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Recently, users complain that they cannot call the PSTN. With the help of testing, you find that the gateway is not switching to the secondary call agent when the primary call agent is unreachable. In order to permit the MGCP gateway to take use of a different call agent once the primary fails, which configuration should you make? A. Add ccm-manager fallback-mgcp command to the gateway.B. Add ccm-manager redundant-host command to the gatewayC. Assign a Cisco Unified CallManager group including the secondary call agent to the gatewayD. Define gateway as a non-gatekeeper-controlled intercluster trunk with the secondary Cisco Unified CallManager defined. Answer: BQUESTION 13Users are making calls from Cisco IP Phones that are registered on a Cisco Unified Communications Manager cluster through an H.323 gateway to the PSTN phones. Sometimes these users report that calls are dropped after they have talked for some time. Different Cisco Unified CM servers have been rebooted for testing during production hours. Currently Cisco IOS 12.0 is used on the H.323 gateway.Which change helps to prevent calls from dropping when a Cisco Unified CM server is rebooted during testing? A. Upgrade the gateway Cisco IOS software to a version later than 12.4(4)9T.B. A misconfiguration of calling search spaces must exist. Reconfigure the CSS.C. There is no scenario where rebooting an H.323 gateway will allow call survivability.D. A misconfiguration of VoIP dial peers to the Cicso Unified CM servers must exist. Reconfigure the VoIP dial peers. Answer: AQUESTION 14Which Cisco troubleshooting tool allows syslog logs to be viewed? A. RTMTB. DNAC. sniffer filtersD. the Triple combo tool Answer: A QUESTION 15Which three troubleshooting tools are available on Cisco Unified Communications Manager? (Choose three.) A. CLIB. SDL and SDI traceC. NetcoolD. alarmsE. SCCP troubleshooting toolsF. RTMT Answer: ABFQUESTION 16Which tool or approach is the most effective to display detailed real-time output of CPU utilization of Cisco Unified Communications Manager servers for troubleshooting purposes? A. RTMTB. DNAC. creating trace filesD. sniffer traces Answer: A QUESTION 17When using trace output to troubleshoot a Cisco Unified CallManager 5.0 problem, how can you collect and view the trace files? A. Download the RTMT plug-in from the Cisco Unified CallManager Serviceability page to view the preconfigured trace files.B. Configure the proper trace settings on the Cisco Unified CallManager Serviceability page and then use the embedded RTMT tool to view the trace files.C. Configure the proper alarms and traces on the Cisco Unified CallManager Administration page and view the output with the RTMT plug-in.D. Configure the proper trace settings on the Cisco Unified CallManager Serviceability page and download the RTMT plug-in from the CallManager Administration page to view the trace output. Answer: DQUESTION 18Before a SAF client begins to forward, publish, subscribe, and update service data, with which option must it establish a relationship? A. SAF service identifierB. SAF domainsC. SAF forwarderD. SAF Cisco Unified Communications Manager Answer: C QUESTION 19Which size is the jitter buffer of the Cisco MTP transcoder? A. 10 - 30 msB. 45 - 60 msC. 20 - 40 msD. 25 - 50 ms Answer: CQUESTION 20You are troubleshooting why a user cannot make calls to the PSTN. You are reviewing trace files and you have found where the user's IP phone initiates the call but you never see the call go out the gateway. What is the next step in troubleshooting this issue? A. Look in the SDL trace file to see if there is a signal to another Cisco Unified CallManager node with the same time-stamp.B. Look in the SDL trace file to see if there is a signal to another Cisco Unified CallManager node with the same TCP handle.C. Look in the IP Voice Media Streaming App trace file to see if an MTP was invoked.D. Look in the MGCP trace file to determine which MGCP gateway the call was sent to. Answer: A !!!RECOMMEND!!! 2016/07 Latest Braindump2go Cisco 300-080 Exam PDF and VCE 196q Dumps Instant Download: http://www.braindump2go.com/300-080.html [100% Exam Pass Guaranteed!] 2016/07 Latest Braindump2go Cisco 300-080 Exam Questions PDF: https://drive.google.com/folderview?id=0B272WrTALRHcRVZjNGRLeDJsNk0&usp=sharing