

70-333 Dumps

Deploying Enterprise Voice with Skype for Business 2015

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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 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 area. 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
192.168.30.0/23	Houston Subnet
192.168.40.0/24	
131.107.100.0/24	
131.107.100.1/32	
131.107.100.3/32	

Answer:

Explanation:

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

NEW QUESTION 3

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 4

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

	▼
^1(\d{5})\$	
^2(\d{4})\$	
^3(\d{4})\$	
^3(\d{5})\$	

	▼
+1\$1	
+1310\$1	
+1310555\$1	
+555\$1	

Answer:

Explanation:

Normalization rule

Match pattern

Translation pattern

Internal Los Angeles

	▼
^1(\d{5})\$	
^2(\d{4})\$	
^3(\d{4})\$	
^3(\d{5})\$	

	▼
+1\$1	
+1310\$1	
+1310555\$1	
+555\$1	

NEW QUESTION 5

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

Answer Area

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:

Explanation:

Actions	Answer Area
Run the Import-CsAnnouncementFile Skype for Business Management Shell cmdlet.	Create a .wav file that contains the announcement.
Run the New-CsAnnouncement Skype for Business Management Shell cmdlet.	Run the Import-CsAnnouncementFile Skype for Business Management Shell cmdlet.
Create an .mp3 file that contains the announcement.	Run the New-CsAnnouncement Skype for Business Management Shell cmdlet.
Create an unassigned number range.	Create an unassigned number range.
Run the Set-CsApplicationServer Skype for Business Management Shell cmdlet.	
Create a .wav file that contains the announcement.	

NEW QUESTION 6

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 7

You plan to modify the firewall in the Montreal office that connects to the Internet.

You need to implement the inbound firewall rules to meet the security requirements for inbound A/V connections.

Which rules should you implement? To answer, configure the appropriate firewall rules in the answer area.

Answer Area

Rule	Source IP	Source Port	Destination IP	Destination Port
Rule1	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>
Rule2	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>

Answer:

Explanation:

Answer Area

Rule	Source IP	Source Port	Destination IP	Destination Port
Rule1	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>
Rule2	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>	<div>10.0.1.1</div> <div>131.107.1.1</div> <div>131.107.1.2</div> <div>ANY</div>	<div>TCP 443</div> <div>TCP 50,000 to 59,999</div> <div>UDP 3478</div> <div>ANY</div>

NEW QUESTION 8

You need to create the Unified Messaging (UM) dial plans and access numbers for subscriber access.

How many UM dial plans and access numbers should you create? To answer, select the appropriate number of UM dial plans and access numbers in the answer area.

Answer Area

Number of UM dial plans:

4

2

1

0

▲▼

Number of access numbers:

0

1

2

4

▲▼

Answer:

Explanation:

Answer Area

Number of UM dial plans:

4
2
1
0

▲▼

Number of access numbers:

0

▲▼

1
2
4

NEW QUESTION 9

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:

	▼
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	

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 10

You plan to implement the response group for the support staff and to create the agent groups shown in the following table.

Group name	Group members
Group1	All IT support users
Group2	All on call managers

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

Formal
Informal

Attendant
Longest idle
Parallel
Round robin
Serial

Group2

Formal
Informal

Attendant
Longest idle
Parallel
Round robin
Serial

Answer:

Explanation:

Answer Area

Group Name

Participation Policy

Routing Method

Group1

Formal
Informal

Attendant
Longest idle

Group2

Formal
Informal

Parallel
Round robin
Serial

Attendant
Longest idle

Parallel

Round robin

Serial

Topic 3, Humongous Insurance

Humongous Insurance currently employs 7,500 people and is based in the United States.

The company has 4,000 users in its main office in Chicago. It also has two branch offices, one in New York with 2,000 users and the other in Los Angeles with 1,500 users. The company has 30 executives located across the offices.

The company recently migrated from Lync Server 2010 to Skype for Business Server 2015.

The company plans to increase the use of audio and video (A/V) conferencing to reduce travel costs. Employees have been asked to use A/V, telephony, and instant messaging (IM) conferencing solutions whenever possible for communication with clients and users in the branch offices.

Humongous Insurance is federated with Trey Research Corporation, which uses a third-party IM solution. Users at both companies need to communicate with each other by using their respective IM client.

The company plans to continuously analyze the network to identify: potential network delay points

availability of adequate bandwidth to meet current requirements and anticipated growth

The company plans to continuously analyze the Skype for Business Server 2015 servers for performance metrics.

Planned Changes

Humongous is in discussion to acquire the Southbridge Video company. Southbridge Video has 2,000 users in two sites.

Employees of Humongous Insurance use a variety of mobile devices, including Windows Phones and iOS phones. All iOS phone users must be notified of events, such as new or missed IM invitations and voice mail messages, even when they are using the corporate wireless network.

All public switched telephone network (PSTN) traffic will use the G.711 codec. All voice will use the RTAudio Wideband codec.

All peer-to-peer (P2P) video sessions will support H.264 and RTVideo codecs. Audio bandwidth between the Los Angeles and New York offices is limited.

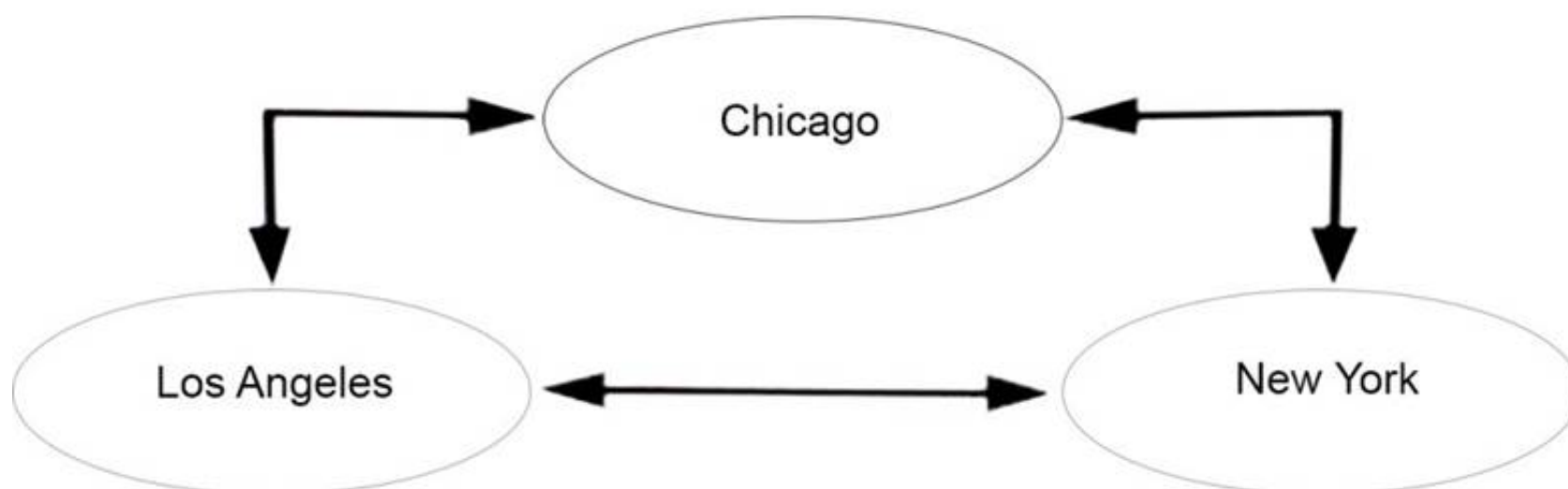
Challenges

All users report poor audio and video quality. Calls are dropped or suffer from jitter and lag. These issues exist when users are using the Skype for Business client or Skype for Business Phone Edition.

Humongous Insurance users currently are unable to IM with external clients that use different IM programs. Currently there is no network oversight for Skype for Business communications.

The company has no plans to increase the bandwidth at any sites. Humongous Insurance does not have a defined remote user access policy. Existing Environment

The three offices are connected to one another as shown in the following network diagram:



Additional information about the existing environment is shown in the Supporting Information section: Overview of data centers

Detail of the Chicago data center

Overview of audio and video codec bandwidths

Employees of Humongous Insurance use various operating systems, including Windows 7 Enterprise, and Windows 8. All users are enabled for Enterprise Voice, and many have Skype for Business Phone Edition.

The concurrent communication modalities usage of the three offices is shown in the following table:

Role	Concurrent usage percentage Chicago office	Concurrent usage percentage New York office	Concurrent usage percentage Los Angeles office
PSTN	5%	5%	5%
Voice	10%	10%	15%
H.264/RTVideo	5%	5%	5%

Humongous Insurance expects to see the number of video users increase by as much as 10 percent. The company expects to see the number of voice users

increase by as much as 15 percent.

All users at Humongous Insurance will be Skype for Business enabled. The network team configures Quality of Service (QoS) as follows:

Media type	Starting port	Port count
Audio	49152	4,174
Video	57501	4,017

The Skype for Business Server 2015 media configuration matches the port settings used on the network. The company uses synthetic transactions to perform call and voice tests.

Many of the employees of Humongous Insurance work from home and sign in securely to the corporate environment in order to use Skype for Business.

Business Requirements

Humongous Insurance would like to review the existing network topology and wants to ensure that network capacity will meet current Skype for Business Server 2015 infrastructure capacity requirements for audio and video usage, as well as meet the anticipated increases.

All existing issues relating to the poor quality of Skype for Business Server 2015 calls and conferencing need to be addressed. Company policy allows for all employees to have the option to use video conferencing. Employees will be limited to VGA video quality. Company executives will not have these limitations.

Quality of Service (QoS) ports must be unique for audio and video traffic.

The company wants to confirm that the current network bandwidth allocated for the network is sufficient for all current and anticipated growth in Skype for Business Server 2015 traffic and would like to guarantee high-quality Skype for Business Server 2015 calls and video conferencing.

Humongous Insurance wants to ensure that the volume of Skype for Business Server 2015 traffic does not exceed 50 percent of WAN bandwidth.

The recommended design solution needs to eliminate the possibility of WAN oversubscription by Skype for Business Server 2015 traffic.

Humongous Insurance wants to use only dedicated computers to monitor the network and perform synthetic transactions to limit the load on the computers.

The proposed Skype for Business Server 2015 solution needs to provide scalability to accommodate the company's future acquisition of Trey Research.

The proposed Skype for Business Server 2015 solution should be readily available for users in all three sites.

Technical Requirements

The company wants to allow only authorized employees to connect remotely.

All remote users will use a secure method when remotely connecting to the network and the Skype for Business environment.

The number of open and listening ports must be kept to a minimum.

All maintenance and tests of network policies must be done outside of normal business hours. Bandwidth between sites cannot be increased.

The quality of voice and video traffic must be increased. The number of dropped calls must be reduced.

The corporate information technology (IT) department must be able to push updates and scan users' computers when they are connected remotely.

All push notifications must be able to be sent mobile clients on all supported cell phone manufactures. Many users do not use Windows mobile phones.

IM users must be able to IM with users who are using other IM programs.

The Skype for Business Server 2015 Bandwidth Calculator must be used to analyze Skype for Business traffic. It is recommended to deploy QoS for end-to-end support.

Mobile client users are supported only with Lync 2013 Mobile clients.

