

Exam Questions 70-333

Deploying Enterprise Voice with Skype for Business 2015

<https://www.2passeasy.com/dumps/70-333/>



NEW QUESTION 1

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

NEW QUESTION 2

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	

Answer:

Explanation:

Range:

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

Application server:

	▼
Server1.contoso.com	
Server2.contoso.com	
Server3.contoso.com	
Pool1.contoso.com	

NEW QUESTION 3

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
Internal Los Angeles	<div style="border: 1px solid black; padding: 2px;">▼</div> <div style="border: 1px solid black; padding: 2px;">^1(\d{5})\$</div> <div style="border: 1px solid black; padding: 2px;">^2(\d{4})\$</div> <div style="border: 1px solid black; padding: 2px;">^3(\d{4})\$</div> <div style="border: 1px solid black; padding: 2px;">^3(\d{5})\$</div>	<div style="border: 1px solid black; padding: 2px;">▼</div> <div style="border: 1px solid black; padding: 2px;">+1\$1</div> <div style="border: 1px solid black; padding: 2px;">+1310\$1</div> <div style="border: 1px solid black; padding: 2px;">+1310555\$1</div> <div style="border: 1px solid black; padding: 2px;">+555\$1</div>

Answer:

Explanation:

Normalization rule	Match pattern	Translation pattern
Internal Los Angeles	<div style="border: 1px solid black; padding: 2px;">▼</div> <div style="border: 1px solid black; padding: 2px;">^1(\d{5})\$</div> <div style="border: 1px dashed green; padding: 2px;">^2(\d{4})\$</div> <div style="border: 1px dashed green; padding: 2px;">^3(\d{4})\$</div> <div style="border: 1px solid black; padding: 2px;">^3(\d{5})\$</div>	<div style="border: 1px solid black; padding: 2px;">▼</div> <div style="border: 1px solid black; padding: 2px;">+1\$1</div> <div style="border: 1px solid black; padding: 2px;">+1310\$1</div> <div style="border: 1px dashed green; padding: 2px;">+1310555\$1</div> <div style="border: 1px solid black; padding: 2px;">+555\$1</div>

NEW QUESTION 4

You need to create the DHCP options required for Skype for Business Phone Edition devices.

Which three actions should you perform? To answer, move the three appropriate actions from the list of actions to the answer area and arrange them in the correct order.

- Actions**
- From a DHCP server, run **dhcpcfgscript.bat**.
 - Copy **Dhcpconfigscript.bat** to a Skype for Business Server 2015 server.
 - From a Skype for Business Server 2015 server, run **dhcputil.exe**.
 - Copy **Dhcpconfigscript.bat** to a DHCP server.
 - From a DHCP server, run **dhcputil.exe**.
 - From a Skype for Business Server 2015 server, run **dhcpcfgscript.bat**.

Answer Area

>
<

Answer:

Explanation:

Actions	Answer Area
From a DHCP server, run <code>dhcpcfgscript.bat</code> .	Copy <code>Dhcpcfgscript.bat</code> to a DHCP server.
Copy <code>Dhcpcfgscript.bat</code> to a Skype for Business Server 2015 server.	From a DHCP server, run <code>dhcputil.exe</code> .
From a Skype for Business Server 2015 server, run <code>dhcputil.exe</code> .	From a DHCP server, run <code>dhcpcfgscript.bat</code> .
Copy <code>Dhcpcfgscript.bat</code> to a DHCP server.	
From a DHCP server, run <code>dhcputil.exe</code> .	
From a Skype for Business Server 2015 server, run <code>dhcpcfgscript.bat</code> .	

NEW QUESTION 5

Remote users report that they fail to access their voice mail by using the Call Voice Mail option in the Skype for 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 firewall
- C. TCP 5062 and 8057 on the external firewall
- D. TCP 443 and UDP 3478 on the internal firewall
- E. UDP 50,000 to 59,999 on the internal firewall
- F. UDP 50,000 to 59,999 on the external firewall

Answer: AD

NEW QUESTION 6

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:

<https://docs.microsoft.com/en-us/skypeforbusiness/plan-your-deployment/enterprise-voice-solution/translation-r>

NEW QUESTION 7

You need to configure the agent groups to meet the response group requirements. How should you configure each agent group? To answer, select the appropriate options for each group in the answer area.

Answer Area

Group name	Participation Policy	Routing Method
Group1	<input type="text" value="Formal"/> <input type="text" value="Informal"/>	<input type="text" value="Attendant"/> <input type="text" value="Longest idle"/> <input type="text" value="Parallel"/> <input type="text" value="Round robin"/> <input type="text" value="Serial"/>
Group2	<input type="text" value="Formal"/> <input type="text" value="Informal"/>	<input type="text" value="Attendant"/> <input type="text" value="Longest idle"/> <input type="text" value="Parallel"/> <input type="text" value="Round robin"/> <input type="text" value="Serial"/>

Answer:

Explanation: References:
<https://docs.microsoft.com/en-us/skypeforbusiness/deploy/deploy-enterprise-voice/create-or-modify-an-agent-gr>

NEW QUESTION 8

You need to allocate bandwidth required for application sharing.

How much bandwidth in kilobits per second (Kbps) should you allocate? To answer, select the appropriate amount of bandwidth for each network connection in the answer area.

From the client computer of the conference presenter to a Front End pool:	<input type="text"/> <input type="text" value="500 Kbps"/> <input type="text" value="50,000 Kbps"/> <input type="text" value="100,000 Kbps"/>
From a Front End pool to an Edge Server:	<input type="text"/> <input type="text" value="500 Kbps"/> <input type="text" value="50,000 Kbps"/> <input type="text" value="100,000 Kbps"/>
From an Edge Server to the Internet:	<input type="text"/> <input type="text" value="500 Kbps"/> <input type="text" value="50,000 Kbps"/> <input type="text" value="100,000 Kbps"/>

Answer:

Explanation:

From the client computer of the conference presenter to a Front End pool:

	▼
500 Kbps	
50,000 Kbps	
100,000 Kbps	

From a Front End pool to an Edge Server:

	▼
500 Kbps	
50,000 Kbps	
100,000 Kbps	

From an Edge Server to the Internet:

	▼
500 Kbps	
50,000 Kbps	
100,000 Kbps	

NEW QUESTION 9

You need to replace the existing call group pickup solution in Denmark with a Skype for Business solution. You need to prepare the Skype for Business Server 2015 environment to provide call group pickup functionality.

Which three Skype for Business Management Shell cmdlets should you run in sequence? To answer, move the appropriate Skype for Business Management Shell cmdlets from the list of cmdlets to the answer area and arrange them in the correct order.

Possible actions

- Enable-CsTopology
- New-CsTrustedApplication
- New-CsCallParkOrbit
- New-CsTrustedApplicationPool
- New-CsTrustedApplicationEndpoint

Actions to perform

⬅

➡

⬆

⬇

Answer:

Explanation: "You need to prepare the Skype for Business Server 2015 environment to provide call group pickup functionality."

We're just preparing the environment. We're not creating the call park yet. After running the three cmdlets above, you can then run the New-CsCallParkOrbit cmdlet.

NEW QUESTION 10

Problem Statements

Two users who recently joined the company's finance department report that they are not able to use their Aries phones even when they tether the phone to their computer.

