

Prestel CPA Series

NETWORK MULTI CHANNEL CLASS D
POWER AMPLIFIER



IMPORTANT SAFETY INSTRUCTIONS

Please read the basic protective measure as below before using.

1. Read these instructions before use.
2. Heed all warnings.
3. Follow all instructions.
4. Do not block any ventilation openings. Install in accordance with the manufacturer's instructions.
5. Be sure to connect to an appropriate outlet with a protective grounding connection. **Improper grounding can result in electrical shock.**
6. Do not defeat the safety purpose of the polarized or grounding-type plug. A polarized plug has two blades with one wider than the other. A grounding type plug has two blades and a third grounding prong. The wide blade or the third prong are provided for your safety. If the provided plug does not fit into your outlet, consult an electrician for replacement of the obsolete outlet.

WARNING -Improper grounding can result in electrical shock.

7. Keep this unit away from the following locations:
 - (1) Locations exposed to oil splashes or steam, such as near cooking stoves, humidifiers, etc.
 - (2) Unstable surfaces, such as a wobbly table or slope.
 - (3) Locations exposed to excessive heat, such as inside a car with all the windows closed, or places that receive direct sunlight.
 - (4) Locations subject to excessive humidity or dust accumulation.
8. Always lower the volume control to minimum before turning on the power to this unit. A sudden blast of sound may damage your hearing.
9. This unit has ventilation holes at the front and rear to prevent the internal temperature rising too high. Do not block them. Blocked ventilation holes are a fire hazard. In particular, do not:
 - (1) Place the unit on its side or upside down.
 - (2) Place the unit in any poorly ventilated location such as a bookcase or closet (other than on the dedicated rack).
 - (3) Cover the unit with a table cloth or place it on a carpet or bed.
10. Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.
11. Connect this unit's power cord only to an AC outlet of the type stated in this Owner's Manual or as marked on the unit. Failure to do so is a fire and electrical shock hazard.

12. Do not scratch, bend, twist, pull, or heat the power cord. A damaged power cord is a fire and electrical shock hazard.
13. In case an abnormality occurs during operation:
 - (1) If the power cord is damaged (i.e., cut or a bare wire is exposed), ask your dealer for a replacement. Using the unit with a damaged power cord is a fire and electrical shock hazard.
 - (2) Should this unit be dropped or the cabinet be damaged, turn the power switch off, remove the power plug from the AC outlet, and contact your dealer. If you continue using the unit without heeding this instruction, fire or electrical shock may result.
 - (3) If you notice any abnormality, such as smoke, odor, or noise, or if a foreign object or liquid gets inside the unit, turn it off immediately. Remove the power cord from the AC outlet. Consult your dealer for repair. Using the unit in this condition is a fire and electrical shock hazard.
14. Always turn the power off when the amplifier is not in use. Illustrations in this manual are for explanatory purposes only, and may not match the actual appearance of the product during operation.



The lightning flash with arrowhead symbol within an equilateral triangle is intended to alert the user to the presence of uninsulated "dangerous voltage" within the product's enclosure that may be of sufficient magnitude to constitute a risk of electric shock to persons.



The exclamation point within an equilateral triangle is intended to alert the user to the presence of important operating and maintenance (servicing) instructions in the literature accompanying the product.



Applies only to altitudes below 2000m area safety.



Applies only in Non-tropical area safety.

Introduction

Features:

- Digital Power Amplifier Adopting Switching Power Technology
- Class-D PA Power Amplifier with Minimal Power Consumption
- Occupies Less Rack Space and Generates Less Heat
- Multi-Channel Power Amplifier in 19-inch Rack-Mount Unit
- Rated Power Output: 120W–650W
- Multi-Channel Separate Speaker Outputs (4~16Ω/100V)
- Dual-Channel Balanced Line Input via Phoenix Connector
- Each Input with Independent Gain Control
- Each Channel Equipped with High-Cut Filter
- Built-in Auto Standby Feature for Power Consumption Savings
- Separate Channel Indicators for Protection, Clipping, Input, and Output
- Comprehensive Protection against Short Circuit, Overload, High Temperature, Clipping, and DC Issues
- Wide-Range 110V&230V AC Input

Description:

The Series digital Class-D power amplifier adopts switching power technology, featuring minimal power consumption and remarkably higher efficiency—up to 85%. Additionally, it helps save installation rack space, generates less heat, and thus extends its performance lifespan.