Supporting Information

Overview of data centers

Role	Chicago data center	New York data center	Los Angeles data center
Skype for Business 2015 Front End Servers	CHI-FE1 CHI-FE2 CHI-FE3	NYC-FE1 NYC-FE2	LA-FE1 LA-FE2
Skype for Business 2015 Mediation Server	CHI-ME1 CHI-ME2	NYC-ME1 NYC-ME2	LA-ME1 LA-ME2

Detail of the Chicago data center

Role	Name
Active Directory Domain Controllers	CHI-DC1 CHI-DC2
Exchange 2013 Servers	CHI-EXCH1 CHI-EXCH2
Skype for Business 2015 Back-End Server	CHI-BE1
Skype for Business 2015 Edge Servers	CHI-SfBEdge1 CHI-SfBEdge2
Windows Server 2008	CHI-Edge1
Windows Server 2012	CHI-Edge2

Overview of audio and video codec bandwidths

Audio Codec	Bandwidth audio payload, IP header, UDP, RTP, and STRP (Kbps)
G.711	92.0
G.722	95.6
G.722 Stereo	159.6
Siren	47.6
RTAudio Narrowband	39.8
RTAudio Wideband	57

Video Codec	Maximum bandwidth (Kbps)
H.264/RTVideo (320x240) 4:3	350
H.264/RTVideo (640x480) 4:3	800
H.264/RTVideo (1280x720) 16:9	2,500
H.264 (1920x288) 20:3	2,000
H.264 (1920x1080) 16:9	4,000

NEW QUESTION 10

You need to configure the Skype for Business Server 2015 environment to meet the technical requirements for push notifications to mobile users. Which two actions should you perform? Each correct answer presents part of the solution.

- A. Open port 23457 on the Enterprise Wi-Fi network.
- B. Open port 5223 on the Enterprise Wi-Fi network.
- C. Configure NAME redirection over HTTPS for the acquired domain.
- D. Configure a web publishing rule for port 80 on the reverse proxy for the acquired domain.

Answer: BC

NEW QUESTION 14

You need to ensure that there is enough bandwidth available to meet the current maximum bandwidth requirements for peer-to-peer (P2P) video calls. How much bandwidth in kilobits per second (Kbps) should you allocate?

- A. 303, 750 Kbps
- B. 399, 750 Kbps
- C. 813, 750 Kbps
- D. 1,503,750 Kbps

Answer: A

NEW QUESTION 19

You need to deploy and enable Quality of Service (QoS) on all workstations to meet the planned changes and performance goals. Which three actions should you perform? Each correct answer presents part of the solution.

- A. Configure the Group Policy object (GPO) that configures QoS and set the value to Enforced.
- B. Create Group Policy objects (GPOs) for Skype for Business audio and video and configure them to use the same port ranges and Differentiated Services Code point (DSCP) values for QoS.
- C. Upgrade all Windows XP and Windows Vista workstations to Windows 7 or above.
- D. Create Group Policy objects (GPOs) for Skype for Business audio and video and configure them to use different port ranges and Differentiated Services Code Point (DSCP) value for QoS.
- E. Apply Service Pack 3 to all Windows XP workstations and Service Pack 2 to all Windows Vista workstations.

Answer: ACD

NEW QUESTION 23

You need to recommend a solution to test users' ability to schedule, join, and conduct an audio/video (A/V) conference. What should you do?

- A. Run the Test-CsAVConference Skype for Business Management Shell cmdlet and then analyze the trace call flow in Snooper.
- B. Run the Test-CsMcxConference Skype for Business Management Shell cmdlet and save the results. View the results by using the Skype for Business Server 2015 Logging Tool.
- C. Run the Test-CsUcwaConference Skype for Business Management Shell cmdlet.
- D. Run the Test-CsAudioConferencingProvider Skype for Business Management Shell cmdlet and save the result
- E. View the results by using the Central Logging Service.

Answer: A

Explanation: References:

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

NEW QUESTION 26

You need to recommend a solution that will improve voice quality for Skype for Business clients. What should you recommend?

- A. Change the Voice802lp value to 46.
- B. Ensure that VoiceDiffservtag values are set to 46 for all Skype for Business Phone Edition devices at all sites.
- C. Configure an Audio and Video Quality of Service (QoS) to use a DSCP value of 40 and a port range of 57501:65535.
- D. Configure a Domain Group Policy object (GPO) configuring Quality of Service (QoS) and set it to Enforced for all Windows 7 clients.

Answer: B

Explanation: References:

[https://technet.microsoft.com/en-us/library/jj205137\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/jj205137(v=ocs.15).aspx)





NEW QUESTION 31

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 32

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 36

You are planning the Peer-to-Peer (P2P) audio bandwidth requirements for the Berlin wide area network (WAN) connection.

You need to calculate the bandwidth usage network impact of P2P audio calls between Berlin and Moscow. What is the minimum bandwidth in megabits per second (Mbps) required for P2P calls between Berlin and Moscow?

- A. 8.7 Mbps
- B. 7.8 Mbps
- C. 9.1 Mbps
- D. 6.1 Mbps

Answer: A

NEW QUESTION 40

You need to deploy the Mediation Server pools, gateways, and trunks that are required to support London, Berlin, and Moscow users.

How many of each component should you deploy? To answer, select the appropriate number of Mediation Server pools, gateways, and trunks and gateways from the list in the answer area.

Component

Number to deploy

Mediation Server pools

	▼
3	
4	
5	
6	

Gateways

	▼
3	
4	
5	
6	

Trunks

	▼
3	
4	
5	
6	

Answer:

Explanation:

Component

Number to deploy

Mediation Server pools

	▼
3	
4	
5	
6	

Gateways

	▼
3	
4	
5	
6	

Trunks

	▼
3	
4	
5	
6	

NEW QUESTION 41

You need to configure the Berlin public switched telephone network (PSTN) usages and voice routes. In which order will the PSTN usage policies be applied? To answer, move all PSTN usage policies from the list of PSTN usage policies to the answer area and arrange them in which they should be applied.

PSTN usage policies

Berlin_National_pstnusage
Moscow_International
Moscow_National_pstnusage
Berlin_International

PSTN usage policy application order



Answer:

Explanation:

PSTN usage policies

Berlin_National_pstnusage
Moscow_International
Moscow_National_pstnusage
Berlin_International

PSTN usage policy application order

Moscow_International
Berlin_National_pstnusage
Moscow_National_pstnusage
Berlin_International

NEW QUESTION 44

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 48

You need to plan a solution to allow masking of a calling number. How can a calling number of a Skype for Business user be changed to match the main number of the company?

- A. Create a new normalization rule and add it to the global dial plan.
- B. Use the Skype for Business control panel to mask the calling number.
- C. Create a trunk translation rule that masks the called number.
- D. Use the voice gateway to mask the called number.

Answer: A

NEW QUESTION 53

You need to configure the trunks defined in the Skype for Business topology based on the requirements. Which two actions should you perform? Each correct answer presents part of the solution.

- A. Enable media bypass.
- B. Enable centralized media processing.
- C. Set Encryption Support Level to required.
- D. Disable REFER support.

Answer: AD

NEW QUESTION 57

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 61

You need to troubleshoot and resolve the issue User1 is reporting. Which action should you perform to resolve the issue?

- A. Troubleshoot the Quality of Service (QoS) configuration.
- B. Verify that User1 has selected the new headset.
- C. Modify the client configuration to not use G.722 codec.
- D. On the public switched telephone network (PSTN) trunk, verify the aLaw/μLaw configuration.

Answer: B

NEW QUESTION 66

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		<div>▼</div> <div>1 – 1024</div> <div>1024 – 65535</div> <div>50020 – 50040</div> <div>66020 – 66040</div>
Video		<div>▼</div> <div>1 – 1024</div> <div>1024 – 65535</div> <div>50030 – 50050</div> <div>58000 – 58020</div>
Application sharing		<div>▼</div> <div>1 – 1024</div> <div>1024 – 65535</div> <div>42000 – 42020</div> <div>58000 – 58020</div>
File transfer		<div>▼</div> <div>1 – 1024</div> <div>1024 – 65535</div> <div>42040 – 42040</div> <div>50025 – 50045</div>

Answer:

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

NEW QUESTION 70

Tailspin Toys completes the implementation of Call Admission Control (CAC) for the entire network. You need to configure the Enterprise Voice policies for executives.

What should you do?

- A. Enable the bandwidth policy override feature in the user level voice policy.
- B. Enable the bandwidth policy override feature in the user level client policy.
- C. Enable the bandwidth policy override feature in the machine level group policy object.
- D. Enable the bandwidth policy override feature in the user level group policy object.

Answer: A

NEW QUESTION 71

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.

1390

1280

1340

1230

2570

2670

Location

Melbourne

Singapore

Number of Lines

Answer:

Explanation:

1390

1280

1340

1230

2570

2670

Location

Melbourne

Singapore

Number of Lines

1340

1230

NEW QUESTION 72

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 73

You need to resolve the issue users are having with the Call Via Work functionality. What should you do?

- A. Create and apply a Call Via Work polic
- B. Enable the policy.
- C. Create and apply a client policy that enables the CallViaWork option.
- D. Create and apply a voice polic
- E. Enable the CallViaWork option.
- F. Configure Call Via Work on the public switched telephone network (PSTN) trunk that connects to the PBX.
- G. Create and apply a Call Via Work policy that specifies a value for the AdminCallBackNumber option.

Answer: A

Explanation: References:
<https://technet.microsoft.com/en-gb/library/dn933908.aspx>

Topic 6, Contoso, LTD 2

Contoso, Ltd. is a Canadian business law firm that currently employs 300 people. The company operates a main office in Vancouver that houses 250 employees and a branch office in Seattle that houses 50 employees. All Contoso users are Skype for Business-enabled. Contoso is launching a hybrid configuration of Office 365 to validate the cloud-based Exchange Online service.

Contoso plans to introduce a remote disaster recovery data center to host remote replicas of key services, including Active Directory Domain Services (AD DS), Skype for Business, Exchange, and files and legal software. The company also plans to acquire a small law firm in Vancouver called Litware, Inc., which has 80 employees. Litware currently uses a 10-megabit per second (Mbps) Internet connection.

Half of Litware employees will use Skype for Business services. All Skype for Business-enabled users in Litware will use Instant Messaging only. Currently, Instant Messaging concurrent usage at Litware is 50 percent during peak hours.

There is a possibility that Contoso will make two more acquisitions during the next three years. Vancouver and Seattle employees have submitted support tickets to report poor quality of outbound Skype for Business calls to external users. A user activity report collected by the help desk department is shown in the User Activity Report section.

There is a concern that in the future, the volume of Skype for Business traffic may reach up to 50 percent of the current bandwidth available in the Vancouver office.

