

Exam Questions 350-801

Implementing and Operating Cisco Collaboration Core Technologies

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NEW QUESTION 1

Which two access layer switches provide support to provide high-quality voice and take advantage of the full voice feature set. To provide high-quality voice and take advantage of the full voice feature set, which two access layer switches provide support? Choose two

- A. Use multiple egress queues to provide priority queuing of RTP voice packet streams and the ability to classify or reclassify traffic and establish a network trust boundary.
- B. Use 808.IQ trunking and 802.Ip for proper treatment of Layer 2 CoS packet marking on ports with phones connected.
- C. Implement IP RTP header compression on the serial interface to reduce the bandwidth requlreq per voice call on point-to-point links.
- D. Deploy RSVP to improve VoIP QoS only where it can have a positive Impact on quality and functionality where there Is limited bandwidth and frequent network congestion.
- E. Map audio and video streams of video calls (AF41 and AF42) to a class-based queue with weighted random early detection.

Answer: AB

NEW QUESTION 2

Which service must be enabled when LDAP on cisco UCM is used?

- A. Cisco AXL Web Service
- B. Cisco CallManager SNMP Service
- C. Cisco DirSync
- D. Cisco Bulk Provisioning Service

Answer: C

NEW QUESTION 3

Which SNMP service must be activated manually on the Cisco Unified Communications Manager after installation?

- A. Cisco CallManager SNMP
- B. SNMP Master Agent
- C. Connection SNMP Agent
- D. Host Resources Agent

Answer: A

NEW QUESTION 4

A company has an excessive number of call transfers to local and long-distance PSTN from Cisco Unity Connection voicemail. Which action in the Cisco Unity Connection restriction table resolves this issue?

- A. Block PSTN patterns on Default Transfe
- B. Default Outdia
- C. and Default System Transfer.
- D. Implement password complexity on voicemail boxes to prevent accounts from being compromised.
- E. Create a custom restriction table ???????????? and block it.
- F. Create a custom restriction table ***** and block it.

Answer: A

NEW QUESTION 5

Which two features of Cisco Prime Collaboration Assurance require advanced licensing? (Choose two.)

- A. real lime alarm browse
- B. multicluster support
- C. call quality monitoring
- D. email notifications
- E. customizable events

Answer: BC

NEW QUESTION 6

A customer is deploying a SIP IOS gateway for a customer who requires that in-band DTMF relay is first priority and out-of-band DTMF relay is second priority. Which 10\$ entry sets the required priority?

- A. dtmf-relay cisco-rtp
- B. dtmf-relay sip-kpml cisco-rtp
- C. sip-notify dtmf-relay rtp-nte
- D. dtmf-relay rtp-nte sip-notify

Answer: D

NEW QUESTION 7

A Cisco Telepresence SX80 suddenly has issues displaying main video to a display over HDMI. Which command can you use from the SX80 admin CLI to check the video output status to the monitor?

- A. xStatus HDMI Output
- B. xStatus video Output

- C. xconfiguration video Output
- D. xcommand video status

Answer: B

NEW QUESTION 8

What happens when a Cisco IP phone loses connectivity to the duster during an active call?

- A. The call continues to be active, but features like transfer or hold do not work.
- B. The call continues and all features work.
- C. The call drops immediately.
- D. The call drops after missing two keepalives from Cisco UCM.

Answer: D

NEW QUESTION 9

Refer to the exhibit.

| Time | Source | Destination | Info |
|-----------|---------------|-------------|---|
| 18.683437 | 10.117.34.222 | 10.0.101.10 | 50310 → 5060 [SYN] Seq=0 Win=64240 Len=0 MSS=1460 WS=256 SACK |
| 18.938881 | 10.117.34.222 | 10.0.101.10 | 50314 → 5060 [SYN] Seq=0 Win=64240 Len=0 MSS=1460 WS=256 SACK |
| 21.686680 | 10.117.34.222 | 10.0.101.10 | [TCP Retransmission] 50310 → 5060 [SYN] Seq=0 Win=64240 Len=0 |
| 21.941993 | 10.117.34.222 | 10.0.101.10 | [TCP Retransmission] 50314 → 5060 [SYN] Seq=0 Win=64240 Len=0 |
| 27.687008 | 10.117.34.222 | 10.0.101.10 | [TCP Retransmission] 50310 → 5060 [SYN] Seq=0 Win=64240 Len=0 |
| 27.942784 | 10.117.34.222 | 10.0.101.10 | [TCP Retransmission] 50314 → 5060 [SYN] Seq=0 Win=64240 Len=0 |

