## [December-2021High Quality Braindump2go 350-801 Exam VCE Dumps 350-801 289Q Free Share[Q253-Q282

<u>December/2021 Latest Braindump2go 350-801 Exam Dumps with PDF and VCE Free Updated Today! Following are some new</u> <u>350-801 Real Exam Questions!</u>QUESTION 253Refer to the exhibit. An engineer must modify the existing QoS policy-map statement to implement LLQ for voice traffic. Which change must the engineer make in the configuration?

ROUTER-1(config)# policy-map ROUTER-1(config-pmap)# class ROUTER-1(config-pmap-c)# ba ROUTER-1(cc -fig-pmap-c)# exi ROUTER-1(config-pmap)# class ROUTER-1(config-pmap-c)# ba ROUTER-1(config-pmap-c)# exit ROUTER-1(config-pmap)# exit

A. bandwidth 170 to LLQ 170B. bandwidth 170 to priority 170C. bandwidth 170 to reserve 170D. bandwidth 170 to percent 170Answer: DQUESTION 254Refer to the exhibit. A collaboration engineer needs to replace the original, single NTP server that was configured during the initial install of a Cisco UCM server. What is the first step to accomplish this task?

admin:utils nt ntpd (pid 1742	<li>8) is running</li>		
VENEV	v.Brai	'n.	E.
*192.168.1.1	17.253.14.125	2	ι
*192.168.1.2	17.253.14.125	2	t

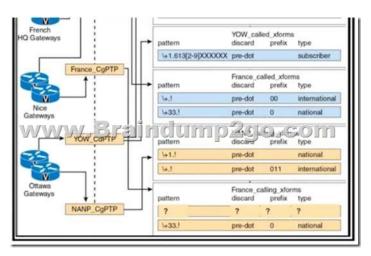
A. Restart the NTP service on Cisco UCM.B. Enable NTP authentication for the new NTP server on Cisco UCM.C. Stop the NTP service on Cisco UCM.D. Delete the original NTP server from Cisco UCMAnswer: DQUESTION 255An administrator executes the debug isdn q931 command while debugging a failed call. After a test call is placed, the logs return a disconnect cause code of 1.What is the cause of this problem?A. The destination number rejects the callB. The destination number is busyC. The media resource is unavailableD. The dialed number is not assigned to an endpointAnswer: DQUESTION 256Refer to the exhibit. There is a call flow between Phone A and Phone B Phone B (holder) places Phone A (holder) on hold. Which MRGL and Audio



A. MRGL A and Audio Source 4B. MRGL B and Audio Source 4C. MRGL B and Audio Source 2D. MRGL A and Audio Source 2Answer: CQUESTION 257A SIP phone has been configured in the system with MAC address 0030 96D2 D5CB.The phone retrieves the configuration file from the Cisco UCM. Which naming format is the file that is downloaded?A. SIPOO3O96D2D5CB.cnf.xmlB. SEP003096D2D5CB.cnf.xmlC. SEP0030406706705.cnfD. SIP0030406706705.cnfAnswer: AQUESTION 258Where in Cisco UCM are codec negotiations configured for endpoints?A. in enterprise parametersB. under device profiles in device settingsC. under regions using preference listsD. in in-service parametersAnswer: DQUESTION 259 Refer to the exhibit. A call from +1 613 555 1234 that is sent out through the Nice Gateways should result in a calling party of 001 613 555 1234 with the numbering type "international".Which configuration accomplishes this goal?

Source are played to Phone A?

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A. +.l pre-dot 00 internationalB. + 0011 pre-dot 1 internationalC. +1 ! none pre-dot 001 internationalD. 613XXXXXX none +011 internationalAnswer: CQUESTION 260Refer to the exhibit. Unanswered calls do not reach the voicemail associated with the phones Instead, callers receive the default greeting. Which action fixes the configuration?

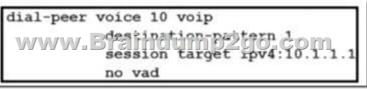
Use Device Pool Called Party Tra	nsformation CSS
Calling Party Transformation CSS	< None >
Use Device Pool Calling Party Tra	ansformation CSS
Calling Party Selection*	Originator
Calling Line ID Presentation*	Default
Calling Name Presentation*	Default
Collins and Conserved Darks Info Fo	matter i an i i
	rmat* Deliver DN only in connected party
	Mittimp2g
Restinction Civersio = atrage	Mittimp2g
Redirection Conversion = Adro 50 k fet h Fa k fa s th fa S Use Device Pool Redirecting Part	Mittimp2g
Redirect to Civirsio * Adro Co A fec. A Pa A fais to C Use Device Pool Redirecting Part Caller Information	Mittimp2g

