

Simple screen



Full-function screen

This Operation Guide provides details on using the ProVisionaire Control template file (Dialer edition) for RM-CR. The simple screen is for USB/Bluetooth/Aux call control, and the full-function screen further allows for call control using SIP.

ProVisionaire Control is an application that allows you to operate a compatible system from a Windows computer via a wired LAN or a Wi-Fi network. Compatible sound systems may consist of CL/QL/TF, MTX/MRX, RM-CG/RM-CR, etc. You can design a control panel by freely arranging various widgets such as buttons and sliders, allowing you to create custom panels to suit your needs.

### Important

- Customize the template file to suit your operating environment.
- In order to understand the relationship between the on-screen controls and the audio route control, also refer to the schematic diagram of the route control at the end of this manual.

### Notice

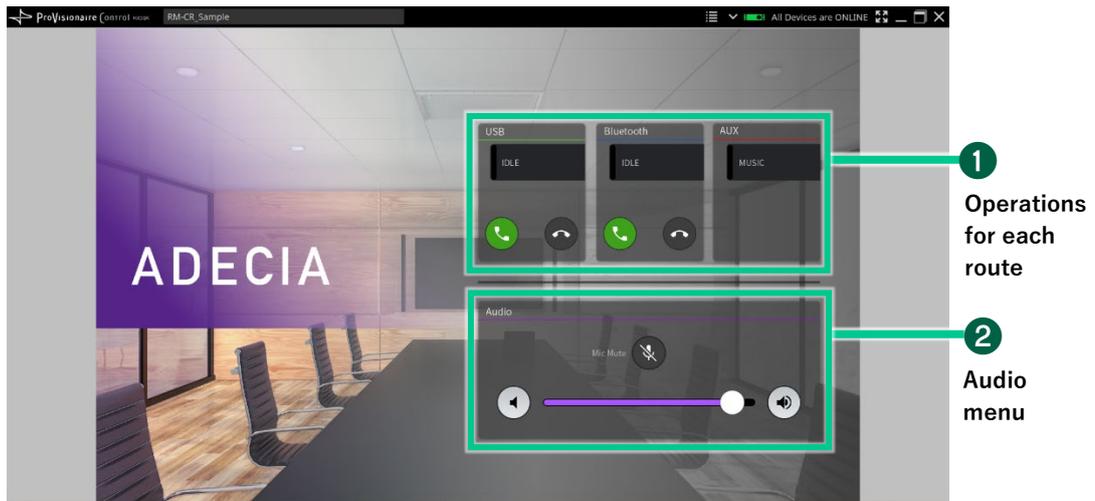
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# Description of Simple Screen

The simple screen is a template screen for call control when using only the USB/Bluetooth/AUX routes.



## 1 Operation menu for each route

### 1 [Line Status]

Indicates the status of the route.

#### ■ Indications for USB/Bluetooth/AUX:

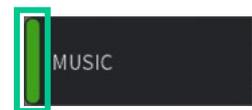
<b>IDLE</b>	No ongoing call
<b>CALLING</b>	Calling another base
<b>FAILED</b>	Line connection error
<b>ACTIVE</b>	Line connected
<b>INCOMING</b>	Call from another base
<b>MUSIC</b>	Playing music (when the channel is set to stereo for the USB or Bluetooth route)



### 2 [Input Indicator]

Lights up in green when audio input is detected.

#### ■ Applicable routes: USB/Bluetooth/AUX



### 3 [Offhook]

This button starts a call.

#### ■ Applicable routes: USB/Bluetooth



### 4 [Hangup]

This button disconnects the call.

#### ■ Applicable routes: USB/Bluetooth



## 2 Audio menu

### ⑤ [Mic Mute]

This button switches between muting and unmuting the audio of Dante's microphone input (RM-CG, etc.) route and the USB output route. Use this when you do not want the audio from your base to be transmitted to the other party.