User Activity Report

Audio Stream (Caller -> Callee)			
Codec:	G722	Sample rate:	16000
Audio FEC:	False	Bandwidth estimates:	8987 Kbps
Packet utilization:	80370		
Avg. packet loss rate:	16%	Max. packet loss rate:	21.00%
Avg jitter:	23 ms	Max. jitter:	29 ms
Avg. round trip:	21 ms	Max. round trip:	24 ms
Burst duration:	0 ms	Burst gap duration:	1606060 ms
Burst density:	0.00%	Burst gap density:	0.00%
Avg. concealed samples ratio:	0.00 %	Avg. stretched samples ratio:	0.00%
Avg. compressed samples ratio:	0.00%		
Avg. network MOS:	2.21	Min. network MOS:	1.89
Avg. network MOS degradation:	0.08	Max. network MOS degradation:	0.40
NMOS degradation (jitter):	0.00%	NMOS degradation (packet loss):	0.00%
Avg. sending MOS:	1.35	Min. sending MOS:	1.00
Audio Stream (Callee -> Caller)			
Codec:	G722	Sample rate:	16000
Audio FEC:	False	Bandwidth estimates:	9011 Kbps
Packet utilization:	7887		
Avg. packet loss rate:	22.00%	Max. packet loss rate:	29.00%
Avg. jitter:	1.2 ms	Max. jitter:	2.2 ms
Avg. round trip:	1.3 ms	Max. round trip:	2.2 ms
Burst duration:	0 ms	Burst gap duration:	156400 ms
Burst density:	0.00%	Burst gap density:	0.00%
Avg. concealed samples ratio:	0.00%	Avg. stretched samples ratio:	0.00%
Avg. compressed samples ratio:	0.00%		
Avg. network MOS:	4.23	Min. network MOS:	4.06
Avg. network MOS degradation:	0.06	Max. network MOS degradation:	0.23
NMOS degradation (jitter):	0.00%	NMOS degradation (packet loss):	0.00%
Avg. listening MOS:	1.58	Min. listening MOS:	1.00

Existing Environment

The core portion of the infrastructure is located in the Vancouver data center, although the Seattle data center also has two servers (listed below) to provide redundancy of Active Directory and data services.

The following services and applications comprise Contoso's infrastructure in Vancouver and Seattle.

Vancouver Data Center:

Role	Name
Active Directory Domain Controllers	DC1-VAN DC2-VAN
Exchange 2013 Servers	EXCH1-VAN EXCH2-VAN
Skype for Business Server 2015 Front-End Servers	FE1-VAN FE2-VAN
Skype for Business Server 2015 Back-End server	BE1-VAN
Skype for Business Server 2015 Edge Server	Edge1-VAN
DFS Server	FileSRV1-VAN
Backup Server	BackupSRV1-VAN
Fax and Print server	PrintSRV1-VAN
Application Server	AppSRV1-VAN

Seattle Data Center:

Role	Name
Active Directory Domain Controller	DC1-SEA
DFS Server	FileSRV1-SEA

An overview of the network infrastructure of Vancouver and Seattle data centers is shown in the exhibit. (Click the Exhibit button.)

All users in the Vancouver and Seattle offices use the Microsoft Windows 8 Enterprise operating system and phones that are enabled for Enterprise voice.

Contoso expects to see a growing number of remote users. They anticipate that up to 20 percent of users in Vancouver might work remotely.

Contoso has defined a team of 35 users who will participate in a pilot deployment of the Office 365 hybrid configuration.

The offices in Vancouver and Seattle are using 5-Mbps Internet connections. Contoso has a site-to-site VPN between the Vancouver office and the Seattle office.

All external outbound calls from the Vancouver and Seattle offices are routed outside through the office in Vancouver.

Business Requirements

Contoso needs to review the existing network topology as well as the Skype for Business Server 2015 infrastructure to ensure that network capacity is sufficient to accommodate the Litware, Inc. Skype for Business users in the future.

Contoso expects to see up to 85 percent of simultaneous instant messaging, 1% of peer-to-peer traffic and 25 percent of public switched telephone network (PSTN) calls over the WAN during business hours.

All existing issues that relate to poor quality of outbound Skype for Business calls need to be addressed as well.

Contoso wants to ensure the following:

Current network bandwidth allocated to the Vancouver office is sufficient to sustain the future volume of Skype for Business traffic and guarantee high quality of Skype for Business calls.

Volume the Skype for Business traffic does not exceed 25 percent of the WAN bandwidth.

Contoso plans to redesign the existing site-to-site VPN solution between the Vancouver and Seattle offices to implement a robust and reliable WAN solution with end-to-end support of Quality of Service (QoS).

The recommended design solution needs to eliminate the possibility of WAN oversubscription by Skype for Business traffic.

The proposed Skype for Business Server 2015 solution needs to be: Scalable to accommodate the company's future acquisitions.

Readily available for Vancouver and Seattle users.

In case of a disaster, a manual switchover of Skype for Business services to the disaster recovery datacenter should be available.

Technical Requirements

All Session Initiation Protocol traffic (SIP) traffic must be encrypted.

All external Skype for Business Server 2015 services must be protected by a firewall.

A commercial third-party certificate must be used on the external interface of Skype for Business Server 2015 Edge servers.

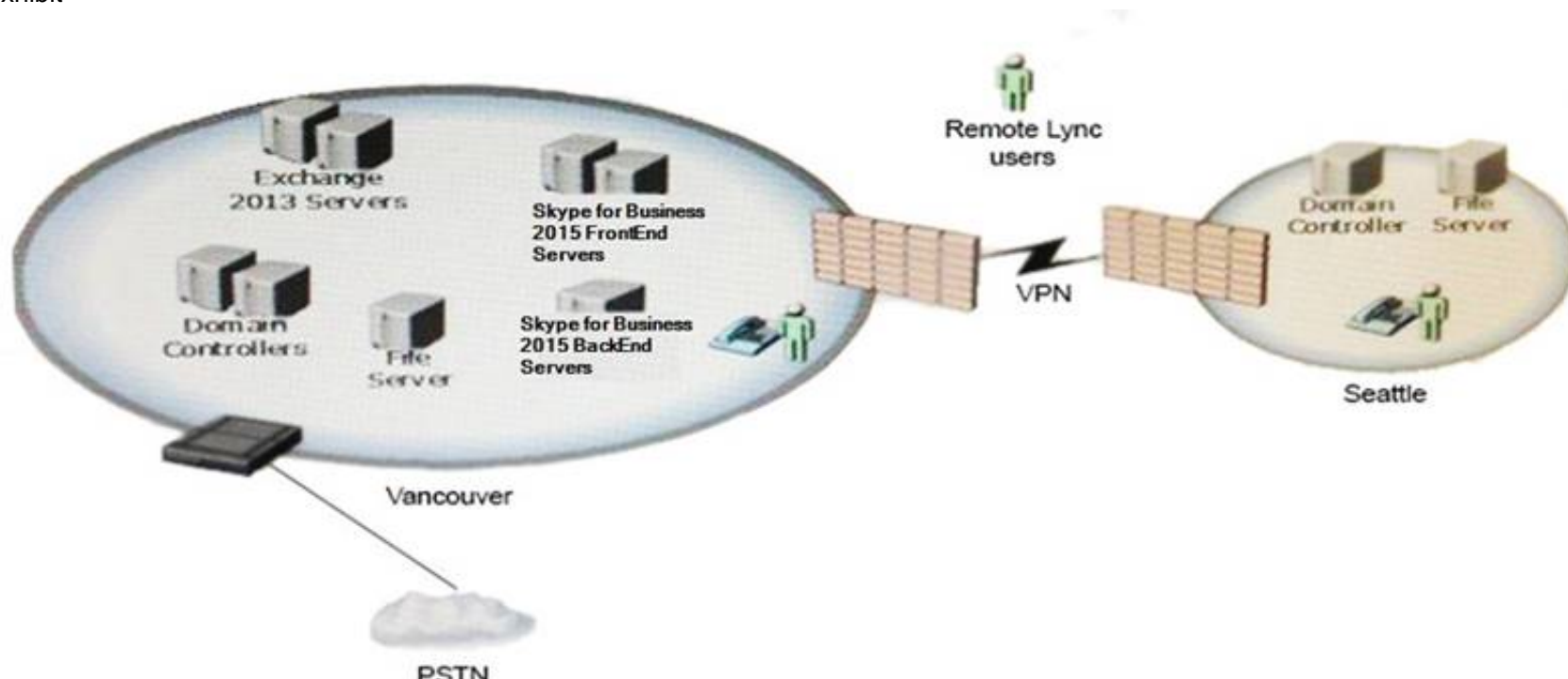
The Role-Based Access Control (RBAC) model will be implemented to delegate basic administrative tasks to the Skype for Business help desk team.

All maintenance of Skype for Business servers must be performed outside of business hours.

The proposed architecture solution must support interoperability between Skype for Business Server 2015 and Exchange 2013.

The Skype for Business Bandwidth Calculator must be used to analyze Skype for Business traffic. The QoS end-to-end support needs to be implemented.

Exhibit



NEW QUESTION 75

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.

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	Answer Area	
	Vancouver office	Seattle office
Configure Call Admission Control (CAC).	Task	Task
Deploy Quality of Service (QoS).	Task	Task
Deploy sFlow on all network switches.	Task	
Create a throttling policy.		
Implement Multiprotocol Label Switching (MPLS).		

Answer:

Explanation:

Tasks	Answer Area	
	Vancouver office	Seattle office
Configure Call Admission Control (CAC).		
Deploy Quality of Service (QoS).	Implement Multiprotocol Label Switching (MPLS).	Implement Multiprotocol Label Switching (MPLS).
Deploy sFlow on all network switches.	Deploy Quality of Service (QoS).	Deploy Quality of Service (QoS).
Create a throttling policy.	Configure Call Admission Control (CAC).	
Implement Multiprotocol Label Switching (MPLS).		

NEW QUESTION 78

You need to estimate the future volume of Skype for Business traffic in the Vancouver office.

Which two employee totals should you use for the estimate? Each correct answer presents part of the solution.

- A. 300 employees at Contoso, Ltd.
- B. 80 employees at Litware, Inc.
- C. 40 employees at Litware, Inc.
- D. 35 hybrid users at Contoso, Ltd.

Answer: AC

NEW QUESTION 82

You need to identify the metric that is causing poor quality in Skype for Business calls by analyzing the user activity report collected by the help desk department.

Which metric should you choose?

- A. burst gap duration
- B. round trip
- C. packet loss
- D. jitter

Answer: C

NEW QUESTION 85

You need to define the goals that you would like to achieve after you perform usage modeling of Skype for Business traffic.

Which two goals should you establish? Each correct answer presents part of the solution.

- A. Confirm that bandwidth is sufficient for 40 new Litware, Inc.
- B. Skype for Business users.
- C. Confirm that the volume of Skype for Business traffic does not exceed 5 megabits per second (Mbps).

- D. Confirm that bandwidth is sufficient for 80 new Litware, In
E. Skype for Business users.
F. Confirm that the volume of Skype for Business traffic does not exceed 25 percent of wide area network (WAN) bandwidth.
G. Confirm that bandwidth is sufficient for Microsoft Office 365 Hybrid Skype for Business deployment.

Answer: CD

NEW QUESTION 86

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.)