With a rated power output ranging from 120W to 650W, these digital class-D amplifiers enable cost-effective deployment in multi-zone, multiple-source public address systems. Their versatile loudspeaker outputs—supporting both high-impedance 100V and low-impedance 4 Ohms—ensure seamless compatibility with both PA fixed installations and Hi-fi stereo sound setups.

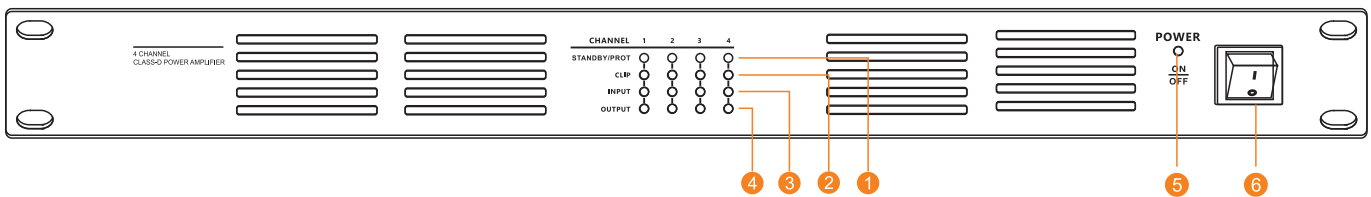
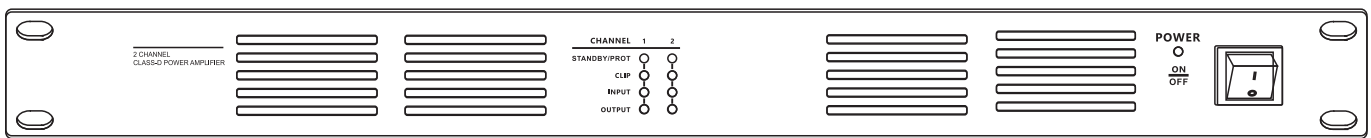
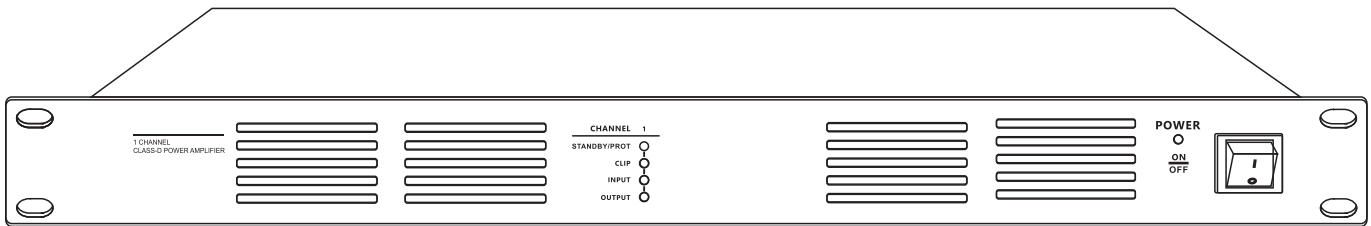
Each channel is equipped with balanced line inputs via a Phoenix connector, featuring gain control. It offers separate zone speaker outputs supporting both 100V and 4 Ohms. The built-in multi-channel separate high-pass filters can be enabled or disabled through dip switch pre-setting.

The amplifier will immediately wake up upon detecting any input. Visual working status indicators—including protection, clipping, input, and output—allow for easy monitoring. It provides comprehensive protection against short circuits, overloads, high temperatures, clipping, and DC issues.

With a wide-range 230V AC power supply, it supports worldwide sound-system installations. Additionally, a 24V

NOTES * DC24V Customization required

FrontPanel



1. PROT

Protection indicator will light on in orange once the amplifier comes into protection status. The protection may be caused by speaker line short circuit, speaker line open circuit, amplifier overload and inner working temperature over 55°C, this indicator gives visual guidance for repair and it will light off once these failures have been repaired. There are separate protection indicators for each channel.

