将Emergency Responder与CUCM集成

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配置Cisco Emergency Responder 创建CER Web用户(可选) 配置组设置 配置电话设置 配置服务器设置 确定Cisco Unified Communications Manager集群 创建现场警报(可选)

简介

本文档介绍如何使用交换机端口电话跟踪将Cisco Emergency Responder与Cisco Unified Communications Manager(CUCM)集成。

先决条件

要求

Cisco 建议您了解以下主题:

- 思科应急响应器(CER)
- CUCM
- 呼叫路由
- 基本简单网络管理协议(SNMP)知识

使用的组件

本文档中的信息基于以下软件版本:

- CER 11.5版
- CUCM版本11.5

本文档中的信息都是基于特定实验室环境中的设备编写的。本文档中使用的所有设备最初均采用原 始(默认)配置。如果您的网络处于活动状态,请确保您了解所有命令的潜在影响。

背景信息

本文档由Cisco TAC工程师编写,并不取代参考配置和设计指南的需要。

带有示例的节表示可能设置的示例值,仅供参考。专用网络的配置取决于专用实体的需求及其内部 设计准则。

在CER与CUCM的集成完成后,有必要与本地公共安全应答点(PSAP)协调紧急呼叫测试。

如果PSAP测试成功,但在测试后对配置进行了更多更改,则必须在完成其他更改后与本地PSAP协 调安排紧急呼叫的测试。简而言之,只要进行了可能影响呼叫路由的更改,就测试紧急呼叫。

CER基础知识

911

用于联系特定国家/地区的公共紧急机构(如警察、消防和医疗机构)的简单、容易记忆的号码。并 非所有国家都使用911拨打紧急电话,因此请注意使用正确的紧急号码。为简单起见,本文档只能 引用911作为紧急电话号码。

PSAP

公共资助设施,用于路由和发送紧急呼叫。 该组织由现场操作员组成,他们应答911呼叫并确定需 要派遣哪个紧急机构(警察、消防等)。

自动位置标识(ALI)

在主叫方的PSAP和地址/位置自动显示。 话务员可以使用此信息定位进行911呼叫的人员。

ALI数据库(ALI-DB)

电话公司有一个用户数据库,将电话号码与姓名和地址进行匹配。当呼叫到达911网络时,此数据 库用于提取与主叫电话号码匹配的地址,并使PSAP话务员更容易找到您。

自动号码识别(ANI)

这是主叫方号码的另一个术语。ALI与ANI的不同之处在于,ALI包含有关呼叫者位置的更多信息。

紧急响应位置(ERL)

发出紧急呼叫的区域。这未必是紧急事件的发生地。如果紧急呼叫者报告一般紧急事件,则实际紧 急事件可能位于不同区域。在CER中,将交换机端口和电话分配给ERL,并且ERL定义包括ALI数据 。 ALI数据由PSAP用于确定发出911呼叫的呼叫者的位置。

紧急位置标识号(ELIN)

PSAP可用于回叫紧急呼叫者的电话号码。如果紧急呼叫被断断或如果PSAP在有意终止紧急呼叫后需要其他信息,则PSAP可能需要呼叫ELIN。 ELIN是ERL配置的一部分。

呼叫流

以下是您可以通过CER使用的不同呼叫流:

911呼叫



现场警报



配置

配置Cisco Unified Communications Manager

- 创建分区
- 创建呼叫搜索空间
- 为电话分配分区和CSS
- 创建CTI路由点(CTI RP)
- 创建 CTI 端口
- 创建路由模式
- 创建转换模式
- 配置PSAP回叫
- 创建JTAPI用户
- 配置SNMP服务

创建分区

创建两个分区。导航到呼叫路由 > 控制类 > 分区:

- 911_PT
- 电话(_P)

- Partitic	Partition Information				
To enter multiple partitions, use one line for each partition entry. You can enter up to 75 partitions; the names and descriptions can have up to a total of 1475 characters. The partition name cannot exceed 50 characters. Use a comma (',') to separate the partition name and description on each line. If a description is not entered, Cisco Unified Communications Manager uses the partition name as the description. For example: << partitionName >> , << description >> CiscoPartition, Cisco employee partition DallasPartition					
Name*	911_PT Phones PT				

💊 注意:

911_PT是存储您的紧急号码的分区。如果已有用于紧急号码的分区,您可以继续使用之前配置的分区。只要在本文档中提到911_PT,就只需替换预配置分区的名称。

Phones_PT是与所有内部目录号码(DN)关联的分区。如果您已经有内部DN的分区,您可以继续使用之前配置的分区。只要在本文档中提到Phones_PT,就只需替换预配置分区的名称。

创建呼叫搜索空间

创建两个呼叫搜索空间。导航到呼叫路由>控制类>呼叫搜索空间:

