

Microsoft

70-333 Exam

Microsoft Deploying Enterprise Voice with Skype for Business Exam

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Questions & Answers

Version: 13.0

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HOTSPOT

You need to ensure that the Miami users can use a five-digit extension when they call the Los Angeles users.

What should you include in the normalization rule? To answer, configure the appropriate match pattern and translation pattern in the dialog box in the answer area.

Normalization rule	Match pattern	Translation pattern
nternal Los Angeles		
	^1(\d{5})\$	+1\$1
	^2(\d{4})\$	+1310\$1
	^3(\d{4})\$	+1310555\$1
	^3(\d{5})\$	+555\$1
		Answer:
Normalization rule	Match pattern	Answer: Translation pattern
	Match pattern	
		Translation pattern
	^1(\d{5})\$	Translation pattern
	^1(\d{5})\$ ^2(\d{4})\$	Translation pattern +1\$1 +1310\$1
Normalization rule nternal Los Angeles	^1(\d{5})\$	Translation pattern

Question: 2

DRAG DROP

You need to configure the unassigned numbers range to support the planned changes.

Which four actions should you perform?

To answer, move the four appropriate actions from the list of actions to the answer area and arrange them in the correct order.

Actions

Run the Import-CsAnnouncementFile Skype for Business Management Shell cmdlet.

Run the
New-CsAnnouncement
Skype for Business
Management Shell cmdlet.

Create an .mp3 file that contains the announcement.

Create an unassigned number range.

Run the Set-CsApplicationServer Skype for Business Management Shell cmdlet.

Create a .wav file that contains the announcement.

Answer Area

(2)







Answer:

Create a .wav file that contains the announcement.

Run the Import-CsAnnouncementFile Skype for Business Management Shell cmdlet.

Run the New-CsAnnouncement Skype for Business Management Shell cmdlet.

Create an unassigned number range.

Question: 3

You collect the following call statistics for users in Los Angeles:

- Peak call concurrency is 25 percent.
- Fifty percent of all placed calls are routed to the public switched telephone network (PSTN)
- Thirty percent of all placed calls are internal calls to users in the other offices.
- Twenty percent of all placed calls are conference calls.
- The average bandwidth for peer-to-peer calls is 65 kilobits per second (Kbps).
- The average bandwidth used for conference calls is 100 Kbps.

You need to allocate the minimum amount of bandwidth required on the wide area network (WAN) link for the planned implementation of the voice gateway in the Los Angeles office. All calls to the PSTN will be routed through the local voice gateway in the Los Angeles office.

How much bandwidth in megabits per second (Mbps) should you allocate?

- A. 7.20 Mbps
- B. 3.95 Mbps
- C. 10.00 Mbps
- D. 2.25 Mbps

Answer: B

Question: 4

DRAG DROP

You need to plan the subnets for Call Admission Control (CAC) and media bypass.

Which subnet or subnets should you use for each office?

To answer, drag the appropriate subnet or subnets to the correct office in the answer are

a. Each subnet may be used once, more than once, or not at all. Additionally, you may need to drag the split bar between panes or scroll to view content.

Subnets		Answer Area		
192.168.0.0/16		New York	Subnet	Subnet
192.168.8.0/21		Los Angeles	Subnet	
192.168.20.0/24		Miami	Subnet	
102.100.20.0724		Houston	Subnet	
192.168.30.0/23		10		
192.168.40.0/24	•			
131.107.100.0/24	•			
131.107.100.1/32	•			
131.107.100.3/32	•			
				Answer:

Answer Area

New York	192.168.8.0/21	131.107.100.3/32	
Los Angeles	192.168.30.0/23		
Miami	192.168.20.0/24		
Houston	192.168.40.0/24		

Question: 5

HOTSPOT

You need to design the Call Park solution.

Which Call Park design should you choose? To answer, select the appropriate range and application server in the dialog box in the answer area.

Range:		_
	0-99	
	100-199	
	200-9999	
	999-10000	

Application server:

	_
Server1.contoso.com	
Server2.contoso.com	
Server3.contoso.com	
Pool1.contoso.com	

Δ	nswer:	
	1134461.	