2. CLIP

Clip indicator will light on in red once the input or output is too high. There are separate CLIP indicators for each channel.

3. INPUT

Input indicator will light on in green once there is any input signal has been detected. There are separate input indicators for each channel.

4. OUTPUT

Output indicator will light on in green once there is output signal has been passed. There are separate output indicators for each channel.

5. POWER

Power indicator will be light on in blue once the unit has been powered whether by AC or DC.

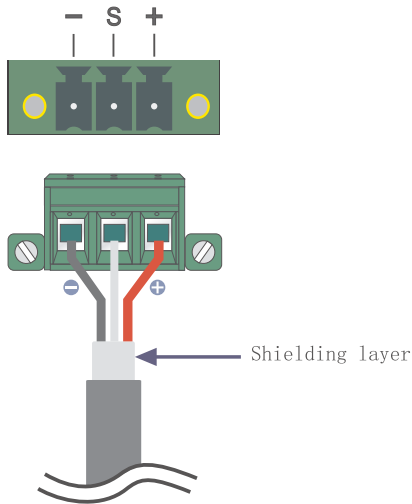
6. POWER SWITCH

Power switch used to power on and power off the equipment.

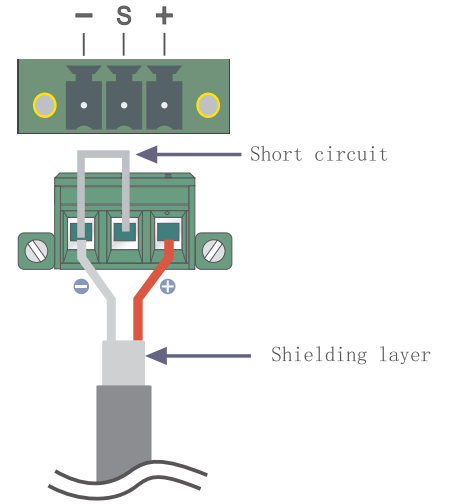
Speaker Connection

- Connecting 4-16 Ω Speaker Systems

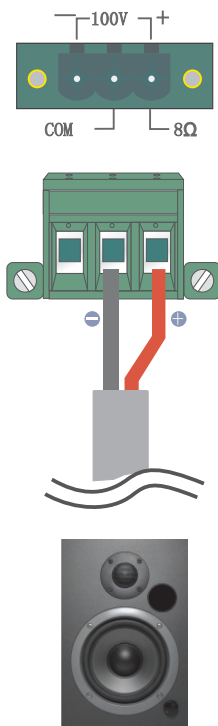
When connecting conventional 4-16 Ω speaker systems, connect the speaker's positive (+) side to the terminal labeled 4-16 Ω . Connect the speaker's negative (-) side to the terminal labeled COM.



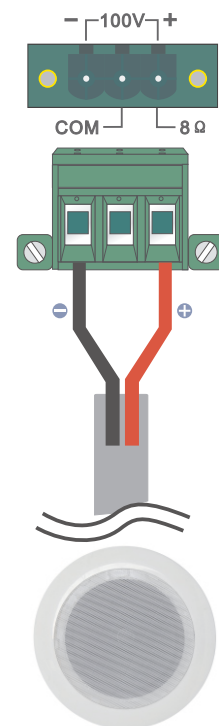
Audio input connection method (balanced input)



Audio input connection method (unbalanced input)

