51 - NEW QUESTION 60: 1.[2016 Latest Cisco 300-080 CTCOLLAB PDF & 300-080 CTCOLLAB VCE 196Q&As Dumps Instant Download:<http://www.braindump2go.com/300-080.html> 2.[2016 New Cisco 300-080 CTCOLLAB Exam Questions PDF:<https://drive.google.com/folderview?id=0B272WrTALRHcRVZjNGRLeDJsNk0&usp=sharing> QUESTION 51 Cisco TelePresence System EX90-A and EX90-B are in a call. EX90-A tries to call EX90-C. When the call is dialed, EX90-B is put on hold. EX90-A and EX90-C are connected, but there is no merge button on the touch panel. What is causing this issue? A. The multisite option key is missing. B. The multisite configuration is missing. C. The conference option key is missing. D. The conference configuration is missing. E. Cisco TelePresence systems cannot make multipoint calls without a Cisco TelePresence Server. F. The multipoint option key is missing. Answer: B QUESTION 52 Which statement indicates something that can cause an inbound PSTN call to an H.323 gateway that is configured in Cisco Unified Communications Manager to fail to ring an IP phone? A. The gateway is not registered in Cisco Unified Communications Manager. B. The gateway IP address that is configured in Cisco Unified Communications Manager does not match the IP address that is configured at the gateway in the h323-gateway voip bind srcaddr command. C. The Cisco Unified Communications Manager does not have a matching route pattern to match the called number. D. The gateway is missing the command allow-connections h323 to h323 under the voice service voip configuration. Answer: B QUESTION 53 Some users report that they cannot dial out from headquarters on their Cisco IP Phones to PSTN users, but others can. Which troubleshooting approach is the most direct to isolate the source of the failure of the users that cannot dial out to the PSTN? A. Use DNA to analyze the dialing permissions of the Cisco IP Phones. B. Use DNA to generate actual calls to the PSTN. C. Use RTMT to analyze the dialing permissions of the Cisco IP Phones. D. Use RTMT to generate actual calls to the PSTN. Answer: A QUESTION 54 You have been presented with a trouble ticket from an end user who works at a remote location that is served by a Cisco Unified Communications Manager Express. The user reports being unable to place calls to international numbers, but all other calls work properly and other users at this location can place international calls. Which two troubleshooting techniques would be helpful in resolving this issue? (Choose two.) A. Cisco IOS debug tools B. Class of Restriction baseline configuration for the user on Cisco Unified Communications Manager Express C. show output of the ephone and ephone-dn configurations D. show output of the voice translation rules in the voice gateway E. show output for the T1 controller and voice port configuration in the voice gateway Answer: AB QUESTION 55 Which Cisco Unified Communications Manager troubleshooting tool can be used to determine the digit manipulation path a call takes within the Cisco Unified Communications Manager system from the perspective of a specific directory number, without having the actual device at hand? A. Cisco Unified Communications Manager Serviceability B. Cisco Unified Communications Manager Dialed Number Analyzer C. Cisco Unified Communications Manager Real Time Monitoring Tool D. Cisco Unified Syslog Viewer E. Cisco IOS debugs Answer: B QUESTION 56 During a business-to-business video call through the Cisco Expressway solution, the internal endpoint can call out to the remote endpoint on the Internet, but it does not receive audio or video. The remote endpoint receives both audio and video. What is causing the issue? A. The Cisco Expressway does not have a Rich Media Session license. B. The firewall is blocking SIP signaling. C. The Cisco Unified Communications Manager is not configured for business-to-business calling. D. The firewall is blocking inbound RTP ports. E. The Advanced Networking option is not installed on the Expressway Edge. Answer: D QUESTION 57 System A at Company 1 is calling System B at Company 2. The call completes, but only audio and video are present on System A from System B. What are two possible causes? (Choose two.) A. System A cannot call System B because it is at a different company. B. There is a firewall in the path that is blocking audio and video traffic from Company 1 to Company 2. C. The firewall at Company 1 is blocking outgoing traffic. D. An access list is blocking video and audio somewhere in the video and audio path between System A and System B. E. System A has turned off the camera and the microphone. Answer: BC QUESTION 58 Which tool allows the administrator to analyze call routing in Cisco Unified Communications Manager without physically placing a call? A. Cisco Unified Communications Manager Dialed Number Analyzer B. Cisco IOS Gateway debug commands C. Cisco Unified Communications Manager RTMT trace output D. base configuration information for this user that specifies Class of Restriction, Partition, and Calling Search Space information E. Cisco Unified Communications Manager Serviceability tools F. Cisco Unified Communications Manager OS Administration Answer: A QUESTION 59 When a user tries to initiate an Ad Hoc conference call from an IP phone, this message appears: No Conference Bridge Available. Which two actions resolve this issue?

(Choose two.) A. Make sure that the Join softkey is assigned to the phone. B. Make sure that a Conference Bridge resource is registered to Cisco Unified Communications Manager. C. Reset the phone, to re-register resources. D. Make sure that a Conference Bridge Resource is assigned to the MRGL on the phone that initiates the conference call. Answer: BD

QUESTION 60

Two phones in the same cluster and at the same site have a call currently connected. The site local H.323 PSTN gateway loses connection with Cisco Unified Communications Manager. Which two results do you expect? (Choose two.) A. SRST is active, and all the phones enter SRST mode. B. No incoming and outgoing calls are possible. C. Cisco Unified SRST is able to receive incoming calls. D. The current call is not disconnected. E. The phones display "CM Fallback Service Operating." Answer: BD

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