AudioCodes Routing Manager (ARM)

# Generating Call Detail Records with the ARM CDR Generator

Version 9.2



		Table of Contents	
1	Intr	oduction	7
2	Cor	nfiguring CdrArmMessage Fields	9
De	fines	the P-Asserted-Identity URI that it sent (following manipulation)	9
3	Ger	nerating CDRs	11
	3.1	Map File Example	13



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4

Configuration Note 1. Introduction

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# **Abbreviations and Terminology**

Each abbreviation, unless widely used, is spelled out in full when first used.



### **Related Documentation**

Manual Name
ARM Installation Manual
ARM User's Manual
Mediant 9000 SBC User's Manual
Mediant 4000 SBC User's Manual
Mediant SE SBC User's Manual
Mediant SE-H SBC User's Manual
Mediant VE SBC User's Manual
Mediant VE-H SBC User's Manual
Mediant 1000B Gateway and E-SBC User's Manual
Mediant 800B Gateway and E-SBC User's Manual
Mediant 500 Gateway and E-SBC User's Manual
Mediant 500 MSBR User's Manual
Mediant 500L Gateway and E-SBC User's Manual
Mediant 500L MSBR User's Manual
MP-1288 High-Density Analog Media Gateway User's Manual
One Voice Operations Center Server Installation, Operation and Maintenance Manual
One Voice Operations Center Integration with Northbound Interfaces
One Voice Operations Center User's Manual
One Voice Operations Center Product Description
One Voice Operations Center Alarms Guide
One Voice Operations Center Security Guidelines

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#### **Document Revision Record**

LTRT	Description
42100	This is the first release of this document.
42101	One Start-Call; CdrMessage > CdrArmMessage; sessionKey; routingRuleId; routingRuleName; discardingByRoutingRule; partial; description; /opt/tomcat/logs/CDR_Json; drop-calls period; myOutputFileName; UnfinishedCdrs.json
42102	New CdrArmMessage fieldsiu optionod option.

Configuration Note 1. Introduction

# 1 Introduction

This Configuration Note shows how to implement the CDR Generator.

The CDR Generator is a utility that converts CdrMessages into CdrArmMessages. The conversions are performed in the following CdrArmMessages formats:

- Clear text
- Jsons

CdrMessages are sent by the Router for each leg which takes part in a call. CdrMessages are of two types:

- Start call
- End call

A CdrArmMessage contains the total quantity of information about a call taken from the CdrMessage. In general, a successful call (without an alternative route) has three CdrMessages:

- One Start-Call
- Two End-Call

The CDR Generator takes all these messages and gathers them into one message.

To display the CDR in more readable format, you can generate a CSV file and define the column name of each CDR field.



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# **2 Configuring CdrArmMessage Fields**

The table below describes the CDR ARM message fields. Use the table as reference when configuring the Map file for the output CSV file.

Table 2-1: CdrArmMessage Field Descriptions

CDR Field	Description		
CdrApplicationType	Defines the node application type:  SBC GW HYBRID THIRD_PARTY		
sessionKey	Defines a unique key identifier.		
sessionId	Defines a unique session identifier.		
nodeld	Defines the ARM node's database ID.		
nodeName	Defines the node name as described in the ARM GUI.		
nodelp	Defines the node's IP address.		
incomingPconOrConnectionName	Defines the incoming leg name as described in the ARM GUI.		
incomingSipInterface	Defines the SIP interface ID of an incoming Connection or Peer Connection in the SBC / Gateway.		
incomingCallId	Defines the Call ID of the incoming leg.		
outgoingPconOrConnectionName	Defines the outgoing leg name as described in the ARM GUI.		
outgoingCallId	Defines the Call ID of the outgoing leg.		
srcUri	Defines the Source URI that is sent (following manipulation).		
srcUriBeforeMap	Defines the Source URI before manipulation.		
from	Defines the From URI that is sent (following manipulation).		
fromBeforeMap	Defines the From URI before manipulation.		
pai	Defines the P-Asserted-Identity URI that it sent (following manipulation).		
paiBeforeMap	Defines the P-Asserted-Identity URI before manipulation.		
ppi	Defines the P-Prefetred-Identity URI sent (following manipulation).		
ppiBeforeMap	Defines the P-Prefetred-Identity URI before manipulation.		
dstUri	Defines the Destination URI sent (following manipulation).		
dstUriBeforeMap	Defines the Destination URI before manipulation.		
armSetupTime	Defines the time at which CALL_START is sent, per the ARM's time.		
armReleaseTime	Defines the time at which CALL_END is sent, per the ARM's time.		
sbcSetupTime	Defines the time at which Gateway / SBC time when starting to handle an Invite message, as reported by the Gateway / SBC.		