You need to implement voice resiliency for international calls.

Which three actions should you perform in sequence? To answer, move the appropriate actions from the list of actions to the answer area and arrange them in the correct order.

Possible actions	Actions to perform	
Create a new public switched telephone network (PSTN) trunk between the London Mediation Server and the Berlin gateway.	1	
Publish the topology.	2	
Create a new public switched telephone network (PSTN) usage and voice route.	3	
Add a new normalization rule.	⤵	⤴
	⤴	⤵
Create a new dial plan.		
Create a new public switched telephone network (PSTN) trunk between the London Mediation Server and the Moscow gateway.		

Answer:

Explanation:

Possible actions	Actions to perform	
Create a new public switched telephone network (PSTN) trunk between the London Mediation Server and the Berlin gateway.	1	Create a new public switched telephone network (PSTN) trunk between the London Mediation Server and the Moscow gateway.
Publish the topology.	2	Create a new public switched telephone network (PSTN) usage and voice route.
Create a new public switched telephone network (PSTN) usage and voice route.	3	Publish the topology.
Add a new normalization rule.	⤵	⤴
	⤴	⤵
Create a new dial plan.		
Create a new public switched telephone network (PSTN) trunk between the London Mediation Server and the Moscow gateway.		

NEW QUESTION 10

Problem Statements

Two users who recently joined the company's finance department report that they are not able to use their Aries phones even when they tether the phone to their computer.

You need to ensure that the new finance department users can sign in to the Aries phone without tethering the phone to their computer.

Which two actions should you perform?

NOTE: Each correct answer presents part of the solution.

- A. Assign a PIN for the users by running the Set-CsClientPin Skype for Business Management Shell command.
- B. Inform users about the new PIN.
- C. Tell users to log on to the device with their Active Directory Domain Services (AD DS) credentials and then set a Device Lock PIN.
- D. Tell users to sign in to the phones with their Active Directory Domain Services (AD DS) credentials.
- E. Tell users to change their PIN by using the dial-in conferencing webpage.

Answer: AD

NEW QUESTION 14

You are troubleshooting audio call issues for users in Copenhagen. You need to resolve the issues. Which three actions should you perform? Each correct answer presents part of the solution.

- A. Create and configure a new network region and assign Copenhagen to this region.
- B. Create an inter-site policy which includes Copenhagen and Berlin and apply the existing BandwidthPolicyProfile "LOW".
- C. Create an inter-site policy which includes Copenhagen and Moscow and specify the existing BandwidthPolicyProfile "LOW".
- D. Add Copenhagen to the Berlin network region.
- E. Create and configure a new network region and assign Berlin and Copenhagen to this region.

Answer: BCE

NEW QUESTION 19

You need to replace the existing automatic call distribution (ACD) helpdesk with Skype for Business Server 2015 solution.

Which three actions should you perform in sequence to recreate the ACD in Skype for Business Server 2015? To answer, move the appropriate actions from the list of actions to the answer area and arrange them in the correct order.

Possible Actions		Actions to Perform
Create a managed workflow.		
Restart the responsegroup service.		
Create and configure the agent group.		
Create and configure queue.	➤	⬆
Create an unmanaged workflow.	⬅	⬇

Answer:

Explanation:

Possible Actions		Actions to Perform
Create a managed workflow.		Create and configure the agent group.
Restart the responsegroup service.		Create and configure queue.
Create and configure the agent group.		Create a managed workflow.
Create and configure queue.	➤	⬆
Create an unmanaged workflow.	⬅	⬇

NEW QUESTION 24

You need to enable all users for voice mail.

Which three configuration steps are required to configure the environment for Exchange Unified Messaging? Each correct answer presents part of the solution.

- A. Run the OcsUMUtil executable on the Skype for Business server.
- B. Run ExchUCUtil.ps1 on the Exchange server.
- C. Run the New-CsHostingProvider Skype for Business Management Shell cmdlet.
- D. Run the New-CsHostedVoicemailPolicy Skype for Business Management Shell cmdlet.
- E. Apply the hosted voice mail policy to all users.

Answer: ABD

NEW QUESTION 27

You need to design the Quality of Service (QoS) configuration for the solution.

For each solution option, which port range should you use? To answer, select the appropriate port range from each list in the answer area.

Answer Area

Options	Port Range
Audio	<ul style="list-style-type: none"> 1 – 1024 1024 – 65535 50020 – 50040 66020 – 66040
Video	<ul style="list-style-type: none"> 1 – 1024 1024 – 65535 50030 – 50050 58000 – 58020
Application sharing	<ul style="list-style-type: none"> 1 – 1024 1024 – 65535 42000 – 42020 58000 – 58020
File transfer	<ul style="list-style-type: none"> 1 – 1024 1024 – 65535 42040 – 42040 50025 – 50045

Answer:

Explanation: All services can use ports 1024-65535. However, for QoS, we need to use port ranges that do not overlap.

NEW QUESTION 32

You need to determine the number of lines that are required for the new public switched telephone network (PSTN) connections.

How many lines are required for each location? To answer, drag the appropriate number of line to the correct location. Each number 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.

Answer Area

- 1390
- 1280
- 1340
- 1230
- 2570
- 2670



Location	Number of Lines
Melbourne	
Singapore	

Answer:

Explanation:

Answer Area

- 1390
- 1280
- 1340
- 1230
- 2570
- 2670



Location	Number of Lines
Melbourne	1340
Singapore	1230

NEW QUESTION 35

You need to manage the bandwidth consumption for the wide area network (WAN) links. Which functionality should you manage?

- A. Inter-site audio
- B. PowerPoint upload
- C. Intra-site audio
- D. Desktop sharing

Answer: A

NEW QUESTION 38

You need to modify the relevant portion of the Persona configuration in the Skype for Business Bandwidth Calculator to estimate the planned volume of Skype for Business traffic for the Contoso office in Vancouver.

In the Skype for Business Bandwidth Calculator, which options should you select? To answer, select the appropriate value in each drop-down list in the Skype for Business Calculator. (Note: The drop-down lists are in the Persona Definitions section.)

Answer:

Explanation:

NEW QUESTION 39

You are developing a design solution to eliminate the possibility of wide area network (WAN) over-subscription by Skype for Business traffic. You need to assign tasks to each office to prevent over subscription and to ensure quality.

Which tasks should you assign to each office? To answer, drag the appropriate task or tasks to the correct office or offices in the answer area. Each task may be used once, more than once, or not at all. You may need to drag the split bar between panels or scroll to view content.

Tasks

- Configure Call Admission Control (CAC).
- Deploy Quality of Service (QoS).
- Deploy sFlow on all network switches.
- Create a throttling policy.
- Implement Multiprotocol Label Switching (MPLS).

Answer Area

Vancouver office

- Task
- Task
- Task

Seattle office

- Task
- Task

Answer:

Explanation:

Tasks

- Configure Call Admission Control (CAC).
- Deploy Quality of Service (QoS).
- Deploy sFlow on all network switches.
- Create a throttling policy.
- Implement Multiprotocol Label Switching MPLS).

Answer Area

Vancouver office

- Implement Multiprotocol Label Switching MPLS).
- Deploy Quality of Service (QoS).
- Configure Call Admission Control (CAC).

