350-801^{Q&As}

Implementing and Operating Cisco Collaboration Core Technologies (CLCOR)

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QUESTION 1

A customer reports that the load balancing solution is not working for two Cisco Unified Border Element routers. The (cube1.abc.com) router should take 60% of the calls, and the (cube2.abc.com) router should take 40% of the calls. Assuming all DNS A records are present, which two SRV configurations would provide the desired outcome? (Choose two.)

A. _sip_udp.abc.com 60 IN SRV 60 1 5060 cube1.abc.com

B. _sip_udp.abc.com 60 IN SRV 1 40 5060 cube2.abc.com

C. _sip_udp.abc.com 60 IN SRV 3 60 5060 cube2.abc.com

D. _sip_udp.abc.com 60 IN SRV 2 60 6060 cube1.abc.com

E. _sip_udp.abc.com 60 IN SRV 1 60 5060 cube1.abc.com

Correct Answer: BE

QUESTION 2

What is used to create a QoS solution that delivers priority to voice and video traffic across a network?

A. CBWFQ

B. traffic shaping

C. DiffServ

D. traffic policing

Correct Answer: A

QUESTION 3

An engineer encounters third-party devices that do not support Cisco Discovery Protocol. What must be configured on the network to allow device discovery?

A. LACP

B. TFTP

C. LLDP

D. SNMP

Correct Answer: C

LLDP (Link Layer Discovery Protocol) is a vendor-neutral network discovery protocol that is used to discover the

topology of a network. LLDP is similar to CDP (Cisco Discovery Protocol), but it is not proprietary to Cisco. LLDP is supported by a wide range of network devices, including switches, routers, and firewalls. To configure LLDP on a network, you must enable LLDP on the devices that you want to discover. You can then use a network management tool, such as Cisco Network Assistant, to view the topology of the network. The other options are incorrect. TFTP (Trivial File Transfer Protocol) is a network protocol that is used to transfer files between devices. LACP (Link Aggregation Control Protocol) is a network protocol that is used to aggregate multiple network links into a single logical link. SNMP (Simple Network Management Protocol) is a network protocol that is used to manage network devices.

QUESTION 4

Callers from a branch report getting busy tones intermittently when trying to reach colleagues in other office branches during peak hours. An engineer collects Cisco CallManager service traces to examine the situation. The traces show:

50805567.000 |07:35:39.676 |SdlSig |StationOutputDisplayNotify |restart0 |StationD(1,100,63,6382) |StationCdpc(1,100,64,4725) |1,100,40,7.101980^*^* |[R:N-H:0,N:0,L:0,V:0,Z:0,D:0] TimeOutValue=10 Status=x807 UnicodeStatus= Locale=1 50805567.001 |07:35:39.676 |AppInfo | StationD: (0006382) DisplayNotify timeOutValue=10 notify=`x807\\' content=`Not Enough Bandwidth\\' ver=85720014.

What should be fixed to resolve the issue?

A. class of service configuration

B. geolocation configuration

C. region configuration

D. codec configuration

Correct Answer: C

QUESTION 5

Refer to the exhibit.

```
Sent:
INVITE sip:2004@192.168.100.100:5060 SIP/2.0
Via: SIP/2.0/UDP 192.168.100.200:5060;branch=z9hG4bKF1FED
From: "7000" <sip:7000@192.168.100.200>;tag=43CDE-1A22
To: <sip:2004@192.168.100.100>
Call-ID: 12BCA00-3C3E11EA-01234567@192.168.100.200
Supported: 100rel,timer,resource-priority,replaces,sdp-anat
Min-SE: 1800
User-Agent: Cisco-SIPGateway/IOS-16.9.5
Allow: INVITE, OPTIONS, BYE, CANCEL, ACK, PRACK, UPDATE, REFER, SUBSCRIBE, NOTIFY, INFO, REGISTER
CSeq: 101 INVITE
Contact: <sip:7000@192.168.100.200:5060>
Expires: 180
Max-Forwards: 68
P-Asserted-Identity: "7000" <sip:7000@192.168.100.200>
Session-Expires: 1800
Content-Type: application/sdp
Content-Length: 254

v=0
o=CiscoSystemsSIP-GW-UserAgent 5871 9974 IN IP4 192.168.100.200
s=SIP Call
c=IN IP4 192.168.100.200
t=0 0
m=audio 8002 RTP/SAVP 0
c=IN IP4 192.168.100.200
a=rtpmap:0 PCMU/8000
a=ptime:20
```

Calls to Cisco Unity Connection are failing across Cisco Unified Border Element when callers try to select a menu prompt Why is this happening and how is it fixed?

A. Cisco Unified Border Element is not sending any support for DTMF. DTMF configuration must be added to the appropriate dial peer.

B. The Cisco Unity Connection Call Handler is configured for a -Release to Switch" transfer type. Transfers must be disabled for the Cisco Unity Connection Call Handler.

C. Cisco Unity Connection is configured on G.729 only. Cisco Unity Connection must be reconfigured to support iLBC.

D. Cisco Unified Border Element is sending the incorrect media IP address. The IP address of the loopback interface must be advertised in the SDP.

Correct Answer: A

m = audio 35904 RTP/AVP 8 101 a = rtpmap:8 PCMA/8000 a = rtpmap:101 telephone-event/8000

101 being the DTMF attribute ETC

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