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 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. An NTP mismatch is preventing the connection of the TCP session between the SIP phone and the Cisco UC
- B. The SIP phone and Cisco UCM must be set with identical NTP sources.
- C. The certificates on the SIP phone are not trusted by the Cisco UC
- D. The SIP phone must generate new certificates.
- E. DNS lookup for the Cisco UCM FQDN is failin
- F. The SIP phone must be reconfigured with the proper DNS server.
- G. An network device is blocking TCP port 5060 from the SIP phone to the Cisco UC
- H. This device must be reconfigured to allow traffic from the IP phone.

Answer: D

NEW QUESTION 10

What are two key features of the Expressway series? (Choose two.)

- A. VPN connection toward the internal UC resources
- B. SIP header modification
- C. B2B calls
- D. device registration over the Internet
- E. IP to PSTN call connectivity

Answer: CD

NEW QUESTION 10

How many minutes does it take for automatic fallback to occur in a Presence Redundancy Group if the primary node lost a critical service?

- A. 5 min
- B. 10 min
- C. 30 min
- D. 60 min

Answer: C

NEW QUESTION 13

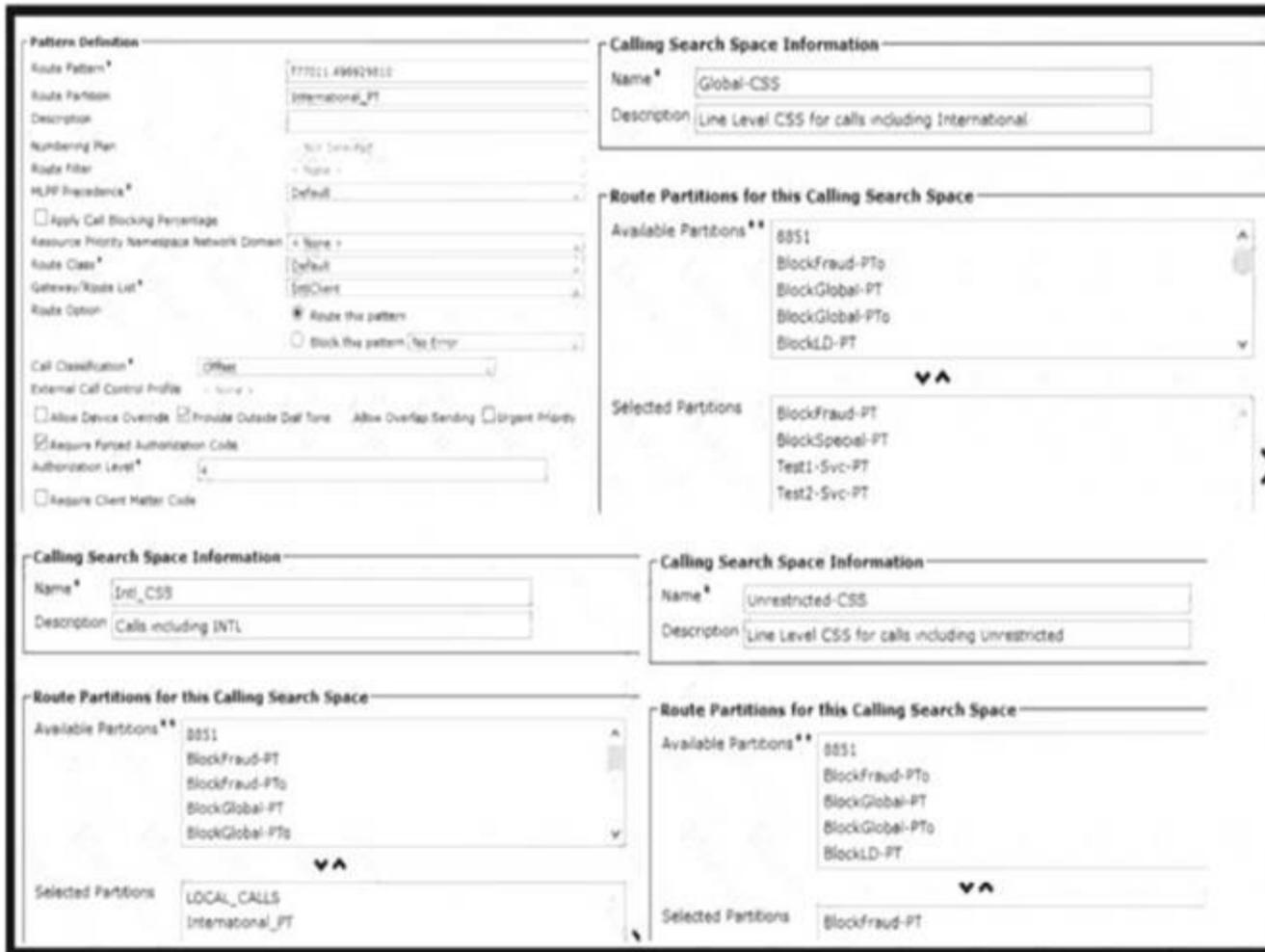
A Cisco UCM administrator wants to enable the Self-Provisioning feature for end users. Which two prerequisites must be met first? (Choose two.)