configured to send the time (duration) in milliseconds.  voiceDuration  Defines how long voice is streamed, in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  completeDuration  Defines how long the entire call takes, from the first incoming Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  voiceStreamed  Determines whether the call was established or not.  sipTerminationReason  Defines the SIP termination reason.  sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq  Defines the number that each route (path) of a call has, starting from 0.  lastNodeId  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  Defines the name of the last Peer Connection or Connection in the path.	CDR Field	Description		
a call ends), as reported by the Gateway / SBC.  sbcAlertTime  Defines the time at which the remote side phone starts to ring, as reported by the Gateway / SBC.  alertDuration  Defines how long the phone rings, in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds, as reported by the Gateway / SBC.  CompleteDuration  Defines how long the entire call takes, from the first incoming Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  VoiceStreamed  Determines whether the call was established or not.  sipTerminationReason  Defines the SIP termination reason.  sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq  Defines the number that each route (path) of a call has, starting from 0.  lastNodeId  Defines the ARM database ID of the last node in the path.  lastNodeName  Defines the name of the last node in the path as described in the ARM GUI.  lastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  Defines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule ID of the matching rule  discardingByRoutingRule  Defines the Routing Rule ID in case of discarding rule  path  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	sbcConnectTime	when the call is established), as reported by the Gateway /		
ring, as reported by the Gateway / SBC.  alertDuration  Defines how long the phone rings, in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  VoiceDuration  Defines how long voice is streamed, in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  CompleteDuration  Defines how long the entire call takes, from the first incoming Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  VoiceStreamed  Determines whether the call was established or not.  sipTerminationReason  Defines the SIP termination reason.  sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq  Defines the number that each route (path) of a call has, starting from 0.  lastNodeId  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  lastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  routingRuleId  Defines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule ID in case of discarding rule  Describes the Routing Rule ID in case of discarding rule  Describes the Path.  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	sbcReleaseTime			
reported by the Ğateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  voiceDuration  Defines how long voice is streamed, in milliseconds, as reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  completeDuration  Defines how long the entire call takes, from the first incoming Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  voiceStreamed  Determines whether the call was established or not.  sipTerminationReason  Defines the SIP termination reason.  sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq  Defines the number that each route (path) of a call has, starting from 0.  lastNodeId  Defines the ARM database ID of the last node in the path.  lastNodeName  Defines the name of the last node in the path as described in the ARM GUI.  lastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  routingRuleId  Defines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule ID in case of discarding rule  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	sbcAlertTime			
reported by the Gateway / SBC. The SBC / Gateway must be configured to send the time (duration) in milliseconds.  completeDuration  Defines how long the entire call takes, from the first incoming Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  voiceStreamed  Determines whether the call was established or not.  sipTerminationReason  Defines the SIP termination reason.  sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq  Defines the number that each route (path) of a call has, starting from 0.  lastNodeId  Defines the ARM database ID of the last node in the path.  lastNodeName  Defines the name of the last node in the path as described in the ARM GUI.  lastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  routingRuleId  Defines the Routing Rule ID of the matching rule  discardingByRoutingRule  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	alertDuration	reported by the Gateway / SBC. The SBC / Gateway must be		
