

- C. AXL
- D. XMPP
- E. QBE

**Correct Answer:** BD

**QUESTION 21**

Which Cisco Unified communications manager configuration is required for SIP MWI integration?

- A. Select "Redirecting Diversion Header Delivery Inbound" on the SIP trunk.
- B. Enable "Accept presence subscription" on the SIP trunk security profile.
- C. Select "Redirecting Diversion Header Delivery outbound" on the SIP trunk.
- D. Enable "Accept unsolicited notification" on the SIP Trunk security profile.

**Correct Answer:** D

**QUESTION 22**

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. TFTP
- B. LACP
- C. LLDP
- D. SNMP

**Correct Answer:** C

**QUESTION 23**

A customer wants a video conference with five Cisco TelePresence IX5000 series system. Which media resource is necessary in the design to fully utilize the immersive functions?

- A. Software conference bridge on Cisco Unified Communications Manager
- B. Cisco PVD4-128
- C. Cisco Webex meetings server
- D. Cisco meeting server

**Correct Answer:** D

**QUESTION 24**

How does Cisco Unified Communications Manager perform a digit analysis on-hook versus off-hook for an outbound call from a Cisco IP phone that is registered to Cisco Unified CM?

- A. On-hook, by pressing the digits and entering "#" to process the call, Unified CM performs a digit-by-digit analysis; off-hook, Unified CM analyses all digits as a string.
- B. On-hook, unified CM performs a digit-by-digit analysis; off-hook, Unified CM considers all digit were dialed and dos not wait for additional digits.
- C. On-hook, no digit analysis is performed; off-hook, Unified CM requires the "#" to start the digit analysis.

- D. On-hook, Unified CM considers all digits were dialed and does not wait for additional digits; off-hook, Unified CM performs a digit-by-digit analysis.

**Correct Answer:** D

**QUESTION 25**

On which protocol and port combination does Cisco prime collaboration receive notifications (Traps and InformRequests) from several network devices in the collaboration infrastructure for which it has requested notifications?

- A. UDP 161
- B. TCP 161
- C. TCP 80
- D. UDP 162

**Correct Answer:** D

**QUESTION 26**

What is a valid class included in the 8-class QoS strategy in a VoIP network?

- A. Assured Forwarding
- B. Broadcast video
- C. Multimedia conferencing
- D. Real-Time Interactive

**Correct Answer:** C

**QUESTION 27**

On which Cisco Unified Communications Manager nodes can the TFTP service be enabled?

- A. Any node
- B. Any subscriber nodes
- C. Only nodes that have Cisco Unified CM service enabled
- D. Any two nodes

**Correct Answer:** D

**QUESTION 28**

Which settings are needed to configure the SIP route pattern in Cisco Unified Communications Manager?

- A. Pattern usage, IPv4 pattern, IPv6 pattern, and description
- B. Pattern usage, IPv4 pattern, and SIP trunk/Route list
- C. Pattern usage, IPv6 pattern, and SIP trunk/Route list
- D. SIP trunk/Route list, description, and IPv4 pattern

**Correct Answer:** B

---

**QUESTION 29**

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

- A. Voice Cos 6 (IP Precedence 4, PHB AF41, or DSCP 24)  
Video Cos 5 (IP Precedence 4, PHB EF, or DSCP 34)
- B. Voice Cos 5 (IP Precedence 5, PHB EF, or DSCP 46)  
Video Cos 4 (IP Precedence 4, PHB AF41, 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 6, PHB AF41, or DSCP 16)  
Video Cos 4 (IP Precedence 5, PHB EF, or DSCP 32)

**Correct Answer: A**

**QUESTION 30**

Refer the exhibit. Given this "debug isdn q921" output, what is the problem with the PRI?

```
000193: Dec 5 14:35:31.054: ISDN Se0/1/0:15 Q921: User TX -> SABMEp sapi=0 tei=0
000193: Dec 5 14:35:32.054: ISDN Se0/1/0:15 Q921: User TX -> SABMEp sapi=0 tei=0
000193: Dec 5 14:35:33.054: ISDN Se0/1/0:15 Q921: User TX -> SABMEp sapi=0 tei=0
000193: Dec 5 14:35:34.054: ISDN Se0/1/0:15 Q921: User TX -> SABMEp sapi=0 tei=0
```

- A. Layer 1 is down on the controller.
- B. PRI does not have an IP address configured on the interface.
- C. Noting, the PRI is sending keepalives.
- D. Layer 2 is down on the controller.

**Correct Answer: D**

**QUESTION 31**

Where is the default for maximum session Bit Rate for a region configured?