- 911_CSS:可以包括911_PT和Phones_PT
- Phones_CSS:可以只包括Phones_PT



为IP电话分配分区和CSS

- IP电话上的DN需要与Phones_PT分区相关联。
- 电话需要使用Phones_CSS来调用转换模式(9.911和911)。

▲ 警告:如果设置更加复杂,则IP电话需要能够拨打911和/或9.911转换模式(稍后配置),而 CTI路由点和CTI端口需要能够呼叫IP电话。

Asso	ciation	Phone Type	
	Modify Button Items	Product Type: Cisco 7861 Device Protocol: SIP	
1	•771 Line [1] - 5003 in Phones PT		
	Unassigned Associated Items	Real-time Device Status	
2	The Line [2] - Add a new DN	Registration: Registered with	Cisco Unified Communications Manager 14.48.62.152
3	Add a new BLF Directed Call Park	IPv4 Address: <u>10.48.62.13</u> Active Load ID: -1-3	18
4	Call Park	Inactive Load ID: sip78xx.10-3-1-1	12
5	Call Pickup	Download Status: None	
6	CallBack		
7	Conference List	Device Information	
8	Do Not Disturb	Device is Active	
9	Forward All	Device is trusted	
10	Group Call Pickup	MAC Address*	38ED18552E6E
11	Hunt Group Logout	Description	Auto 5003
12	Intercom [1] - Add a new Intercom	Device Pool*	Default 👻
13	Malicious Call Identification	Common Device Configuration	< None >
14	Meet Me Conference	Phone Button Template*	Universal Device Template Button Layout
15	Mobility	Softkey Template	< None >
16	Other Pickup	Common Phone Profile*	Standard Common Phone Profile
17	Quality Reporting Tool	Calling Search Space	
18	Redial	Coming Search Space	Phones_CSS

创建CTI路由点

- 911和912 CTI RP需要与911_PT分区关联并使用911_CSS。
- 911和9.911转换模式需要能够到达911 CTI RP。
- 911 CTI RP需要能够调用为CER配置的路由模式。
- 913 CTI RP还需要与911_PT分区关联并使用911_CSS。
- PSAP回叫转换模式需要能够到达913 CTI RP。

	Device Name	Description	Device Pool	Calling Search Space	Partition	Extension
	CER 911	CTI RP for Primary CER Server	Default	911_CSS	911 PT	911
	CER 912	CTI RP for Seconday CER Server	Default	911_CSS	911 PT	912
	CER 913	CTI RP for PSAP Callbacks	Default	911 CSS	911 PT	913XXXXXXXXXXXX

CTI路由点 — 911

- 对于内部和外部呼叫, Forward Busy、Forward No Answer、Forward No Coverage、 Forward Unregistered和Forward on Failure需要目标为912(如果您有辅助CER服务器)、现 场安全号码或路由模式,以便911呼叫仍可以传到PSAP。
- 对于本文档中的示例,911 CTI RP会在需要时将呼叫转发到912。确保呼叫转移和呼叫代答设置的呼叫搜索空间使用911_CSS,以便转发的呼叫可以到达912 CTI RP。

all Forward and Call Pickup Settings				
	Voice Mail	Destination		Calling Searc
Calling Search Space Activation Policy			Use System Default	-
Forward All	i or		< None >	•
econdary Calling Search Space for Forward	All		< None >	•
orward Busy Internal	🖾 or	912	911_CSS	•
orward Busy External	i or	912	911_CSS	•
orward No Answer Internal	🖾 or	912	911_CSS	-
rward No Answer External	I or	912	911_CS5	
rward No Coverage Internal	🖾 or	912	911_CSS	•
rward No Coverage External	i or	912	911_CSS	•
rward on CTI Failure	🖾 or	912	911_CSS	-
rward Unregistered Internal	🖾 or	912	911_CSS	-
orward Unregistered External	🖾 or	912	911_CSS	
Answer Ring Duration (seconds)				
all Pickup Group < None >		*		

CTI路由点—912

- 对于内部和外部呼叫, Forward Busy、Forward No Answer、Forward No Coverage、 Forward Unregistered和Forward on Failure需要将呼叫路由到现场安全号码或路由模式,以便 911呼叫仍然可以传到PSAP。
- 对于本文档中的示例,如果需要,912 CTI RP会将呼叫转发到用于默认ERL的路由模式。确保呼叫转移和呼叫代答设置的呼叫搜索空间使用911_CSS,以便转发的呼叫可以到达路由模式。

Call Forward and Call Pickup Settings				
	Voice Mail	Destination		Calling Search Space
Calling Search Space Activation Policy			Use System Default	
Forward All	C or		< None >	•
Secondary Calling Search Space for Forward All			< None >	•
Forward Busy Internal	🖾 or	10911	911_CSS	*
Forward Busy External	III or	10911	911_CSS	
Forward No Answer Internal	🖾 or	10911	911_CSS	•
Forward No Answer External	m or	10911	911_CSS	•
forward No Coverage Internal	🖾 or	10911	911_CSS	•
orward No Coverage External	🖾 or	10911	911_CSS	•
orward on CTI Failure	🖾 or	10911	911_CSS	
Forward Unregistered Internal	i or	10911	911_CSS	•
Forward Unregistered External	🖾 or	10911	911_CSS	
o Answer Ring Duration (seconds)				
Call Pickup Group < None >		•		