Invite until the call is ended, in milliseconds, as reported by the Gateway / SBC.  voiceStreamed Determines whether the call was established or not.  sipTerminationReason Defines the SIP termination reason.  sipTerminationReasonDesc Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  routeSeq Defines the number that each route (path) of a call has, starting from 0.  lastNodeId Defines the ARM database ID of the last node in the path.  lastNodeName Defines the name of the last node in the path as described in the ARM GUI.  lastPconOrConnectionName Defines the name of the last Peer Connection or Connection in the path.  routingRuleId Defines the Routing Rule ID of the matching rule  routingRuleName Defines the Routing Rule name of the matching rule  discardingByRoutingRule Defines the Routing Rule ID in case of discarding rule  path Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	voiceDuration	reported by the Gateway / SBC. The SBC / Gateway must be		
sipTerminationReason  Defines the SIP termination reason.  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  Defines the number that each route (path) of a call has, starting from 0.  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  Defines the name of the last Peer Connection or Connection in the path.  routingRuleId  Defines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	completeDuration	Invite until the call is ended, in milliseconds, as reported by		
sipTerminationReasonDesc  Defines a more detailed and descriptive SIP termination reason than the field 'sipTerminationReason'.  Defines the number that each route (path) of a call has, starting from 0.  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  Defines the name of the last Peer Connection or Connection in the path.  ToutingRuleId  Defines the Routing Rule ID of the matching rule  ToutingRuleName  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	voiceStreamed	Determines whether the call was established or not.		
reason than the field 'sipTerminationReason'.  Poefines the number that each route (path) of a call has, starting from 0.  IastNodeId  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  IastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  Poefines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule name of the matching rule  discardingByRoutingRule  Defines the Routing Rule ID in case of discarding rule  path  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	sipTerminationReason	Defines the SIP termination reason.		
starting from 0.  Defines the ARM database ID of the last node in the path.  Defines the name of the last node in the path as described in the ARM GUI.  IastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  routingRuleId  Defines the Routing Rule ID of the matching rule  routingRuleName  Defines the Routing Rule name of the matching rule  discardingByRoutingRule  Defines the Routing Rule ID in case of discarding rule  path  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	sipTerminationReasonDesc			
lastNodeName  Defines the name of the last node in the path as described in the ARM GUI.  Defines the name of the last Peer Connection or Connection in the path.  Poefines the Routing Rule ID of the matching rule  Defines the Routing Rule name of the matching rule  Defines the Routing Rule ID in case of discarding rule  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	routeSeq			
the ARM GUI.  IastPconOrConnectionName  Defines the name of the last Peer Connection or Connection in the path.  Poefines the Routing Rule ID of the matching rule  Defines the Routing Rule name of the matching rule  Defines the Routing Rule ID in case of discarding rule  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	lastNodeId	Defines the ARM database ID of the last node in the path.		
in the path.  routingRuleId Defines the Routing Rule ID of the matching rule  routingRuleName Defines the Routing Rule name of the matching rule  discardingByRoutingRule Defines the Routing Rule ID in case of discarding rule  path Describes the path.  partial Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	lastNodeName			
routingRuleName  Defines the Routing Rule name of the matching rule  Defines the Routing Rule ID in case of discarding rule  path  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	lastPconOrConnectionName			
discardingByRoutingRule  Defines the Routing Rule ID in case of discarding rule  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	routingRuleId	Defines the Routing Rule ID of the matching rule		
path  Describes the path.  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	routingRuleName	Defines the Routing Rule name of the matching rule		
partial  Defines whether all CdrMessages applicable to this route are found in the input file. Will be 'True' for in-progress calls.	discardingByRoutingRule	Defines the Routing Rule ID in case of discarding rule		
found in the input file. Will be 'True' for in-progress calls.	path	Describes the path.		
description Briefly describes a partial call.	partial			
	description	Briefly describes a partial call.		

