



# Call Progress Analysis Over IP-to-IP Media Session

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The Call Progress Analysis Over IP-IP Media Session feature enables the detection of automated answering systems and live human voices on outbound calls and communicates the detected information to the external application. Typically, call progress analysis (CPA) is extensively used in contact center deployments in conjunction with the outbound Session Initiation Protocol (SIP) dialer, where CPA is enabled on the Cisco Unified Border Element (Cisco UBE), and digital signal processors (DSP) perform the CPA functionality.

- [Finding Feature Information, page 1](#)
- [Restrictions for Call Progress Analysis Over IP-to-IP Media Session, page 1](#)
- [Information About Call Progress Analysis Over IP-IP Media Session, page 2](#)
- [How to Configure Call Progress Analysis Over IP-to-IP Media Session, page 3](#)
- [Configuration Examples for the Call Progress Analysis Over IP-to-IP Media Session, page 6](#)
- [Feature Information for Call Progress Analysis Over IP-IP Media Session, page 7](#)

## Finding Feature Information

Your software release may not support all the features documented in this module. For the latest caveats and feature information, see [Bug Search Tool](#) and the release notes for your platform and software release. To find information about the features documented in this module, and to see a list of the releases in which each feature is supported, see the feature information table at the end of this module.

Use Cisco Feature Navigator to find information about platform support and Cisco software image support. To access Cisco Feature Navigator, go to [www.cisco.com/go/cfn](http://www.cisco.com/go/cfn). An account on Cisco.com is not required.

## Restrictions for Call Progress Analysis Over IP-to-IP Media Session

- Only SIP-to-SIP Early Offer (EO-to-EO) call flows are supported.
- Session Description Protocol (SDP) passthrough and flow-around media calls are not supported.

- Only the G711 flavor of codec is supported.
- Skinny Client Control Protocol (SCCP)-based digital signal processor (DSP) farm is not supported.

## Information About Call Progress Analysis Over IP-IP Media Session

### Call Progress Analysis

Call progress analysis (CPA) is a DSP algorithm that analyzes the Real-Time Transport Protocol (RTP) voice stream to look for special information tones (SIT), fax or modem tones, human speech, and answering machine tones. CPA also passes the voice information to Cisco IOS or Cisco Unified Border Element (Cisco UBE).

CPA is initiated on receiving a new SIP INVITE with x-cisco-cpa content. While a call is in progress, the DSP or the Xcoder analyzes the incoming voice or media stream. The DSP identifies the type of voice stream based on statistical voice patterns or specific tone frequencies and provides the information to the Cisco UBE. The Cisco UBE notifies the dialer with a SIP UPDATE with x-cisco-cpa content along with the detected event. Based on the report, the caller (dialer) can decide to either transfer the call or terminate the call.

To use the CPA functionality, you must enable CPA and configure CPA timing and threshold parameters.

**Table 1: X-cisco-cpa content meaning**

SIP Message	Direction of Message	Meaning
18x or 200	Cisco IOS to dialer	Cisco UBE informs the dialer if CPA is enabled for a call or not.
New INVITE	Dialer to Cisco IOS	Dialer requests Cisco IOS or the Cisco UBE to activate the CPA algorithm for this session.
UPDATE	Cisco IOS to dialer	Cisco IOS or the Cisco UBE notifies the dialer about the detected event.

### CPA Events

**Table 2: CPA Event Detection List**

CPA Event	Definition
Asm	Answer machine
AsmT	Answer machine terminate tone

CPA Event	Definition
CpaS	Start of the Call Progress Analysis
FT	Fax/Modem tone
LS	Live human speech
LV	Low volume or dead air call
SitIC	Special information tone IC -- Intercept -- Vacant number or Automatic Identification System (AIS)
SitNC	SIT tone NC—No Circuit (NC), Emergency, or Trunk Blockage
SitVC	SIT tone VC—Vacant Code
SitRO	SIT tone RO—Reorder Announcement
SitMT	Miscellaneous SIT Tone

# How to Configure Call Progress Analysis Over IP-to-IP Media Session

## Enabling CPA and Setting the CPA Parameters

Perform the following task to enable CPA and set the CPA timing and threshold parameters:

### SUMMARY STEPS

1. **enable**
2. **configure terminal**
3. **dspfarm profile *profile-identifier* transcode**
4. **call-progress-analysis**
5. **exit**
6. **voice service voip**
7. **cpa timing live-person *max-duration***
8. **cpa timing term-tone *max-duration***
9. **cpa threshold active-signal *signal-threshold***
10. **end**