Seattle office

- Implement Multiprotocol Label Switching MPLS).
- Deploy Quality of Service (QoS).

Topic 7, Mix Questions set

NEW QUESTION 42

Your company has a Skype for Business Server 2015 infrastructure that contains a single site. You subscribe to a third-party SIP provider that provides E911 services to the company. During testing, you discover that location information is not being passed to the SIP provider. You have the following trunk configuration:

```

Identity : Global
OutboundTranslationRulesList : {}
SipResponseCodeTranslationRulesList : {}
OutboundCallingNumberTranslationRulesList : {}
PstnUsages : {}
Description :
ConcentratedTropology : True
EnableBypass : False
EnableMobileTrunkSupport : False
EnableReferSupport : False
EnableSessionTimer : False
EnableSignalBoost : False
MaxEarlyDialogs : 20
RemovePlusFromUri : False
RTCPActiveCalls : True
RTCPActiveCallsOnHold : True
SRMode : Required
EnablePIDFLOSupport : False
EnableRTPLatching : False
EnableOnlineVoice : False
ForwardCallHistory : False
Enable3pccRefer : False
ForwardPAI : False
EnableFastFailoverTimer : True
EnableLocationRestriction : False
NetworkSiteID :
    
```

You need to ensure that the location information is sent to the third-party SIP provider. What property should you set to True?

- A. EnableReferSupport
- B. EnablePIDFLOSupport
- C. ForwardPAI
- D. EnableBypass

Answer: B

Explanation: References:

<https://docs.microsoft.com/en-us/powershell/module/skype/set-cstrunkconfiguration?view=skype-ps>

NEW QUESTION 43

Fabrikam has recently deployed Skype for Business Server for all modalities including Enterprise Voice which they manage from central location in Sydney. A recent change in their work from home policies requires users to work from their homes at least 3 days a week. Fabrikam's IT help desk has 10 members in their tier 1 TeleHelp group dedicated to receive incoming support calls of the internal employees. Fabrikam wants to deploy a Skype for Business solution for the TeleHelp group with minimum administrative effort.

TeleHelp group members want to get a notification of all the incoming calls to any member of the TeleHelp group. Each member should be able to answer the call as per their availability.

You need to deploy Skype for Business solution for TeleHelp group.

What two actions should you perform? Each Answer presents part of the solution.

- A. Configure Group Call Pickup for TeleHelp group.
- B. For every TeleHelp group member configure Forward My Calls to Team-Call Group consisting of all 10members.
- C. For every TeleHelp group member configure Forward My Calls to My Delegates and add each member as a delegate to each other.
- D. For every TeleHelp group member configure Simultaneously Ring to My Delegates and add each member as a delegate to each other.
- E. Configure a response group and add TeleHelp group members as formal agents.
- F. Configure a response group and add TeleHelp group members as informal agents.

Answer: AD

NEW QUESTION 48

You have a Skype for Business Server 2015 infrastructure. The infrastructure contains one central site. The infrastructure contains a Standard Edition server that has the Mediation Server role installed and is located in the central site.

You configure a trunk to an IP-PBX which is qualified for integration with Skype for Business. You have 300 users who are enabled for Enterprise Voice.

You need to provide public switched telephone network (PSTN) services to the Enterprise Voice users if the IP-PBX fails.

What should you do?

- A. Install a Survivable Branch server that connects directly to the PST
- B. Configure a new trunk and add the trunk to the existing rout
- C. Add the route to the PSTN usage and add the PSTN usage to the voice policy.
- D. Install a PSTN gateway that connects directly to the PST
- E. Configure a new trunk to the new PSTN gatewa
- F. Create a new route and a new PSTN usag
- G. Add the route to the PSTN usage and add the PSTN usage to the voice policy.
- H. Install an additional IP-PBX that connects directly to the PST
- I. Configure the existing trunk to use both IP-PBXs in resiliency mode.
- J. Install a new trunk to a certified SIP trunk provide
- K. Add the trunk to the existing PSTN usag
- L. Add the PSTN usage to the voice policy.

Answer: B

NEW QUESTION 49

Your company has an on-premises Skype for Business Server 2015 infrastructure. The company purchases a subscription to Microsoft Office 365.

You plan to implement a hybrid solution between the on-premises infrastructure and Office 365.

You implement Active Directory Federation Services (AD FS) and you configure directory synchronization between your environment and Office 365.

You need to configure federation to support the hybrid configuration.

What Skype for Business Management Shell command should you run? To answer, select the appropriate options in the answer area.

Answer Area

New-CsHostingProvider -Identity "Skype for Business Online" -ProxyFqdn

-Enabled \$True -EnabledSharedAddressSpace

-HostsOCSUsers

Answer:

Explanation: References:

<https://technet.microsoft.com/en-us/library/jj205126.aspx>

NEW QUESTION 50

You support a Skype for Business Server 2015 environment that consists of the following:

A single Skype for Business Server 2015 Standard Edition pool that is located in a main office.
Four branch offices that are connected by a 9 megabits per second (Mbps) dedicated wide area network (WAN) link, each of which has 300 users.
You need to be able to support 75 concurrent calls.
Users in the branch offices report poor quality audio during conferences. You perform the following troubleshooting steps:
Implemented Quality of Service (QoS) and Call Admission Control (CAC). In CAC, defined the main office and each branch office as a site In CAC, associated subnets to the relevant sites. Provisioned policy-based QoS for media traffic.
Poor audio quality persists.
You run the Skype for Business Management Shell Get-CsNetworkSite cmdlet. The relevant portion of the results are shown below:

```
C:\...\Get-CsNetworkSite

Identity : DataCenter1
NetworkSiteID : DataCenter1
Description :
NetworkRegionID : Corporate
BypassID : bec84b6e-1874-456d-89eb-b57dac78a2be
BWPolicyProfileID :
LocationPolicy :

Identity : DataCenter2
NetworkSiteID : DataCenter2
Description :
NetworkRegionID : Corporate
BypassID : 2b18787d-699e-44d0-9ee6-d57e4edda60d
BWPolicyProfileID :
LocationPolicy :

Identity : Redmond
NetworkSiteID : Redmond
Description :
NetworkRegionID : Corporate
BypassID : 36b4aadc-4e96-440c-aaf8-00bc22607493
BWPolicyProfileID :
LocationPolicy :

Identity : London
NetworkSiteID : London
Description :
NetworkRegionID : Corporate
BypassID : 8c207f05-4deb-42c6-bb2a-78e39fe84ec0
BWPolicyProfileID :

LocationPolicy :C:\...\
```

You need to improve audio quality. What should you do?

- A. Create a region route to route network traffic between the CAC sites.
- B. Create a bandwidth policy profile with an Audio Session Limit of 40 Kbp
- C. Assign the profile to the central site.
- D. Create a bandwidth policy profile with an Audio Session Limit of 40 Kbps and an Audio Limit of 3 Mbps. Assign the profile to each branch site.
- E. Create a bandwidth policy profile with an Audio Session Limit of 40 Kbps and an Audio Limit of 1 Mbps. Assign the profile to each branch site.

