



QUESTION & ANSWER

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Exam : 920-432

Title : Communication Server 1000 RIs.5.0-BCM RIs.4.0 Multi-site

Version : DEMO

- 1. What may be a cause of echo on a VoIP network?
- A. the VoIP products on the network
- B. the CODEC used on the VoIP network
- C. poor-quality headphones used on the VoIP network
- D. MCDN network features configured on the VoIP network

Answer: C

- 2. All systems on centralized voicemail must use what kind of dialing plan?
- A. CDP
- B. FNP
- C. ESN
- D. UDP

Answer: A

3. Click on the Exhibit button.

Based on the exhibit showing the H.323 settings, what is causing problems with the Communication

Server (CS) 1000 - Business Communications Manager (BCM) integration?

3 GW Settings										
Primary gatekeeper IP address	47.104.36.76									
Alternate gatekeeper IP address	0.0.0.0									
Primary Network Connect Server IP address	47.104.36.76									
Primary Network Connect Server Port number	16500	Range: 1024	to 65535							
Alternate Network Connect Server IP address	0.0.0.0									
Iternate Network Connect Server Port number	16500	Range: 1024	to 65535							
Primary Network Connect Server timeout	10	Range: 1 to	30							
phony Resources										
1 1	ual Type	Dip Sw	State	Devices	Low	High	ITC	otal E	Busy	_
	unks	N/A	N/A	Lines		1	60	N/A	N/A	
A static	App Sets	N/A	Enabled	Sets	- 0 -	N/A	N/A	8	Ö	
Disable Enable										
tails for Module: 0 Routing Table IP Trunk Settings H3 Telephony Settings Fallback to circuit-switched	323 Settings H3; Disabled	-	ers SIP Settings eeper digits	SIP Media Param	eters SIP	URI Map	_			
Routing Table IP Trunk Settings H3 Telephony Settings Fallback to circuit-switched Gateway protocol] Gateke		SIP Media Param	eters SIP	URI Map				
Routing Table IP Trunk Settings H Telephony Settings Fallback to circuit-switched Gateway protocol	Disabled _] Gateke	eeper digits 📄 er wildcard 📄	SIP Media Param	eters SIP	URI Map				
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Routing Table IP Trunk Settings H3 Telephony Settings Fallback to circuit-switched Gateway protocol Configuration Call signaling Gate Enable H245 tunnelling	Disabled] Gateke	eeper digits er wildcard Ca	Il signaling port	1720	URI Map				
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- A. An alternate gatekeeper has not been defined.
- B. The RAS port has not been defined on the BCM.
- C. The Primary Gatekeeper IP address does not match.
- D. The Primary Network Connect Server Port number does not match the Call signaling port.

Answer: C

4. When should you change the RTP over UDP port range configuration on a Business Communications Manager (BCM)?

A. when you are configuring a CS 1000 and BCM VoIP network integration

- B. when only absolutely necessary in instances where port configurations are causing conflicts
- C. when you have multiple BCM systems in a CS 1000 and BCM VoIP network integration
- D. when you are configuring a multisite BCM VoIP network integration

Answer: B

- 5. Click on the Exhibit button.
- The Message Wait Indication is not working on a Business Communications Manager remote site.

Based on the exhibit showing Centralized Voice Messaging, what is the issue?

Task Navigation Panel	Voice Messac	jing / Contact Cente	e.	
Configuration Administration				
Velcome	Centralized	Voice Messaging —		
System	Voice Mes	sage Centers		
Administrator Access	Center	External Number 🔺	Message Waiting Indication String	Message Waiting Cancellation String
Resources Telephony	1	2000	2001#	AN*0#
Clobal Settings		2000		
i Sets	2		AN*1#	AN*0#
🚉 Lines	3		AN*1#	AN*0#
Active Physical Lines	4		AN*1#	AN*0#
 Active VolP Lines Target Lines 	5		AN*1#	AN*0#
 Dialing Plan Ring Groups Call Security Hospitality Hunt Groups Call Detail Recording Data Services Applications Voice Messaging/Contact Ce 				
Mailbox				
	Mailbox	Number*: 2000		
	Mailt	oox Class: Regu	lar User 💌 <u>Class Details</u>	
		Language: Englis	sh(American) 😒	
	Locat	ion Name: CP-20)4 🔽	
Mailbox	∈File System \	/olume ID: 103		
Linl	ked to external	Directory: Not lin	ked Link	

A. The Link to the external directory is not active.

- B. The External Number does not contain the proper access code.
- C. The Message Waiting Cancellation String has not been configured.
- D. The Message Wait Indication String must use the mailbox number assigned by CallPilot.

Answer: D

6. Click on the Exhibit button.

The remote Business Communications Manager is having difficulty getting its voicemail to work.

Based on the exhibit showing Capabilities for DN 4000, what is the problem?

roperties Capabilities SVVCA C	all Group	Preferences	Button Programming Table	Button	Programming	User Speed Dial	
Handsfree	Auto	~	HF answ	erback			
Pickup group			DND or	n Busy		Allow redirect	
Page zone	1			Paging		Redirect ring	
Direct dial	1		Auto hold for incomin	g page		Receive short tones	
Intrusion protection level	None	*	Prior	ity call		Silent monitor supervisor	
			Au	to hold			

- A. Redirect ring is enabled.
- B. Priority call is not enabled.
- C. Allow redirect is not enabled.
- D. Pickup group has not been defined.

Answer: C

7. Click on the Exhibit button.

There is an error with the SIP settings of a Communication Server 1000 - Business Communications Manager integrated network.

Based on the information in the exhibit, what is the error?

P Server Settings	
Mode	Redirect 🖌
UDP transport enabled	
UDP port	5060
UDP maximum transmission unit (MTU)	1500
TCP transport enabled	
TCP port	5060
TCP maximum transmission unit (MTU)	1500

A. The TCP MTU is set too low.

- B. UDP transport is not enabled.
- C. UDP and TCP ports are the same port.
- D. The SIP server mode should be set to Direct.

Answer: B

8. How do jitter buffers help enhance Quality of Experience (QoE)?

A. Jitter buffers transmit voice frames at a fixed rate, enhancing the QoE.

B. Jitter buffers assign packets to the appropriate queue, allowing for the packets to be played immediately.

C. Jitter buffers can be configured to discard packets that are low quality thereby playing only high quality packets.

D. Jitter buffers hold arriving packets in a buffer long enough to allow the slowest packets to arrive, enabling the packets to be played in the correct sequence.

Answer: D

9. What happens if a Business Communications Manager (BCM) is not configured for fallback and the IP call quality is below threshold?

A. The IP call fails.

B. The IP call is rerouted to a PSTN line pool.

- C. The IP call proceeds with low-quality transmission.
- D. The IP call is held until call quality is above threshold.

Answer: C

- 10. Which components are the two main components within SIP architecture?
- A. the INVITE and the REGISTER
- B. the SIP user agent and the SIP network server
- C. the User Agent Client and the User Agent Server
- D. the SIP proxy server and the SIP redirect server

Answer: B