## DETAILED STEPS

	Command or Action	Purpose
<b>Step 1</b>	<b>enable</b>  <b>Example:</b> Device> enable	Enables privileged EXEC mode. <ul style="list-style-type: none"> <li>• Enter your password if prompted.</li> </ul>
<b>Step 2</b>	<b>configure terminal</b>  <b>Example:</b> Device# configure terminal	Enters global configuration mode.
<b>Step 3</b>	<b>dspfarm profile <i>profile-identifier</i> transcode</b>  <b>Example:</b> Device(config)# dspfarm profile 15 transcode	Enters DSP farm profile configuration mode, defines a profile for DSP farm services, and enables the profile for transcoding.
<b>Step 4</b>	<b>call-progress-analysis</b>  <b>Example:</b> Device(config-dspfarm-profile)# call-progress-analysis	Enables call progress analysis (CPA) on Cisco UBE. <ul style="list-style-type: none"> <li>• You must configure this command to activate the CPA feature and set CPA parameters.</li> </ul>
<b>Step 5</b>	<b>exit</b>  <b>Example:</b> Device(config-dspfarm-profile)# exit	Exits DSP farm profile configuration mode and enters global configuration mode.
<b>Step 6</b>	<b>voice service voip</b>  <b>Example:</b> Device(config)# voice service voip	Enters voice service configuration mode.
<b>Step 7</b>	<b>cpa timing live-person <i>max-duration</i></b>  <b>Example:</b> Device(conf-voi-serv)# cpa timing live-person 2501	(Optional) Sets the maximum waiting time (in milliseconds) that the CPA algorithm uses to determine if a call is answered by a live human.
<b>Step 8</b>	<b>cpa timing term-tone <i>max-duration</i></b>  <b>Example:</b> Device(conf-voi-serv)# cpa timing term-tone 15500	(Optional) Sets the maximum waiting time (in milliseconds) that the CPA algorithm uses to wait for the answering machine termination tone after the answering machine is detected.

	Command or Action	Purpose
<b>Step 9</b>	<b>cpa threshold active-signal</b> <i>signal-threshold</i>  <b>Example:</b> Device(conf-voi-serv)# cpa threshold active-signal 18db	(Optional) Sets the threshold (in decibels) of an active signal that is related to the measured noise floor level. <ul style="list-style-type: none"> <li>• If a signal threshold configured by this command is greater than the measured noise floor level, then the signal is considered as active. The active signal thresholds that you can configure are 9, 12, 15, 18, and 21 decibels.</li> </ul>
<b>Step 10</b>	<b>end</b>  <b>Example:</b> Device(conf-voi-serv)# end	Exits voice service configuration mode and returns to privileged EXEC mode.

## Verifying the Call Progress Analysis Over IP-to-IP Media Session

Perform this task to verify that call progress analysis has been configured for a digital signal processor (DSP) farm profile.

### SUMMARY STEPS

1. **enable**
2. **show dspfarm profile** *profile-identifier*

### DETAILED STEPS

**Step 1**     **enable**  
Enables privileged EXEC mode.

**Example:**  
Device> **enable**

**Step 2**     **show dspfarm profile** *profile-identifier*  
Displays the configured DSP farm profile information for a selected Cisco Call Manager group. In the following sample output, the Call Progress Analysis field shows that CPA is enabled.

**Example:**  
Device# **show dspfarm profile 3**

```

Profile ID = 3, Service =Universal TRANSCODING, Resource ID = 3
Profile Description :
Profile Service Mode : Non Secure
Profile Admin State : UP
Profile Operation State : ACTIVE

```

```

Application : CUBE    Status : ASSOCIATED
Resource Provider : FLEX_DSPRM    Status : UP
Number of Resource Configured : 4
Number of Resources Out of Service : 0
Number of Resources Active : 0
Codec Configuration: num_of_codecs:4
Codec : g711ulaw, Maximum Packetization Period : 30
Codec : g711alaw, Maximum Packetization Period : 30
Codec : g729ar8, Maximum Packetization Period : 60
Codec : g729abr8, Maximum Packetization Period : 60
Noise Reduction : ENABLED
Call Progress Analysis : ENABLED

```

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## Troubleshooting Tips

Use the following commands to troubleshoot the call progress analysis for SIP-to-SIP calls:

- `debug ccsip all`
- `debug voip ccapi inout`
- `debug voip hpi all`
- `debug voip ipipgw`
- `debug voip media resource provisioning all`

## Configuration Examples for the Call Progress Analysis Over IP-to-IP Media Session

### Example: Enabling CPA and Setting the CPA Parameters

The following example shows how to enable CPA and set a few timing and threshold parameters. Depending on your requirements, you can configure more timing and threshold parameters.

```

Device> enable
Device# configure terminal
Device(config)# dspfarm profile 15 transcode
Device(config-dspfarm-profile)# call-progress-analysis
Device(config-dspfarm-profile)# exit
Device(config)# voice service voip
Device(conf-voi-serv)# cpa timing live-person 2501
Device(conf-voi-serv)# cpa timing term-tone 15500
Device(conf-voi-serv)# cpa threshold active-signal 18db
Device(conf-voi-serv)# end

```

# Feature Information for Call Progress Analysis Over IP-IP Media Session

The following table provides release information about the feature or features described in this module. This table lists only the software release that introduced support for a given feature in a given software release train. Unless noted otherwise, subsequent releases of that software release train also support that feature.

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**Table 3: Feature Information for Call Progress Analysis Over IP-IP Media Session**

Feature Name	Releases	Feature Information
Call Progress Analysis Over IP-to-IP Media Session	15.3(2)T	The Call Progress Analysis Over IP-to-IP Media Session feature enables detection of automated answering systems and live human voices on outbound calls and communicates the detected information to an external application.  The following command was introduced: <b>call-progress-analysis.</b>
Call Progress Analysis Over IP-to-IP Media Session	Cisco IOS XE Release 3.9S	The Call Progress Analysis Over IP-to-IP Media Session feature enables detection of automated answering systems and live human voices on outbound calls and communicates the detected information to an external application.  The following command was introduced: <b>call-progress-analysis.</b>