CTI路由点— 913

- 对于内部和外部呼叫, Forward Busy、Forward No Answer、Forward No Coverage、 Forward Unregistered和Forward on Failure需要将呼叫路由到现场安全号码。
- 对于本文档中的示例,913 CTI RP将呼叫转发到60003,这是现场安全号码。确保呼叫转移和 呼叫代答设置的呼叫搜索空间使用可以到达现场安全号码的CSS。

Call Forward and Call Pickup Settings				
	Voice Mail	Destination		c
Calling Search Space Activation Policy			Use System Default	
Forward All	or		< None >	
Secondary Calling Search Space for Forward All			< None >	
Forward Busy Internal	i or	60003	911_CSS	
Forward Busy External	i or	60003	911_CSS	
Forward No Answer Internal	i or	60003	911_CSS	
Forward No Answer External	i or	60003	911_CSS	
Forward No Coverage Internal	i or	60003	911_CSS	
Forward No Coverage External	i or	60003	911_CSS	
Forward on CTI Failure	i or	60003	911_CSS	
Forward Unregistered Internal	C or	60003	911_CSS	
Forward Unregistered External	i or	60003	911_CSS	
No Answer Ring Duration (seconds)				
Call Pickup Group < None >		•		

创建 CTI 端口

- CTI端口仅用于电话现场警报。
- CTI端口需要能够呼叫现场报警号码(可以是内部或外部号码,只要呼叫到达现场安全人员)。
- CTI端口DN必须按顺序排列。
- CER仅支持G.711,因此CTI端口和电话之间用于现场安全性的区域关系不得低于64 kbps。

Association	Phone Type		
1 <u>Line [1] - 60010 (no partition)</u>	Product Type: CTI Port		
2 ems Line [2] - Add a new DN	Device Protocol: SCCP		
3 Intercom [1] - Add a new Intercom	Real-time Device Status		
	Registration: Unknown		
	IPv4 Address: None		
	Device Information		
	Device is Active		
	Device is trusted		
	Device Name*	CER_CTIPort_1	
	Description		
	Device Pool*	Default	✓ <u>View Details</u>
	Common Device Configuration	< None >	✓ <u>View Details</u>
	Common Phone Profile*	Standard Common Phone Profile	✓ <u>View Details</u>
	Calling Search Space	911_CSS	-

提示:对于内部呼叫,您可以在屏幕上显示紧急呼叫,或您喜欢的其他内容。这样,人们就能 在接听电话之前了解其紧迫性。对于外部呼叫,您可以将外部电话号码掩码配置为所有人都可 以识别的号码。

Display (Internal	Emergency Call	Display text for a line appearance is intended for displaying text such
Caller ID)	as a name instead of a directory num the proper identity of the caller.	ber for internal calls. If you specify a number, the person receiving a call may not see
ASCII Display	Emergency Call	
ine Text Label		
and foxe Edbor		
SCII Line Text abel		
ixternal Phone Jumber Mask		
/isual Message Maiting Indicator Policy*	Use System Policy	
1onitoring Calling	< None >	×

创建路由模式

默认ERL路由模式

- 需要在911_PT中。
- 您可以将Calling Party Transform Mask设置为ERL中ELIN的编号。无论如何,CER可以将其 更改为ELIN。
- 将Discard Digits设置为PreDot。

