Mida eFramework

Mida LiteCallCenter SBC Configuration Guide

Mida Solutions

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1.Introduction

1.1 Legal Statements

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1.2 Preface

This document is part of the official documentation of Mida Solutions products and details functionalities, user interface, option and working modes in detail. The system allows the user to configure all system functions using a simple and intuitive WEB interface. Please refer to the reference table for a complete list of documents relevant for system configuration.

1.3 Audience

The present document addresses both end users and system administrators of the products.

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1.4 Notations



This document highlights, where possible, the main parameters and operations through **bold** or *italics* text and all parts that might be critical during system configuration or use. Critical parts are also marked with Warning symbol reported here on the left.

1.5 References

This manual includes references to the following list of documents:

- [1] Mida_Unified_Portal-Administration_&_User_Manual
- [2] Mida_Appliance-Administration_Manual
- [3] MidaRec Gateway-Administration_Manual
- [4] <u>www.midasolutions.com/browsercompatibility</u>

2.SBC configurations



SBC to Mida eFramework configurations require to have the SBC Number Transformation Table **already set. Do that follow** <u>**Ribbon configuration guide</u></u> or see Appendix A of this guide for a brief step-by-step guide.</u>**

2.1 Mida LiteCallCenter to SBC connection

To set up the connection between Mida LiteCallCenter and the SBC, follow the next steps and insert values as stated in the screenshots, if no other values are specified.

1. Go to *Call Routing > Transformation* and create a new *Transformation table* (in the example below we called it "*DN to E.164 Map*").

This transformation will change the call destination with the proper Teams number.

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 Go to Call Routing > Call Routing Table and create a new call route entry (in the example below, "From Lite Call Center")





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3. Go to SIP > SIP Server Tables and create a new SIP Server ("Lite Call Center")



4. Go to SIP > SIP Profiles and create a new entry ("Lite Call Center")

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5. Go to **Signaling Groups** and create a new entry ("Signaling Group")

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2.2 PSTN to SBC connection

To set up the connection between PSTNMida LiteCallCenter and the Mida LiteCallCenter, follow the next steps and insert values as stated in the screenshots, if no other values are specified.

1. Go to *Call Routing > Transformation* and create a new *Transformation table* (in the example below we called it "*From PSTN to Lite Call Center"*).

This transformation will lead desired calls from the PSTN to the Mida LiteCallCenter.

In *Value*, it is possible to insert the desired prefix. All calls incoming to the SBC with that prefix will be redirected to Mida LiteCallCenter.

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3. Go to **SIP > SIP Server Tables** and create a new SIP Server ("PSTN").



4. Go to **SIP** > **SIP Profiles** and create a new entry ("*PSTN"*). Leave everything as default.

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5. Go to *Signaling Groups* and create a new entry ("PSTN").

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2.3 SBC to Teams – Teams Direct Routing

To configure Teams Direct Routing, follow the SBC vendor guide:

Ribbon: Best Practice - Configuring SBC Edge for Microsoft Teams Direct Routing •