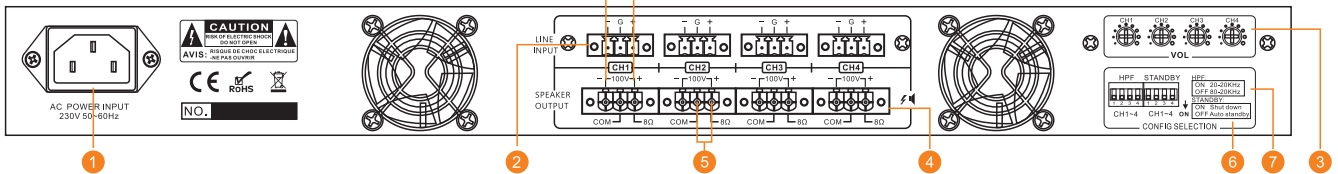
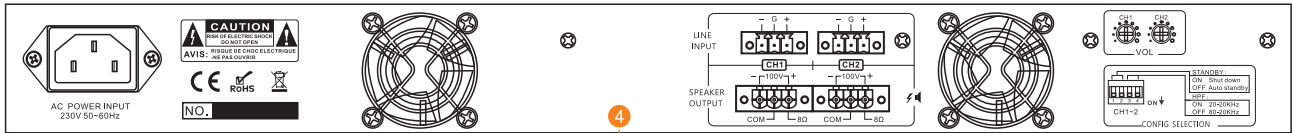
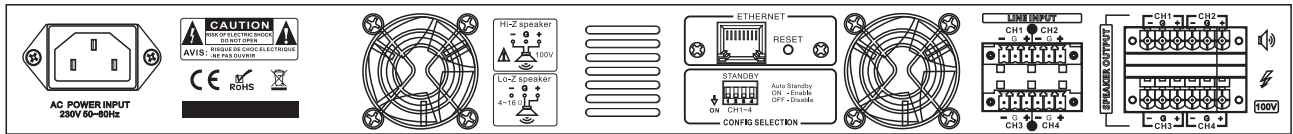


FOR 4~8 Ω LOW IMPEDANCE SPEAKER CONNECTION



FOR 100V SPEAKER WITH TRANSFORMER CONNECTION

Rear Panel



1. AC POWER SOCKET

AC power cord for input of 230V, 50-60Hz for main power input.

2. LINE INPUT(1-4 Channel)

The line input support both balanced or unbalanced inputs from audio sources or mixer, these input are supplied in phoenix connectors. There are two channel separate line inputs which could be managed respectively. (**NOTICE:** For unbalanced line input, "-" and "GND" shall be short-circuited)

3. GAIN CONTROL

The gain control is used to adjustment the input sensitivity from various inputs so as to have the best response. There are separate gain control for each channel. (**NOTICE:** gain control is not output volume control. Units with volume control are available upon request)

4. 100V SPEAKER OUTPUT(PIN1=COM & PIN3=100V +)

The speaker outputs shall be connected to 100V line loudspeakers, the negative end to the speaker negative end and the positive end to the speaker positive end. There are two channel separate 100V speaker outputs which could be managed respectively. These speaker output are supplied in phoenix connector.

Never mixed the negative end and the positive one, which may cause the short circuit protection.

Never connect low impedance loudspeaker to these 100V ends, which may burn up the speaker or have heavy distortion.

5. 8Ω SPEAKER OUTPUT(PIN2=COM & PIN3=4~16 Ohms)

The speaker outputs shall be connected to low impedance 8Ω loudspeakers, the negative end to the speaker negative end and the positive end to the speaker positive end. There are two channel separate 8Ω speaker outputs which could be managed respectively. These speaker output are supplied in phoenix connector.

Never mix the negative and positive ends and the positive one, which may cause the short circuit protection.

Never connection the 100V line loudspeaker to these 8Ω ends, which may burn up the speaker or have heavy distortion.

6. AUTO-STANDBY

The Auto standby function will be enabled or disabled by pre-set the two channel separate dipswitch. The enabled auto-standby function will automatically power come into standby once no input has been detected within 2 minutes. The amplifier will immediately wake up once there is detection of input signal. There are two channel separate auto-standby dipswitch which could be managed respectively.

7. HIGH -PASS FILTER

The high-pass filter function is enabled or disabled by the separate dipswitches for each channel.