在本示例中,RTP位置的默认ERL相同

Route Partition 911_PT Description Route Pattern used by CER for RTP Location Numbering Plan Not Selected Route Filter < None > MLPP Precedence* Default Apply Call Blocking Percentage	Route Pattern*		10.911		
Description Route Pattern used by CER for RTP Location Numbering Plan Not Selected Route Filter < None > MLPP Precedence* Default Image: Second Control Profile None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option Image: Route this pattern Image: Second Control Profile None > Call Classification* OffNet External Call Control Profile < None > Image: Require Forced Authorization Code Allow Overlap Sending Authorization Level* 0 Image: Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Default Image: Use Calling Party's External Phone Number Mask Calling Party Transform Mask Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Number Mask Calling Party Number Mask Calling Party Number Mask Default Calling Party Number Mask Calling Party Number Mask	Route Partition		911_PT		
Numbering Plan Not Selected Route Filter < None > MLPP Precedence* Default Image: Apply Call Blocking Percentage Image: Apply Call Blocking Percentage Resource Priority Namespace Network Domain < None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option Image: Require Class if (ation * OffNet Image: Require Forced Authorization Code Authorization Level* O Image: Require Client Matter Code Image: Require Client Matter Code Image: Image: Summer (used by Emergency Call Handler) Calling Party Transformations Image: Use Calling Party's External Phone Number Mask Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Number Type* Cisco CallManager Calling Party Number Type* Cisco CallManager Connected Ine ID Presentation* Default	Description		Route Pattern used by CER for RTP Location		
Route Filter < None > MLPP Precedence* Default MLPP Precedence* Default Resource Priority Namespace Network Domain < None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option Image: Route this pattern Image: Route Option Image: Route this pattern Image: Route Option Image: Route Option Image: Route Cleant Matter Code Image: Route Cleant Matter Code Image: Require Client Matter Code Image: Route Option Image: Require Client Matter Code Image: Route Option Image: Route Claing Party Transform Mask Image: Route Option Image: Route Claing Party Transform Mask Image: Route Option Calling Name Present	Numbering Plan		Not Selected		
MLPP Precedence* Default Apply Call Blocking Percentage Resource Priority Namespace Network Domain < None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option Route this pattern Block this pattern Allow Device Override Provide Outside Dial Tone Allow Overlap Sending Urger Allow Device Override Provide Outside Dial Tone Allow Overlap Sending Urger Require Forced Authorization Code Authorization Level* 0 Require Forced Authorization Code Authorization Level* 0 Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Number Type* Cisco CallManager Calling Party Transformations Connected In ID Presentation* Default Connected Name	Route Filter		< None >		
□ Apply Call Blocking Percentage Resource Priority Namespace Network Domain < None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option	MLPP Precedence*		Default		
Resource Priority Namespace Network Domain < None > Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option Route Option Route this pattern Block this pattern Block this pattern Block this pattern Call Classification* OffNet Require Coverride Provide Outside Dial Tone Allow Overlap Sending Urger Require Forced Authorization Code Authorization Level* 0 Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Transformations Connected Party Transformations* Default Connected Name Presentation* Default Connected Name Presentation* Default Connected Name Presentation* Default Calling Party Transformations Connected Name Presentation* Default Calling Party Transformations Connected Name Presentation* Default Connected Name Presentation Connected	Apply Call Blocking Percent	tage			
Route Class* Default Gateway/Route List* SIPTrunkPSTN Route Option	Resource Priority Namespace I	Network Domain	< None >		
Gateway/Route List* SIPTrunkPSTN Route Option	Route Class*		Default		
Route Option Route this pattern Block this pattern No Error Call Classification* OffNet External Call Control Profile None > Allow Device Override Provide Outside Dial Tone Allow Device Override Default Calling Party Transformations Calling Party Numbering Plan* Cisco CallManager Calling Party Transformations* Connected Darty Transformations* Connected Line ID Presentation* Default Connected Name Presentation*	Gateway/Route List*		SIPTrunkPSTN		
Block this pattern No Error Call Classification* OffNet External Call Control Profile None > Allow Device Override Provide Outside Dial Tone Allow Device Override Internation Calling Party Transformation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Connected Name Presentation* Default Connected Name Presentation* Default	Route Option		Route this pattern		
Call Classification * OffNet External Call Control Profile None > Allow Device Override Provide Outside Dial Tone Allow Overlap Sending Urger Require Forced Authorization Code Authorization Level* Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default Connec			Block this pattern No Error		
External Call Control Profile < None > Allow Device Override Provide Outside Dial Tone Allow Overlap Sending Urger Require Forced Authorization Code Authorization Level* 0 Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Conne	Call Classification*	OffNet	-		
 Allow Device Override Provide Outside Dial Tone Allow Overlap Sending Urger Require Forced Authorization Code Authorization Level* 0 Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Name Presentation* Default Cisco CallManager Cisco CallManager Connected Party Transformations* Connected Line ID Presentation* Default Connected Name Presentation* 	External Call Control Profile	< None >	-		
Require Forced Authorization Code Authorization Level* 0 Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Transformations Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation*	Allow Device Override	rovide Outside D)ial Tone 🔲 Allow Overlap Sending 🔲 Urgent		
Authorization Level*	Require Forced Authorizati	on Code			
Require Client Matter Code Is an Emergency Services Number (used by Emergency Call Handler) Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Transformations Connected Party Transformations Connected Line ID Presentation*	Authorization Level*	0			
Calling Party Transformations Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default Connected Name Presentation* Default	Require Client Matter Code Is an Emergency Services	e Number (used b ₎	y Emergency Call Handler)		
Use Calling Party's External Phone Number Mask Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default	Calling Party Transformations				
Calling Party Transform Mask Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default Connected Name Presentation* Default	Use Calling Party's Externa	al Phone Number	Mask		
Prefix Digits (Outgoing Calls) Calling Line ID Presentation* Default Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default	Calling Party Transform Mask				
Calling Line ID Presentation* Default Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Calling Party Transformations Connected Party Transformation* Default Connected Name Presentation* Default Connected Name Presentation* Default	Prefix Digits (Outgoing Calls)				
Calling Name Presentation* Default Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default	Calling Line ID Presentation*	Default			
Calling Party Number Type* Cisco CallManager Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default	Calling Name Presentation*	Default			
Calling Party Numbering Plan* Cisco CallManager Connected Party Transformations Connected Line ID Presentation* Default Connected Name Presentation* Default	- Calling Party Number Type*	Cisco CallMana			
Connected Party Transformations Connected Line ID Presentation [*] Default Connected Name Presentation [*] Default	Calling Party Numbering Plan* Cisco CallMana		ager 🗸		
Connected Line ID Presentation* Default Connected Name Presentation* Default	Connected Party Transformat	tions			
Connected Name Presentation* Default					
	Connected Name Presentation* Default		•		
Called Party Transformations					
Discard Digits PreDot 🗸	Discord Disits				