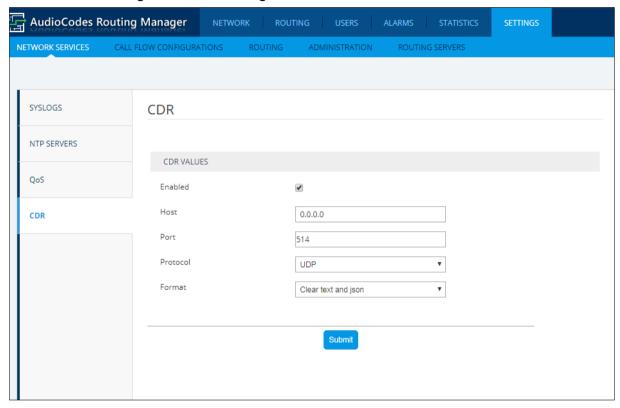
Configuration Note 3. Generating CDRs

# 3 Generating CDRs

The ARM allows you to generate CDRs using the CDR Generator utility.

- To generate CDRs using the CDR Generator utility:
- Open the CDR page in the ARM GUI (Settings > Network Services > CDR).
- Enable CDR and select Json or Clear text and Json format from the 'Format' dropdown menu.

Figure 3-1: Generating CDRs with the ARM CDR Generator



- 3. Collect all the CDRs. You can do this in two ways:
  - If you have a syslog server, take the syslog file from the syslog server.
  - If you do not have a syslog server, collect the logs from All Routers and create
    one large file: ARM saves the CDRs in Json format under the
    /opt/tomcat/logs/CDR\_Json directory. The name of the CDR file will be
    CDR\_Json-\*.log (up to 10 files can be saved). Create one file: 'CDR\_Json.log'.
- 4. Generate a CdrArmMessages file from CDRs:
  - Copy the one CDR file you got into one of the Routers.
  - Run the following command for generating the CDR file:

```
cdrUtil -i CDR Json.txt
```

The output files are:

- CdrArm\_ CDR\_Json\_[timestamp].txt
- CdrArm\_ CDR\_Json \_[timestamp]. json
- **5.** To set the output file name, use the -o option. For example:

```
cdrUtil -i CDR_Json.log -o myOutputFileName
```

**6.** To set the drop-calls period in hours, use the -d option. For example:

```
cdrUtil -d 12 CDR Json.log
```



All partial calls created more than 12 hours before are exported to the file 'DroppedCdrs.json'. The other partial calls are part of the input CDR in the next cdrUtil that is run and are exported to the file 'UnfinishedCdrs.json'.

When running cdrUtil again, it uses the 'myOutputFileName' input and the file 'UnfinishedCdrs.json'.

'UnfinishedCdrs.json' is overwritten with new cdrUtil results.

The file 'UnfinishedCdrs.json' will then no longer be used as input and each cdrUtill UnfinishedCdrs message will be added to it.

The default value for drop-calls is 24 hours.

7. To set the input 'UnfinishedCdrs.json' file location, use the -iu option. For example:

```
cdrUtil CDR Json.log -iu /root/UnfinishedCdrs.json
```

**8.** To set the output directory, use the -od option. For example:

```
cdrUtil CDR Json.log -od /root/tempDir
```

The output directory will contain the following files:

- CdrArm CDR Json [timestamp].txt
- CdrArm\_ CDR\_Json \_[timestamp]. Json
- UnfinishedCdrs.json
- DroppedCdrs.json'
- 9. To set the date format, use the -f option. For example:

```
cdrUtil CDR Json.log -f "yyyy-MM-dd HH:mm:ss.SSSSZZ"
```