Specifications

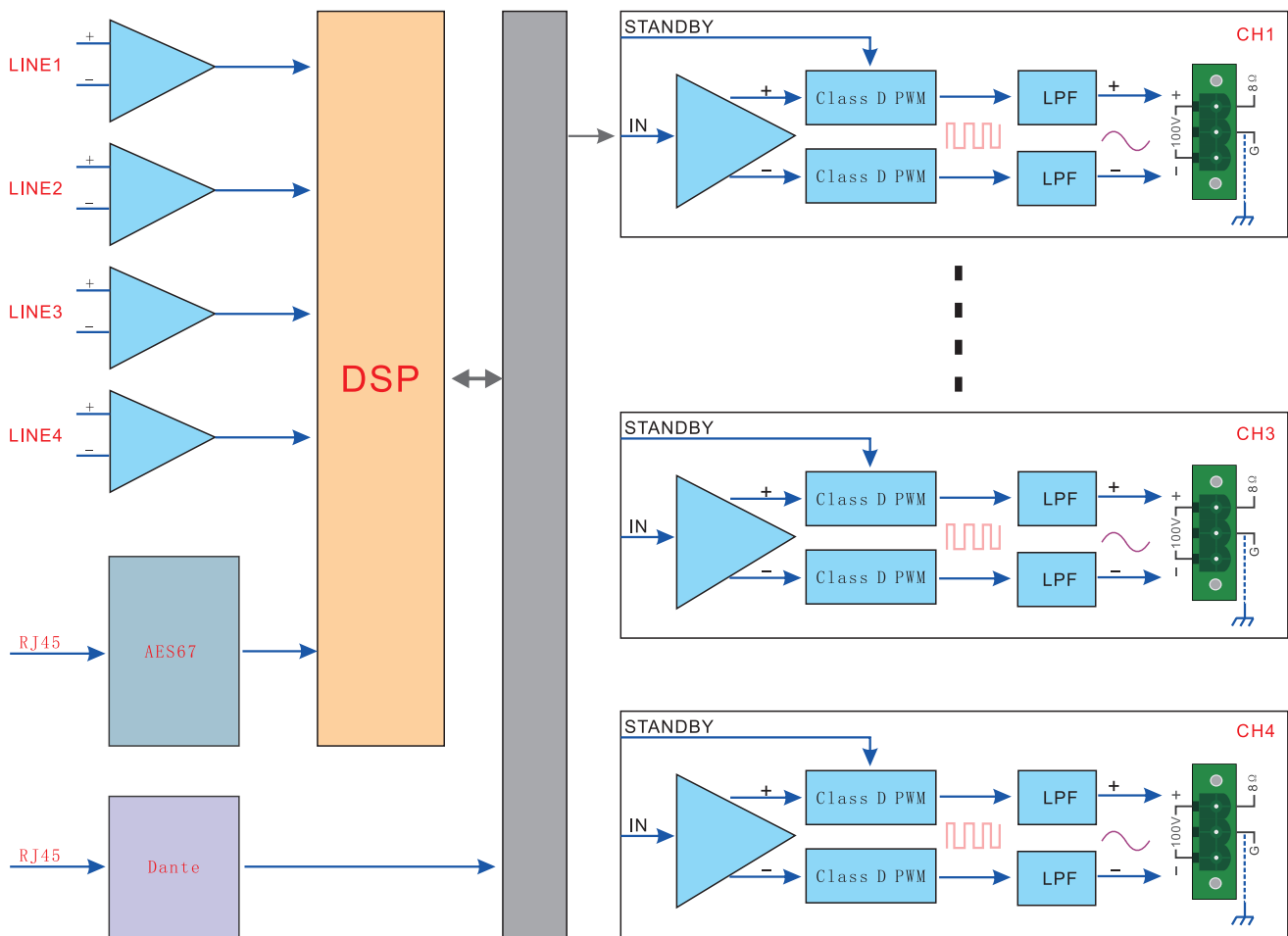
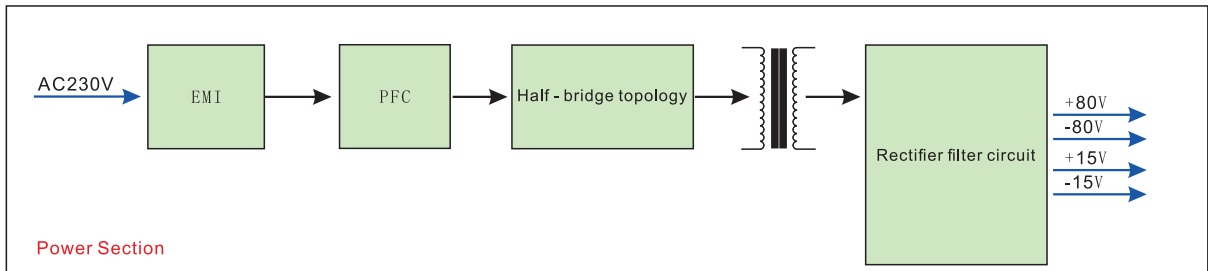
Model	120W	240W	350W	500W
Description	CLASS-D POWER AMPLIFIER			
RatedPowerOutput	1x120W 2x120W 4x120W	1x240W 2x240W 4x240W	1x350W 2x350W 4x350W	1x500W 2x500W 4x500W
SpeakerOutput	4~16Ω & 100V			
FrequencyResponse	L/H Cut OFF 20Hz-20KHz(+1/-2dB) L/H Cut ON 70Hz-20KHz(+1/-3dB)			
Input	0.775V,0dBu,balanced phoenix connector by 1~4 CH			
InputImpedance	10KΩ			
THD	<0.1%(1KHz/-3dBv,100W)			
S/NRatio	>85dB			
Crosstalk	>65dB,1KHz,Maxoutput			
PowerConsumption	150~550W	260~1100W	400~1500W	550~2200W
PowerSupply	Wide AC input: 110V & 230V ,50-60H,			
Dimension	482(W)x420(D)x44(H)mm			
Weight	6.5kg	7.5kg	8kg	8.5kg