- A. End users must have a secondary extension.
- B. Cisco Extended Functions service must be running
- C. End users must belong to Standard CCM Admin Users group, the Standard CCM End Users group, and the Standard CCM Self-Provisioning group.
- D. End users must have a primary extension.
- E. End users must be associated to a user profile or feature group template that includes a universal line template and universal device template.

Answer: DE

NEW QUESTION 17

Refer to the exhibit.



How must the +E.164 translation pattern be configured to reach international number 496929810?

- Pattern= \+.496929810, CSS=Unrestricted-CSS, PreDot, Prefix=777011
- Pattern= \+.777011496929810, CSS=Intl_CSS
- Pattern= \+.011496929810, CSS=Global-CSS, PreDot, Prefix=777
- Pattern= \+.496929810, CSS=Intl_CSS, PreDot, Prefix=777011

- A. Option A
- B. Option B
- C. Option C
- D. Option D

Answer: D

NEW QUESTION 21

An engineer is asked to implement on-net/off-net call classification in Cisco UCM. Which two components are required to implement this configuration? (Choose two.)

- A. CTI route point
- B. SIP route patterns
- C. route group
- D. route pattern
- E. SIP trunk

Answer: DE

NEW QUESTION 23

What is required for Cisco UCM to accept SIP calls with a URI in the format of 'sip:2001@cucmpub.cisco.com'?

- A. Define Cluster Fully Qualified Domain Name under Servers in Cisco UCM.
- B. Change the Destination Address to a Fully Qualified Domain Name on the SIP trunk.
- C. Define Cluster Fully Qualified Domain Name in Enterprise Parameters.
- D. Set the SIPS URI Handling to True in CallManager Service Parameters.

Answer: C

NEW QUESTION 26

A Cisco IP Phone 7841 that is registered to a Cisco Unified Communications Manager with default configuration receives a call setup message. Which codec is negotiated when the SDP offer includes this line of text?

M=audio 498181 RTP/AVP 0 8 97

- A. G.711ulaw
- B. iLBC

C. G.711alaw
D. G.722

Answer: A

NEW QUESTION 28

An engineer configures a Cisco Unified Border Element and must ensure that the codecs negotiated meet the ITSP requirements. The ITSP supports G.711ulaw and G.729 for audio and H.264 for video. The preferred voice codec is G.711. Which configuration meets this requirement?

