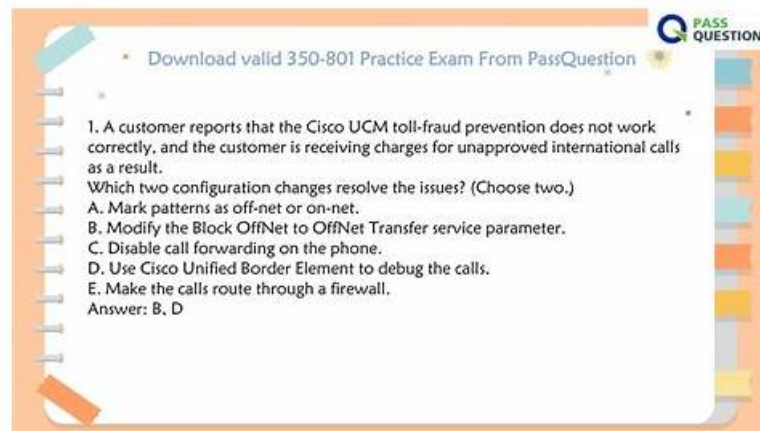


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Cisco Implementing and Operating Cisco Collaboration Core Technologies Sample Questions (Q175-Q180):

NEW QUESTION # 175

Refer to the exhibit.

Region Configuration

Save Delete Reset Apply Config Add New

Region Information

Name: Dallas-REG

Region Relationships

Region	Audio Codec Preference List	Maximum Audio Bit Rate	Maximum Session Bit Rate for Video Calls
SanJose-REG	Use System Default (Factory Default low loss)	24 kbps (AMR-WB)	Use System Default (384 kbps)
NOTE: Regions not displayed	Use System Default	Use System Default	Use System Default

Modify Relationship to other Regions

Regions	Audio Codec Preference List	Maximum Audio Bit Rate	Maximum Session Bit Rate for Video Calls
Austin-REG			
Dallas-REG			
Default			
SanJose-REG			

Keep Current Setting

Which codec should an engineer select for a call made between "Dallas-REG" & "Austin-REG"?

- A. G.729
- B. G.711
- C. MP4A-LATM
- D. OPUS

Answer: A

Explanation:

The codec preference list for the "Dallas-REG" region is "Factory Default low loss". This list includes the following codecs in order of preference:

- * G.729
- * G.711
- * OPUS
- * MP4A-LATM

The codec preference list for the "Austin-REG" region is "Factory Default low loss". This list includes the following codecs in order of preference:

- * G.729
- * G.711
- * OPUS
- * MP4A-LATM

Since both regions have the same codec preference list, the codec that will be used for a call made between "Dallas-REG" and "Austin-REG" is G.729.

G.729 is a narrowband speech codec that was developed by the ITU-T in 1988. It is a low-bitrate codec that provides good quality speech at a bitrate of 8 kbps. G.729 is widely used in VoIP applications and is the default codec for many VoIP systems.

G.711 is a wideband speech codec that was developed by the ITU-T in 1972. It is a high-bitrate codec that provides excellent quality speech at a bitrate of 64 kbps. G.711 is not as widely used as G.729 due to its high bitrate requirements.

OPUS is a lossy audio codec that was developed by the IETF in 2012. It is a low-bitrate codec that provides good quality speech at a bitrate of 6 kbps. OPUS is widely used in VoIP applications and is the default codec for many VoIP systems.

MP4A-LATM is a lossy audio codec that was developed by the IETF in 1999. It is a high-bitrate codec that provides excellent quality speech at a bitrate of 24 kbps. MP4A-LATM is not as widely used as G.729 or OPUS due to its high bitrate requirements.

NEW QUESTION # 176

Which two conditions must a user meet to provision a new device using the self-provisioning feature? (Choose two.)

- A. The user must be part of "standard CCM Super User".
- B. The user must have at least one user device profile assigned.
- C. At least DN's must be assigned to the user device.
- D. The user must have the appropriate universal device template linked to the linked to the user profile.
- E. The user must have a primary extension.

Answer: D,E

NEW QUESTION # 177

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

- A. traffic shaping
- **B. CBWFQ**
- C. DiffServ
- D. traffic policing

Answer: B

NEW QUESTION # 178

An administrator is trying to change the default LINECODE for a voice ISDN T1 PRI. Which command makes this change?

- A. linecode esf
- B. linecode hdb3
- **C. linecode b8zs**
- D. linecode ami

Answer: C

Explanation:

Explanation/Reference: <https://www.cisco.com/en/US/docs/ios/dial/configuration/guide/>

dia_cfg_isdn_pri_external_docbase_0900e4b1806c752c_4container_external_docbase_0900e4b18216dd1b.html

NEW QUESTION # 179

Refer to the exhibit.



```
10.10.10.2:5060;branch=a8bH5bK7954A198F
From:
<sip:012345678@10.10.10.2>;tag=8D79AF62-DB2
To: <sip:90123456@10.10.4.14>;
tag=811681~ffa80926-5fac-4cc5-b802-
2dbde74ae7w2-
v=0
o=CiscoSystemsCCM-SIP 811681 1 IN IP4
10.10.4.14
s=SIP Call
c=IN IP4 10.5.4.3
t=0 0
m=audio 27839 RTP/AVP 0 101
a=rtpmap:0 PCMU/8000
a=ptime:20
a=rtpmap:101 telephone-event/8000
a=fmtr:101 0-15
```

Which Codec is negotiated?

- **A. G.711ulaw**
- B. G.729
- C. ILBC
- D. G.728

Answer: A

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