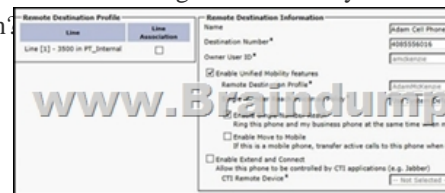


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<https://drive.google.com/drive/folders/0B75b5xYLjSSNQ3VqWUNLRU1XVik?usp=sharing> QUESTION 101 Which Cisco Unified Communications Manager tool verifies configured route patterns, calling search spaces, and route groups? A. Cisco Unified Reporting B. Cisco Unified Communications Manager CDR Analysis and Reporting C. Cisco Unified Communications Manager Dialed Number Analyzer D. Cisco Unified Real-Time Monitoring Tool Answer: QUESTION 102 Refer to the exhibit. A CUCM user has been configured to use the Mobility feature and is expecting their home phone 408-555-6016 to ring simultaneously when their office phone is called. What configuration change needs to be made to allow this happen?



A. Select the Enable Move to Mobile check box. B. Select the Enable Extend and Connect check box. C. Select the Line Association check box. D. Deselect the Enable Single Number Reach check box. Answer: QUESTION 103 Refer to the exhibit.

How many high-complexity transcoding sessions can this Cisco ISR G2 support?

```
1#show inventory
NAME: "CISCO2901/K9 chassis", DESCR: "CISCO2901/K9 chassis"
PID: C1502901/K9 , VID: V01 , SN: F0L16021F8

NAME: "VWIC3-IMFT-T1/E1 - 1-Port R3-48 Multiflex Trunk - T1/E1 on slot 0 SubSlot 0", DESCR: "VWIC3-IMFT-T1/E1"
PID: VWIC3-IMFT-T1/E1 , VID: V01 , SN: F0C16020N8F

NAME: "PVDM3 DSP DMM with 32 channels on slot 0 Subslot 4", DESCR: "PVDM3 DSP DMM with 32 Channels"
PID: PVDM3-32 , VID: V01 , SN: F0C16021J05

NAME: "C1941/C2901 AC Power Supply", DESCR: "C1941/C2901 AC Power Supply"
PID: PWR-1941-2901-AC , VID: , SN:

1#
1#show voice dsp capabilities slot 0 dsp 1
DSP Type: SP2600 -32
Card 0 dsp id 1 Capabilities:
Credits 480 , G711credits 15, MC Credits 34, MC credits 22,
FC Channel 14, MC Channel 14, MC Channel 21,
Conference & Party Credits
Security:
SEC LC Xcode 20, SEC MC Xcode 74,
SEC MC Xcode 26, SEC LC UNIV Xcode 20,
SEC MC UNIV Xcode 68, SEC MC UNIV Xcode 40,
SEC G729 conf 120, SEC G722 conf 120, SEC ILBC conf 160,
SEC G711 conf 68 ,
Max Conference Parties per DSP:
G711 104, G729 40, G722 40, ILBC 32,
SEC G711 56, SEC G729 32,
SEC G722 32 SEC ILBC 24,
voice Channels:
g711perdsp = 32, g726perdsp = 21, g729perdsp = 14, g729aperdsp = 21,
g723perdsp = 0, g728perdsp = 14, g711_ilbcpersdp = 22, gsmarrbperdsp = 14,
ilbcpersdp = 14, fsacpersdp = 7, modemrelayerpersdp = 14,
g7224apersdp = 21, h324persdp = 14,
m.f.tlrsuperdsp = 32, faxrelayerpersdp = 21,
maxchpersdp = 32, minchpersdp = 14,
srtp_maxchpersdp = 18, srtp_minchpersdp = 9, faxrelay_srtp_persdp = 9 ,
g711_srtp_persdp = 18, g729_srtp_persdp = 9, g729a_srtp_persdp = 16
gnx64_srtp_persdp = 18
```

A. 21 B. 14 C. 60 D. 15 Answer: B QUESTION 104 After a Cisco Unified Communications Manager system is installed, users report problems when more than four users attempt to join a Meet-Me conference. Which parameter should you increase? A.

Maximum Ad Hoc Conference, Call Manager Service Parameter B. Maximum Ad Hoc Conference, Enterprise Parameters Configuration

C. Maximum Meet-Me Conference, Call Manager Service Parameter D. Maximum Meet-Me Conference, Enterprise Parameters Configuration

Answer: QUESTION 105 Two Cisco DX80 collaboration endpoints that support high-definition video are registered to a Cisco Unified Communications Manager in a single-site deployment model. However, the video that is displayed is pixelated and appears as sub-HD. To enable the Cisco DX80s to make high-definition video calls, which configuration option do you check? A. Check the configuration in the Cisco DX80 settings menu, and enable high-definition video.

B. Check the Maximum Session Bit Rate for Video Calls in the Cisco Unified Communications Manager Regions Configuration.

C. Check that Video Capabilities are enabled under the Product Specific Configuration area for the device in Cisco Unified Communications Manager.

D. Check the DSCP for Video Calls value in the Cisco Unified Communications Manager clusterwide parameters QoS configuration. Answer: B QUESTION 106 You are experiencing bad video quality in calls, and you suspect packet loss in the network. Both the network switch and the collaboration endpoint network interfaces are set to autonegotiation, but the reported port speed and duplex settings do not match. Which action should you take? A. Set the collaboration endpoint to use a manual speed and duplex setting of 100 Mbps and full duplex, and keep the switch port setting as "auto."

