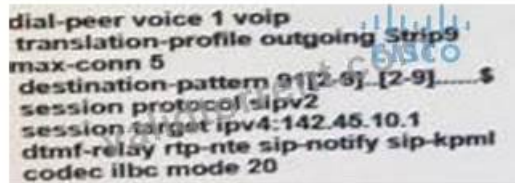


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```
dial-peer voice 1 voip
translation-profile outgoing Strip9
max-conn 5
destination-pattern 91[2-9] [2-9]...$
session protocol sipv2
session target ipv4:142.45.10.1
dtmf-relay rtp-nte sip-notify sip-kpmi
codec ilbc mode 20
```

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As you all know that practicing with the wrong preparation material will waste your valuable money and many precious study hours. So you need to choose the most proper and verified preparation material with caution. Preparation material for the Implementing and Operating Cisco Collaboration Core Technologies (350-801) exam questions from Dumpcollection helps to break down the most difficult concepts into easy-to-understand examples. Also, you will find that all the included questions are based on the last and updated 350-801 exam dumps version.

Cisco 350-801 exam, also known as Implementing and Operating Cisco Collaboration Core Technologies, is designed for professionals who want to validate their knowledge and expertise in collaborating core technologies. 350-801 exam is part of the Cisco Certified Network Professional (CCNP) Collaboration certification track and is intended for network engineers, administrators, and technicians who want to advance their careers in the field of collaboration technologies.

Cisco 350-801 Exam and Its Peculiar Features

Cisco doesn't usually reveal the details of its exams. The same works for 350-801 CLCOR. However, there are some basic details related to this certification test that are known from the previous examinees. For example, it is known that Cisco 350-801 is made up of multiple-choice questions. The candidates must complete all these questions within the allocated time of 120 minutes. The test is delivered in the English language only. It is available for the applicants at the price of \$400. You can register for your exam on the Pearson VUE website.

To prepare for the Cisco 350-801 exam, the students can opt for different options, including the official instructor-led training course. It is available on the Cisco site. Implementing Cisco Collaboration Core Technologies (CLCOR) is a paid course designed for those individuals who already have some level of experience in computer networking.

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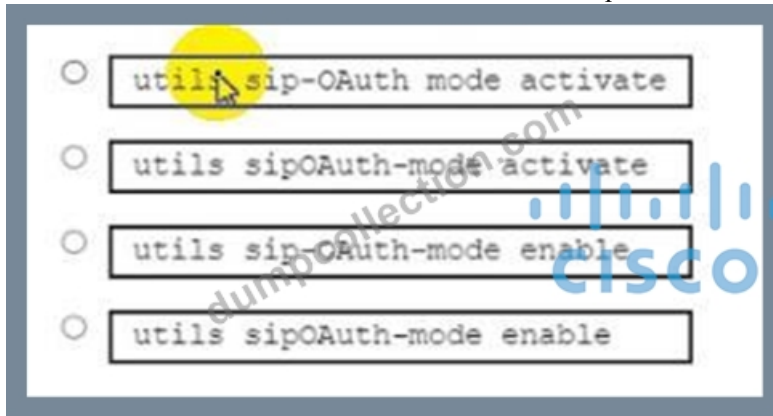
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Cisco 350-801 Exam is a certification test that validates the skills and knowledge of IT professionals in implementing and operating Cisco Collaboration Core Technologies. 350-801 exam is designed for those who wish to become certified in Cisco Collaboration technologies and are expected to perform tasks related to day-to-day operation, configuration and troubleshooting of such technologies.

Cisco Implementing and Operating Cisco Collaboration Core Technologies Sample Questions (Q56-Q61):

NEW QUESTION # 56

Which command enables SIP OAuth mode on a Cisco UCM publisher node?



- A. Option B
- B. Option A
- C. Option C
- **D. Option D**

Answer: D

NEW QUESTION # 57

Location-based call admission control is used in Cisco UCM to control the amount of bandwidth allowed between predefined locations. What must be considered when calculating the number of calls that can be made?