Answer: C

Explanation: References: <https://technet.microsoft.com/en-us/library/gg412785.aspx>

NEW QUESTION 51

Your network environment includes Skype for Business Server 2015 with Enterprise Voice enabled. The current voice configuration settings include the following:

```
Dial Plan: North America
Normalization rules:
Internal: ^(\d{4})$ => +14034998$1
Local: ^1?((403)\d{7})$ => +1$1
NA Long Distance: ^(1\d{11})$ => +$11
Voice policy: NorthAmerica
Calling features enabled:
    call forwarding, delegation, call transfer, simultaneous
    ring, team call, PSTN reroute, Calling features disabled:
    call park, bandwidth policy override, malicious call
    tracing
PSTN usage record:
NorthAmerica-PSTN, associated route PSTN =^\+ (?! (61))

Routes:
NorthAmerica-Local-Route (^\+1(403) \d{7})$
NorthAmerica-LongDistance = ^\+1\d{10}$

Trunk Configuration: NorthAmerica
Encryption support level = Required
Enabled: Media bypass, Centralized media processing, refer
support
```

When users attempt to call to an outside number, they receive the following error: "User is not authorized." You need to ensure that users are able to dial outside numbers.

Which two actions should you perform? Each correct answer presents part of the solution.

- A. Modify public switched telephone network (PSTN) usage record NorthAmerica-PSTN and its associated route PSTN for voice policy from ^\+(?!(61)) to ^\+1\d{10}\$.
- B. Modify route NorthAmerica-LongDistance- from ^\+1\d{10}\$ to ^\+1\d{11}\$.
- C. Under Dial Plan, modify normalization rule Local from ^1?((403)\d{7})\$ => +1\$1 to ^(3\d{7})\$ =>+403\$1.
- D. Under Dial Plan, modify normalization rule NA Long Distance from ^(1\d{11})\$ => +\$11 to^(1\d{10})\$ => +\$1.

Answer: AD

NEW QUESTION 54

A company has offices in Montreal and New York. The company has an existing phone system. The company plans to migrate to Skype for Business 2015. Users dial number 3 followed by a four-digit extension to reach users in the other office. The company would like to keep the same user experience after they migrate to Skype for Business 2015.

You to create a dial plan that allows users to dial between the two offices.

Which pattern match and translation rule should you use? To answer, select the appropriate pattern match and translation rule in the answer area.

Answer Area

Pattern match:

▼

^(3\d{4})\$1
^3(\d{3})\$1
^3(\d{4})\$
^4(3\d{4})\$

Translation rule:

▼

+1720553\$1
+17205554\$1
+1720555\$
+1720555\$1

Answer:

Explanation:

Answer Area

Pattern match:

- ^(3\d{4})\$1
- ^3(\d{3})\$1
- ^3(\d{4})\$
- ^4(3\d{4})\$

Translation rule:

- +1720553\$1
- +17205554\$1
- +1720555\$
- +1720555\$1

NEW QUESTION 59

You have a Skype for Business Server 2015 infrastructure that contains a voice named Policy1. Policy1 contains local and national public switched telephone network (PSTN) usages.

You have a dial plan named DialPlan1. DialPlan1 is configured as shown in the following table.

Pattern	Translation rule
^(d{7})\$	+3120\$1
^0(6\d{8})\$	+31\$1
^0((([123457]\d{4,8}88\d{7}))\$	+31\$1
^0?0([2-9]\d*)\$	+\$1

The Skype for Business Server infrastructure contains four routes. The routes are configured as shown in the following table.

Route name	Pattern to match	PSTN usage	Trunk name
PBXUsers1	+3120123	Local	LegacyPBX1
PBXUser2	+3188(?!(123))	Local	LegacyPBX2
National1	+31(?!(88))	National	SIPTrunk1
National2	+31	National	SIPTrunk2

You create a test user who has the following configurations:

Line URL: +31201234567

Voice policy: Policy1

Dial plan: DialPlan1

You need to assign the trunks that will be used when the following numbers are dialed by the test user:

0881234567

0884567890

Which trunk should you use for each number? To answer, select the appropriate trunk for each number in the answer area.

Answer Area

0881234567

- LegacyPBX1
- LegacyPBX2
- SIPTrunk1
- SIPTrunk2

0884567890

- LegacyPBX1
- LegacyPBX2
- SIPTrunk1
- SIPTrunk2

Answer:

Explanation:

Answer Area

0881234567

- LegacyPBX1
- LegacyPBX2
- SIPTrunk1
- SIPTrunk2

0884567890

- LegacyPBX1
- LegacyPBX2
- SIPTrunk1
- SIPTrunk2

NEW QUESTION 63

You support a Skype for Business environment. You deploy System Center Operations Manager. You implement Enterprise Voice. Some users use a desktop phone to place and receive public switched telephone network (PSTN) calls with people outside of Skype for Business. Some of these users report intermittent connectivity issues with incoming and outgoing calls. You need to periodically confirm that users are able to place and receive calls with people outside of Skype for Business. Which two actions should you perform? Each correct answer presents part of the solution.

- A. Create a watcher node and configure it for an extended, non-default synthetic transaction.
- B. In the synthetic transaction, run the Test-CsP2PAV Skype for Business Management Shell cmdlet.
- C. In the synthetic transaction, run the Test-CsPstnPeerToPeerCall Skype for Business Management Shell cmdlet.
- D. Create a watcher node and configure it for a default synthetic transaction.

Answer: AC

Explanation: References:
[https://technet.microsoft.com/en-us/library/jj204943\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/jj204943(v=ocs.15).aspx)

NEW QUESTION 64

A company has a Skype for Business Server 2015 environment. All users are enabled for Enterprise Voice. The infrastructure has a single route that applies to all users of (.*).

You need to prevent the users from calling external numbers that start with 1900 or 1976.
 What are two possible ways to achieve this goal? Each correct answer presents a complete solution.

- A. Create an unassigned number range of +19000000000 to +19009999999 and +19760000000 to+19769999999.
- B. Modify the dial plan rules to exclude the normalization of dialed numbers that start with 1900 or 1976.
- C. Create a voice route that has the following match pattern: ^\+(!1900 | 1976))
- D. Create a voice route that has the following match pattern: ^\+(?!1900 | 1976))

Answer: AD

NEW QUESTION 65

You have a Skype for Business Server 2015 infrastructure. The infrastructure contains one central site. The infrastructure contains a Standard Edition server that has the Mediation Server role installed and is located in the central site.
 You configure a trunk to an IP-PBX. The IP-PBX is qualified for integration with Skype for Business. You have 300 users who are enabled for Enterprise Voice.
 You need to provide public switched telephone network (PSTN) services to the Enterprise Voice users if the IPPBX fails.
 Which two actions will achieve the goal? Each Answer presents a complete solution.

- A. Install a Survivable Branch appliance that connects directly to the PST
- B. Configure a new trunk and add thetrunk to the existing rout
- C. Add the route to the PSTN usage and add the PSTN usage to the voice policy.
- D. Install an additional IP-PBX that connects directly to the PST
- E. Configure the existing trunk to use both IP-PBXs in resiliency mode.
- F. Install a PSTN gateway that connects directly to the PST
- G. Configure a new trunk to the new PSTNgatewa
- H. Create a new route and a new PSTN usag
- I. Add the route to the PSTN usage and add the PSTN usage to the voice policy.
- J. Install a new trunk to a certified SIP trunk provide
- K. Add the trunk to the existing PSTN usag
- L. Add thePSTN usage to the voice policy.