B. Configure the switch port to use a manual speed and duplex setting of 100 Mbps and full duplex, and keep the collaboration endpoint setting as "auto." C. Keep both the switch port and collaboration endpoint settings as "auto," because this configuration is not responsible for the mismatched settings that were reported. D. Set both the collaboration endpoint and the switch port to use manual speed and duplex

settings of 100 Mbps and full duplex. Answer: DQUESTION 107 Refer to the exhibit. A new phone was installed and currently has issues registering with Cisco Unified Communications Manager. The phone received its IP address via DHCP, with options 3 and 150.

Phone URL Parameters	
URL Authentication	http://cucm.cisco.com:8080/ccmip/authenticate.jsp
URL Directories	http://cucm.cisco.com:8080/ccmip/xmldirectory.jsp
URL Idle	
URL Idle Time	
URL Messages	http://cucm.cisco.com:8080/ccmip/getservicesmenu.jsp
IP Phone Proxy Address	
URL Services	http://cucm.cisco.com:8080/ccmip/getservicesmenu.jsp

Which two actions would enable the phone to download its configuration automatically? (Choose two.)
 A. Change the phone URL parameters to IP addresses.
 B. Provide DHCP option 6 to the phone via DHCP.
 C. Reboot the phone.
 D. Change the VLAN that the phone is assigned to.
 E. Manually assign the phone settings.
 F. Provide DHCP option 15 to the phone via DHCP.
 Answer: ABQUESTION 108 Refer to the exhibit. You are trying to establish a multipoint call via a Cisco TelePresence SX20 that is registered to Cisco Unified Communications Manager. When you attempt to bring a third party into the call, you receive an error that the call cannot be completed. You confirm that your MCU resources are configured correctly on Cisco Unified Communications Manager and that other devices are able to establish multipoint calls. What is the cause of this issue?

Conference 1	
ActiveControl Mode	Auto
CallProtocolIPStack	IPv4
Encryption Mode	BestEffort
IncomingMultisite Call Mode	Allow
MaxReceiveCallRate	6000 (64 to 6000)
MaxTotalReceiveCallRate	10000 (64 to 10000)
MaxTotalTransmitCallRate	10000 (64 to 10000)
MaxTransmitCallRate	6000 (64 to 6000)
Multipoint Mode	MultiSite
AutoAnswer	
Delay	1 (0 to 50)
Mode	On
Mute	On
DefaultCall	
Protocol	Sip
Rate	6000 (64 to 6000)

A. The multipoint mode should be set to Multiway.
 B. The call protocol should be H.323.
 C. The MCU is at capacity.
 D. The multipoint mode should be set to Cisco Unified Communications Manager Resource Group.
 E. The multipoint mode should be set to None.
 Answer: DQUESTION 109 You are integrating a new video endpoint with Cisco VCS, but you find that the unit is failing to register. You assign extension 4000 to the device in the "vc.cisco.com" SIP domain, and you set its registration proxy to the IP address of 10.1.1.10 as the Cisco VCS. In order for the device to register via SIP, which format must you use when you set the SIP address of the device?
 A. 4000@vc.cisco.com
 B. 4000C. 4000@10.1.1.10D. 4000@cisco
 Answer: AQUESTION 110 Refer to the exhibit. According to the log diagram, what is the reason that the call ended?

```

May 17 22:02:45.043: //-1/xxxxxxxxxxxx/SIP/Msg/ccsipDisplayMsg:
Sent:
SIP/2.0 200 OK
Via: SIP/2.0/UDP 203.84.23.194:5060;branch=z9hG4bK1sdq7c20dgrhm310u2n1cdhpevej2.1
From: "John Ford" <sip:734561290@10.88.124.108;user=phone>;tag=433298465-1431900233
To: "882931316 882931316" <sip:882931316@cisco.com>;tag=15428C-15CF
Date: Sun, 17 May 2015 22:02:45 GMT
Call-ID: BW080353544180515-1909915730810.83.154.138
Server: Cisco-SIPGateway/IOS-15.4.3.M
CSeq: 838061991 BYE
Reason: Q.850;cause=16
Content-Length: 0

May 17 22:02:45.043: //-1/xxxxxxxxxxxx/SIP/Msg/ccsipDisplayMsg:
Sent:
BYE sip:882931316@10.0.10.2:5060 SIP/2.0
Via: SIP/2.0/UDP 10.0.10.8:5060;branch=z9hG4bKA2281
From: "John Ford" <sip:00734561290@cisco.com>;tag=15424C-1FB
To: <sip:882931316@10.0.10.2>;tag=1130116-90b05bfa-5f5d-4e05-8607-205de3dbe498-207
Date: Sun, 17 May 2015 22:02:26 GMT
Call-ID: 3B92C356-FC1711E4-8014AC6D-9A5A5580@10.0.10.8
User-Agent: Cisco-SIPGateway/IOS-15.4.3.M
Max-Forwards: 70
Timestamp: 1431900165
CSeq: 103 BYE
Reason: Q.850;cause=16
Content-Length: 0
    
```

A. The call was put on hold.B. The call was experiencing one-way audio.C. The call was completed successfully.D. The call was transferred.E. The call dropped out.
Answer: C
QUESTION 111
You recently configured a system for B2B SIP URI calls, and users confirmed that they could make calls. You are receiving multiple reports that inbound calls are failing and that users are not receiving calls to their URI. You confirm that all zones between expressways are active, and the trunk between Cisco Unified Communications Manager and Cisco VCS Expressway is active. You also see that the inbound call is sent from Cisco VCS Expressway C to Cisco Unified Communications Manager. Why are the calls failing?
A. The Cisco Unified Communications Manager FQDN was not set.
B. The cluster FQDN was not set in Enterprise Parameters.
C. The certificate is not valid.
D. The cluster FQDN was not set in Service Parameters.
E. The FQDN was not registered in DNS.
Answer: B!!!**RECOMMEND!!!**
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