Range:

	-
0-99	1
100-199	
200-9999	
999-10000	

Application server:

	_
Server1.contoso.com	
Server2.contoso.com	
Server3.contoso.com	
Pool1.contoso.com	

Question: 6

HOTSPOT

You need to deploy the Skype for Business components that are necessary to support the planned changes for voice resiliency.

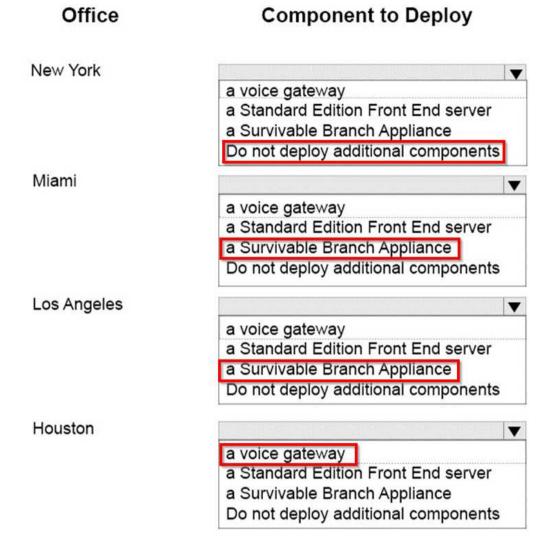
Which component should you deploy at each office? To answer, select the appropriate component for each office in the answer area.

Answer Area

Office

Component to Deploy New York a voice gateway a Standard Edition Front End server a Survivable Branch Appliance Do not deploy additional components Miami a voice gateway a Standard Edition Front End server a Survivable Branch Appliance Do not deploy additional components Los Angeles a voice gateway a Standard Edition Front End server a Survivable Branch Appliance Do not deploy additional components Houston • a voice gateway a Standard Edition Front End server a Survivable Branch Appliance Do not deploy additional components **Answer:**

Answer Area



Question: 7

DRAG DROP

You need to ensure that least cost routing is used in the Los Angeles and Houston offices for local calls after the gateway is deployed. The solution must meet the technical requirements. Which route should you assign to each public switched telephone network (PSTN) usage? To answer, drag the appropriate route to the correct PSTN usage in the answer are a. Each route may be used once, more than once, or not at all. Additionally, you may need to drag the split bar between panes or scroll to view content.

Routes

LocalHouston	
LocalLA	1
LDNY	
LDHouston	9
LDLA	0

Answer Area

LocalDialing	Route	Route
PrimaryLongDistanceDialing	Route	
BackupLongDistanceDialing	Route	

Routes

0 0

Answer Area

LocalDialing	LocalLA	LocalHouston
PrimaryLongDistanceDialing	LDLA	
BackupLongDistanceDialing	LDNY	

Question: 8

You need to create a rule to meet the technical requirements for international dialing from the SIP trunk in the New York office.

Which type of rule should you create?

- A. called number translation
- B. normalization
- C. dialing
- D. calling number translation

Answer: A

Explanation:

References:

 $\underline{\text{https://docs.microsoft.com/en-us/skypeforbusiness/plan-your-deployment/enterprise-voice-}} \underline{\text{solution/translation-rules}}$

Question: 9

You need to create the voice routes. How many voice routes should you create?

- A. 1
- B. 2
- C. 4
- D. 40
- E. 270

Answer: C

Question: 10

Remote users report that they fail to access their voice mail by using the Call Voice Mail option in the Skype tor Business client. Remote users can access voice mail by using the subscriber access number. Internal users do not report any issues accessing voice mail.

You need to enable voice mail access without manually dialing the subscriber number. Which two sets of ports should you open? Each correct answer presents part of the solution.

- A. TCP 443 and UDP 3478 on the external firewall
- B. TCP 5062 and 8057 on the internal firewallC. TCP 5062 and 8057 on the external firewall
- C. TCP 443 and UDP 3478 on the internal firewall
- D. UDP 50,000 to 59,999 on the internal firewall
- E. UDP 50,000 to 59,999 on the external firewall

Answer: AD

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