Skype for Business Bandwidth Calculator

Return to Start

Client and Deployment Options

Client: Lync 2010, Lync 2013, Skype for Business, Skype for Business Mobile, Lync Room System, Lync Room Edition
Deployment: Onpremises, Online, Online-Dedicated

Usage Models Definitions (Max Concurrency at Peak Time per Modality)

Usage	none	low	medium	high	Custom Threshold
IM/Presence	0%	65.00%	80.00%	90.00%	85.00%
Presence	0%	0.50%	1.50%	2.50%	1.00%
P2P audio	0%	0.10%	0.30%	0.50%	1.00%
P2P video	0%	1.00%	3.00%	5.00%	1.00%
conf audio	0%	0.10%	0.50%	1.00%	1.00%
conf video	0%	0.50%	1.00%	1.50%	1.00%
desktop share	0%	5.00%	10.00%	15.00%	25.00%
PSTN audio	0%	0%	0%	0%	0%

Persona Definitions

Client	Deployment	IM/Presence	P2P audio	P2P video	conf audio	conf video	desktop share	PSTN audio	Lync 2010 RTV Type	Remote Users	Skype for Business/Lync 2013 Stereo Audio	Skype for Business/Lync 2013 Video Quality
Desktop Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Mobile Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Administrator	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Small Branch Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Vancouver Employees	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best

Drop-down menus for Persona Definitions:

- IM/Presence: None, Low, Medium, High, Custom1
- P2P audio: None, Low, Medium, High, Custom1
- P2P video: None, Low, Medium, High, Custom1
- conf audio: None, Low, Medium, High, Custom1
- conf video: None, Low, Medium, High, Custom1
- desktop share: None, Low, Medium, High, Custom1
- PSTN audio: None, Low, Medium, High, Custom1
- Remote Users: 10%, 20%, 30%, 40%, 50%

Answer:

Explanation:

Skype for Business Bandwidth Calculator

Return to Start

Client and Deployment Options

Client: Lync 2010, Lync 2013, Skype for Business, Skype for Business Mobile, Lync Room System, Lync Room Edition
Deployment: Onpremises, Online, Online-Dedicated

Usage Models Definitions (Max Concurrency at Peak Time per Modality)

Usage	none	low	medium	high	Custom Threshold
IM/Presence	0%	65.00%	80.00%	90.00%	85.00%
Presence	0%	0.50%	1.50%	2.50%	1.00%
P2P audio	0%	0.10%	0.30%	0.50%	1.00%
P2P video	0%	1.00%	3.00%	5.00%	1.00%
conf audio	0%	0.10%	0.50%	1.00%	1.00%
conf video	0%	0.50%	1.00%	1.50%	1.00%
desktop share	0%	5.00%	10.00%	15.00%	25.00%
PSTN audio	0%	0%	0%	0%	0%

Persona Definitions

Client	Deployment	IM/Presence	P2P audio	P2P video	conf audio	conf video	desktop share	PSTN audio	Lync 2010 RTV Type	Remote Users	Skype for Business/Lync 2013 Stereo Audio	Skype for Business/Lync 2013 Video Quality
Desktop Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Mobile Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Administrator	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Small Branch Information Worker	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best
Vancouver Employees	Skype for Business	Onpremises	High	medium	medium	medium	medium	medium	CIF	10%	0%	Best

Drop-down menus for Persona Definitions:

- IM/Presence: None, Low, Medium, High, Custom1
- P2P audio: None, Low, Medium, High, Custom1
- P2P video: None, Low, Medium, High, Custom1
- conf audio: None, Low, Medium, High, Custom1
- conf video: None, Low, Medium, High, Custom1
- desktop share: None, Low, Medium, High, Custom1
- PSTN audio: None, Low, Medium, High, Custom1
- Remote Users: 10%, 20%, 30%, 40%, 50%

NEW QUESTION 87

You need to estimate the volume of Skype for Business traffic.

In the Skype for Business Calculator, which options should you select? To answer, select the appropriate branch site or sites in the Skype for Business Bandwidth Calculator.

Return to Start

Display Units: Kbps

Site definitions

Site Name	Total Users in Site	Central Site Providing User Services (for Onprem users)	Internet Site (for Onprem users)	OS65 Site Providing User Services (for Online users)	Internet Access (for Online users)	Number of Sites like this	Local PSTN Breakout? (for Onprem Users)
Seattle	0					1	Yes
Vancouver	0					1	No

WAN link Info

Low delay WAN ? (Default=No)	WAN Link Speed	Input Units	WAN Link BW Allocated to RTC Traffic	Input Units
No	5	Mbps	1.25	Mbps
	10	Mbps	5	Mbps
		Mbps	10	Mbps

Answer:

Explanation:

Return to Start		Site definitions					WAN link Info			
Display Units	Kbps									
Site Name	Total Users in Site	Central Site Providing User Services (for Onpremise users)	Internet Site (for Onpremise users)	OS65 Site Providing User Services (for Online users)	Internet Access (for Online users)	Number of Sites like this	Local PSTN Breakout? (for Onpremise Users)	Low delay WAN ? (Default=No)	WAN Link Speed	WAN Link BW Allocated to RTC Traffic
Seattle	0	0	Vancouver	Vancouver		1	Yes No	No	5 10	1.25 5 10

NEW QUESTION 91

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 94

You deploy Skype for Business for a company that has three office locations. The locations and total number of users per location are shown below:

Office Location	Total number of users
Boston	2,000
Chicago	1,500
New York	4,000

You must deploy Enterprise Voice. You have the following requirements:

All New York and Boston users that have IP PBX phones must be able to call each other without using the public switched telephone network (PSTN).

Audio and video conferencing quality must be improved between the New York and Chicago offices. You need to ensure that Enterprise Voice is deployed correctly.

How many of each component or server should you deploy? To answer, select the appropriate number for each component or server in the answer area.

Answer Area

Server or component

Number to add

Skype for Business Server 2015 Mediation Servers:

▼

0
1
2
3
4

PSTN Gateways:

▼

0
1
2
3
4

Skype for Business Server 2015 Trunks:

▼

0
1
2
3
4

Answer:

Explanation:

Answer Area

Server or component

Number to add

Skype for Business Server 2015 Mediation Servers:

▼

0
1
2

3

4

PSTN Gateways:

▼

1

0
2
3
4

Skype for Business Server 2015 Trunks:

▼

2

0
1
3
4

NEW QUESTION 98

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
RTCPCallsOnHold : True
SRTMode : 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 99

Tailspin Toys deploys a Skype for Business Enterprise Voice environment that has two paired pools. The pools are located in Sydney and Singapore for high availability (HA).

You must collect user opinions about the quality of audio calls. You must survey 60% of the calls for feedback.

You need to configure end users ratings for the audio calls. What should you do?

- A. Use SQL Management studio to run the SQL query [QoeMetrics].[dbo].[CallQualityFeedbackTokenDef].
- B. Deploy Call Quality Dashboard (CQD) to see the Rate My Call reports.
- C. Configure Rate My Call percentage to 60% and deploy Call Quality Dashboard (CQD) to see the reports.
- D. Use SQL Management studio to run the SQL query [QoeMetrics].[dbo].[CallQualityFeedbackToken].

Answer: B

Explanation: References:

<https://technet.microsoft.com/en-us/library/dn951407.aspx>

<http://blogs.technet.com/b/jenstr/archive/2015/05/05/rate-my-call-in-skype-for-business-2015.aspx>

NEW QUESTION 102

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 adelegate to each other.
- D. For every TeleHelp group member configure Simultaneously Ring to My Delegates and add each memberas 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 104

Tailspin Toys deploys Skype for Business Enterprise Voice.

Tailspin Toys executives must be able to call each other directly at any time irrespective of their presence status.

You need to configure the Enterprise Voice solution. What should you do?

- A. Set the value of the privacy relationship setting for each executive team contact to the value Workgroup.
- B. Set the value of the simultaneous ring setting for all executives to Ring my delegates after 10 seconds.
- C. Set the value of the simultaneous ring setting for all executives to Ring my team call group
- D. Add all of the members of the leadership team to the team call group.
- E. Create a private telephone line for each executive and implement Call Via Work.

Answer: A

Explanation: References:

<http://blogs.technet.com/b/seanearp/archive/2011/09/06/leave-me-alone-using-lync-to-not-communicate.aspx>

NEW QUESTION 108

A company implements Enterprise Voice. All users are assigned a direct inward dialing (DID) phone number. Due to the increased fees for direct inward dialing (DID) phone ranges, the company must make the following changes:

Reduce the DID phone number range from 2000 to 1000 phone numbers by implementing internal extensions.

Transfer all calls to unassigned phone numbers to an operator after informing callers that the phone number is not currently in use.

Transfer all calls to phone number range replaced by internal extensions to an operator after informing callers that the phone number has been replaced.

Ensure that any changes to the system will not have adverse effects.

You need to implement a solution.

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

Select and Place:

Actions		Answer Area
Import two announcement files.		
Create two announcements.	➤	⬆
Modify the TEL URI of the users.	⬅	⬇
Create two announcement number ranges.		

Answer:

Explanation: References:

<https://technet.microsoft.com/en-us/library/gg412783.aspx> [https://technet.microsoft.com/en-us/library/gg412748\(v=ocs.15\).aspx](https://technet.microsoft.com/en-us/library/gg412748(v=ocs.15).aspx)

NEW QUESTION 113

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 118

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 centralsite.
- 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 121

You perform a network readiness assessment for a Skype for Business deployment.

The organization's network includes a head office and 400 branch offices. All branch offices are connected to the head office with a Multiprotocol Label Switching (MPLS) link.

You have completed the traffic simulation. The Usage Modeling Data for the Beijing site is shown in the following table:

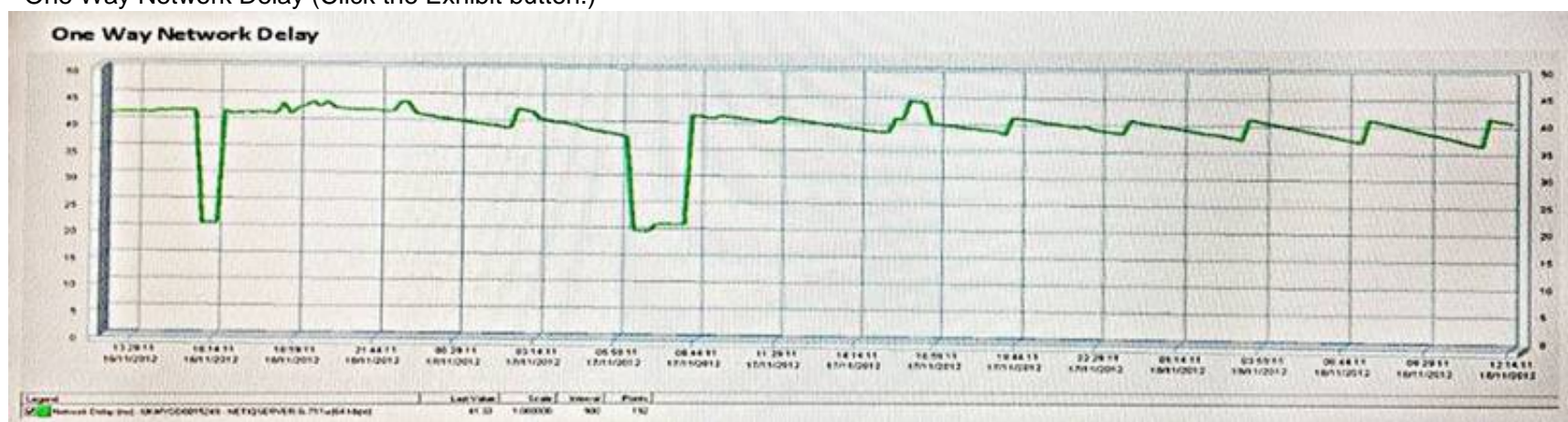
Metric	Value
Total users	500
Wide area network (WAN) link speed megabits per second (Mbps)	20
Peak users signed in	320
Total WAN BW	3,728.4
Total WAN BW no video	2,848.4
% of WAN link	18.21%
% of WAN link no video	13.91%
% of WAN BW for RTC traffic	182.05%
% of WAN BW for RTC traffic no video	139.08%