- A. Region configuration
- B. Enterprise Phone configuration
- C. Service parameter configuration
- D. Enterprise parameters configuration

**Correct Answer: A**

**QUESTION 32**

You are adding regions in Cisco Unified Communications Manager. Which codec(s) are selected when a call is placed if you set up the max audio bit rate to use 8 kbps?

- A. G.722
- B. G.729
- C. G.729 and G.711ualw
- D. G.711ulaw and G.711alaw

**Correct Answer: B**

**QUESTION 33**

How are E.164 called-party numbers normalized on a globalized call-routing environment is Cisco Unified Communications Manager?

- A. Normalization is achieved by stripping or translating the called numbers.
- B. Call ingress must be normalized before the call being routed.
- C. Normalization is not required.
- D. Normalization is achieved by setting up calling search and partitions at the SIP trunk for PSTN connection.

**Correct Answer: A**

**QUESTION 34**

A customer has Cisco Unity Connection that is integrated with LDAP. As a unity connection administrator, you have received a request to change the first for VM user. Where must the change be performed?

- A. Cisco unity connection
- B. Cisco IM and presence
- C. Cisco Unified Communications Manager end user
- D. Active directory

**Correct Answer: D**

**QUESTION 35**

An engineer is configuring a BOT device for a Jabber user is Cisco Unified Communication Manager. Which phone type must be selected?

- A. Cisco Dual mode for Android
- B. Cisco Unified Client services Framework
- C. third-party SIP device
- D. Cisco Dual Mode for iPhone

**Correct Answer: A**

**QUESTION 36**

Which command in the MGCP gateway configuration defines the secondary Cisco Unified Communications Manager server?

- A. ccm-manager redundant-host
- B. Mgcg call-agent
- C. Mgcpgapp
- D. ccm-manager fallback-mgcp

**Correct Answer: A**

**QUESTION 37**

Which access control group is required on an end user to allow Jabber to do deskphone mode?

- A. Standard CTI Enabled

- B. Standard CTI Allow Reception of SRTP key Material
- C. Allow control of device from CTI
- D. Standard CTI Secure Connection

**Correct Answer: A**

#### QUESTION 38

Refer to the exhibit. A call is failing to establish between two SIP Devices. The Called device answer with this SDP. Which SDP parameter causes this issue?

```
v=0
o=Cisco-SIPUA 13439 0 IN IP4 10.10.10.10
s=SIP Call
b=AS:4064
t=0 0
m=audio 0 RTP/AVP 114 9 124 113 115 0 8 116 18 101
c=IN IP4 10.10.10.10
b=TIAS:64000
a=rtpmap:114 opus/48000/2
a=fmtp:114 maxplaybackrate=16000;prop-
maxcapture=16000;maxaveragebitrate=64000;stereo=0;prop-stereo=0;usedtx=0
a=rtpmap:9 G722/8000
a=rtpmap:124 ISAC/16000
a=rtpmap:113 AMR-WB/16000
a=fmtp:113 octet-align=0;mode-change-capability=2
a=rtpmap:115 AMR-WB/16000
a=fmtp:115 octet-align=1;mode-change-capability=2
a=rtpmap:0 PCMU/8000
a=rtpmap:8 PCMA/8000
a=rtpmap:116 iLBC/8000
a=fmtp:116 mode=20
a=rtpmap:18 G729/8000
a=fmtp:18 annexb=yes
a=rtpmap:101 telephone-event/8000
```

```
b=TIAS:64000
a=rtpmap:114 opus/48000/2
a=fmtp:114 maxplaybackrate=16000;prop-
maxcapture=16000;maxaveragebitrate=64000;stereo=0;prop-stereo=0;usedtx=0
a=rtpmap:9 G722/8000
a=rtpmap:124 ISAC/16000
a=rtpmap:113 AMR-WB/16000
a=fmtp:113 octet-align=0;mode-change-capability=2
a=rtpmap:115 AMR-WB/16000
a=fmtp:115 octet-align=1;mode-change-capability=2
a=rtpmap:0 PCMU/8000
a=rtpmap:8 PCMA/8000
a=rtpmap:116 iLBC/8000
a=fmtp:116 mode=20
a=rtpmap:18 G729/8000
a=fmtp:18 annexb=yes
a=rtpmap:101 telephone-event/8000
a=fmtp:101 0-15
a=sendrecv
```

- A. The RTP port is set to 0.
- B. The payload for G.711ulaw must be 18.
- C. The media stream is set to sendonly.
- D. The calling device did not offer aptime value.

**Correct Answer: A**