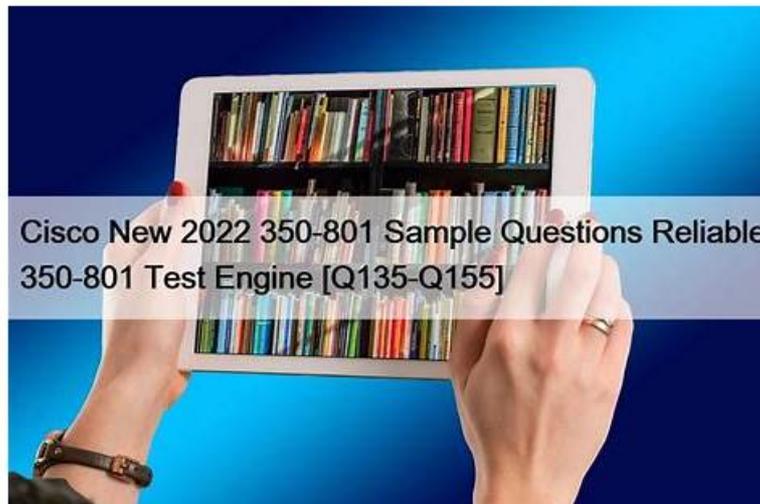


# Reliable Cisco 350-801 Braindumps Free & Certification 350-801 Dumps



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As the talent competition increases in the labor market, it has become an accepted fact that the 350-801 certification has become an essential part for a lot of people, especial these people who are looking for a good job, because the certification can help more and more people receive the renewed attention from the leaders of many big companies. So it is very important for a lot of people to gain the 350-801 Certification. We must pay more attention to the certification and try our best to gain the 350-801 certification.

Cisco 350-801 exam is a challenging test that requires a comprehensive understanding of the Cisco Collaboration Core Technologies. 350-801 exam consists of multiple-choice questions, simulation questions, and testlets. The simulation questions require candidates to perform real-world tasks on virtual equipment, while the testlets evaluate their ability to solve complex problems in a simulated environment.

## Understanding functional and technical aspects of Implementing Cisco Collaboration Core Technologies (350-801 CLCOR) Protocols, codecs, and endpoints

The following will be discussed in **CISCO 350-801 exam dumps**:

- Self provisioning
- DTMF
- Implementing Cisco Unified Communications Manager Provisioning Features
- Implementing Globalized Call Routing
- Deploy SIP endpoints
- Enabling Cisco Jabber
- Configure codec negotiations
- Call set up and tear down
- Implementing Voice Gateways
- Implement classification and marking
- Analyze traffic patterns and quality issues in converged IP networks supporting voice, video, and data traffic
- Identify the appropriate collaboration codecs for a given scenario
- Bulk Administration Tool (BAT)
- Configure and troubleshoot Cisco Unity Connection call handlers
- Configuring Calling Privileges in Cisco Unified Communications Manager
- Describing Dial Plans and Endpoint Addressing
- SDP
- Troubleshoot these elements of a SIP conversation
- Exploring Call Signaling over IP Networks

- Manual
- Troubleshoot collaboration endpoints
- Describing the Cisco Collaboration Solutions Architecture
- Exploring Codecs
- Describe how Mobile Remote Access (MRA) is used to allow endpoints to work from outside the company
- Integrating Cisco Unified Communications Manager LDAP
- Implementing MGCP Gateways
- Implementing and Troubleshooting Media Resources in Cisco Unified Communications Manager

## Understanding functional and technical aspects of Implementing Cisco Collaboration Core Technologies (350-801 CLCOR) QoS

The following will be discussed in **CISCO 350-801 Exam Dumps Pdf**:

- Describe problems that can lead to poor voice and video quality
- Describe the class models for providing QoS on a network
- CS3
- AF41
- Latency
- Implement Toll Fraud Prevention on Cisco Unified Communications Manager
- Bandwidth
- Packet loss
- CS4
- Deploy an On-Premise Cisco Jabber Client for Windows
- Configure Calling Privileges
- EAI: Configure QOS
- Implement Globalized Call Routing
- EF
- Describe the purpose and function of these DiffServ values as it pertains to collaboration
- 4/5 Class model
- Configure the Integration Between Unity Connection and Cisco UCM
- Configure and verify LLQ (class map, policy map, service policy)
- Describe and determine location-based CAC bandwidth requirements
- 8 Class model
- Manage Unity Connection Users
- Jitter

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## Certification 350-801 Dumps & 350-801 Top Questions

The 350-801 Mock Exams not just give you a chance to self-access before you actually sit for the certification exam, but also help you get an idea of the Cisco exam structure. It is well known that students who do a mock version of an exam benefit from it immensely. Some Cisco certified experts even say that it can be a more beneficial way to prepare for the Implementing and Operating Cisco Collaboration Core Technologies exam than spending the same amount of time studying.