- **10.** Generate a CSV file of the CdrArmMessages file from CDRs:
- 11. To generate a CSV file, use the -m option. For example:

```
cdrUtil -i CDR Json.log -m myMapFileName
```

The map file is a text file in json format that defines which fields will be displayed in the CSV file and what their new name (if required by the customer) will be. The file contains a list of name pairs:

- armCdrFieldName [the name of the field from the CdrArmMessage]
- convertedCdrFieldName [the customer-required name of the field that will be displayed in the CSV file]

#### Note:



- The generated CSV file contains only the 'armCdrFieldName' field from the mapping json file but you can dynamically build a CSV file, customize CSV file column names and determine where the CSV file will take its columns from. Use the example in the next section as reference.
- To create a new column with a new name which isn't displayed in the CdrArmMessage, set 'armCdrFieldName' to empty and 'convertedCdrFieldName' to the new column name.
   See convertedCdrFieldName = Direction in the section below.

Configuration Note 3. Generating CDRs

# 3.1 Map File Example

Use the example below as reference if you need to *dynamically build* a CSV file. Note the list of name pairs contained in the file:

- "armCdrFieldName" [the name of the field from the CdrArmMessage]
- "convertedCdrFieldName" [the customer-required name of the field that will be displayed in the CSV file]

```
{
    "name": "Puzzel",
    "cdrFieldNameJsonList": [
   {
            "armCdrFieldName": "srcUriBeforeMap",
            "convertedCdrFieldName": "ANUM before Manipulation"
   },
    {
            "armCdrFieldName": "srcUri",
            "convertedCdrFieldName": "ANUM after Manipulation"
   },
   {
            "armCdrFieldName": "dstUriBeforeMap",
            "convertedCdrFieldName": "BNUM before Manipulation"
   },
            "armCdrFieldName": "dstUri",
            "convertedCdrFieldName": "BNUM after Manipulation"
   },
            "armCdrFieldName": "sbcConnectTime",
            "convertedCdrFieldName": "START"
   },
            "armCdrFieldName": "sbcReleaseTime",
            "convertedCdrFieldName": "FINISH"
   },
    {
            "armCdrFieldName": "nodeName",
            "convertedCdrFieldName": "EXI"
   },
            "armCdrFieldName": "lastNodeName",
             "convertedCdrFieldName": "FXI"
   },
            "armCdrFieldName": "incomingPconOrConnectionName",
            "convertedCdrFieldName": "ITI"
   },
            "armCdrFieldName": "lastPconOrConnectionName",
            "convertedCdrFieldName": "OTI"
```



```
},
{
        "armCdrFieldName": "sipTerminationReason",
        "convertedCdrFieldName": "Diagnostic"
},
{
        "armCdrFieldName": "voiceStreamed",
        "convertedCdrFieldName": "ToBeCharged"
},
        "armCdrFieldName": "",
        "convertedCdrFieldName": "Direction"
},
        "armCdrFieldName": "incomingCallId",
        "convertedCdrFieldName": "SipCallId"
},
        "armCdrFieldName": "sbcSetupTime",
        "convertedCdrFieldName": "SetupAt"
},
{
        "armCdrFieldName": "sbcAlertTime",
        "convertedCdrFieldName": "AlertAt"
},
        "armCdrFieldName": "completeDuration",
        "convertedCdrFieldName": "msSetup"
},
        "armCdrFieldName": "alertDuration",
        "convertedCdrFieldName": "msAlert"
]
```

#### **International Headquarters**

1 Hayarden Street, Airport City Lod 7019900, Israel Tel: +972-3-976-4000

Fax: +972-3-976-4040

AudioCodes Inc.

200 Cottontail Lane, Suite A101E, Somerset NJ 08873

Tel: +1-732-469-0880 Fax: +1-732-469-2298

Contact us: <a href="https://www.audiocodes.com/corporate/offices-worldwide">https://www.audiocodes.com/corporate/offices-worldwide</a>

Website: www.audiocodes.com

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