所有其他ERL路由模式

- 需要在911_PT中。
- 您可以将Calling Party Transform Mask设置为ERL中ELIN的编号。无论如何,CER可以将其 更改为ELIN。
- 将Discard Digits设置为PreDot。

	110.911
	911 PT
	Route Pattern used by CER for SJ Location
	Not Selected
	< None >
	Default
1200	
Network Domain	< None >
	Default
	SIPTrunkPSTN2
	Route this pattern
	Block this pattern No Error
OffNet	•
< None >	•
Provide Outside D	Dial Tone 🔲 Allow Overlap Sending 🔲 Urge
on Code	
0	
-	
e Number (used by	y Emergency Call Handler)
e Number (used by	y Emergency Call Handler)
e Number (used by	y Emergency Call Handler)
e Number (used by	y Emergency Call Handler) ^r Mask
e Number (used by	y Emergency Call Handler) r Mask
e Number (used by al Phone Number	y Emergency Call Handler) r Mask
e Number (used by al Phone Number Default	y Emergency Call Handler) r Mask
e Number (used by al Phone Number Default Default	y Emergency Call Handler)
e Number (used by al Phone Number Default Default Cisco CallMana	y Emergency Call Handler)
e Number (used by al Phone Number Default Default Cisco CallMana * Cisco CallMana	y Emergency Call Handler)
e Number (used by al Phone Number Default Default Cisco CallMana Cisco CallMana	y Emergency Call Handler)
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e Number (used by befault Default Cisco CallMana Cisco CallMana tions tions Default Default Default	y Emergency Call Handler) r Mask ager ager v ager v v v v v v v v v v v v v v v v v v v
e Number (used by Default Default Cisco CallMana Cisco CallMana tions on* Default Default	y Emergency Call Handler) r Mask ager ager
	tage Network Domain OffNet < None > Provide Outside D on Code

911和9.911转换模式之间的唯一区别是9.911转换模式上的丢弃PreDot。

Translation Pattern	911
Partition	Phones_PT
Description	911 TP for CER
Numbering Plan	< None > v
Route Filter	< None > v
MLPP Precedence*	Default 🗸
Resource Priority Namespace Network Domain	< None > 🔹
Route Class*	Default 🗸
Calling Search Space	911_CSS 👻

Translation Dattorn				
Translation Pattern		9.911		
Partition		Phones_PT		
Description		911 TP for CER		
Numbering Plan		< None >		
Route Filter		< None >		
MLPP Precedence*		Default		
Resource Priority Namespace Network Domain		< None >		
Route Class*		Default		
Calling Search Space		911_CSS		
Use Originator's Calling Sea	rch Space			
External Call Control Profile		< None >		
Route Option		Route this pattern		
		Block this pattern No Error		
Provide Outside Dial Tone				
Urgent Priority	Vrgent Priority			
Do Not Wait For Interdigit Ti	imeout On Subs	sequent Hops		
Route Next Hop By Calling F				
Is an Emergency Services N	umber (used by	/ Emergency Call Handler)		
	,	,,		
Calling Party Transformations				
Use Calling Party's External	Phone Number	Mask		
Calling Party Transform Mask				
Prefix Digits (Outgoing Calls)				
Calling Line ID Presentation*	Default 🗸			
Calling Name Presentation*	Default 🗸			
Calling Party Number Type*	Cisco CallMana	iger 🗸 🗸		
Calling Party Numbering Plan*	Cisco CallManager 🗸			
Connected Party Transformation	ons			
Connected Line ID Presentation	Default	▼		
Connected Name Presentation*	Default	· •		

Called Party Transformations

Discard Digits

PreDot

- JTAPI用户必须是应用用户(而不是最终用户)。
- JTAPI用户需要将CTI路由点和CTI端口与其关联。否则,这些CTI设备无法注册,呼叫也无法 工作。
- 需要将JTAPI用户添加到Standard CTI Allow Calling Number Modification和Standard CTI Enabled组。

Application User Informat	ion	
User ID*	CER	Edit Credential
Password	•••••	
Confirm Password	•••••	
Digest Credentials		
Confirm Digest Credentia	s	
BLF Presence Group*	Standard Presence group	~
Accept Presence Sub	scription	
Accept Out-of-dialog	REFER	
Accept Unsolicited No	tification	
Accept Replaces Head	Jer	
Device Information		
Available Devices	ATA34D850190314	
Available Devices	ATADBED18021A	Device Association
	Auto-registration Template	
	ExtConnDevice	Find more Route Points
	∠	
Controlled Devices	CER 911	
	CER_912	
	CER_913	
	SEP001BD5122EB5	+
Available Profiles	8841DP	
	~~	+
CTI Controlled Device Pr	ofiles	
		*
		*
		Ŧ
CAPF Information		
Associated CAPF Profile	15	A
		View Details
Permissions Information		
Groups Standard CTLE	nabled	
Standard CTI A	llow Calling Number Modificati	Add to Access Control Group
		Remove from Access Control Group
	View Details	
Roles Standard CTI A Standard CTI E	llow Calling Number Modification	