DHCP Server Information	
Host Server*	192.168.10.240
Primary DNS IPv4 Address	192.168.99.1
Secondary DNS IPv4 Address	
Primary TFTP Server IPv4 Address(Option 150)	192.168.10.244
Secondary TFTP Server IPv4 Address(Option 150	)
nie www.astain	dump2
TFTP Server Name(Option 66)	
ARP Cache Timeout(sec)*	0
IP Address Lease Time(sec)*	0
Renewal(T1) Time(sec)*	0
Rebinding(T2) Time(sec)*	0

A. Restart the TFTP service under Cisco Unified ServiceabilityB. Add a DHCP subnet to the DHCP server under Cisco UCM AdministrationC. Restart the Cisco DHCP Monitor Service under Cisco Unified Serviceability.D. Add the new DHCP server to

the primary DNS serverAnswer: DQUESTION 264Which action is required for a firewall configuration on a Mobile and Remote Access through Cisco Expressway deployment?A. The traversal zone on Expressway-C points to Expressway-E through the peer address field on the traversal zone, which specifies the Expressway-E server address. For dual NIC deployments, set the Expressway-E address using an FQDN that resolves the IP address of the internal interfaceB. The internal firewall must allow these inbound and outbound connections between Expressway-C and Expressway-E. SIP: HTTPS (tunnelled overSSH between C and E): TCP 2222; TCP 7001; Traversal Media: UDP 2776 to 2777 (or 36000 to 36011 for large VM/appliance): XMPP: TCP 7400. C. The external firewall must allow these inbound connections to Expressway: SIP: TCP 5061; HTTPS: TCP 8443: XMPP: TCP 5222; Media: UDP 36002 to 59999.D. Do not use a shared address for Expressway-E and Expressway-C, as the firewall cannot distinguish between them If static NAT for IP addressing on Expressway-E is used, ensure that any NAT operation on Expressway-C does not resolve the same traffic IP address Shared NAT is not supported. Answer: BQUESTION 265Which characteristic of distributed class-based weighted fair queueing addresses jitter prevention?A. It provides additional granularity by allowing a user to create classes B. It minimizes jitter by implementing a priority queue for voice trafficC. It uses a priority queue for voice traffic to avoid jitterD. It provides additional granularity by allowing a user to define custom classesAnswer: AQUESTION 266A company hosts a conference call with no local users. How does the administrator stop the conference from continuing?A. modifies the Block OffNet to OffNet Transfer service parameterB. removes the transcoderC. changes the codecs that are supported on the conference resourceD. modifies the Drop Ad Hoc Conference service parameterAnswer: AQUESTION 267What happens to voice packets from a Cisco 8845 IP Phone in the QoS trust boundary?A. The phone and access layer switch negotiate the classification of packetsB. The voice packets are classified by the phone, and the classification is accepted.C. Cisco UCM determines how the voice packets are classifiedD. The voice packets are not trusted, and the access layer switch reclassifies the packets. Answer: BQUESTION 268Refer to the exhibit. An engineer configures a VoIP dial peer on a Cisco gatewa. Which codec

is used?



A. No codec is used (missing codec command).B. G.729r8C. G.711ulawD. G.711alawAnswer: BQUESTION 269A collaboration engineer adds a voice gateway to Cisco UCM. The engineer creates a new gateway device in Cisco UCM, selects VG320 as the device type and selects MGCP as the protocol. What must be done next to add the gateway to the Cisco UCM database?A. Configure the module in slot 0 of the new gatewayB. Select the DTMF relay type for the gatewayC. Select a device pool for the new gatewayD. Add the FQDN or hostname of the deviceAnswer: AQUESTION 270An administrator configures Cisco UCM to use UDP for SIP signaling and finds that an endpoint cannot make calls. Which action resolves this issue?A. Change the phone security profile.B. Change the SIP dial rules.C. Change the SIP profileD. Change the common phone profile. Answer: AQUESTION 271What is a capability of Cisco Expressway?A. It provides access to on-premises Cisco Unified Communications infrastructure for remote endpointsB. It has remote endpoint enrollment with Certificate Authority Proxy FunctionC. It functions as an analog telephony adapterD. It gives directory access for remote users via Cisco Directory IntegrationAnswer: AQUESTION 272After an engineer implements the FAC and CMC features together, users report that calls take almost one minute to complete and that they occasionally hear the reorder tone. Which two actions address this issue? (Choose two.)A. Advise the user to press the "#" button after dialling the FAC and CMC codesB. Change the code if the problem persists i/i-Advise the user to hang up and try againC. Adjust the T302 timer from the default of 15 seconds to 5 seconds to shorten the interdigit timerD. Do not wait for the tones immediately dial the FAC and CMCAnswer: ACQUESTION 273An administrator troubleshoots call flows and suspects that there are issues with the dial plan. Which tool enables a quick analysis of the dial plan and provides call flows of dialed digits? A. Digit Analysis AnalyzerB. Dialed Number AnalyzerC. Cisco Dial Plan AnalyzerD. Dial Plan AnalyzerAnswer: BQUESTION 274A company's employees have been complaining that they have been unable to select options on the internal IVR of the help desk IT support has been given Cisco UCM traces and below is the snippet of the SDP of the INVITE packet.