Unmuted



Muted

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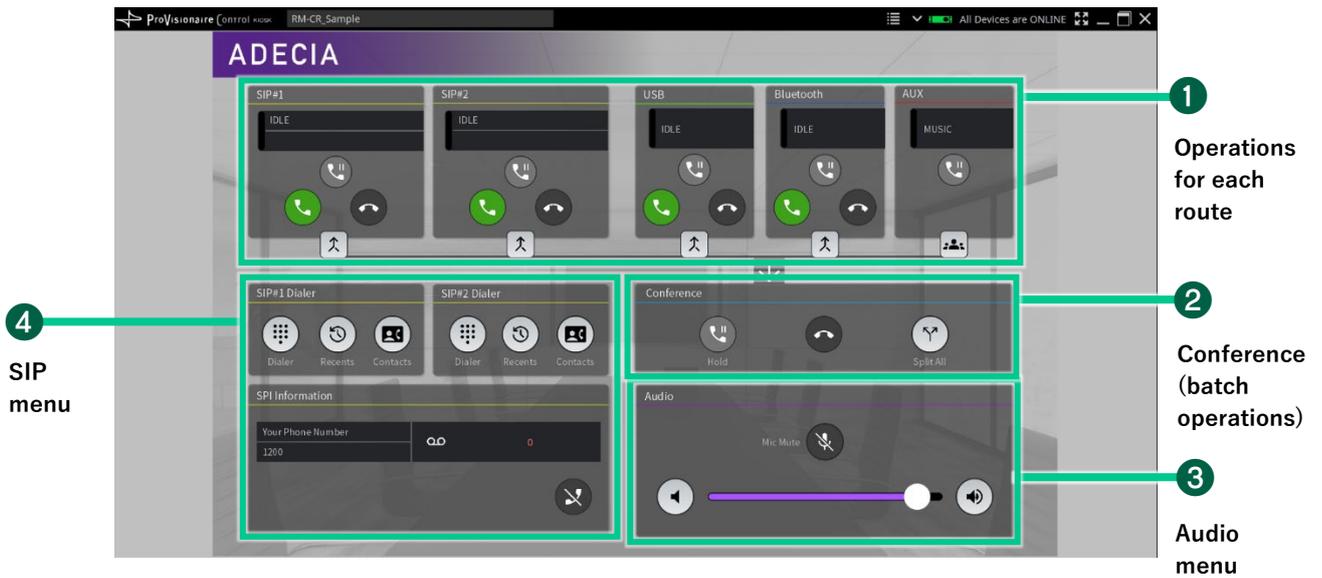
### ⑥ [Volume]

This slider adjusts the volume of the speaker output. The setting can be adjusted to one of 20 levels.



# Description of Full-Function Screen

The full-function screen is a template screen for call control when using an SIP1/SIP2 route in addition to the USB/Bluetooth/AUX routes. Multi-point calls (conferences) can be controlled by mixing the various routes.



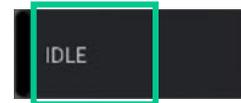
## 1 Operation menu for each route

### ① [Line Status]

Indicates the status of the route.

- Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX

IDLE	No ongoing call
CALLING	Calling another base
FAILED	Line connection error
ACTIVE	Line connected
INCOMMING	Call from another base
ONHOLD	Call on hold
MUSIC	Playing music (when the channel is set to stereo for the USB or Bluetooth route)



### ② [Phone Number]

Displays the phone number of the other party during the call.

If the phone number is registered in the "Contact List", the name will be displayed.

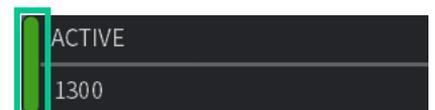
- Applicable routes: SIP1/SIP2



### ③ [Input Indicator]

Lights up in green when audio input is detected.

- Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX



④ [Hold or Resume]

This button switches between holding and resuming the call on the corresponding route.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX



On hold



Resumes call

⑤ [Offhook]

This button starts the call on the corresponding route. With SIP1/SIP2, it is used to make or answer a call.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth



⑥ [Hangup]

This button disconnects the call on the corresponding route.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth



⑦ [Join or Split]

This button mixes input from the corresponding route, allowing you to join or leave a multi-point call.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX



Leave



Join

## 2 Conference (batch operations) menu

Operations for multi-point calls (conferences) that mix SIP1/SIP2/USB/Bluetooth/AUX routes can be performed as a batch.