Answer: CD

Explanation: References:
<https://technet.microsoft.com/en-us/library/jj721915.aspx>
<http://www.gecko-studio.co.uk/sip-trunk-lync-sir/>

NEW QUESTION 67

You have a Skype for Business Server 2015 infrastructure. You have a Virtual Desktop Infrastructure (VDI) that uses Hyper-V.
 You use VDI templates based on Windows 8.1. All thin clients run Windows 8.1. Each thin client has a Skype for Business-qualified headset.
 You need to prepare all of the thin clients and all of the VDI templates so that users can place calls by using the headset.
 Which components should you install on the thin clients and the VDI templates? To answer, drag the appropriate components to the correct location in the answer area. Each component 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.
 Each correct answer is worth one point.

Components	Answer Area
Skype for Business Phone Edition	Thin clients: Component
Skype for Business	Thin clients: Component
Skype for Business VDI plug-in	VDI templates: Component
Windows 8 RDS client	

Answer:

Explanation: References:
<https://itbasedtelco.wordpress.com/2013/06/17/lync-deployment-using-virtual-desktops-vdi/> [https://technet.microsoft.com/en-us/library/jj205304\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/jj205304(v=ocs.15).aspx)

NEW QUESTION 69

A company plans to integrate their Skype for Business 2015 on-premises environment with Skype for Business Online.
 You have the following requirements:
 ? Skype for Business Online users must be able to find on-premises users when they perform a search in the Skype for Business client.
 ? Skype for Business on-premises users and Skype for Business Online users must be able to communicate by using instant messaging.
 Which two components should you include in the recommendation? Each correct answer presents part of the solution.

- A. a Reverse Proxy
- B. The Microsoft Online Services Directory Synchronization tool
- C. a Session Border Controller
- D. a Skype for Business Server 2015 partner application

E. a Skype for Business Server 2015 Edge pool

Answer: BE

NEW QUESTION 71

You have deployed Skype for Business Server 2015 in three locations, Boston, New York and Chicago. Users from the Boston location are reporting a delay in the audio stream during conferences and that the audio drops frequently. Users in the other two locations do not report the same issues. You need to recommend a solution to resolve the audio issues for the users in the Boston office. Which three possible recommendations should you include in the solution? Each correct answer presents a complete solution.

- A. Call Admission Control (CAC)
- B. user services
- C. Quality of Service (QoS)
- D. a location policy
- E. a conferencing policy

Answer: ACE

NEW QUESTION 74

You have a Skype for Business Server 2015 Infrastructure. The infrastructure contains two sites and is configured for Enterprise Voice. The sites are located in Chicago and Toronto. Each site has a distinct voice policy, public switched telephone network (PSTN) usage, voice route, and trunk transaction rule as shown in the following table:

Site name	Dial plan name	Voice policy name	PSTN usage name	Voice route name	Trunk name
Toronto	Toronto-DP	Toronto-VP	Toronto-Usage	Toronto-AllCalls	Toronto-PSTN
Chicago	Chicago-DP	Chicago-VP	Chicago-Usage	Chicago-AllCalls	Chicago-PSTN

You need to ensure that all outbound voice calls to the PSTN from the users in the Toronto site are routed through the trunk at the Toronto site. Calls must use the Mediation Server at the Chicago site if the Mediation Server at the Toronto site is unavailable. What should you do?

- A. Add the Chicago-PSTN trunk to the Toronto-AllCalls voice route.
- B. Add the Chicago-AllCalls voice route to the Toronto-Usage PSTN usage record.
- C. Add the Chicago-Usage PSTN usage record to the Toronto-PSTN trunk.
- D. Add the Chicago-Usage PSTN usage record to the Toronto-VP voice policy.

Answer: A

NEW QUESTION 79

You support a customer whose network environment includes Skype for Business Server 2015 servers. All of the company servers are located in the company datacenter. All users have Skype for Business client installed on their local computers. You deploy a virtual desktop infrastructure (VDI) environment to a pilot group of users. You install the Lync VDI plug-in and plug Skype for Business compatible devices to the users' computers. Users report that Skype for Business fails intermittently when pairing with the Lync VDI plug-in. You need to ensure reliable access to Skype for Business Server 2015 for this pilot group of users. What should you do?

- A. For all VDI users, set the value of the EnableMediaRedirection property to \$FALSE.
- B. Configure the remote desktop client settings so that audio plays on the local computer and remote recording is disabled.
- C. Run Windows 10 or higher on the local computers.
- D. Uninstall Skype for Business from the local users' computers.

Answer: D

Explanation: References: [https://technet.microsoft.com/en-us/library/jj204713\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/jj204713(v=ocs.15).aspx)

NEW QUESTION 82

You have a Skype for Business Server 2015 infrastructure. You configure the Skype for Business Server to require that each user has a location. You configure a location for each subnet. Users report that when they sign in, they are prompted for their location information. You need to prevent the users from being prompted for their location information when they sign in to Skype for Business on the internal network. Which Skype for Business Management Shell cmdlet should you run?

- A. Publish-CsLISConfiguration
- B. Publish-CsTopology
- C. Set-CsNetworkSubnet
- D. Export-CsLISconfiguration

Answer: A

Explanation: References: <https://technet.microsoft.com/en-us/library/gg398364.aspx> References: <https://technet.microsoft.com/en-us/library/gg398364.aspx>

NEW QUESTION 83

A company plans to implement Skype for Business Server 2015. Before they start deploying the solution, they want to ensure that the network can handle the media traffic that the solution produces.

You must perform a network traffic simulation on their network.

You need to ensure that the correct network metrics are collected during the network traffic simulation. Which three network metrics should you collect? Each Answer presents part of the solution.

- A. packet loss
- B. one-way network delay
- C. average jitter
- D. forward error correction overhead
- E. number of media streams

Answer: ABC

Explanation: References:

http://i.crn.com/custom/UC_Strategies_Report.pdf Page 4

NEW QUESTION 84

You are performing a network readiness assessment for Skype for Business deployment for a customer's organization.

The customer's organization includes five sites that are connected by site-to-site virtual private network (VPN) links. Each site includes from 20 to 100 client computers.

You need to collect network traffic metrics.

Which three network traffic metrics should you collect? Each correct answer presents part of the solution.

- A. average packet loss
- B. average jitter
- C. TCP/IP port statistics
- D. one-way network delay
- E. minimum jitter
- F. peer-to-peer (P2P) available bandwidth

Answer: ABD

NEW QUESTION 85

You create a topology that includes Enterprise Voice. You configure the system to play a message when unassigned numbers are dialed.

You need to add the voice announcements.

Which two file formats can you use? Each Answer presents a complete solution.

- A. WAVE Audio file (*.WAV).
- B. Windows Media Audio File (*.WMA)
- C. MPEG-4 Audio File (*.M4A)
- D. MPEG-2 Audio File (*.MPA)
- E. MP3 Audio File (*.MP3)

Answer: AB

Explanation: References: [https://technet.microsoft.com/en-us/library/gg398649\(v=ocs.14\).aspx](https://technet.microsoft.com/en-us/library/gg398649(v=ocs.14).aspx)

NEW QUESTION 90

Your company has a Skype for Business Server 2015 infrastructure.