A.

```
voice class codec 10
  codec preference 1 g711ulaw
  codec preference 2 g729r8

dial-peer voice 101 voip
  session protocol sipv2
  destination e164-pattern-map 1
  voice-class codec 10
```

B.

```
voice class codec 10
  codec preference 1 g729r8
  codec preference 2 g711ulaw
  video codec h264

dial-peer voice 101 voip
  session protocol sipv2
  destination e164-pattern-map 1
  voice-class codec 10
```

C.

```
voice class codec 10
  codec preference 1 g711ulaw
  codec preference 2 g729r8
  video codec h264

dial-peer voice 101 voip
  session protocol sipv2
  destination e164-pattern-map 1
  voice-class codec 100
```

D.

```
voice class codec 10
  codec preference 1 g711ulaw
  codec preference 2 g729r8
  video codec h264

dial-peer voice 101 voip
  session protocol sipv2
  destination e164-pattern-map 1
  voice-class codec 10
```

Answer: D

NEW QUESTION 30

An end user at a remote site is trying to initiate an Ad Hoc conference call to an end user at the main site. The conference bridge is configured to support G.711. The remote user's phone only supports G.729. The remote end user receives an error message on the phone: "Cannot Complete Conference Call." What is the cause of the issue?

- A. The remote phone does not have the conference feature assigned.
- B. A software conference bridge is not assigned.
- C. A Media Termination Point is missing.
- D. The transcoder resource is missing.

Answer: D

NEW QUESTION 33

An administrator needs to create a partial PRI consisting of the first seven timeslots available. Which configuration snippet configures the ISDN E1 PRI for this task?

- A.

```
config t
2900(config)#isdn switch-type primary-ni
2900(config)#interface Serial0/0/0:15
2900(config-controller)#pri-group timeslots 1-7
```
- B.

```
config t
2900(config)#isdn switch-type primary-ni
2900(config)#controller e1 0/0/0
2900(config-controller)#pri-timeslots 1-7
```
- C.

```
config t
2900(config)#isdn switch-type primary-ni
2900(config)#controller e1 0/0/0
2900(config-controller)#pri-group timeslots 1-7
```
- D.

```
config t
2900(config)#isdn switch-type primary-ni
2900(config)#pri-group timeslots 1-7
```

Answer: C

NEW QUESTION 36

Which behavior occurs when Cisco UCM has a CallManager group that consists of two subscribers?

- A. Endpoints attempt to register with the bottom subscriber in the list.
- B. Endpoints attempt to register with the top subscriber in the list.
- C. Endpoints attempt to register with both subscribers in a load-balanced method.
- D. If a subscriber is rebooted, endpoints deregister until the rebooted system is back in service.

Answer: B

NEW QUESTION 41

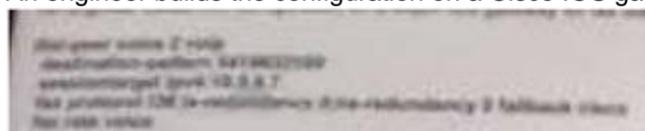
Which two protocols are proxied over an Expressway-E/C pair when a Mobile and Remote Access login including phone services is performed? (Choose two.)

- A. HTTPS
- B. H.323
- C. SIP
- D. SCCP
- E. SRTP

Answer: AC

NEW QUESTION 44

An engineer builds the configuration on a Cisco IOS gateway for the dial-peers:



Which command is required to complete the configuration?

- A. Codec g726r32
- B. Codec g729cr81
- C. Codec g723ar63
- D. Codec g711ulaw

Answer: D

NEW QUESTION 47

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

Answer: D

NEW QUESTION 52

An engineer deploys a Cisco Expressway-E server for a customer who wants to utilize all features on the server. Which feature does the engineer configure on the Expressway-E?

- A. H.323 endpoint registrations
- B. Mobile and Remote Access
- C. SIP gateway for PSTN providers
- D. VTC bridge

Answer: B

NEW QUESTION 53

How are E.164 called-party numbers normalized on a globalized call-routing environment in Cisco UCM?

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

Answer: C

NEW QUESTION 56

Endpoint A is attempting to call endpoint B. Endpoint A only supports G.711ulaw with a packetization rate of 20 ms, and endpoint B supports packetization rate of 30 ms for G.711ulaw. Which two media resources are allocated to normalize packetization rates through transrating? (Choose two.)