NOTES * Customization required

DSP Parameters

Frequency Resp	20Hz~20.0kHz
Dynamic Range	
Distortion	0.01%(THD)
Mixman Delay	7ms
Output gain	-40dB~+6dB 0.5dB
Input gain	-40dB~+6dB 0.5dB
Parametric Equalizer	
Filter	
Gain	±12dB in 0.1dB steps
Central Freq.	20Hz~20kHz 360 ISO
Q value	0.5~128
Shelving	
Lo-shelf	20Hz~1kHz
Hi-shelf	1kHz~20kHz
Shelf gain	±12dB in 0.1dB steps
HPF&LPF	
Filter	
Frequency (high pass)	<10Hz~16.0kHz
Frequency (low pass)	35Hz~22.0kHz
Response curve	Butterworth 6dB 18dB 24dB 48dB Bessel 12dB 18dB 24dB 48dB Linkwitz-Riley 24dB 48dB
Limiter	
Level	-40~+3dB
Attach time	1~100ms
Holding time	0~100ms
Decay time	10~1000ms
Other Functions	SIGNALGEN、NOISEGATE、FBE、DELAY、EXPAND、FIR、EQ、UCKER、LIMIT

Circuit Signal Flowchart



Software User Manual

AES67 Controller 3 performs audio routing, DSP processing and network monitoring on DL module embedded devices and **AES67 Virtual Sound Card**

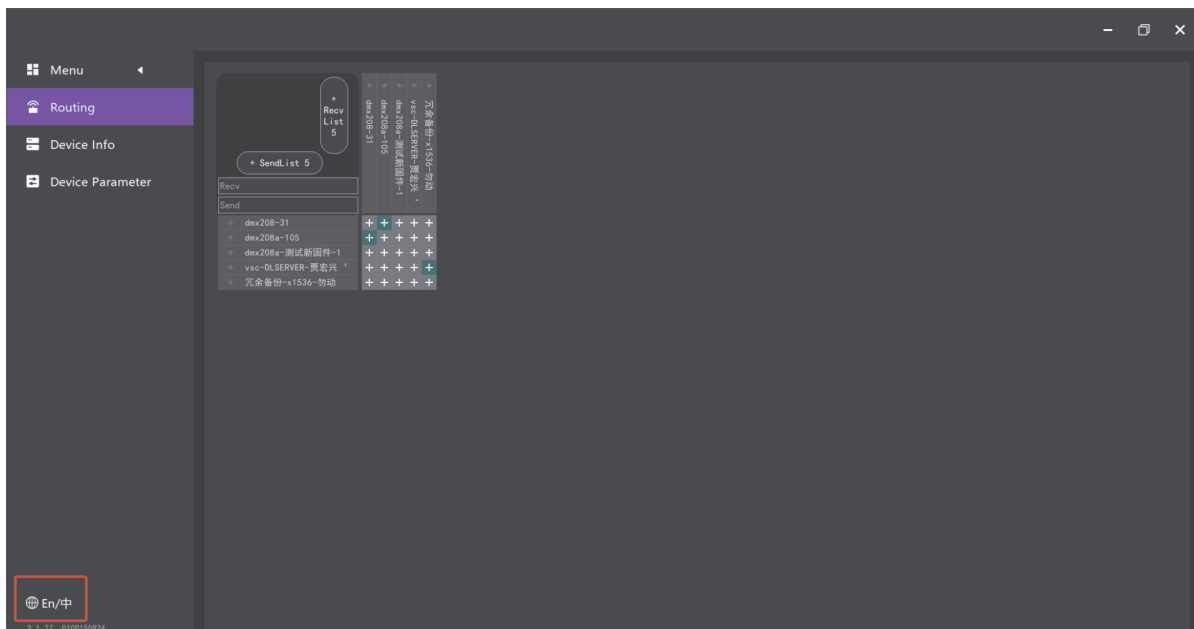
The **AES67 Controller 3** system is optimized for large, demanding audio systems and is an ideal choice for any location that requires a powerful audio processing system and full compliance with broadcasting standards.

