[New 210-060 Dumps210-060 VCE and PDF 254Q Instant Download in Braindump2go[Q226-Q236

2018 January New Cisco 210-060 Exam Dumps with PDF and VCE Free Updated Today! Following are some new added 210-060 Exam Questions: 1.|2018 New 210-060 Exam Dumps (PDF & VCE) 254Q&As Download: **https://www.braindump2go.com/210-060.html**2.|2018 New 210-060 Exam Questions & Answers Download: **https://drive.google.com/drive/folders/0B75b5xYLjSSNcHNlT3VrNy1CbUU?usp=sharing**QUESTION 226What allows a phone user to log a phone out of all line groups to which the phone directory numbers belong?A. Cisco IPPAB. Cisco FinesseC. HLog softkeyD. Cisco Agent DesktopAnswer: CExplanation: **https://supportforums.cisco.com/t5/other-collaboration-voice-and/allowing-cfa-for-a-phone-in-a-hunt-group/td-p/1264535** QUESTION 227Several users in the same Call Manager group reported that they receive a fast busy tone when they go off hook. They reported that the issue went away after a few minutes. Where should the voice systems administrator begin troubleshooting?A. SIP gatewayB. Cisco Unified Communications Manager subscriberC. network cableD. 7945 IP phone**Answer: B**QUESTION 228Which two explanations of DTMF dialing are true? (Choose two.)A. DTMF dialing consists of simultaneous voice-band tones generated when a button is pressed on a telephone.B. The use of DTMF enables support for advanced telephony services.C. DTMF dialing uses INVITE messages to signal when the first digit is pressed in a new call.D. DTMF dialing consists of a simultaneous digital-band pulse generated when a button is pressed on a telephone.Answer: ABExplanation: **https://www.cisco.com/c/en/us/td/docs/ios-xml/ios/voice/sip/configuration/12-4t/sip-12-4t-book/voi-sip-dtmf.html#GUID-A23A02E9-3FD2-4C3A-8ED4-78BDC82AAEAE**QUESTION 229How are calls routed in an MGCP call agent?A. by route patterns on Cisco Unified Communications ManagerB. by dial peers on the gatewayC. by static routes on the gatewayD. by a translation pattern on Cisco Unified Communications Manager**Answer: A**QUESTION 230A voice engineer has installed an XML-based phone application from a third party and subscribed a user's 7945 IP phone to the application. Which action does the user take to launch the new service on the IP phone?A. Select the Applications button on the 7945 IP phone.B. Select Settings > Applications.C. Select Settings > Network > Applications.D. Select the Services button on the 7945 IP phone.**Answer: D** QUESTION 231An administrator must keep CDR data for a longer period of time and wants to modify the configured value of the CDR/CMR Files Presentation Duration Days). Which menu options does the engineer navigate?A. Unified CM Administration > System > Enterprise ParametersB. Unified Serviceability > Tools > Serviceability Reports ArchiveC. Unified CM Administration > Call Routing > Route Plan ReportD. Unified Serviceability > Tools > CDR Management**Answer: A**QUESTION 232IP WAN failure has occurred. Which two configurations must be made to allow calls to or from an IP phone at a branch location to complete to a destination outside of that branch? (Choose two.)A. Survivable Remote Site Telephony is configured on the branch router.B. LAN failover configuration is set to Survivable Remote Site Telephony.C. Call Forward UnRegistered option is configured on each branch phone profile.D. POTS WAN failover is set to Survivable Remote Site Telephony.E. An ISDN is configured as a primary survivable backup siteAnswer: CEExplanation: **https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab10/collab10/models.html**QUESTION 233A voicemail user reports that he cannot access Cisco Unity Connection from the IP phone message button. Which option must you perform to resolve the problem?A. In Cisco Unified Communications Manager, navigate to Advanced Features > Voicemail to confirm that the user has the correct voicemail profileB. In Cisco Unified Communications Manager, navigate to Device > Phone to confirm that the device has the correct voicemail profileC. In Cisco Unified Communications Manager, navigate to Device > Phone to confirm that the directory number has the correct voicemail profileD. In Cisco Unified Communications Manager, navigate to User Management > End User to confirm that the user has the correct voicemail profileE. In Cisco Unity Connection, navigate to Users > Users to confirm that the user web password is correct.**Answer: C**QUESTION 234A user reports that during calls they hear excessive hissing when neither party is talking.Which option is one cause of this noise?A. QoSB. LoPSC. VADD. EPLE. SRST**Answer: C**QUESTION 235A network administrator finds directory number 77555 in the <none> partition. Which numbers can reach 77555 in this situation?A. Only numbers in the <none> partition can reach 77555.B. All numbers in every calling search space can reach 77555.C. Only internal directory numbers can reach 77555.D. Directory number 77555 cannot be reached by any number.**Answer: A**QUESTION 236What is the maximum amount of packet loss an engineer should allow for voice traffic on an IP network?A. 5 percentB. 2 percentC. 3 percentD. 1 percent**Answer: A**!!!RECOMMEND!!!1.|2018 New 210-060 Exam Dumps (PDF & VCE) 254Q&As Download:**https://www.braindump2go.com/210-060.html**2.|2018 New 210-060 Study Guide Video: YouTube Video: YouTube.com/watch?v=antJcSoL9k0