- A. software MTP on Cisco IOS Software
- B. software MTP on Cisco UCM
- C. software transcoder on Cisco UCM
- D. hardware transcoder on Cisco IOS Software
- E. hardware MTP on Cisco IOS Software

Answer: BE

NEW QUESTION 58

Which command in the MGCP gateway configuration defines the secondary Cisco UCM server?

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

Answer: A

NEW QUESTION 62

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# copy running-config startup-config
- B. Switch(config)# no lldp run
- C. Switch# configure terminal
- D. Switch(config)# interface GigabitEthernet1/0/1
- E. Switch(config)# no lldp transmit

Answer: BC

NEW QUESTION 67

According to the QoS Baseline Model, drag and drop the applications from the left onto the Per-Hop Behavior values on the right.



- A. Mastered
- B. Not Mastered

Answer: A

Explanation:



NEW QUESTION 72

An employee of company ABC just quit. The IT administrator deleted the employee's user id from the active directory at 10 a. m. on March 4th The nightly sync occurs at 10 p.m. daily. The IT administrator wants to troubleshoot and find a way to delete the user id as soon as possible How is this issue resolved?

- A. Wait until 10 pm on March 4th when the user is automatically removed from Cisco UCM.
- B. Wait until 10 pm on March 5th when the user is automatically removed from Cisco UCM.
- C. Wait until 3 15 a.
- D. on March 6th for garbage collection to remove the user from Cisco UCM.
- E. Wait until 315am on March 5th for garbage collection to remove the user from Cisco UCM.

Answer: C

NEW QUESTION 77

When setting a new primary DNS server in the Cisco UCM CLI what is required for the change to take affect?

- A. restart of CallManager service
- B. restart of DirSync service
- C. restart of the network service
- D. restart of TFTP service

Answer: C

NEW QUESTION 78

An administrator needs to help a remote employee make a free call to an international destination. The administrator calls the employee, then conferences in the international party. The administrator drops the call, and the employee and the international party continue their conversation. Which action prevents this type of toll fraud in the Cisco UCM?

- A. Set service parameter 'Advanced Ad Hoc Conference' to FALSE.
- B. Set service parameter "Drop Ad Hoc Conference" to "When Conference Controller leaves."
- C. Set service parameter "Advanced Ad Hoc Conference" to 2.
- D. Set service parameter "Drop Ad Hoc Conference" to "Do not allow outside parties."

Answer: B

NEW QUESTION 79

An engineer implements QoS in the enterprise network. Which command is used to verify the classification and marking on a Cisco IOS switch?

- A. show class-map interface GigabitEthernet 1/0/1
- B. show policy-map interface GigabitEthernet 1/0/1
- C. show access-lists
- D. show policy-map

Answer: B

NEW QUESTION 83

Which certificate does the Disaster Recovery System in Cisco UCM use to encrypt its communications?

- A. Cisco Tomcat
- B. CAPF
- C. Cisco CallManager
- D. IPsec

Answer: D

NEW QUESTION 87

Which two recommendations are made to optimize Cisco UCM configuration to reduce the number of toll fraud incidents in an organization? (Choose two.)

- A. Classify all route patterns as on-net and prohibit on-net to on-net call transfers in Cisco UCM service parameters.
- B. Classify all route patterns as on-net or off-net and prohibit off-net to off-net call transfers in Cisco UCM service parameters.
- C. Classify all route patterns as off-net and prohibit off-net to off-net call transfers in Cisco UCM service parameters.
- D. Inbound CSS on any gateway typically should have access to internal destinations and PSTN destinations.
- E. Inbound CSS on any gateway typically should have access to internal destinations only and not PSTN destinations.

Answer: BE

NEW QUESTION 92

An engineer must configure a SIP route pattern using domain routing with destination +13135551212. The domain ciscocm1.jupiter.com resolves to 192.168.1.3. How must the IPV4 Pattern be configured?

