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https://drive.google.com/drive/folders/0B75b5xYLjSSNcHNlT3VrNy1CbUU?usp=sharingQUESTION 104A field technician must reset a single 7965 IP phone so that it will be discovered on the network again and request an IP address from DCHP. Which steps should the network engineer provide?A.   Have field technician select the Settings button on phone, then dial # # * * *B.   Have field technician select the Settings button on phone, then dial # # * # #C.   Have field technician select the Settings button on phone, then dial * * # * *D.   Have field technician select the Settings button on phone, then dial * * # # #Answer: CQUESTION 105An end user is experiencing performance issues with their IP phone. Which phone feature could the administrator configure to allow the end user to notify them when issues occur?A.   Cisco Quality Report ToolB.   System Log ManagementC.   Cisco CDR Analysis and ReportingD.   Cisco JTAPIAnswer: AQUESTION 106An end user has an 7945 IP phone, but is unable to launch any of the IP phone applications. A network engineer has confirmed that the device is subscribed to the appropriate applications in Cisco Unified Communications Manager.How should the engineer instruct the user to launch the IP phone applications?A.   Select the "?" button on the IP phone.B.   Select the globe button on the IP phone.C.   Select Settings > User Preferences.D.   Select Settings > Device Configuration.Answer: BQUESTION 107Which component is needed for a voice call to be processed between the enterprise Cisco Unified Communications system and a cell phone via the PSTN?A.   Cisco Analog Voice Gateway VG224B.   Cisco Unified Communications Manager call processing nodeC.   Cisco Unified Communications Manager Music On Hold nodeD.   Cisco Integrated Service Router with digital signal processor resourcesAnswer: DQUESTION 108What is needed to support SIP Early Media?A.   Media Termination PointB.   TranscoderC.   AnnunciatorD.   Conference BridgeE.   DSPF.   Route ListAnswer: AQUESTION 109The IP phone of user A is registered with Cisco Unified Communications Manager subscriber1 while the IP phone of user B is registered with subscriber2. User A is speaking to user B on an active call. A junior network engineer mistakenly reboots subscriber1.What effect does this have on the call?A.   User A can hear B, but B cannot hear A.B.   User A cannot hear B, but B can hear A.C.   This action drops the call.D.   This action does not affect the call.E.   The call remains active, but quality may suffer.Answer: DQUESTION 110A company has invested in an on-premises VoIP solution. For design simplicity purposes, network engineers have opted to use a single VLAN for both data and voice traffic. Shortly after implementing IP phones, customers are reporting problems.Which two potential problems may be reported about the calls as a result of this design decision? (Choose two.)A.   jitter detected in voice callsB.   problems with email latencyC.   slow internet download speedsD.   choppy voice callsE.   PCs are getting IP addresses but phones are notAnswer: ADQUESTION 111A network engineer receives a report about poor quality on an active call between the IP phone of user A over the WAN to the IP phone of user B. Using web access to the phone, the network engineer remotely checks call statistics such as jitter, network delay, and packet loss. Calculated packet loss is 3%, average jitter is 20 ms, network delay is 1 ms, and conceal seconds is 7.What is the most likely problem with this call?A.   Calculated packet loss is too high.B.   Average jitter is too high.C.   Network delay is too high.D.   Conceal seconds are too high.Answer: AQUESTION 112Which command allows the telephony service of a Cisco Unified Communications Manager Express router to be associated to loopback address 192.168.143.44?A.   !telephony-servicemax-ephones 4max-dn 8ip source-address 192.168.143.44!B.   !telephony-servicemax-ephones 4max-dn 8ip bind src-addr 192.168.143.44!C.   !telephony-servicemax-ephones 4max-dn 8source-address ipv4:192.168.143.44!D.   !telephony-servicemax-ephones 4max-dn 8ip address 192.168.143.44!Answer: AQUESTION 113Which four devices can be used to provide analog ports, traditional phones, and fax machines? (Choose four.)A.   Cisco VG224 Analog Voice GatewayB.   Foreign Exchange Station Voice Interface CardC.   Cisco High Density VoiceFax Network ModuleD.   Cisco ATA190 Analog Telephone AdapterE.   Cisco VG350 Analog Voice GatewayF.   Cisco Unified Border ElementG.   Foreign Exchange Office Voice InterfaceAnswer: ABDEQUESTION 114Client A in X site uses an IP phone to call client B in Y site. Engineers have selected SCCP as the default VoIP signaling protocol. Which network path will the call signaling take when client A calls client B?A.   IP phone X > CUCM subscriber > IP phone Y using TCP port 1000B.   IP phone X > CUCM subscriber > IP phone Y using TCP port 2000C.   IP phone X > CUCM subscriber > IP phone Y using UDP port 1000D.   IP phone X > IP phone Y using TCP port 2000E.   IP phone X > IP phone Y using TCP port 1000Answer: B**!!!RECOMMEND!!!**]1.|2017 New 210-060 Exam Dumps (PDF & VCE) 225Q&As Download:

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