You need to ensure that users can answer calls placed to other users by using the Group Call Pickup feature.

Which Skype for Business Management Shell cmdlet should you run?

- A. New-CsDialPlan
- B. New-CsRoutingConfiguration
- C. New-CsCallParkOrbit
- D. New-CsVoiceRoutingPolicy

Answer: C

Explanation: References: <https://technet.microsoft.com/en-us/library/jj945627.aspx>

NEW QUESTION 92

Your company has a single office located in the United States.

The phone numbers in the United States have a country code of 1 followed by 10 digits.

You have a Skype for Business Server 2015 infrastructure configured for Enterprise Voice. All normalization rules ensure that dialed numbers conform to E.164 standards.

You need to create a called number translation rule that meets the following requirements:

Which pattern match and translation rule should you use? To answer, select the appropriate pattern match and translation rule in the answer area.

Answer Area

Pattern match:

- 1(\d{10})
- ^1(\d{10})\$
- \+1(\d{10})
- ^1(\d{10})\$
- ^1(\d\d\d)(\d{7})\$

Translation rule:

- \$1
- \$2\$3
- +\$1
- +\$1\$10

Answer:

Explanation:

Answer Area

Pattern match:

- 1(\d{10})
- ^1(\d{10})\$
- \+1(\d{10})
- ^1(\d{10})\$
- ^1(\d\d\d)(\d{7})\$

Translation rule:

- \$1
- \$2\$3
- +\$1
- +\$1\$10

NEW QUESTION 95

You support Skype for Business Server 2015 servers in your customer's network. The network includes two branches connected with an 8 megabits per second (Mbps) network circuit. The circuit has 1 Mbps of dedicated bandwidth for Skype for Business voice traffic. The network is required to handle 50 peer-to-peer (P2P) calls at a minimum.

Users report that calls between the branches are being dropped, and voice quality is degraded. Skype for Business audio codecs:

Codec	Typical stream bandwidth kilobits per second (kbps)
RTAudio Wideband	39.8
RTAudio Narrowband	29.3
SILK Wideband	44.3
G.722	46.1
Siren	25.5

You need to calculate appropriate bandwidth for the P2P call volume required. What is the minimum dedicated bandwidth that should you allocate?

- A. 1.275 Mbps
- B. 1.465 Mbps
- C. 1.990 Mbps
- D. 2.215 Mbps
- E. 2.305 Mbps

Answer: B

Explanation: References:
[https://technet.microsoft.com/en-us/library/jj688118\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/jj688118(v=ocs.15).aspx)

NEW QUESTION 96

You have a Microsoft Exchange Server 2013 organization and a Skype for Business Server 2015 infrastructure. You need to design a Unified Messaging (UM) solution to meet the following requirements: Support a peak concurrency of 500 calls. Connect to two Skype for Business Server 2015 pools.

What is the minimum number of mailbox servers that are required?

- A. 1
- B. 2
- C. 3
- D. 5

Answer: C

Explanation: References:
<http://technet.microsoft.com/en-us/library/aa997637>

NEW QUESTION 99

You have a Skype for Business infrastructure that contains four servers. The servers are configured as shown in the following table.

Server name	Role	Average CPU utilization
Server1	Front End Server	60 percent
Server2	Front End Server	60 percent
Server3	SQL Server Back End	50 percent
Server4	SQL Server Back End	50 percent

The infrastructure contains a hardware load balancer named HLB1. The company has an IP-PBX system. The IP-PBX supports media bypass. The IP-PBX has a Session Initiation Protocol (SIP) trunk configured. Each user has a desktop phone that connects to the IP-PBX. Each computer has a Skype for Business-qualified headset.

The company plans to implement Enterprise Voice and to use existing SIP trunks.

You need to recommend a Mediation Server deployment that meets the following requirements: What are two possible ways to achieve the goal? Each correct answer presents a complete solution.

- A. Install the Mediation Server role on Server3 and Server4.
- B. Install the Mediation Server role on Server1 and Server2.
- C. Create a Mediation Server pool that contains Server2 and Server4.
- D. Create a Mediation Server pool that contains HLB1.
- E. Create a Mediation Server pool that contains two new servers.

Answer: BE

NEW QUESTION 104

A company has a main office and five branch offices located in different cities. You configure Skype for Business Enterprise Voice for the main office.

You deploy a Survivable Branch Appliance (SBA) at a branch office and connect it to a public switched telephone network (PSTN) by using a T1 trunk line. You configure the SBA voice gateway to use fully qualified domain names (FQDN).

You test the branch isolation by disconnecting the wide area network (WAN) from the branch office and make an outbound call. The call was successful. Next, you make an inbound call to the branch office and the call fails.

You perform the following troubleshooting steps:

Trace the IncomingAndOutgoingCalls setting.

Use Snooper to analyze logs.

Review the syslog file from the SBA voice gateway.

You discover that messages are sent to the Skype for Business Mediation Server pool, but not received. You reconnect the WAN from the branch office and make an inbound call. The call is successful.

You need to identify the configuration error that causes inbound calls to fail. What should you check?

- A. the Dynamic Host Configuration Protocol (DHCP) options for the branch DHCP server
- B. the route settings for the SBA voice gateway
- C. the status of the DNS services for the branch office domain controller
- D. the DNS settings for the SBA voice gateway

Answer: B

NEW QUESTION 109

A company has several offices that are located in the office building in Seattle. The company also has employees visiting from other countries. You implement Skype for Business Server Enterprise Voice.

The company has a phone at a security desk. You need to implement call routing for emergency calls. The solution must meet the following requirements:

If a user dials 112, the call must be routed to a phone extension of 0911.

If a user dials 0911, the call must be routed to the phone at the security desk.

The phone at the security desk must be used to forward calls to the emergency number in the Seattle offices only.

What technologies should you implement? To answer, drag the appropriate technology to the correct requirement in the answer area. Each technology 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.

Technologies

- a conferencing policy
- a Contact object
- an external user access policy
- a normalization rule
- a voice policy
- an unassigned number

Answer Area

If a user dials 112, the call must be routed to a phone extension of 0911.

If a user dials 0911, the call must be routed to the phone at the security desk.

The phone at the security desk must be used to forward calls to the emergency number in the Seattle office only.

- Technology
- Technology
- Technology

Answer:

Explanation:

Technologies

- a conferencing policy
- a Contact object
- an external user access policy
- a normalization rule
- a voice policy
- an unassigned number

Answer Area

If a user dials 112, the call must be routed to a phone extension of 0911.

If a user dials 0911, the call must be routed to the phone at the security desk.

The phone at the security desk must be used to forward calls to the emergency number in the Seattle office only.

- a normalization rule
- a normalization rule
- a voice policy

NEW QUESTION 114

Your company has a Skype for Business Server 2015 infrastructure that contains two topology sites named London and Seattle. Each site contains an Enterprise Edition Front End pool and an Edge Server. The external DNS records are configured as shown in the following table.