- A. A limit of 900 kb is the total amount of bandwidth used for the sum of calls. The use of compression does not impact the amount of calls because only the codec definition is considered when calculating the CAC in Cisco UCM
- **B. A limit of 900 kb is the total amount of bandwidth used for the sum of the media and signaling in the calls. The system allows for an actual of 20% overhead of IP packet headers. This is to compare the use of ISDN lines with no IP overhead.**
- C. A limit of 900 kb is the total amount of actual used RTP traffic so the oversubscription can be done for multiple calls adding its maximum RTP stream requirements.
- D. A limit of 900 kb is the total amount of bandwidth used for the sum of calls of the defined codec including headers and media traffic

Answer: B

NEW QUESTION # 58

What is a description of the DiffServ model used for implementing QoS?

- A. RTP traffic from voice and video calls is marked EF and placed in the same queue.
- B. AF43 has higher drop precedence than AF42 but lower drop precedence than AF41.
- **C. Voice and video calls are marked with different DSCP values and placed in different queues.**
- D. AF41 has higher drop precedence than AF42, which has higher drop precedence than AF43.

Answer: C

Explanation:

Assured Forwarding

RFC 2597 icon_popup_short.gif defines the assured forwarding (AF) PHB and describes it as a means for a provider DS domain to offer different levels of forwarding assurances for IP packets received from a customer DS domain. The Assured Forwarding PHB guarantees a certain amount of bandwidth to an AF class and allows access to extra bandwidth, if available. There are four AF classes, AF1x through AF4x. Within each class, there are three drop probabilities. Depending on a given network's policy, packets can be selected for a PHB based on required throughput, delay, jitter, loss or according to priority of access to network services.

Classes 1 to 4 are referred to as AF classes. The following table illustrates the DSCP coding for specifying the AF class with the probability. Bits DS5, DS4 and DS3 define the class; bits DS2 and DS1 specify the drop probability; bit DS0 is always zero.

Drop	Class 1	Class 2	Class 3	Class 4
Low	001010 AF11 DSCP 10	010010 AF21 DSCP 18	011010 AF31 DSCP 26	100010 AF41 DSCP 34
Medium	001100 AF12 DSCP 12	010100 AF 22 DSCP 20	011100 AF32 DSCP 28	100100 AF42 DSCP 36
High	001110 AF13 DSCP 14	010110 AF23 DSCP 22	011110 AF33 DSCP 30	100110 AF43 DSCP 38

<https://www.cisco.com/c/en/us/support/docs/quality-of-service-qos/qos-packet-marking/10103-dscpvalues.html>

NEW QUESTION # 59

An engineer troubleshoots outbound call failure on an ISDN-PRI circuit. The engineer is suspecting the "Incomplete Destination". Which debugs or commands are run in the voice gateway to troubleshoot the issue?

- A. debug isdn q931
show isdn status
- B. debug isdn q921
debug voip ecapi inout
- C. debug isdn q921
term mon
- D. debug voip ecapi inout
show controller ti

Answer: A

Explanation:

The engineer should run the following debugs or commands in the voice gateway to troubleshoot the issue:

debug isdn q931 - This debug will show the ISDN Q.931 messages that are being exchanged between the voice gateway and the ISDN switch. This can be used to identify the cause of the "Incomplete Destination" error.

show isdn status - This command will show the status of the ISDN PRI circuit. This can be used to verify that the circuit is up and running.

The other options are not correct. The debug isdn q921 command will show the ISDN Q.921 messages that are being exchanged between the voice gateway and the ISDN switch. This is not necessary for troubleshooting the issue. The term mon command will show the terminal monitor output. This is not necessary for troubleshooting the issue. The debug voip ecapi inout command will show the VoIP ECAP messages that are being exchanged between the voice gateway and the VoIP server. This is not necessary for troubleshooting the issue. The show controller ti command will show the status of the T1 controller. This is not necessary for troubleshooting the issue.

NEW QUESTION # 60

Where is Directory Connector hosted in a Cisco Webex Hybrid Services deployment?

- A. on a Cisco Expressway-C connector host server
- B. on a server in the Webex Data Center
- C. on a dedicated on-premises server
- D. on an on-premises Microsoft Active Directory server

Answer: C

Explanation:

The Cisco Directory Connector is a software application that is installed on a dedicated on-premises server. It synchronizes user identities between the on-premises directory and the Cisco Webex cloud.

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