- A. +13135551212@192.168.1.3
- B. ciscocm1.jupiter.com
- C. \+13135551212@192.168.1.3
- D. 192.168.1.3

Answer: B

NEW QUESTION 96

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

- A. On-hoo
- B. by pressing the digits and entering "#" to process the cal
- C. UCM performs a digit-by-digit analysis; off-hoo
- D. UCM analyzes all digits as a string.
- E. On-hoo
- F. no digit analysis is performed; off-hoo
- G. UCM requires the '#' to start the digit analysis
- H. On-hoo
- I. UCM performs a digit-by-digit analysis; off-hoo
- J. UCM considers all digits were dialed and does not wait for additional digits.
- K. On-hoo
- L. UCM considers all digits were dialed and does not wait for additional digits; off-hoo
- M. UCM performs a digit-by-digit analysis.

Answer: D

NEW QUESTION 97

A user dials 9011841234567 to reach Vietnam. Which steps send the call to the PSTN provider as 011841234567?

A)

in the Called Party Transformation Pattern Configuration section,
 configure the Pattern as 9.011841234567
 configure the Discard Digits as Predot

B)

in the Calling Party Transformation Patterns section,
 configure the Pattern as 9.011841234567
 configure the Discard Digits as Predot 10-10-Dialing

C)

in the Calling Party Transformation Patterns section,
 configure the Pattern as 9.011841234567
 configure the Discard Digits as Predot

D)

in the Called Party Transformation Pattern Configuration section,
 configure the Pattern as 9.011841234567
 configure the Discard Digits as Predot 10-10-Dialing

- A. Option A
- B. Option B
- C. Option C
- D. Option D

Answer: A

NEW QUESTION 100

What happens to voice packets from a Cisco 8845 IP phone in the QoS trust boundary?

- A. The voice packets are not trusted, and the access layer switch reclassifies the packets.
- B. The voice packets are classified by the phone, and the classification is accepted
- C. The voice and access layer switch negotiate the classification of packets
- D. Cisco UCM determines how the voice packers are classified.

Answer: B

NEW QUESTION 103

Where in Cisco UCM is restrictions on audio bandwidth configured?

- A. location
- B. partition
- C. region
- D. serviceability

Answer: C

NEW QUESTION 104

An administrator would like to set several Cisco Jabber configuration parameters to only apply to mobile clients (iOS and Android). How does the administrator accomplish this with Cisco Jabber 12.9 and Cisco UCM 12.5?

- A. Assign the desired configuration file to "Mobile" Jabber Client Configuration in the Service Profile.
- B. Upload the jabber-config.enc file to TFTP
- C. Create a user profile in Jabber Policies.
- D. Deploy jabber-config-user.xml on iOS and Android devices.

Answer: A

NEW QUESTION 105

How does traffic policing respond to violations?

- A. Excess traffic is dropped.
- B. Excess traffic is retransmitted.
- C. All traffic is treated equally.
- D. Excess traffic is queued.

Answer: A

NEW QUESTION 108

Which value should be changed when each Cisco UCM node does not allow for more than 5000 phones to be registered?

- A. Maximum Number of Registered and Unregistered Devices service parameter on each node
- B. Minimum Number of Phones service parameter on each node
- C. Maximum Number of Registered Devices service parameter on each node

D. Maximum Number of Phones service parameter on the Publisher

Answer: C

NEW QUESTION 110

Given this H.323 gateway configuration and using Cisco best practices, how must the called party transformation pattern be configured to ensure that a proper ISDN type of number is set?

```
voice translation-rule 40
 rule 1 /3...$/ /408555&/
!
voice translation-profile INT
 translate calling 40
!
dial-peer voice 9011 pots
 translation-profile outgoing INT
 destination-pattern 9011T
 port 0/1/0:23
```

A.

Pattern Definition

Pattern *

Partition

Description

Numbering Plan

Route Filter

Urgent Priority

MLPP Preemption Disabled

Called Party Transformations

Discard Digits

Called Party Transformation Mask

Prefix Digits

Called Party Number Type *

Called Party Numbering Plan *

B.

