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QUESTION 1

Which recommendation is the best practice for marking video and voice media in a Cisco Unified Communications network?

A. Voice Cos 5 (IP Precedence 6, PHB AF41, or DSCP 16) Video Cos 4 (IP Precedence 5, PHB EF, or DSCP 32)

B. Voice Cos 6 (IP Precedence 4, PHB AF41, or DSCP 24) Video Cos 5 (IP Precedence 4, PHB EF, or DSCP 34)

C. Voice Cos 5 (IP Precedence 2, PHB EF, or DSCP 48) Video Cos 4 (IP Precedence 4, PHB AF41, or DSCP 46)

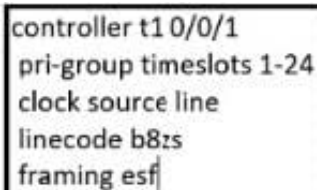D. Voice Cos 5 (IP Precedence 5, PHB EF, or DSCP 46) Video Cos 4 (IP Precedence 4, PHB AF41, or DSCP 34)

Correct Answer: D

Reference: https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/srnd/collab10/collab10/netstruc.html

QUESTION 2

Refer to the exhibit.

```
controller t1 0/0/1
pri-group timeslots 1-24
clock source line
linecode b8zs
framing esf
```

An administrator is attempting to register a SIP phone to a Cisco UCM but the registration is failing. The IP address of the SIP Phone is 10.117.34 222 and the IP 1 address of the Cisco UCM is 10.0.101.10. Pings from the SIP phone to the Cisco UCM are successful. What is the cause of this issue and how should it be resolved?

A. The certificates on the SIP phone are not trusted by the Cisco UCM. The SIP phone must generate new certificates.

B. DNS lookup for the Cisco UCM FQDN is failing. The SIP phone must be reconfigured with the proper DNS server.

C. An NTP mismatch is preventing connection of the TCP session between the SIP phone and the Cisco UCM. The SIP phone and Cisco UCM must be set with identical NTP sources.

D. An network device is blocking TCP port 5060 from the SIP phone to the Cisco UCM. This device must be reconfigured to allow traffic from the IP phone.

Correct Answer: D

QUESTION 3

When a call is delivered to a gateway, the calling and called party number must be adapted to the PSTN service requirements of the trunk group. If a call is destined locally, the + sign and the explicit country code must be replaced

with a national prefix. For the same city or region, the local area code must be replaced by a local prefix as applicable. Assuming that a Cisco UCM has a SIP trunk to a New York gateway (area code 917), which two combinations of solutions localize the calling and called party for a New York phone user? (Choose two.)

A. Configure the gateway to translate the calling number and apply it to the dial peer. Combine it with a translation profile for called numbers. ! voice translation-rule 1 rule 1 /^1917/ // rule 2 /^ [+]1917/ // ! voice translation-profile strip+1 translate calling 1 !

B. Configure two calling party transformation patterns: \+1917.CCCCCC, strip pre-dot, numbering type: subscriber \+!, strip pre-dot, numbering type: national

C. Configure two called party transformation patterns: \+1917.XXXXXXX, strip pre-dot, numbering type: subscriber \+1.!, strip pre-dot, numbering type: national

D. Configure the gateway to translate called numbers and apply it to the dial peer. Combine it with a translation profile for calling numbers. ! voice translation-rule 1 rule 1 /^1917!/ // rule 2 /^[+]1917!/ // ! voice translation-profile strip+1 translate called 1 !

E. Configure two calling party transformation patterns: \+1917.XXXXXXX, strip pre-dot, numbering type: subscriber \+1.!, strip pre-dot, numbering type: national

Correct Answer: DE

QUESTION 4

Refer to the exhibit. What is the registration state of the analog port in this debug output?


```
dial-peer voice 2 voip
destination-pattern 5555678
sessiontarget ipv4:10.5.6.7
codec g729r8
fax protocol t38 ls-redundancy 0 hs-redundancy 0 fallback cisco
fax rate voice
```

A. The MGCP Gateway is not communicating with the Cisco UCM.

B. The analog port failed to register to Cisco UCM with an error code 200.

C. The analog port is currently shut down.

D. The analog port is registered to Cisco UCM.

Correct Answer: D

CUCM sends an AUEP (Audit Endpoint) to the Gateway to determine the status of the given Endpoint. The response

from the Gateway is an ACK with the endpoints capabilities. Once this is complete the Endpoint is registered with the CUCM. https://www.cisco.com/c/en/us/support/docs/voice/media-gateway-control-protocol-mgcp/214635-configure-and-troubleshoot-mgcp-gateways.html

QUESTION 5

The IP phones at a customer site do not pick an IP address from the DHCP. An engineer must temporarily disable LLDP on all ports of the switch to leave only CDP. Which two commands accomplish this task? (Choose two.)

A. Switch(config)# no lldp transmit

B. Switch# configure terminal

C. Switch# copy running-config startup-config

D. Switch(config)# no lldp run

E. Switch(config)# interface GigabitEthernet1/0/1

Correct Answer: BD

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