DNS record name	Record type	Destination
sip.contoso.com	Host (A)	Edge Server in the Seattle site
sip2.contoso.com	Host (A)	Edge Server in the London site
rp.contoso.com	Host (A)	Reverse proxy in the Seattle site
rp2.contoso.com	Host (A)	Reverse proxy in the London site
_sip._tls.contoso.com	Service location (SRV)	sip.contoso.com

All users use the Skype for Business client. Some of the users use the Skype for Business client from their personal computer. You need to ensure that the users in the London site use the Edge Server in the London site for authentication and the users in the Seattle site use the Edge Server in the Seattle site for authentication. The solution must minimize administrative effort. What should you do?

- A. Create an A record for lyncdiscover.contoso.com that points to the server proxy in the Seattle site.
- B. Create an SRV record named _sipexternal._tls.contoso.com that points sip2.contoso.com.
- C. Create an SRV record named _sip._tls.contoso.com that points sip2.contoso.com.
- D. Configure a Group Policy object (GPO) for the London site that configures the Skype for Business client to use an external server name of sip2.contoso.com.

Answer: C

Explanation: References: [https://technet.microsoft.com/en-us/library/gg398758\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/gg398758(v=ocs.15).aspx)

NEW QUESTION 119

Your company has a Skype for Business Server 2015 infrastructure. All users are enabled for Enterprise Voice. All network components support the Link Layer Discovery Protocol (LLDP). A client computer named Computer1 runs Windows 10. Computer1 connects to a wireless network connection and a wired network connection. You need to locate Computer1.

Which element should you use to locate Computer1?

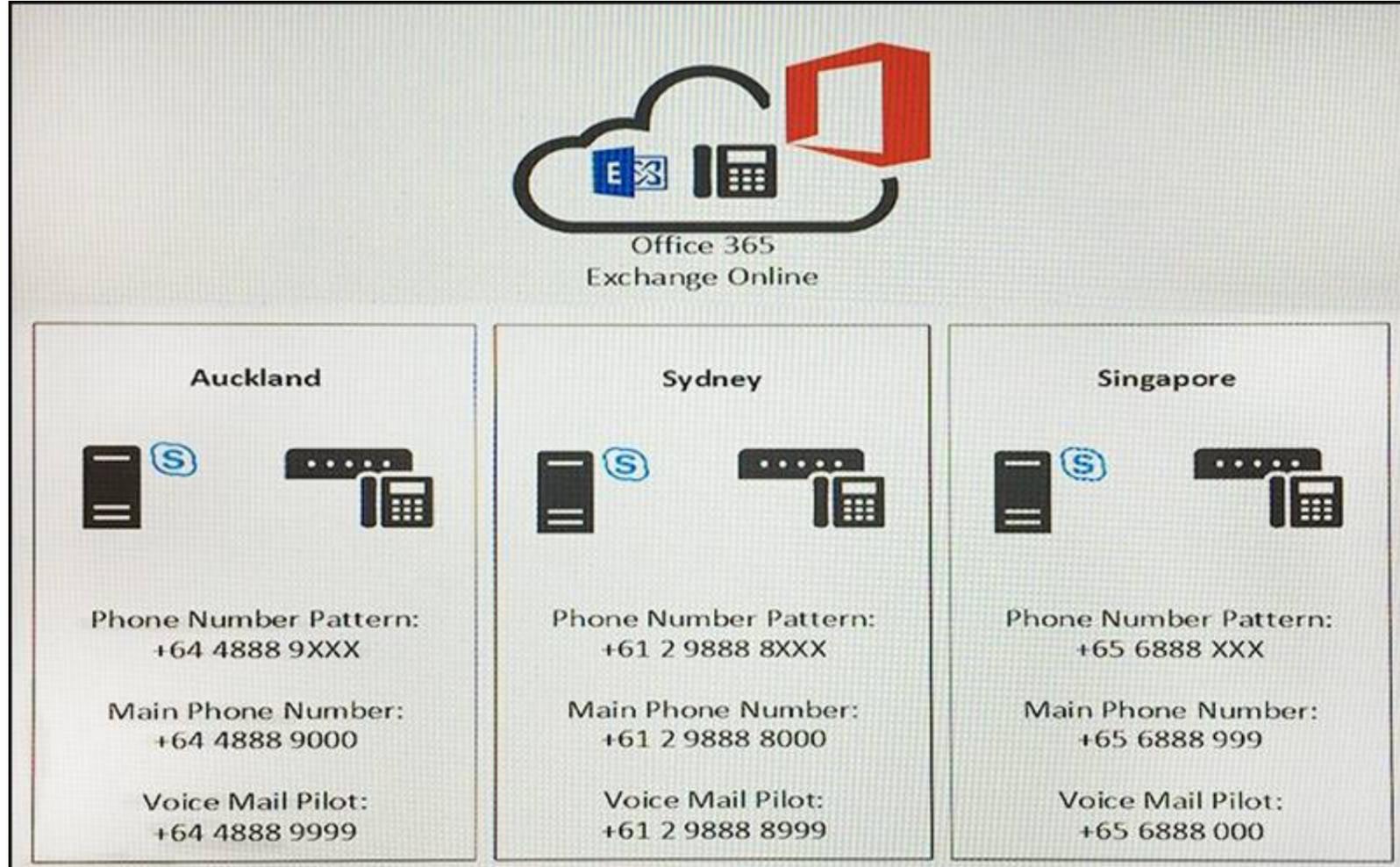
- A. the Basic Service Set Identification (BSSID) address of the wireless access point (WAP)
- B. the IP subnet of the wireless network
- C. the LLDP switch port
- D. the MAC addresses of the wired network connection
- E. the LLDP switch chassis ID

Answer: C

NEW QUESTION 120

WingTip Toys is in the process of completing a rollout of Skype for Business Server 2015 Enterprise Voice. They want to provide voice mail capabilities to all users.

The information below was provided to facilitate the design of Exchange Online Unified Messaging (UM) implementation.



Use the drop-down menus to select the answer choice that answers each question. Each correct answer is worth one point.

Answer Area

To design Outlook Voice Access for Sydney, what is the most appropriate design option that will accomplish the goal with minimum change to Exchange Online configuration?

- Create a Sydney Exchange Unified Messaging Dial Plan, assign +61298889000 as subscriber access number
- Create a Sydney Exchange Unified Messaging Dial Plan, assign +61298889999 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +61298889000 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +61298889999 as subscriber access number

Choose the appropriate design option for Singapore while minimizing the change to Exchange Online configuration?

- Create a Singapore Exchange Unified Messaging Dial Plan, assign +656888000 as subscriber access number
- Create a Singapore Exchange Unified Messaging Dial Plan, assign +656888999 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +656888000 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +656888999 as subscriber access number

Answer:

Explanation:

Answer Area

To design Outlook Voice Access for Sydney, what is the most appropriate design option that will accomplish the goal with minimum change to Exchange Online configuration?

- ▼
- Create a Sydney Exchange Unified Messaging Dial Plan, assign +61298889000 as subscriber access number
- Create a Sydney Exchange Unified Messaging Dial Plan, assign +61298889999 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +61298889000 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +61298889999 as subscriber access number

Choose the appropriate design option for Singapore while minimizing the change to Exchange Online configuration?

- ▼
- Create a Singapore Exchange Unified Messaging Dial Plan, assign +656888000 as subscriber access number
- Create a Singapore Exchange Unified Messaging Dial Plan, assign +656888999 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +656888000 as subscriber access number
- Modify the existing Exchange Unified Messaging Dial Plan, assign +656888999 as subscriber access number

NEW QUESTION 121

You have a Skype for Business Server 2015 infrastructure and a Microsoft Exchange Server 2013 organization. Unified Messaging (UM) integration is enabled. You need to prevent the use of call answering rules. Which Skype for Business Management Shell cmdlet should you run?