A summary of the simulation results is shown in the table below:

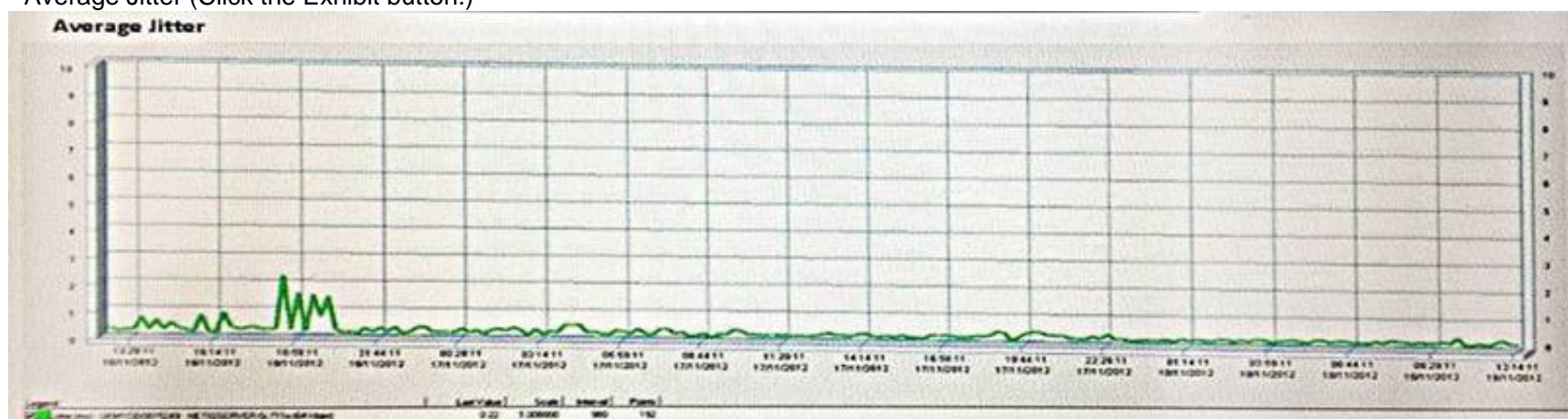
Metric	Value
Traffic Simulated	Beijing: 18 calls
Average MOS	4.36
Average Delay	39.57 ms
Average Jitter	0.31 ms
Maximum Jitter	30 ms
Average Packet Loss	0%
Maximum Consecutive Datagram Loss	1
General Comments	Overall the network performed extremely well, with optimal scores throughout.

Detailed simulation results are shown in the following exhibits:

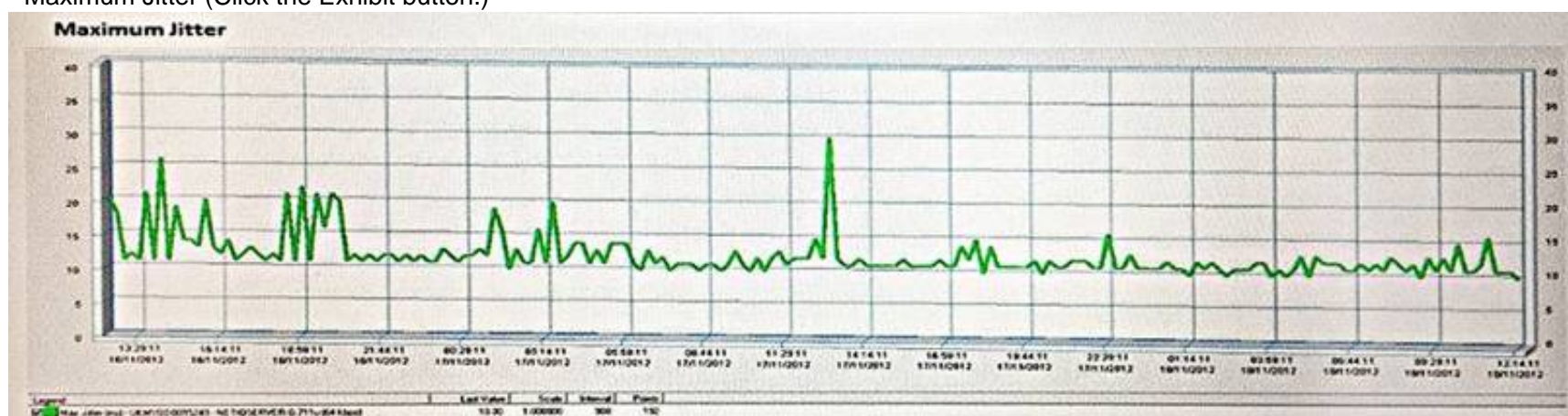
- One Way Network Delay (Click the Exhibit button.)



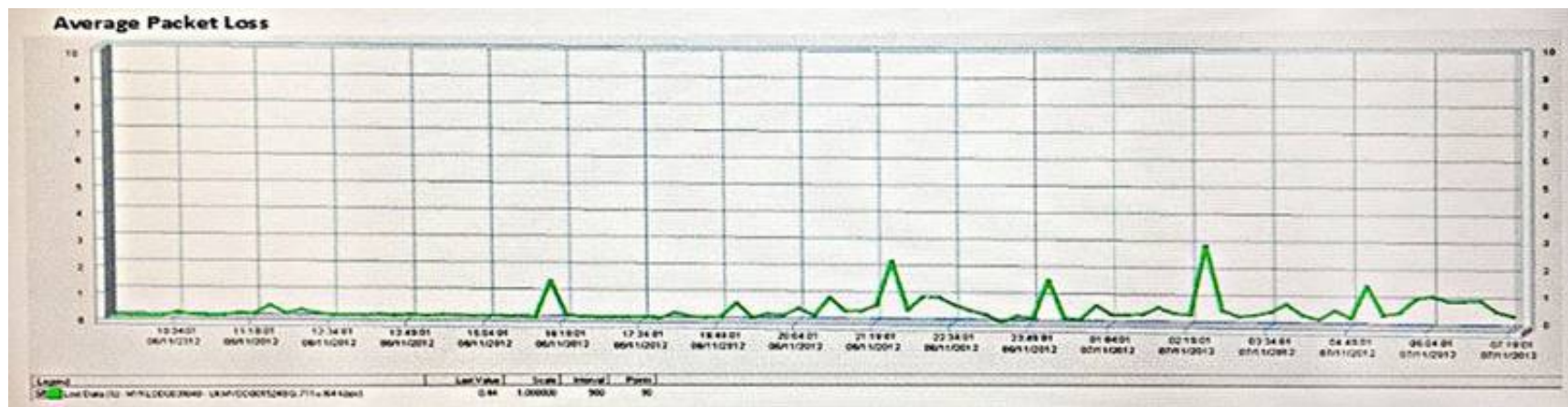
- Average Jitter (Click the Exhibit button.)



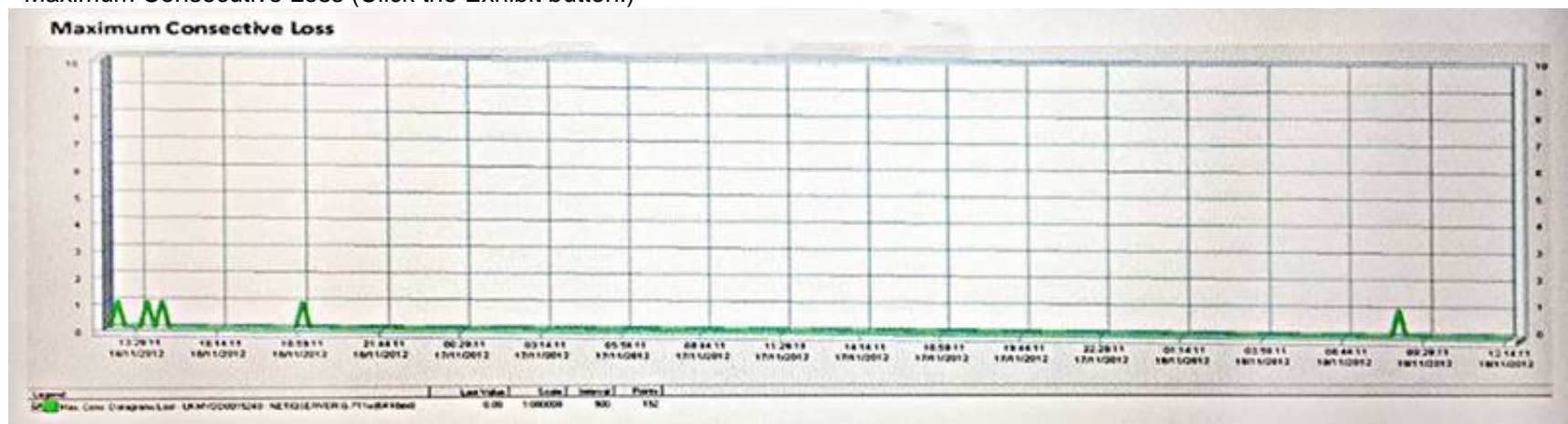
- Maximum Jitter (Click the Exhibit button.)



- Average Packet Loss (Click the Exhibit button.)



- Maximum Consecutive Loss (Click the Exhibit button.)



You need to provide recommendations based on the simulation results. What should you recommend?

- A. Increase WAN link bandwidth.
- B. Monitor Quality of Experience (QoE).
- C. Implement Call Admission Control (CAC) on Skype for Business for the affected sites.
- D. Restrict video capability or reduce quality codec usage for certain user groups.

Answer: A

NEW QUESTION 124

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 126

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 127

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 129

A customer deploys a Skype for Business Server 2015 Enterprise Edition pool and configures Enterprise Voice. The customer deploys Survivable Branch Appliances (SBAs) at each of the remote office locations.

The customer has reported that multiple users in one of the remote offices are not able to dial certain local numbers using Skype for Business. The users receive the error “+1 (425) 555-0100 is unavailable or may not be valid,” and the call fails.

The customer has provided the following information from the Local Number normalization rule from the Dial Plan from the remote office that is experiencing the issue:

Pattern to Match: ^1((425 | 206)\d{7})\$ Translation Rule: +1\$1

You need to ensure that all users at the remote office are able to dial local 10–digit numbers with or without a leading 1.

What should you do?

- A. Change the translation pattern from +1\$1 to +1.
- B. Change the translation pattern from +1\$1 to +\$1.
- C. Change the pattern to match from ^1((425 | 206)\d{7})\$ to ^1?((425 | 206)\d{7})
- D. Change from the pattern to match from ^1((425 | 206)\d{7})\$ to ^((1?425 | 206)\d{7})\$

Answer: C

NEW QUESTION 130

Your company has a Skype for Business Server 2015 infrastructure and a Microsoft Exchange Server 2013. The Skype for Business Server 2015 infrastructure has Enterprise Voice enabled and uses a global four-digit dial plan.