配置SNMP配置

- 确保在所有Callmanager上激活并启动SNMP服务(Cisco Unified Serviceability > Tools > Control Center Feature Services)。
- 在CUCM上配置的SNMP社区字符串名称需要与CER上配置的字符串相同
- 确保Community String Name设置为ReadOnly

Server* 10.122.138.22--CUCM Voice/Video 🤿

lost IP Addresses Information	
Accept SNMP Packets from any host	Accept SNMP Packets only from these hosts Host IP Address
	Insert
	Host IP Addresses
	A
	Remove Remove
Access Privileges	
Access Drivilages* ReadOnly	
Access Privileges* ReadOnly	▼

Image: the second se

配置Cisco Emergency Responder

- 创建CER Web用户(可选)
- 配置组设置
- 配置电话设置
- 配置服务器设置
- 输入许可证
- 确定Cisco Unified Communications Manager集群
- 创建现场警报(可选)
- 创建紧急响应位置(ERL)
- 配置SNMP
- 识别LAN交换机
- 通过交换机端口进行电话跟踪
- 电话跟踪计划
- 电话跟踪(交换机端口、未分配电话、手动配置的电话和基于IP子网的)
- 升级CCM(可选)

创建CER Web用户(可选)

- 如果要限制某人对CER网页的访问,可以通过用户管理创建用户,并将用户添加到具有特定 角色的用户组
- 不同的安全级别/组包括:

用户 ERL管理员 管理实用程序 网络管理员可维护性 系统管理员

配置组设置

System > Cisco ER Group Setting

- 可以选择设置SMTP Mail Server、Source Mail ID和System Administrator Mail ID。
- 如果需要现场邮件警报,则需要配置SMTP邮件服务器和源邮件ID。
- 如果要接收有关重要系统警报的电子邮件,请配置SMTP邮件服务器和管理员邮件ID。可以同时配置有关关键系统警报的现场电子邮件警报和电子邮件,并同时运行。

- Status	
Changes Saved	
-Specify server group attributes	1981-0960-00 00 - 1985-
Cisco ER Group Name *	CERServerGroup
Peer TCP Port *	17001
Heart beat Count *	3
Heart beat Interval (in sec) *	30
Active Call Time out (in min) *	190
SMTP Mail Server	10.48.39.230
Source Mail ID	CER_Admin@d-e2k-41-1.cisc
System Administrator Mail ID	Network_Admin@d-e2k-41
SysLog	disable
Syslog Server	
Notes	

配置电话设置

System > Telephony Settings

您不必更改此页面上的任何内容;但是,此处所做的更改必须与CUCM上配置的CTI路由点匹配。

Ready		
Specify telephony attributes		
Route Point for Primary Cisco ER Server *	911	
Route Point for Standby Cisco ER Server	912	
PSAP Callback Route Point Pattern *	913XXXXXXXXXX	
ELIN Digit Strip Pattern *	913	
UDP Port Begin *	32000	
Inter Cisco ER Group Route Pattern		
IP Type of service (00-FF) *	0x b8	
Onsite Alert Prompt Repeat Count *	1	
Use IP Address from call signaling		

配置服务器设置

System > Server Settings

最好在"调试程序包列表"和"跟踪程序包列表"的所有框中打勾。这增加了确定系统出现问题的根本原因的机会。启用所有这些调试和跟踪对服务器性能的影响最小,因为CER是服务器上唯一的东西。

S tatus eady		
Select Server		
Publisher (primary)		
Modify Server Settings—		
Server Name * Publis	her	
lost Name CER-2	20	
Debug Package List	elect All Clear	All -
CER_DATABASE		CER_SYSADMIN
CER_REMOTEUPDATE		CER_TELEPHONY
CER_PHONETRACKINGE		CER_AGGREGATOR
CER_ONSITEALERT		CER_GROUP
CER_CALLENGINE		CER_CLUSTER
Trace Package List	lect All Clear	
CER_DATABASE		CER_SYSADMIN
CER_REMOTEUPDATE		CER_TELEPHONY
CER_PHONETRACKINGE	NGINE 🗹	CER_AGGREGATOR
CER_ONSITEALERT		CER_GROUP
CER CALLENGINE		CER_CLUSTER

确定Cisco Unified Communications Manager集群

Phone Tracking > Cisco Unified Communications Manager

- 每个运行CallManager服务的CUCM节点也必须运行SNMP服务。
- 指定为Cisco Unified Communications Manager的Callmanager必须运行CallManager服务。
- 如果所有配置都正确,并且SNMP工作正常,则点击显示在如下所示图像右上角的Cisco Unified Communications Managers List超链接可以查看所有Callmanager节点。