m=audio 25268 RTP/AVP 18 101 a=rtpmap:0 PCMU/8000 a=rtpmap:18 G729/8000 a=tptives2@raindump2go.com a=fmtp:18 annexb=no a=rtpmap:101 telephone-event/8000 a=fmtp:101 0-15

How is this issue resolved?A. Configure CODEC for G.729B. Configure DTMF for RFC 2833.C. Configure CODEC for G 722D. Configure DTMF for KPMLAnswer: DQUESTION 275Which Cisco Unity Connection handler plays a greeting that announces the option to dial a user extension by default?A. the Goodbye call handlerB. the Operator call handlerC. the Directory handlerD. the Interview handlerAnswer: CQUESTION 276Refer to the exhibit. Which two codec permutations should be transcoded by this dspfarm? (Choose two )

dspfarm profile 1 transcode universal codec g729r8 codec g729ar8 WWWWCETAINGUMP2go.com codec g711alaw maximum sessions 8 associate application SCCP

A. G.729ar8 to G.711alawB. iLBC to G.711ulawC. G.729r8 to G711ulawD. G.722 to G 729r8E. G.729br8 to G711alawAnswer: BEQUESTION 277Which two steps are required for bulk configuration transactions on the Cisco UCM database utilizing BAT? (Choose two)A. A data file in comma-separated values format must be uploaded to Cisco UCMB. A server template must be created in Cisco UCM.C. A device template must be created in Cisco UCM.D. A data file in Abstract Syntax Notation One format must be uploaded to Cisco UCME. A data file in Extensible Markup Language format must be uploaded to Cisco UCMAnswer: ACQUESTION 278A collaboration engineer troubleshoots issues with a Cisco IP Phone 7800 Series. The IPv4 address of the phone is reachable via ICMP and HTTP and the phone is registered to Cisco UCM. However, the engineer cannot reach the CLI of the phone. Which two actions in Cisco UCM resolve the issue? (Choose two)A. Enable FIPS Mode under Product Specific Configuration Layout in Cisco UCMB. Enable Settings Access under Product Specific Configuration Layout in Cisco UCM.C. Enable SSH Access under Product Specific Configuration Layout in Cisco UCMD. Disable Web Access under Product Specific Configuration Layout in Cisco UCME. Set a username and password under Secure Shell Information in Cisco UCMAnswer: ACQUESTION 279What are two characteristics of jitter in voice and video over IP communications? (Choose two.)A. The packets arrive at varying time intervalsB. The packets arrive out of sequenceC. The packets arrive with frame errorsD. The packets arrive at uniform time intervals.E. The packets never arrive due to tail dropAnswer: ABQUESTION 280 Refer to the exhibit. An administrator configures a secure SIP trunk on Cisco UCM. Which value is needed in the Secure Certificate Subject or Subject Alternate Name field to accomplish this task?

Name*	Secure SIP Trunk Profile	
Description	Non Secure SIP Trunk Profile a	uthenticated by null String
Device Security Mode	Encrypted	
Incoming Transport Type*	TLS	
Outgoing Transport Type	TLS	•
Secure Cettificate Sabjet <del>: ज</del> ्युकेट्ट <del>े</del> क	ind µmp2ge	919911

A. the fully qualified domain name of the remote device that is configured on the SIP trunkB. the common name of the Cisco UCM CallManager certificateC. the common name of the remote device certificatesD. the fully qualified domain name of all Cisco UCM nodes that run the CallManager serviceAnswer: DQUESTION 281An administrator works with an ISDN PRI that is connected to a third-party PBX. The ISDN link does not come up; and the administrator finds that the third-party PBX uses the QSIG signalling method. Which command enables the Cisco IOS Gateway to use QSIG signalling on the ISDN link?A. isdn switch-type basic-qsigB. isdn switch-type basic-niC. isdn switch-type primary-qsigD. isdn incoming-voice voiceAnswer: AQUESTION 282Which type of input is required when configuring a third-party SIP phone?A. digest userB. serial numberC. manufacturerD. authorization codeAnswer: DResources From:1.2021 Latest Braindump2go 350-801 Exam Dumps (PDF & VCE) Free Share:https://www.braindump2go.com/350-801.html2.2021 Latest Braindump2go 350-801 PDF and 350-801 VCE Dumps Free Share:https://drive.google.com/drive/folders/1pALdeSmYT0960h3st7rEH4WExsy2Mjgq?usp=sharing3.2021 Free Braindump2go 350-801 Exam Questions Download:

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