

300-815サンプル問題集 & 300-815日本語版問題解説

Cisco Unified Communications Manager Dialed Number Analyzer

DNA Analysis Output Save the Displayed Output Save

Cisco Unified Communications Manager Dialed Number Analyzer Results

Expand All Collapse All

- Results Summary
 - Calling Party Information
 - Calling Party = 4125551212
 - Partition =
 - Device CSS =
 - Line CSS = ONNET-CSS
 - AAR Group Name =
 - AAR CSS =
 - Dialed Digits = 914125550000
 - Match Result = RouteThisPattern
 - Matched Pattern Information
 - Pattern = 9.1[2-9]XX[2-9]XXXXXX
 - Partition =
 - Time Schedule =
 - Called Party Number = 14125550000
 - Time Zone = Etc/GMT
 - End Device = HQ-RL
 - Call Classification = OffNet
 - InterDigit Timeout = NO
 - Device Override = Disabled
 - Outside Dial Tone = NO
- Call Flow
 - Alternate Matches

- Call Flow
- Route Pattern : Pattern=9.1[2-9]XX[2-9]XXXXXX
- Positional Match List =
- DialPlan =
- Route Filter
 - Filter Name =
 - Filter Clause =
- Require Forced Authorization Code= NO
- Authorization Level= 0
- Require Client Matter Code=NO
- Call Classification =
- PreTransform Calling Party Number = 4125551212
- PreTransform Called Party Number = 914125550000
- Calling Party Transformations
 - External Phone Number Mask = NO
 - Calling Party Mask =
 - Prefix =
 - CallingLineId Presentation = Default
 - CallingName Presentation = Default
 - Calling Party Number = 4125551212
- ConnectedParty Transformations
 - ConnectedLineId Presentation = Default
 - ConnectedName Presentation = Default
- Called Party Transformations
 - Called Party Mask =
 - Discard Digits Instruction = PreDot
 - Prefix =
 - Called Number = 14125550000
- Route List : Route List Name= HQ-RL
 - RouteGroup: RouteGroup Name= Standard Local Route Group
 - PreTransform Calling Party Number = 4125551212
 - PreTransform Called Party Number = 914125550000
 - Calling Party Transformations
 - External Phone Number Mask = Default
 - Calling Party Mask =
 - Prefix =
 - Calling Party Number = 4125551212
 - Called Party Transformations
 - Called Party Mask =
 - Discard Digits Instruction =
 - Prefix =
 - Called Number = 914125550000
 - Alternate Matches
 - Note: Information Not Available

2026年Japancertの最新300-815 PDFダンプおよび300-815試験エンジンの無料共有: <https://drive.google.com/open?id=1Z3gBaB5RG9od1x8DkM4JNcXfx1OmlHlg>

300-815テスト教材は、主に3つの学習モード（Pdf、オンライン、ソフトウェア）をそれぞれ使用します。その中でも、ソフトウェアモデルはコンピューターユーザー向けに設計されており、ユーザーがWindowsインターフェイスを使用して学習の300-815テスト準備を開くことができます。ユーザーが読むのに便利です。300-815テスト教材には、オンライン学習プラットフォームとは異なる最大の利点があります。300-815クイズトレ

ントは、クライアントにログインして同時に詳細を学習することができ、人々は300-815あらゆる種類の電子機器のテスト準備。

Cisco Advanced Call Control and Mobility Services試験の実装は、60~70の質問で構成される90分のテストです。この試験は英語で実施され、オンラインの提案や対面テストなど、さまざまな形式で利用できます。この試験では、Cisco Unified Communications Manager、Cisco Unified Communications Manager Express、Cisco Unity Connection、およびCisco Unified Communications Manager IM&Pをよく理解する必要があります。

>> 300-815サンプル問題集 <<

真実的な300-815サンプル問題集試験-試験の準備方法-100%合格率の 300-815日本語版問題解説

試験を受けることでCisco認定を取得することを期待する人が増えています。ただし、多くの人にとって試験は非常に困難です。特に正しい学習教材を選択せずに適切な方法を見つけた場合、300-815試験に合格して関連する認定を取得することはより困難になります。関連する認定を効率的な方法で取得したい場合は、当社の300-815学習教材を選択してください。弊社の300-815学習教材が試験に合格し、簡単に認定を取得するのに役立ちます。

Cisco Implementing Cisco Advanced Call Control and Mobility Services 認定 300-815 試験問題 (Q192-Q197):

質問 # 192

Refer to the exhibit. The Cisco Unified Border Element receives an INVITE matching inbound dial peer 5002. The outbound dial peer supports only iLBC, and a Local Transcoding Interface is allocated. Based on the configuration and SDP from the INVITE message, which codec is chosen by Cisco Unified Border Element for the inbound call leg?

- A. G.729r8
- **B. G.711 U-law**
- C. G.729br8
- D. G.711 A-law

正解: B

質問 # 193

Which configuration element of a hunt group allows for changing Calling Party Transformations settings?

- **A. hunt pilot**
- B. line group
- C. route group
- D. hunt list

正解: A

質問 # 194

Cisco SIP IP telephony is implemented on two floors of your company. Afterward, users report intermittent voice issues in calls established between floors. All calls are established, and sometimes they work well, but sometimes there is one-way audio or no audio. You determine that there is a firewall between the floors, and the administrator reports that it is allowing SIP signaling and UDP ports from 20000 to 22000 bidirectionally.

What are two possible solutions? (Choose two.)

- A. Go to System Parameters in Cisco Unified Communications Manager and change the range of media ports to 20000-22000.
- **B. Ask the firewall administrator to change the range of UDP ports to 16384-32767.**
- C. Ask the firewall administrator to change the ports to TCP.
- **D. Go to the SIP profile assigned to these IP phones in Cisco Unified CM and change the range of media ports to 16384-**

32767

- E. Go to the SIP profile assigned to these IP phones in Cisco Unified CM and change the range of media ports to 20000-22000.

正解: B、D

質問 # 195

A network engineer designs a new dial plan and wants to block a certain range of numbers (8135100 through 8135105). What is the most specific route pattern that can be configured to block only the numbers in this range?

- A. 813510[012345]
- B. 813510[12345]
- C. 81XXXXX
- D. 813510[

さらに、Japancert 300-815ダンプの一部が現在無料で提供されています: <https://drive.google.com/open?id=1Z3gBaB5RG9od1x8DkM4JNcXfx1OmnlHg>