The Exchange Server organization has Unified Messaging (UM) enabled and uses a four-digit UM dial plan named DialPlan1. All dialing restrictions are set to allow All.

You need to ensure that only the users in the research department can use the Play on Phone feature. Which two actions should you perform? Each correct answer presents part of the solution.

- A. Create a new hunt group.
- B. Modify the default UM mailbox policy.
- C. Modify DialPlan1.
- D. Create a new UM mailbox policy.
- E. Create a new UM dial plan.

Answer: BD

NEW QUESTION 134

You support a Skype for Business Server 2015 environment that has Enterprise Voice (EV).

The company has opened two new satellite branches and configured one with a Survivable Branch Appliance (SBA) and the other with a Survivable Branch Server (SBS).

A wide area network (WAN) outage occurs and EV stopped working for the branch users. You need to ensure that EV functionality continues in the event of WAN outages,

What are two possible ways to achieve this goal? Each correct answer presents a complete solution.

- A. Enable backup Registrar discovery by configuring Dynamic Host Configuration Protocol (DHCP) option 55 on the branch site's DHCP server.
- B. Enable backup Registrar discovery by configuring Dynamic Host Configuration Protocol (DHCP) option 120 on the branch site's DHCP server to point to the fully qualified domain name (FQDN) of the SBA or SBS.
- C. Enable backup Registrar discovery by configuring Dynamic Host Configuration Protocol (DHCP) option 60 on the branch site's DHCP server.
- D. Enable backup Registrar discovery by configuring the SBA or SBS to respond to Dynamic Host Configuration Protocol (DHCP) 120 queries.

Answer: BD

NEW QUESTION 135

You support a Skype for Business Server 2015 environment.

A customer reports that they are receiving alerts from the voice gateway and that outbound calls are failing due to no trunks being available.

You need to configure performance counters to determine the reason that calls are failing.

How should you configure the performance counters? To answer, drag the appropriate counter and server to the correct location or locations in the Solution area. Each voice counter and server may be used once, more than once, or not at all. You may need to drag the split bar between panes or scroll to view content. Each correct answer is worth one point.

Servers and Counters		Solution
Mediation Server		Server or Counter
Edge Server		Server or Counter
Front End Server		
LS:Mediation Server - Total Calls	●	Server or Counter
LS:Mediation Server - Inbound Calls	●	
LS:Mediation Private - MediationServer	●	Server or Counter
LS:Edge Server - Total Calls		Server or Counter
LS:Edge Server - Inbound Calls		
LS:Edge Private - EdgeServer		

Answer:

Explanation:

Servers and Counters		Solution
Mediation Server		Server: Mediation Server
Edge Server		Counter: LS:Mediation Server - Total Calls
Front End Server		
LS:Mediation Server - Total Calls	●	
LS:Mediation Server - Inbound Calls	●	Server: Mediation Server
LS:Mediation Private - MediationServer	●	Counter: LS:Mediation Server - Inbound Calls
LS:Edge Server - Total Calls		
LS:Edge Server - Inbound Calls		
LS:Edge Private - EdgeServer		

NEW QUESTION 139

Parnell Aerospace has a Skype for Business Server 2015 environment. The company plans to implement response groups for their internal Help Desk and Accounting teams.

The two teams have the following requirements for their individual response groups: Help Desk Response Group

*Agents must sign in to the response group before they can accept calls.

*Must offer new calls to the agents in the order in which they are listed in the Agent list.

*Need to share queues with other Response Groups. Accounting Response Group

*Agents must not need to sign in to the response group before they can accept calls.

*Must offer new calls to each agent in turn.

*Must be managed by a dedicated Response Group Manager.

You need to configure the response groups for the Help Desk and Accounting teams.

How should you configure the response groups? To answer, select the appropriate configuration option from each list in the answer area.

Answer Area

Response group configuration options

Participation Policy

Routing Method

Workflow Type

Help Desk response group

Formal

	▼
Longest Idle	
Parallel	
Round Robin	
Serial	

	▼
Managed	
Unmanaged	

Accounting response group

Informal

	▼
Longest Idle	
Parallel	
Round Robin	
Serial	

	▼
Managed	
Unmanaged	

Answer:

Explanation:

Answer Area

Response group configuration options

Participation Policy

Routing Method

Workflow Type

Help Desk response group

Formal

	▼
Longest Idle	
Parallel	
Round Robin	
Serial	

	▼
Managed	
Unmanaged	

Accounting response group

Informal

	▼
Longest Idle	
Parallel	
Round Robin	
Serial	

	▼
Managed	
Unmanaged	

NEW QUESTION 143

You have a Skype for Business Server 2015 infrastructure. The infrastructure contains three servers. The servers are configured as shown in the following table:

Server name	Role
Server1	Standard Edition server
Server2	Mediation Server
Owa01	Office Web Apps Server

You plan to provide users with the ability to use desktop sharing and to display Microsoft PowerPoint presentations during conference. You need to create the firewall rules.

How should you configure the firewall rules? To answer, select the appropriate servers and ports for each feature in the answer area.

Answer Area

Feature

Desktop Sharing

Server

	▼
Server1	
Server2	
Owa1	

Ports

	▼
443	
5061	
5350 to 5389	
49,152 to 65,535	

Display PowerPoint presentations

	▼
Server1	
Server2	
Owa1	

	▼
443	
5061	
5350 to 5389	
49,152 to 65,535	

Answer:

Explanation:

Answer Area

Feature	Server	Ports
Desktop Sharing	<div> <div>▼</div> <div>Server1</div> <div>Server2</div> <div>Owa1</div> </div>	<div> <div>▼</div> <div>443</div> <div>5061</div> <div>5350 to 5389</div> <div>49,152 to 65,535</div> </div>
Display PowerPoint presentations	<div> <div>▼</div> <div>Server1</div> <div>Server2</div> <div>Owa1</div> </div>	<div> <div>▼</div> <div>443</div> <div>5061</div> <div>5350 to 5389</div> <div>49,152 to 65,535</div> </div>

NEW QUESTION 147

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 152

You support a customer's Skype for Business Server 2015 environment. Your customer has granted User1 Conferencing Policy A, and User 2 Conferencing Policy B.

When User2 joins a video conference that is hosted by User1, User2 sees only a single video window. Other users who are joined to the same conference see up to five concurrent video streams.

You need to configure the environment to allow User2 to view multiple concurrent video streams. You need to achieve the goal with the fewest changes to the environment.

Which Skype for Business Management Shell command should you run?

- A. Set-CSClientPolicy -EnableFullScreenVideo \$True
- B. Get-CSConferencingPolicy -Identity 'Conferencing Policy B' | Set-CSConferencingPolicy-EnableMultiViewJoin \$True
- C. Get-CSConferencingPolicy -Identity 'Conferencing Policy B' | Set-CSConferencingPolicy-AllowMultiView \$True
- D. Set-CSClientPolicy -EnableMediaRedirection \$True

Answer: B

Explanation: References:

<https://technet.microsoft.com/en-us/library/gg425788.aspx>

NEW QUESTION 154

Your network contains four servers. The servers are configured as shown in the following table:

Server name	Software	Role
Server1	Skype for Business Server 2015	Front End Server
Server2	Skype for Business Server 2015	Edge Server
Server3	Microsoft Exchange Server 2013	Edge Transport Server
Server4	Microsoft Exchange Server 2013	Mailbox server Client Access server

You need to configure backend integration between Skype for Business Server 2015 and the Exchange Server 2013 Unified Messaging (UM).

You create a dial plan and an auto attendant in the Exchange Server organization. What should you do next?

- A. On Server1, run ocsumutil.exe.
- B. On Server2, run ocsumutil.exe.
- C. On Server3, run exchucutil.ps1.

D. On Server4, run exchucutil.ps1.

Answer: D

NEW QUESTION 157

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

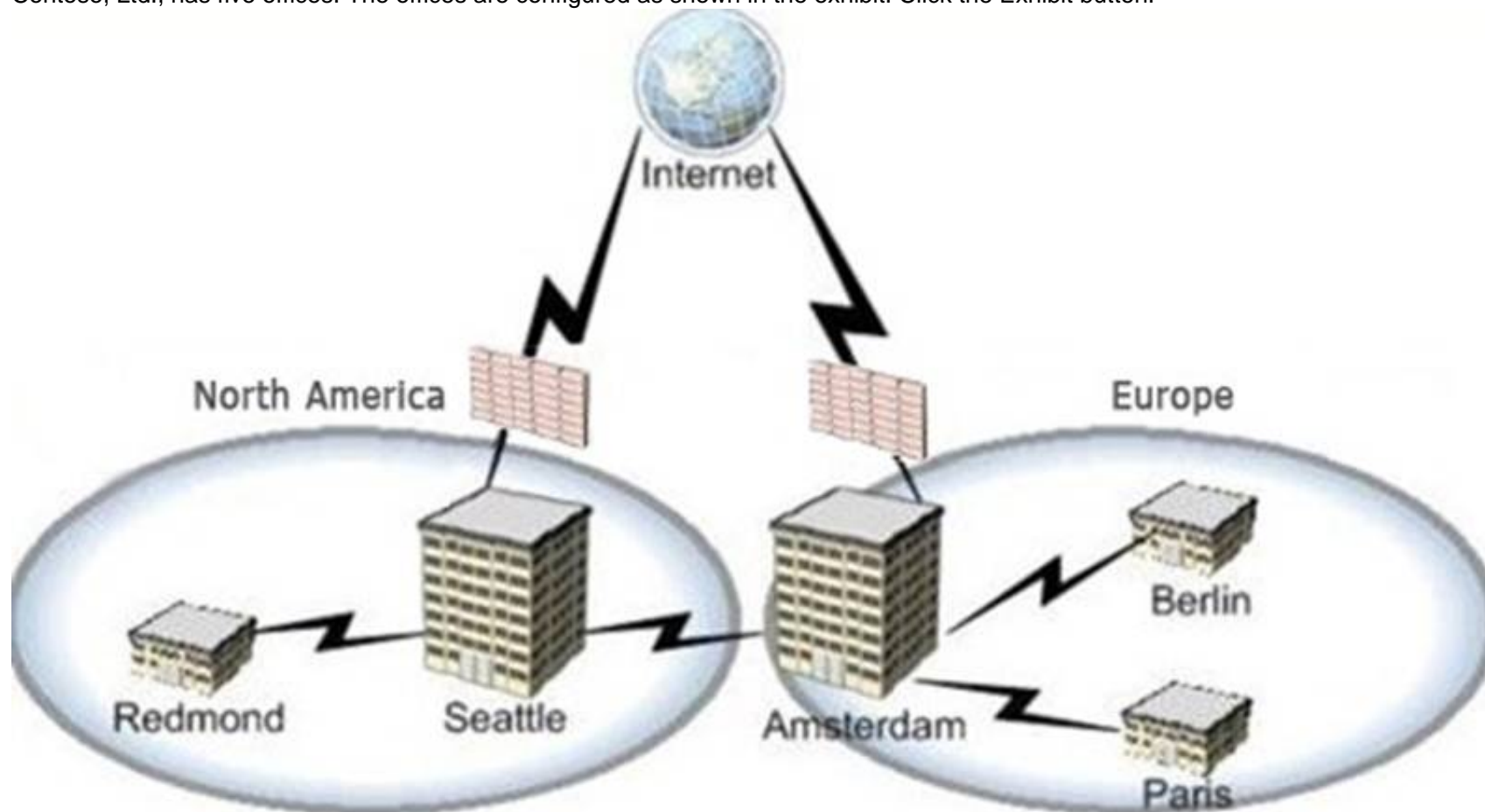
You need to ensure that all outbound voice calls to the PSTN from the users in the Toronto site are routed through the trunk at the Toronto site. Calls must use the Mediation Server at the Chicago site if the Mediation Server at the Toronto site is unavailable. What should you do?