Pattern Definition

Pattern *

Partition

Description

Numbering Plan

Route Filter

Urgent Priority

MLPP Preemption Disabled

Called Party Transformations

Discard Digits

Called Party Transformation Mask

Prefix Digits

Called Party Number Type *

Called Party Numbering Plan *

C.

Pattern Definition

Pattern*

Partition

Description

Numbering Plan

Route Filter

Urgent Priority

MLPP Preemption Disabled

Called Party Transformations

Discard Digits

Called Party Transformation Mask

Prefix Digits

Called Party Number Type*

Called Party Numbering Plan*

D. **Pattern Definition**

Pattern*

Partition

Description

Numbering Plan

Route Filter

Urgent Priority

MLPP Preemption Disabled

Called Party Transformations

Discard Digits

Called Party Transformation Mask

Prefix Digits

Called Party Number Type*

Called Party Numbering Plan*

Answer: C

NEW QUESTION 115

When a new SIP phone is registered to Cisco Unified communications Manager, it keeps failing and showing an “unprovisioned” error message in the phone display. Which problem is a possible cause of this issue?

- A. Auto-registration is disabled on the Cisco Unified Communications Manager nodes and the phone device does not have a DN configured.
- B. The DHCP settings are incorrectly and the phone does not have an alternate TFTP defined.
- C. The phone cannot download and install the latest firmware.
- D. The DN assigned to the phone is already in use by another SIP phone.
- E. The DN configuration for this phone is shared with SCCP phone, which is not supported.

Answer: A

NEW QUESTION 116

An engineer is configuring a Cisco Unified Border Element to allow the video endpoints to negotiate without the Cisco Unified Border Element interfering in the process. What should the engineer configure on the Cisco Unified Border Element to support this process?

- A. Configure path-thru content sdp on the voice service.
- B. Configure a hardcoded codec on the dial peers.
- C. Configure a transcoder for video protocols.
- D. Configure codec transparent on the dial peers.

Answer: D

NEW QUESTION 117

Users dial a 9 before a 10-digit phone number to make an off-net call. All 11 digits are sent to the Cisco Unified Border Element before going out to the PSTN. The PSTN provider accepts only 10 digits. Which configuration is needed on the Cisco Unified Border Element for calls to be successful?

- A. voice translation-rule 1 rule 1 /^9/ //
- B. voice translation-rule 1 rule 1 /^9(.....)/ //
- C. voice translation-rule 1 rule 1 /^9.+/ //
- D. voice translation-rule 1 rule 1 /^9...../ //

Answer: A

NEW QUESTION 122

Refer to the exhibit.

```
05:50:14.102: ISDN BR0/1/1 Q921: User TX -> IDREQ ri=21653 ai=127
05:50:14.134: ISDN BR0/1/1 Q921: User RX <- SABMEp sapi=0 tei=0
05:50:14.150: ISDN BR0/1/1 Q921: User TX -> IDREQ ri=19004 ai=127
05:50:14.165: ISDN BR0/1/1 Q921: User RX <- SABMEp sapi=0 tei=0
```

A customer submits this debug output, captured on a Cisco IOS router. Assuming that an MGCP gateway is configured with an ISDN BRI interface, which BRI changes resolve the issue?

A. C:\Users\wk\Desktop\mudassar\Untitled.jpg

```
interface BRI0/1/0
no ip address
isdn switch-type basic-net3
isdn point-to-multipoint-setup
isdn incoming-voice voice
isdn send-alerting
isdn static-tei 0
```

B. C:\Users\wk\Desktop\mudassar\Untitled.jpg

```
interface BRI0/1/1
no ip address
isdn switch-type basic-net3
isdn point-to-multipoint-setup
isdn incoming-voice voice
isdn send-alerting
isdn static-tei 0
```

C. interface BRI0/1/1

```
no ip address
isdn switch-type basic-net3
isdn point-to-point-setup
isdn incoming-voice voice
isdn send-alerting
isdn static-tei 0
```

D. C:\Users\wk\Desktop\mudassar\Untitled.jpg

```
interface BRI0/1/1
no ip address
isdn switch-type basic-net3
isdn incoming-voice voice
isdn send-alerting
isdn static-tei 0
```

Answer: C

NEW QUESTION 125

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