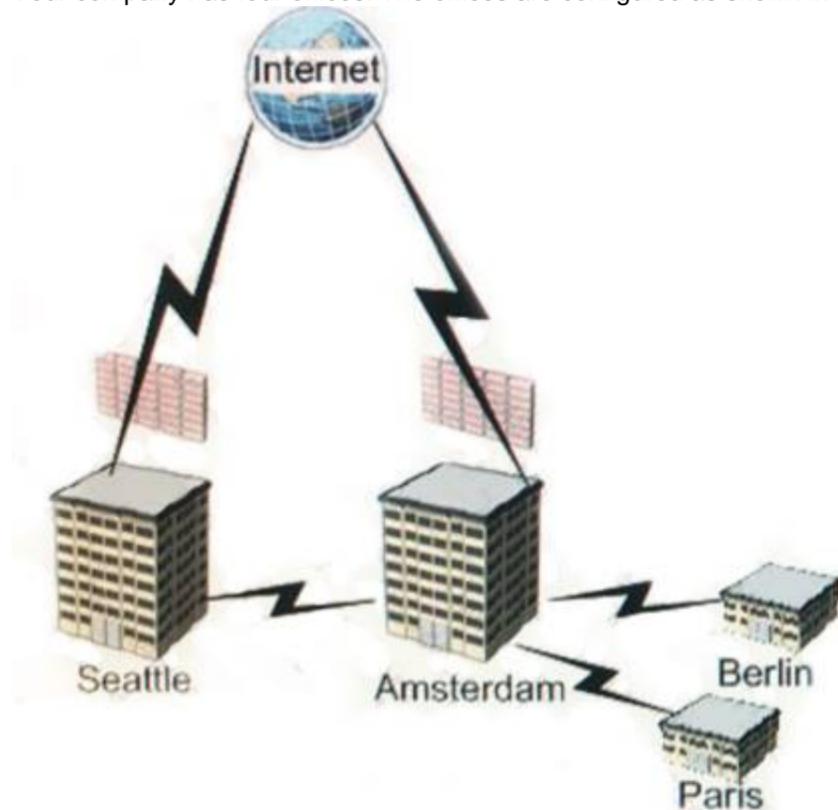
- A. Set-CSVoicePolicy
- B. Remove-UMCallAnsweringRule
- C. Set-CSDialPlan
- D. Set-UMDialPlan

Answer: D

Explanation: References: [https://technet.microsoft.com/en-us/library/bb124408\(v=exchg.160\).aspx](https://technet.microsoft.com/en-us/library/bb124408(v=exchg.160).aspx)

NEW QUESTION 122

Your company has four offices. The offices are configured as shown in the exhibit. Click the Exhibit button.



You have a Skype for Business Server 2015 infrastructure that has the following configuration:

Two regions named EMEA and US. The Paris, Berlin, and Amsterdam offices are assigned to EMEA. The Seattle office is assigned to US.

A bandwidth policy named Policy1 that has a limit of 1,024 kilobits per second (Kbps) for audio and video.

A bandwidth policy named Policy2 that has a limit of 2,048 Kbps for audio and video. You need to configure Call Admission Control (CAC) to meet the following requirements:

Limit the Skype for Business server-related wide area network (WAN) traffic between the Seattle and Amsterdam offices to 2,048 Kbps.

In the Paris office, limit the Skype for Business server-related WAN traffic for audio to 1,024 Kbps and limit the Skype for Business server-related WAN traffic for video to 1,024 Kbps.

Which three actions should you perform? Each correct answer presents part of the solution.

- A. Assign Policy2 to the network in the Amsterdam office.
- B. Assign Policy1 to the network in the Paris office.
- C. Assign Policy2 to the network in the Seattle office.
- D. Create a region link and assign Policy2 to the region link.
- E. Modify region EMEA and region US to enable an Audio/Video alternate path.
- F. Create a region route.

Answer: BDF

Explanation: References:

<https://docs.microsoft.com/en-us/skypeforbusiness/deploy/deploy-enterprise-voice/create-network-interregional->

NEW QUESTION 125

Your company has two offices. The offices are located in Amsterdam and Seattle. The company has the phone number ranges shown in the following table.

Office	Total phone number range
Amsterdam	+31 20 1234 5 00 to +31 20 1234 5 99
Seattle	+1 425 555 1200 to +1 425 555 1299

The range used for each office is shown in the following table.

Office	Used phone number range
Amsterdam	+31 20 1234 5 01 to +31 20 1234 5 55
Seattle	+1 425 555 1201 to +1 425 555 1240

You plan to implement Skype for Business Server 2015 and include an unassigned number range for each office.

You need to identify which unassigned number ranges must be included. The solution must use the minimum amount of administrative effort when users are added to or removed from the infrastructure.

Which number range should you use for each unassigned number range? To answer, select the appropriate number range for each unassigned number range in the answer area. Each correct answer is worth one point.

Answer Area

Amsterdam unassigned number range

- +31201234500 - +31201234599
- +31201234501 - +31201234555
- +31201234556 - +31201234599

Seattle unassigned number range

- +14255551200 - +14255551299
- +14255551201 - +14255551240
- +14255551241 - +14255551299

Answer:

Explanation: References:

<https://docs.microsoft.com/en-us/skypeforbusiness/deploy/deploy-enterprise-voice/create-or-modify-an-unassign>

NEW QUESTION 128

Contoso recently completed a migration to Exchange Online and would like to enable Unified Messaging integration with their on-premises deployment of Skype for Business Server 2015.

Outlook Web App Instant Messaging integration is already working, and Exchange Online users are able to schedule Skype meetings from Outlook Web App.

You need to identify the first step to be performed on both Exchange Online and Skype for Business Server 2015 to enable Unified Messaging capabilities for all users.

In the table below, identify the action that must be taken in each service.

Note: Make only one selection in each column. Each correct answer is worth one point.

Action	Exchange Online	Skype for Business Server 2015
Configure hosted voice mail policy	<input type="radio"/>	<input type="radio"/>
Enable users for hosted voice mail	<input type="radio"/>	<input type="radio"/>
Configure contact object for hosted Exchange Unified Messaging	<input type="radio"/>	<input type="radio"/>
Configure Exchange Unified Messaging dial plan	<input type="radio"/>	<input type="radio"/>
Configure Exchange Unified Messaging policies	<input type="radio"/>	<input type="radio"/>

Answer:

Explanation:

Action	Exchange Online	Skype for Business Server 2015
Configure hosted voice mail policy	<input type="radio"/>	<input checked="" type="radio"/>
Enable users for hosted voice mail	<input checked="" type="radio"/>	<input type="radio"/>
Configure contact object for hosted Exchange Unified Messaging	<input type="radio"/>	<input checked="" type="radio"/>
Configure Exchange Unified Messaging dial plan	<input checked="" type="radio"/>	<input type="radio"/>
Configure Exchange Unified Messaging policies	<input checked="" type="radio"/>	<input type="radio"/>

NEW QUESTION 129

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