- 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 162

Contoso, Ltd., has five offices. The offices are configured as shown in the exhibit. Click the Exhibit button.



Each office has a local connection to the public switched telephone network (PSTN). The peak usage statistics are shown in the following table:

Office	Number of calls	Percentage of calls to North America	Percentage of calls to Europe
Redmond	10	50 percent	50 percent
Seattle	100	80 percent	20 percent
Amsterdam	50	20 percent	80 percent
Berlin	20	0 percent	100 percent
Paris	20	50 percent	50 percent

Every audio call consumes 100 kilobits per second (Kbps).

You plan to replace the PSTN connection with a centralized SIP trunking service. The service will contain a Sip trunk in the Seattle office and a SIP trunk in the Amsterdam office. The solution must use least cost routing.

You need to calculate the maximum amount of bandwidth that will be used by each SIP trunk.

What is the maximum amount of bandwidth that will be used by each SIP trunk? To answer, drag the appropriate bandwidth amount to the correct SIP trunk in the answer area. Each bandwidth amount 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.

Bandwidth		Answer Area
8,500 Kbps	Amsterdam SIP trunk:	Bandwidth
9,500 Kbps	Seattle SIP trunk:	Bandwidth
10,000 Kbps		
10,500 Kbps		
12,500 Kbps		

Answer:

Explanation:

Bandwidth		Answer Area
8,500 Kbps	Amsterdam SIP trunk:	9,500 Kbps
9,500 Kbps	Seattle SIP trunk:	10,500 Kbps
10,000 Kbps		
10,500 Kbps		
12,500 Kbps		

NEW QUESTION 165

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 166

Your network contains a Skype for Business Server 2015 infrastructure and a Microsoft Exchange Server 2013 organization.

You have three distribution groups. The groups are configured as shown in the following table.

Group name	Members
Team1@contoso.com	Users who work on Team1
Team2@contoso.com	Users who work on Team2
AllTeams@contoso.com	All users from Team1 and Team2

You need to ensure that all of the users on Team1 and Team2 can answer calls that are in the same queue. Users must be able to choose when they want to participate in the queue.

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.

Actions

Create a voice policy.

Create a new formal agent group.

Create a new response group queue.

Create a new informal agent group.

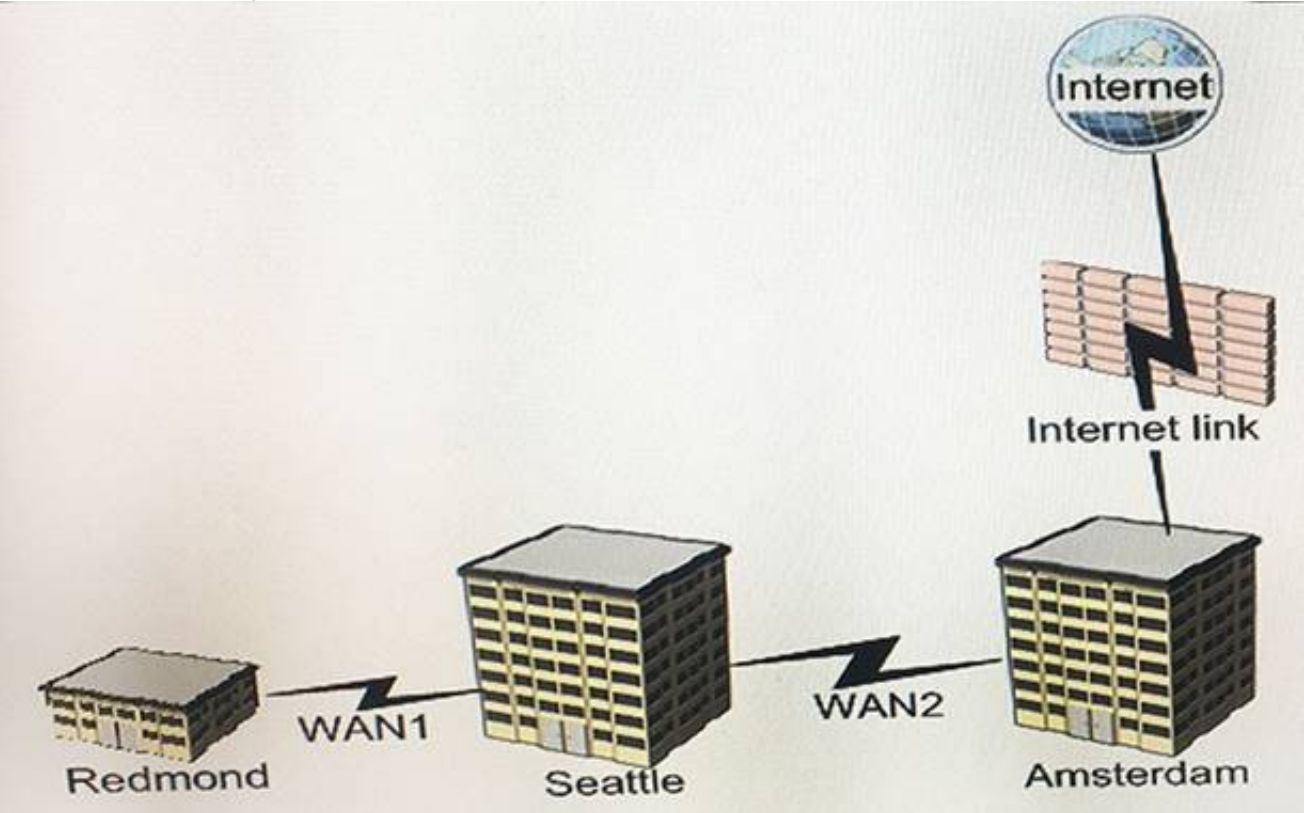
Create a new workflow.

Answer Area

Answer:

Explanation: References:
<https://docs.microsoft.com/en-us/skypeforbusiness/plan-your-deployment/enterprise-voice-solution/response-gro>

NEW QUESTION 171
Your company has three offices.
The network is configured as shown in the exhibit. Click the Exhibit button.



You have a Skype for Business Server 2015 infrastructure. All of the Skype for Business Server components are located in the Amsterdam office. The conference call will contain the attendees shown in the following table:

Attendee location	Number of attendees
Seattle	50
Redmond	10
Amsterdam	100
Internet (Remote)	20

Each audio call consumes 100 kilobits per second (Kbps).
You need to allocate network bandwidth for each attendee location.
How much bandwidth should you allocate for each attendee location? To answer, drag the appropriate bandwidth amounts to the correct attendee locations in the answer area. Each bandwidth amount 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.

Bandwidths

1,000 Kbps

2,000 Kbps

5,000 Kbps

6,000 Kbps

10,000 Kbps

15,000 Kbps

Answer Area

●

WAN1:

Bandwidth

●

WAN2:

Bandwidth

●

Internet link:

Bandwidth

Answer:

Explanation:

Bandwidths

1,000 Kbps

2,000 Kbps

5,000 Kbps

6,000 Kbps

10,000 Kbps

15,000 Kbps

Answer Area

●

WAN1:

1,000 Kbps

●

WAN2:

6,000 Kbps

●

Internet link:

2,000 Kbps

NEW QUESTION 173

You have a Skype for Business Server 2015 infrastructure that contains two offices. The offices are located in Amsterdam and Seattle. The offices have the phone numbers ranges shown in the following table:

Office	Total phone number range
Amsterdam	+31 20 1234 5 00 to +31 20 1234 5 99
Amsterdam	+31 20 1234 7 00 to +31 20 1234 7 99
Seattle	+1 425 555 1200 to +1 425 555 1299

You plan to deploy Enterprise Voice.
You also plan to use a global dial plan that will use three-digit dialing. You need to create the normalization rules for the planned deployment.
Which match patterns should you use? To answer, drag the appropriate match patterns to the correct translation patterns on the answer area. Each match pattern may be used once, more than once, or not at all. Additionally, you may need to drag the split bar or scroll to view content.

Match Patternns

Answer Area

^(5\d{2})\$

^([57]\d{2})\$

^([57])(\d{2})\$

^(2\d{2})\$

^12(\d{2})\$

^1(2\d{2})\$



Translation pattern	Match pattern
+31201234\$1	Match pattern
+14255551\$1	Match pattern

Answer:

Explanation:

Match Patternns

Answer Area

^(5\d{2})\$

^([57]\d{2})\$

^([57])(\d{2})\$

^(2\d{2})\$

^12(\d{2})\$

^1(2\d{2})\$



Translation pattern	Match pattern
+31201234\$1	^([57]\d{2})\$
+14255551\$1	^(2\d{2})\$

NEW QUESTION 174

A company has offices in New York and Seattle. The offices contain a total of five subnets. The subnets are configured as shown in the following table.

Office	Floor	Subnet	Subnet mask
New York	First	172.16.1.0	255.255.255.0
New York	Second	172.16.2.0	255.255.255.0
New York	Third	172.16.3.0	255.255.255.0
Seattle	First	172.16.4.0	255.255.255.0
Seattle	Second	172.16.5.0	255.255.255.0

You have a Skype for Business Server 2015 infrastructure. You deploy the Skype for Business client to all client computers. You plan to deploy Call Admission Control (CAC). You must minimize the number of regions, sites, and subnets for the deployment.

You need to design the network topology for the planned deployment. The solution must ensure that different bandwidth policies can be applied to each floor of each office.

How many regions, site and subnets should you deploy? To answer, select the appropriate number of regions, sites, and subnets in the answer area.

Answer Area

Regions:	<table><tr><td></td><td>▼</td></tr><tr><td>1</td><td></td></tr><tr><td>2</td><td></td></tr><tr><td>3</td><td></td></tr><tr><td>4</td><td></td></tr><tr><td>5</td><td></td></tr></table>		▼	1		2		3		4		5	
	▼												
1													
2													
3													
4													
5													
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Answer:

Explanation: References:
<https://technet.microsoft.com/en-us/library/gg412844.aspx>

NEW QUESTION 179

You have a Skype for Business Server 2015 infrastructure that has Enterprise Voice enabled. When a caller dials an unassigned number, the caller hears a custom message, and then the caller is forwarded to a user named User1. Several months later, User1 leaves the company. You need to ensure that when a caller dials an unassigned number, the caller hears the custom message and is forwarded to a user named User2. Which Skype for Business Management Shell cmdlet should you run?