## Cisco Implementing and Operating Cisco Collaboration Core Technologies Sample Questions (Q227-Q232):

### NEW QUESTION # 227

Which two steps should be taken to provision after the Self-Provisioning feature was configured for end users?

- **A. Plug the phone into the network.**
- B. Dial the hunt pilot extension and associate the phone to an end user.
- C. Enter settings menu on the phone and press \*,\*,# (star, star, pound).
- **D. Dial the self-provisioning IVR extension and associate the phone to an end user.**
- E. Ask the Cisco UCM administrator to associate the phone to an end user.

**Answer: A,D**

Explanation:

Self-Provisioning Phones:

When the feature is configured, you can provision a phone by doing the following:

\* Plug the phone into the network.

\* Dial the self-provisioning IVR extension.

\* Follow the prompts to configure the phone, and associate the phone to an end user. Depending on how you have configured self-provisioning, the end user may enter the user password, PIN, or an administrative authentication code.

[https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/12\\_0\\_1/systemConfig/cucm\\_b\\_system-configuration-guide-1201/cucm\\_b\\_system-configuration-guide-1201\\_chapter\\_01001000.pdf](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/12_0_1/systemConfig/cucm_b_system-configuration-guide-1201/cucm_b_system-configuration-guide-1201_chapter_01001000.pdf)

### NEW QUESTION # 228

When a phone is registered over Mobile and Remote Access, where does it register?

- A. Expressway-E
- B. Expressway-C
- C. Cisco Unified Communications Manager
- D. Cisco Unified Presence Server

**Answer: A**

Explanation:

Explanation/Reference:

Reference: [https://www.cisco.com/c/en/us/td/docs/voice\\_ip\\_comm/cucm/admin/12\\_0\\_1/systemConfig/cucm\\_b\\_system-configuration-guide-1201/cucm\\_b\\_system-configuration-guide-1201\\_chapter\\_01011010.html](https://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/admin/12_0_1/systemConfig/cucm_b_system-configuration-guide-1201/cucm_b_system-configuration-guide-1201_chapter_01011010.html)

### NEW QUESTION # 229

An administrator works with an ISDN PRI that is connected to a third-party PBX. The ISDN link does not come up; and the administrator finds that the third-party PBX uses the QSIG signalling method. Which command enables the Cisco IOS Gateway to use QSIG signalling on the ISDN link?

- A. isdn switch-type basic-ni
- B. isdn switch-type primary-qsig
- C. isdn switch-type basic-qsig
- D. isdn incoming-voice voice

**Answer: C**

### NEW QUESTION # 230

Which location must be assigned to the SIP trunk to replicate enhanced location information via a SIP trunk?

- A. shadow
- B. replica
- C. hub\_none
- D. phantom

**Answer: A**

### NEW QUESTION # 231

Refer to the exhibit.

<https://i.postimg.cc/C57TkcZG/image.png>

```

v=0
o=Cisco-SIPUA 13439 0 IN IP4 10.10.10.10
s=SIP Call
b=AS:4064
t=0 0
m=audio 0 RTP/AVP 114 9 124 113 115 0 8 116 18 101
c=IN IP4 10.10.10.10
b=TIAS:64000
a=rtpmap:114 opus/48000/2
a=fmtp:114 maxplaybackrate=16000;sprop-
maxcapture=16000;maxaveragebitrate=64000;stereo=0;sprop-stereo=0;usedtx=0
a=rtpmap:9 G722/8000
a=rtpmap:124 ISAC/16000
a=rtpmap:113 AMR-WB/16000
a=fmtp:113 octet-align=0;mode-change-capability=2
a=rtpmap:115 AMR-WB/16000
a=fmtp:115 octet-align=1;mode-change-capability=2
a=rtpmap:0 PCMU/8000
a=rtpmap:8 PCMA/8000
a=rtpmap:116 iLBC/8000
a=fmtp:116 mode=20
a=rtpmap:18 G729/8000
a=fmtp:18 annexb=yes

```

A call is failing to establish between two SIP Devices The called device answers with these SOP Which SOP parameter causes issue?

- A. The media stream is set to send only
- **B. The RTP port is set to 0.**
- C. The payload for G.711ulaw must be 18.
- D. The calling device did not offer aptime value

**Answer: B**

Explanation:

The RTP port is used to send and receive media packets during a call. If the RTP port is set to 0, the called device will not be able to send or receive media packets, and the call will fail.

The other options are not correct because:

- \* A. The calling device did not offer aptime value: Theptime value is used to specify the amount of time between each media packet. If the calling device does not offer aptime value, the called device will use the default value of 20 milliseconds.
- \* B. The media stream is set to sendonly: The media stream is set to sendonly when the called device is only able to send media packets, and not receive them. This is not a problem, and the call will still succeed.
- \* C. The payload for G.711ulaw must be 18: The payload for G.711ulaw is the type of media packet that is used. The payload must be set to 18 for G.711ulaw, but this is not a problem, and the call will still succeed.

## NEW QUESTION # 232

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