0.122.138.22 0.122.138.22 CER 0.122.138.23	Cisco Unified Communications Managers List
0.122.138.22 CER 0.122.138.23	
CER 0.122.138.23	
0.122.138.23	
0.122.138.23	
60010	
L	
59	
3804	
administrator	
8443	Test AXL Connectivity
Undate Caracit	Changes
Cancel	unanges
	0.122.138.23 0010 9 1804 1dministrator 1443 Update Cancel 1

注意:执行此步骤后,CTI路由点和CTI端口可显示为在CUCM上注册。

创建现场警报(可选)

ERL > Onsite Alert Settings

- 只要CUCM上的CTI端口能够发出外部呼叫,外部号码(如手机)就可用于现场警报。
- 要使邮件警报正常工作,必须在组设置下配置SMTP邮件服务器。

💊 注意:电子邮件地址字段是可选的。其他所有选项均是必需的。

提示:可以指定电子邮件别名,以便多人获取该电子邮件。如果您的安全团队有电子邮件别名, 这很有用。

Ready			
Add new Onsite Alert (contact		
Onsite Alert ID *			
)nsite Alert Name *			
)nsite Alert Number *			7
Onsite Alert Email Address			
		L	
		Incert Cancel	Changes 1
		Insert Cancel	Changes
		Insert Cancel	Changes
Available Onsite Alert	s	Insert Cancel	Changes
Available Onsite Alert	5	Insert Cancel	Changes
Available Onsite Alert	5	Insert Cancel	Changes
Available Onsite Alert Add New Onsite Alert ID	s Onsite Alert Name	Insert Cancel	Changes Onsite Alert Email Address
Available Onsite Alert Add New Onsite Alert ID	5 Onsite Alert Name SJ	Insert Cancel Cancel Onsite Alert Number 85261234	Changes Onsite Alert Email Address Onsite Security@d-e2k-41-1.cisco.cor

创建紧急响应位置(ERL)

ERL > Conventional ERL

- ERL可以根据需要进行细化(建筑、楼层、象限、房间、工作站等)。
- 如果通过CER发出呼叫,并且没有与该电话关联的ERL,则使用默认ERL,因此最好配置默认 ERL。
- 如果多个ERL使用相同的网关/路由列表,则可以使用相同的路由模式(即10.911)。在CER
 1.x中,必须在组设置下启用主叫方修改,才能使用相同的路由模式。在CER 2.0中,该选项
 不存在,因为默认情况下该选项处于启用状态。
- 当呼叫被路由回CUCM时,路由模式成为被叫号码,而ELIN成为主叫号码。

ERL (1 - 1 of 1)									
Configure Default ERL	Add New ERL								
ERL Name	Route/Translation PatternELIN	Onsite Alert Ids.	Street Name	Community Name	State	Edit	Сору	Delete	Audit Trail
RTP	10.911919537	TestOnsite	Kit Creek	RTP	NC	/	c\$	8	view
Configure Default ERL	Add New ERL							N 4 (<u>Go</u> 1 of 1 ▶ ▶

ERL Settings			
ERL Name *	RTP		
Description	RTP		
Test ERL (Used for Synthetic Testing)		Ν	
ELIN Settings Route/Translation pattern ELIN	Add Update Remove	10.9119195375855	*

-Onsite Alert Settings		
Available Onsite Alert IDs Add Remove	Onsite Alert IDs for the ERL OnSiteAlert	*

配置SNMP

Phone Tracking > SNMP V2

必须在此处配置所有交换机和CallManager服务器,SNMP电话跟踪才能正常工作



..*
10
2
CER
Insert Cancel Changes

识别LAN交换机

Phone Tracking > LAN Switch

- 所有连接电话的交换机都需要在此处进行配置
- 如果是非Cisco交换机或CDP被禁用,请选中Enable CAM based Phone Tracking复选框。

ų	- LAN Switch Details				
	Switch Host Name / IP Address *				
	Description				
	Enable CAM based Phone Tracking				
	Use port description as port location				
	Use SNMPV3 for Discovery				
		Insert Cancel Changes			
	LAN Switches				
	Add LAN Switch				
	Switch Host Na	Switch Host Name / IP Address			
	10.48.38.251				
	10.48.62.250				

通过交换机端口进行电话跟踪

ERL成员>交换机端口

- 电话跟踪完成后,将ERLS分配给交换机端口
- 可以指定位置;但并非必需
- 在下面显示的示例中,通过IP电话子网跟踪电话60002,但交换机正在运行SNMP,因此电话 仍然显示在此处。

Switch(s) (1 to 2	of 2)			Last phone i				racking was done at May 30, 2017 3:28:54 PM (
Assign ERL to Selected S	Switch Ports		Assign ERL Search ERL					Edit Vi		
Switch IP Address		ERL Name	me Switch IP Address		Name Locatio	on Phone Extension	Phone IP /	Address Phone Type		
■ [□] 10.48.38.251										
10.40.50.201		RTP	10.48.38.251	Fa0/1	View		10 49 29 22			
	RTP 10.48.38.	10.48.38.251	Fa0/2	View	1052002	10,40,30,33	Cisco 7975			