- A. Set-CsUnassignedNumber
- B. Set-CsUser
- C. Set-CsVoiceRoute
- D. Set-CsAnnouncement

Answer: D

Explanation: References:
<https://technet.microsoft.com/en-us/library/gg425752.aspx>

NEW QUESTION 181

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 182

Trey Research currently leverages Exchange Online to provide voice mail capabilities for their existing IP-PBX. An Exchange Online Unified Messaging (UM) dial plan with type of E.164 Uniform Resource Identifier (URI) is currently in use. Skype for Business Server 2015 is already deployed and integrated with Exchange Online. Users can see presence in Outlook Web App and schedule Skype meetings from Outlook Web App. Trey Research is deploying Skype for Business Server 2015 Enterprise Voice and would like to retain voice mail capabilities in Exchange Online. You need to enable voice mail for Enterprise Voice users. 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 the in the correct order.

Actions

Steps to perform in sequence

Create Exchange Online UM dial plan.
Modify Exchange Online UM dial plan.
Modify UM settings on users' Exchange mailbox.
Delete Exchange Online UM dial plan.
Disable UM on users' Exchange mailbox.
Enable UM on users' Exchange mailbox.

➤
➡



Answer:

Explanation:

Actions

Steps to perform in sequence

Create Exchange Online UM dial plan.
Modify Exchange Online UM dial plan.
Modify UM settings on users' Exchange mailbox.
Delete Exchange Online UM dial plan.
Disable UM on users' Exchange mailbox.
Enable UM on users' Exchange mailbox.

➤
➡

Modify Exchange Online UM dial plan.
Enable UM on users' Exchange mailbox.
Modify UM settings on users' Exchange mailbox.

⬆
⬇

NEW QUESTION 185

You support a customer whose network environment includes Skype for Business Server 2015 servers, an Active Directory Domain Services (AD DS) domain, and Microsoft Exchange 2013. All client computers have the Skype for Business client and Outlook 2016 installed.

You deploy USB-tethered Skype for Business Phone Edition devices for a group of users. Users are able to access all dialing features, calendar, and contacts.

Users report that the Call Logs feature is unavailable for them. You need to make Call Logs available to all users.

What should you do?

- A. Configure the Exchange Unified Messaging server role to use the Exchange organization.
- B. Add an Exchange Autodiscover SRV record to the company DNS servers.
- C. Configure the Skype for Business Phone Edition devices to use only PIN authentication.
- D. Configure Skype for Business Phone Edition devices to use only client certificate authentication.

Answer: A

NEW QUESTION 186

Litware, Inc is an international air transportation company that has 700 employees. They use contractors for all back office functions.

Litware, Inc has an on-premises Skype for Business Server 2015 deployment used by all users. The company acquires a new Office 365 tenant and would like to move all non-permanent employees to Exchange Online.

You need to ensure that users can access Outlook Voice Access and receive voice mails from callers. Which Skype for Business Management cmdlet should you run?

- A. New-CsSipProxyCustom
- B. New-CsSipDomain
- C. New-CsExUmContact
- D. New-CsVoicemailReroutingConfiguration

Answer: C

Explanation: References:
<https://docs.microsoft.com/en-us/powershell/module/skype/new-csexumcontact?view=skype-ps>

NEW QUESTION 188

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 191

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 196

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 Sever 2015 pools.

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

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

Answer: C

NEW QUESTION 197

Your company's corporate telephone environment is a combination of Skype for Business Server 2015 and traditional Private Branch Exchange (PBX). The sites are connected by a Multiprotocol Label Switching (MPLS) topology. There are 6,500 users enabled for Skype for Business Server 2015 video conferencing. You use the RTVideo and H.264 codecs for video conferencing. On average, 12 percent of users are on a video conference at any one time. The company plans to add additional bandwidth for all sites.

The company is concerned about the quality of video conferences.

As part of a bandwidth upgrade process, you model real-time transport protocol (RTP) media call flows for video communications that use both RTVideo and H.264. You have determined the average number of video conferences and estimated the bandwidth required for the video portion.

You need to allocate the additional bandwidth required when modeling real-time transport control protocol (RTCP).

Which bandwidth option in kilobits per second (Kbps) should you allocate?

- A. 10 Kbps per stream
- B. 15 Kbps per stream
- C. 10 Kbps total
- D. 30 Kbps total.

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 201

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 203

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		Answer Area
a conferencing policy		
a Contact object		
an external user access policy	●	If a user dials 112, the call must be routed to a phone extension of 0911. Technology
a normalization rule	●	If a user dials 0911, the call must be routed to the phone at the security desk. Technology
a voice policy	●	The phone at the security desk must be used to forward calls to the emergency number in the Seattle office only. Technology
an unassigned number		

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 208

Your network contains the servers shown in the following table.

Server name	Software	Role
Server1	Skype for Business Server 2015	Standard Edition server
Server2	Skype for Business Server 2015	Edge Server
Server3	Microsoft Exchange Server 2013	Mailbox server Client Access server

You configure the infrastructure to use Exchange Unified Messaging (UM). You need to create a new contact object for the UM auto attendant. What should you do?

- A. From Server1, run ocsumutil.exe.
- B. From Server1, run the New-CsExUmContact Skype for Business Management Shell cmdlet.
- C. From Server3, run the New-MailContact Exchange Management Shell cmdlet.
- D. From Server3, run the exchucutil.ps1 script.

Answer: D

Explanation: References:
<http://blog.schertz.name/2015/10/skype-for-business-and-exchange-um-integration/>

NEW QUESTION 209

You have a Skype for Business Server 2015 infrastructure that has Enterprise Voice enabled.

When a caller dials an unassigned number, the caller hears a custom message, and then the caller is forwarded to a user named User1.

Several months later, User1 leaves the company.

You need to ensure that when a caller dials an unassigned number, the caller hears the custom message and is forwarded to a user named User2.

Which Skype for Business Management Shell cmdlet should you run?

- A. Set-CsVoiceRoute
- B. Set-CsAnnouncement
- C. Set-CsUnassignedNumber
- D. Set-CsUser

Answer: B

Explanation: References: <https://technet.microsoft.com/en-us/library/gg425752.aspx>

NEW QUESTION 212

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 213

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 217

You support a Microsoft Skype for Business Server 2015 on-premises environment that includes the following: The environment supports 10,000 users and runs in High Availability mode. Some users report that they are unable to make outbound calls to specific public switched telephone network (PSTN) numbers. When they attempt to do so, the call appears to be connected but then drops with a fast busy tone, and then they receive the following error message: "Due to network error your call cannot be completed at this time. Please try your call again." You need to troubleshoot the issue. Which centralized logging scenario should you use?

- A. LYSSAndUCS
- B. HybridVoice
- C. MediaConnectivity
- D. IncomingAndOutgoingCall

Answer: D

NEW QUESTION 218

You have a Skype for Business Server 2015 infrastructure. You need to prevent the users in the legal department from forwarding calls. Other users must be able to forward calls. Which two Skype for Business Management Shell cmdlets should you run? Each correct answer presents part of the solution.

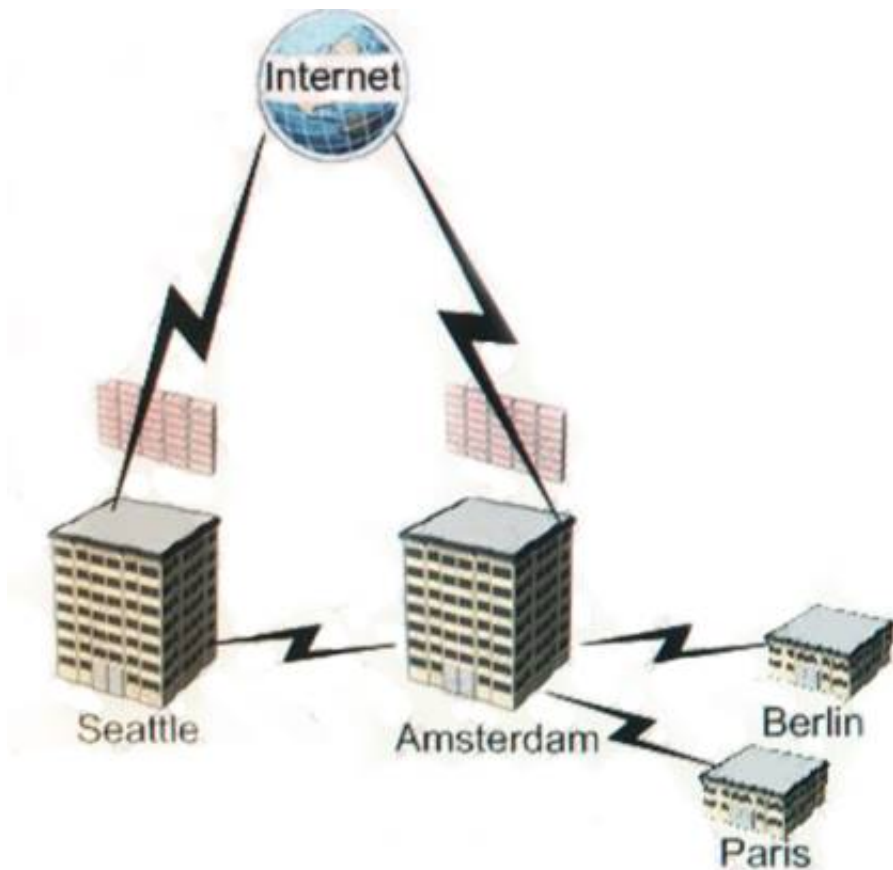
- A. Grant-CsMobilityPolicy
- B. Set-CsVoicePolicy
- C. New-CsMobilityPolicy
- D. New-CsVoicePolicy
- E. Grant-CsVoicePolicy

Answer: DE

Explanation: References: <https://technet.microsoft.com/en-us/library/gg425856.aspx>

NEW QUESTION 222

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 223

You have a Skype for Business Server 2015 infrastructure. You plan to configure Exchange Online to host voice mail. You need to perform the following actions:

- ? Configure the hosted voice mail policy.
- ? Set the external access number for the voice mail.
- ? Enable a user for Unified Messaging (UM).
- ? Configure the Contact object used for the hosted voice mail.
- ? Define a Unified Messaging (UM) IP gateway.

Where should you perform each action? To answer, select the appropriate environment for each task in the answer area.

Answer Area

Configure the hosted voice mail policy.

▼

Skype for Business Server 2015

Exchange Online

Set the external access number for the voice mail.

▼

Skype for Business Server 2015

Exchange Online

Enable a user for Unified Messaging (UM).

▼

Skype for Business Server 2015

Exchange Online

Configure the Contact object used for the hosted voice mail.

▼

Skype for Business Server 2015

Exchange Online

Define a Unified Messaging (UM) IP gateway.

▼

Skype for Business Server 2015

Exchange Online

Answer:

Explanation:

Answer Area

Configure the hosted voice mail policy.

▼

Skype for Business Server 2015

Exchange Online

Set the external access number for the voice mail.

▼

Skype for Business Server 2015

Exchange Online

Enable a user for Unified Messaging (UM).

▼

Skype for Business Server 2015

Exchange Online

Configure the Contact object used for the hosted voice mail.

▼

Skype for Business Server 2015

Exchange Online

Define a Unified Messaging (UM) IP gateway.

▼

Skype for Business Server 2015

Exchange Online

NEW QUESTION 228

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 231

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 233
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