[2016-July-NEWBraindump2go Cisco 300-080 PDF and VCE Dumps Free Download[NQ91-NQ100

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PDF:https://drive.google.com/folderview?id=0B272WrTALRHcRVZjNGRLeDJsNk0&usp=sharing QUESTION 91If you need to avoid choppy speech, what is the maximum tolerable round-trip delay between two VoIP endpoints? A. 100 msB. 200 msC. 300 msD. 400 msE. 500 msF. 800 ms Answer: C QUESTION 92Where does an IP phone obtain the extension number and speed-dial settings from? A. the settings that are configured on the physical phoneB. the registration file that the phone receives from the Cisco Unified Communications ManagerC. the device and line configuration in Cisco Unified Communications Manager, during the registration processD. the default device profile that is configured in Cisco Unified Communications, Manager Answer: C QUESTION 93You are a network technician with many years' experience. Many users complain that they can hear echo when their calls go out an H.323 gateway. You have made some testing for the gateway and have changed the configuration. So the ERL level turns to be 6 dB. Furthermore, the echo-cancel coverage value is raised to 64 ms. Please choose the effect on the voice quality after this modification. A. Consonants will be chopped by the echo canceller.B. The increase in echo-cancel coverage will have no effect on voice quality.C. The ends of sentences will be chopped by the echo canceller.D. The echo canceller will take 2-3 seconds longer to converge at the beginning of the call. Answer: D QUESTION 94You have placed all DNs in the Phones partition. During testing you discover that you cannot place calls between IP phones, but you can place calls to the PSTN and voice mail. What is one possible cause of this issue? A. A database replication issue is preventing calls between phones.B. An access list is blocking RTP streams in your voice VLAN.C. An access list is blocking SCCP packets in your voice VLAN.D. The IP phones have not been assigned a CSS. Answer: D QUESTION 95Refer to the exhibit. According to the output of the debug voip ccapi inout

command, why was this call dropped?

//ROSSA/THINC THEODOSIC HEL SPECIAL DEFFECT by "ROSSA/THINC THEODOSIC 1418: Jun 26 50: NO:22, 248: cc_apt_get_scode_stream ; 4803 1440: Jun 26 50: NO:22, 248: //62006/scoxxxxxxxxxx/CAPt/cc_apt_get_scode_ 1665: Jun 26 10:30:22.248: cc_apt_get_scode_stream : 4800 465; Jun 26 50:30:22.212; cc_apt_pat_xcode_stream ; 4803 008: 2un 26 50:30:22,712: cc_apt_get_xcode_stream : 4800 ct0: 3un 26 50:30:32,712: //6000/wxxxxxxxxxxxxxxxcc.4CAPLroc_apt_get_xcode_s |un |4 |0:10:22.712: cc_epi_get_scode_stream | 4803 |un |4 |0:10:22.784 | /coppin lesk 80000000,ccapi_cc_epi_call_fac |ac_epi_call_fac |ac_epi_call_fac |un |4 |un 76 10:30:32,268: //92009/389C36000000. to 10 february and the control of th te) CallOfsconnect: Capse=0, Disconnect Ca CallOfsconnect: Telstor: Call.edpps 2, Correctioned SMCMORODOCOMMACIAACO, pergetionerishabless, priormedicane so , discoverativat normal call.clar.elsevierscane so , discovera H88: Jun 26 50:30:29.809: :Deside cc_build_feature_vsa H89: Jun 26 50:30:29.809: //-5/commonocom/CGP5/cc_build_feature_vs Jun 26 50:30:24.800: vsacount in free is 3 Jun 26 50:30:25.800: //-1/xxxxxxxxxxx/CLP1/coverPosTtSfree=1s - 16/20014 06: Jun 25 10: 10:25. N.S. Feature call basis

A. Normal call clearingB. User busyC. Call rejectedD. Invalid numberE. No circuitF. No resource Answer: A QUESTION 96You need to increase the maximum number of Meet-Me conference participants on Cisco Unified Communications Manager. Where do you configure this increase? A. System > Service Parameters > [Publisher Server] > Cisco CallManager (Active) > Clusterwide Parameters (Feature ?Conference)B. Call Routing > Meet-Me Number/PatternC. Media Resources >

Conference BridgeD. Media Resources > Media Resource GroupE. Device > Conference Bridge Answer: A QUESTION 97You are receiving complaints of pixilation, smearing, and pulsing of video calls between two offices that are connected by a WAN. Assuming that QoS is implemented on the WAN connection, which classification should you use to mark the video traffic, according to the Cisco QoS baseline? A. CS6B. CS2C. AF41D. AF31E. EFF. CS3 Answer: C QUESTION 98Which Cisco TelePresence Management Suite feature provides proactive suggestions for resolving error conditions and also provides a centralized view of all statuses and configuration errors for managed devices? A. Cisco WebEx in CiscoTelePresence Management SuiteB. CiscoTelePresence schedulingC. graphical monitoringD. ticketing serviceE. audit logging Answer: D QUESTION 99Cisco Unified Communications Manager is deployed in a head office and also provides telephony services to remote regional offices. What is a possible cause of call setup issues with phones in the regional offices? A. remote office firmware for phonesB. codec mismatch in region information settingsC. end user roles and groupsD. physical location configuration Answer: B QUESTION 100Refer to the exhibit. Cisco Unified Communication Manager has been configured with a SAF Security Profile and a SAF Forwarder. In which section of the R1 router configuration should the SAF Forwarder password be entered?

A. interface loopback 0B. router eigrp safC. service-family external clientD. service-family ipv4 Answer: C !!!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