交换机上的SNMP配置

router(config)#

snmp-server community <community string> ro

Sets the SNMP Community string on the switch to Read-Only (RO)

```
lsegnini#show run | b snmp
snmp-server community CER RO
```

验证

- 1. 可以注册CTI路由点和CTI端口。
- 2. 连接到交换机的IP电话必须由CER自动发现。
- 3. IP电话可以呼叫911并通过CER获得呼叫路由。

4. PSAP回叫可以路由到最后一个呼叫PSAP的电话。

故障排除

911呼叫

- 验证主叫电话的CSS与911/9.911转换模式的分区关联。
- 确认911/9.911转换模式已选中紧急优先级,并且其CSS与911 CTI RP的分区相关联。
- 确保为9.911转换模式配置了点前条带。
- 验证911 CTI RP的注册状态,并确保它已注册到主CER。
- 在911 CTI RP上配置的呼叫转移设置可以指向故障切换场景的912 CTI RP。
- 验证在CER中配置的ERL是否根据911呼叫的来源进行了正确的RP/ELIN修改。
- 确保911/912 CTI RP的CSS与来自CER的重定向呼叫的路由模式的分区相关联。

PSAP回叫

- 网关的传入CSS可以到达为回叫呼叫配置的转换模式的分区。
- 根据在GW中发送的有用位数(带有或不带任何前缀),使用正确位数配置的转换模式。
- 转换模式前缀913,以及其余最高有效位数。TP的CSS可以到达913 CTI RP的分区。
- CER条913(ELIN数字条字段)。回叫是在活动呼叫超时中指定的时间内(以分钟为单位)。
- 913 CTI RP的CSS可以到达原始主叫方电话DN的分区。

现场警报

— 为每个ERL正确配置了现场警报联系人。

- CTI端口已注册,其CSS可以到达Onsite警报人员的电话DN的分区。

— 确保有足够的CTI端口处理现场警报的同时呼叫。

911运算符未获取正确的ELIN值

— 确保System > Cisco ER Group Settings > Calling Party Modification value设置为启用。

— 用于CUCM和CER之间交互的应用用户启用了标准CTI并启用了标准CTI允许修改主叫号码用户 组。

- 在911呼叫的路由模式上,未选中Use Calling Party's External Phone Number Mask复选框。
- 在RP/RL/RG/网关级别没有主叫方修改。
- 如果之前的所有设置看起来都正确,请在网关上运行debug以检查911呼叫的主叫方号码(例如

: "debug isdn q931" for a PRI gateway) 。

电话跟踪完成后交换机端口未显示

— 检查CER上的SNMP配置、交换机上的SNMP配置以及交换机是否在CER中配置。

— 确保支持在该版本的CER上跟踪交换机。如果交换机不受支持,您可以在电话跟踪日志中看到 "This device is not supported <ip address>"错误消息。 - CER支持的设备列表列在cisco.com交换机端口显示但电话不显示。

— 检查CER和CCM上的SNMP配置。

— 在每个CUCM上,需要激活和启动Cisco Unified Serviceability > Tools > Control Center - Feature Services > Cisco CallManager SNMP Service。

— 在每个CUCM上,确保网络服务SNMP Primary Agent正在运行。

— 确保所有CUCM服务器都有需要在CER中跟踪的电话,显示在M列表中。可以通过转至Phone Tracking > Cisco Unified Communications Manager >点击Cluster >然后点击Cisco Unified Communications Manager List来检查列表。这可以显示运行CCM服务的CUCM集群中的所有节点 。

— 您可以运行SNMP走路以确认CER能够从CUCM和交换机提取IP电话信息:

输入社区字符串: cer

输入服务器的ip地址,将127.0.0.1用于localhost。请注意,需要提供IP地址,而不是主机名。提示 :10.48.62.250

对象ID(OID):1.3.6.1.4.1.9.9.23.1.2.1.1.6

输入参数作为"file",将输出记录到文件中。[nofile]:

此命令可能会暂时影响CPU性能。

是否继续(y/n)?y

iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10101.1 =字符串: "rtp12-calo-363-gw.cisco.com" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10102.6 =字符串: "SEPF09E636EE825" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10104.8 =字符串: "SEP74A02FC0AD11" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10107.7 =字符串: "SEP6C416A369525" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10108.12 =字符串: "SEP1C1D862F3EDF" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10109.9 =字符串: "SEP6899CD85AE21" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10111.10 =字符串: "SEP84B5170993E8" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10113.11 =字符串: "SEP88908D737AC7" iso.3.6.1.4.1.9.9.23.1.2.1.1.6.10115.2 =字符串: "SEP00235EB7A757"

关于此翻译

思科采用人工翻译与机器翻译相结合的方式将此文档翻译成不同语言,希望全球的用户都能通过各 自的语言得到支持性的内容。

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