⑧ [Hold or Resume]

This button allows you to switch between holding and resuming the calls on various routes.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX



On hold



Resumes call

⑨ [Hangup]

This button disconnects the calls on various routes.

■ Applicable routes: SIP1/SIP2/USB/Bluetooth/AUX



⑩ [Split All]

This button splits the connections of multi-point calls mixing input and output of various routes.



## 3 Audio menu

⑪ [Mic Mute]

This button switches between muting and unmuting the audio of Dante's microphone input (RM-CG, etc.) route and the USB output route. Use this when you do not want the audio from your base to be transmitted to the other party.



Unmuted



Muted

⑫ [Volume]

This slider adjusts the volume of analog speakers and the Dante output. The setting can be adjusted to one of 20 levels.



④ SIP menu

■ Applicable routes: SIP1/SIP2

⑬ [Dialer]

Click this button to display the “Dialer” window for SIP calls.



⑭ [Recent List]

Click this button to display the SIP call history window “Recent List”.



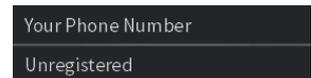
⑮ [Contact List]

Click this button to display the SIP call contacts window “Contact List”.



⑯ [Server Registration]

Displays the SIP number of your device (RM-CR) registered with the SIP server and the status of registration with the SIP server.



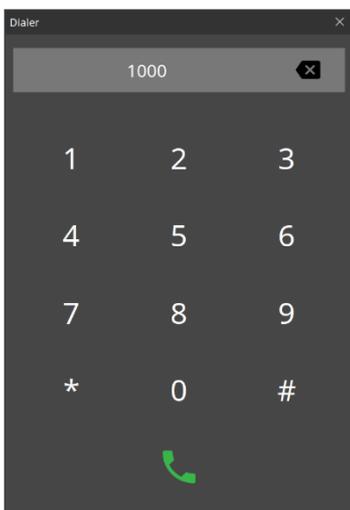
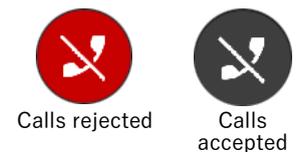
⑰ [Voice Mail Count]

The number of voicemails left on the SIP server is displayed to the right of the icon.

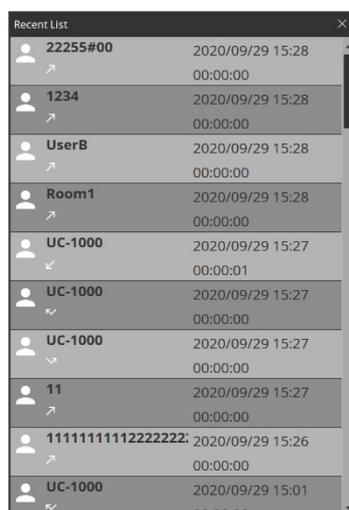


⑱ [Do not Disturb]

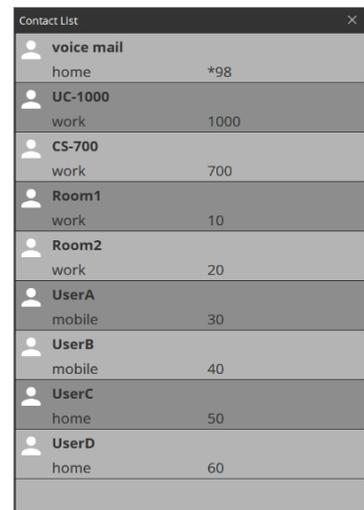
This button switches between turning on and off SIP call rejection.



“Dialer” window



“Recent List” window



“Contact List” window

## Note

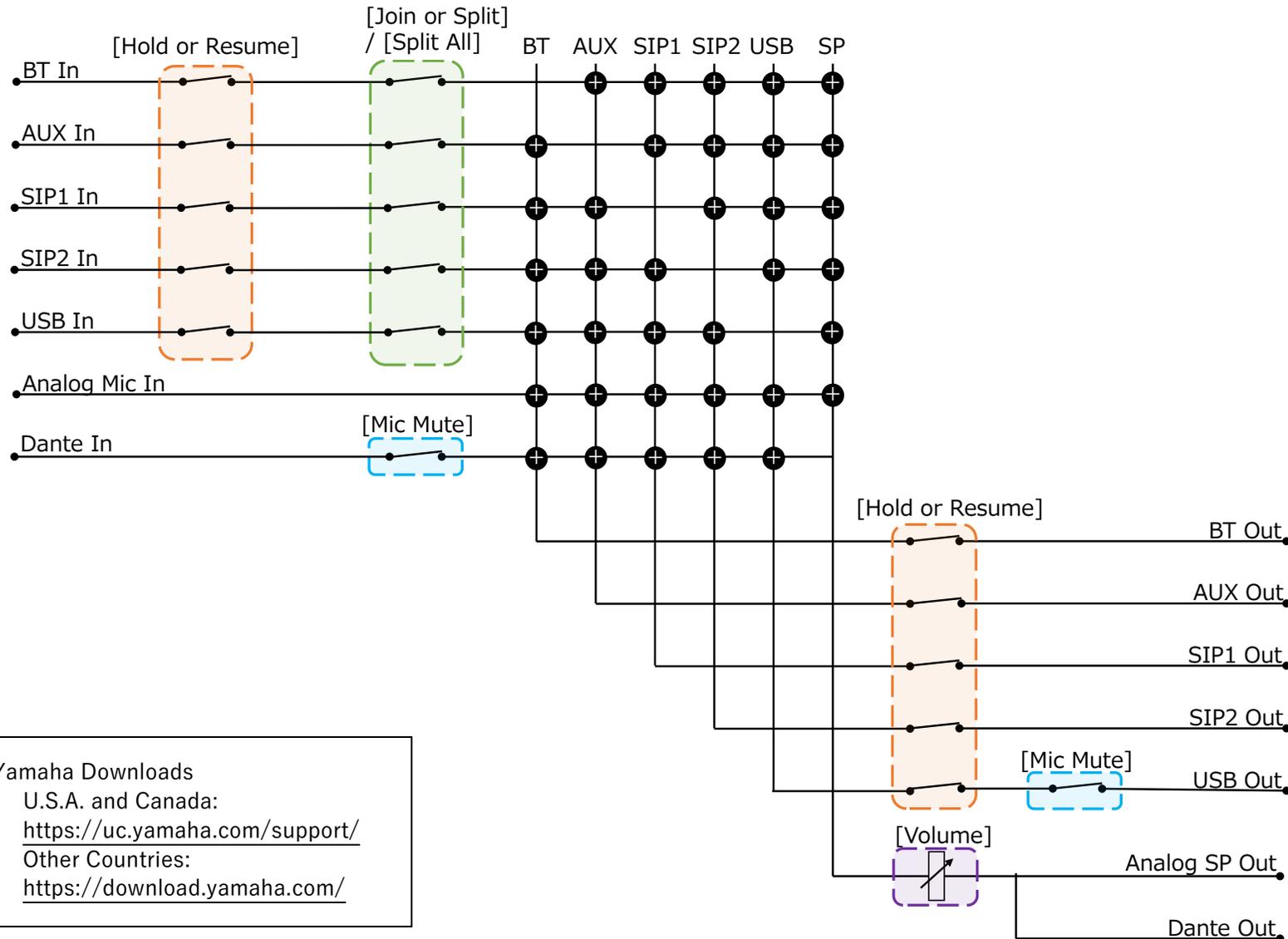
### About automatic connection with [Join or Split] (connection setting for a route used for calls)

- [Join or Split] and [Hold or Resume] cannot be used when [Enable auto-join] is selected (default setting).
- [Enable auto-join] is a parameter that allows you to select whether to automatically join a call. If [Enable auto-join] is selected, [Join] will be automatically performed to make a multi-point call.
- [Enable auto-join] is set via [SETTINGS]→[Dialer]→[Enable auto-join] in the Web GUI “RM-CR Device Manager”.

# Schematic Diagram of Route Control for RM-CR Dialer Template

The schematic diagram below shows the relationship between on-screen controls and audio route control.

For example, it shows that [Join or Split] / [Split All] mixes input from the corresponding route to control the output (such as mixing the SIP input with the USB output).



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