1. Software installation

- Double-click the software installation package to install. You can choose the installation language. Simplified Chinese/English is supported. After selection, click .
- Choose whether you need a shortcut. After selection, click .
- If you need to reselect, you can click the previous step to return. If not, click .
- Wait for about 5 seconds for the installation to be completed.
- Click to enter the software.

2. Language switching

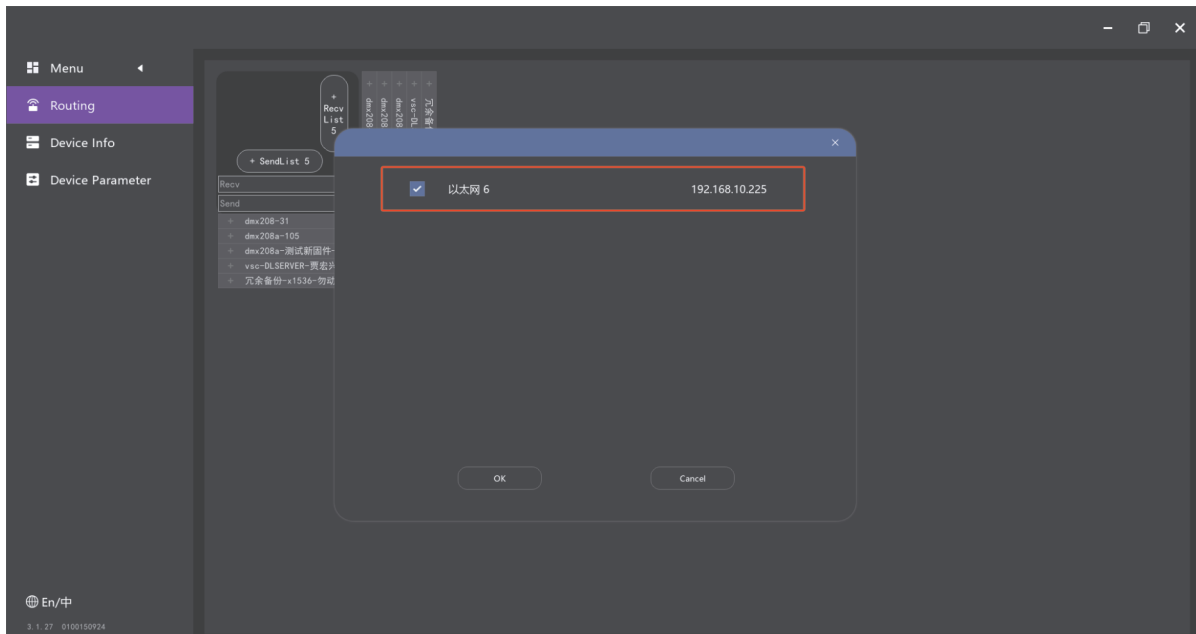
At the lower left corner of the software, click the language button to switch languages. Simplified Chinese/English is supported



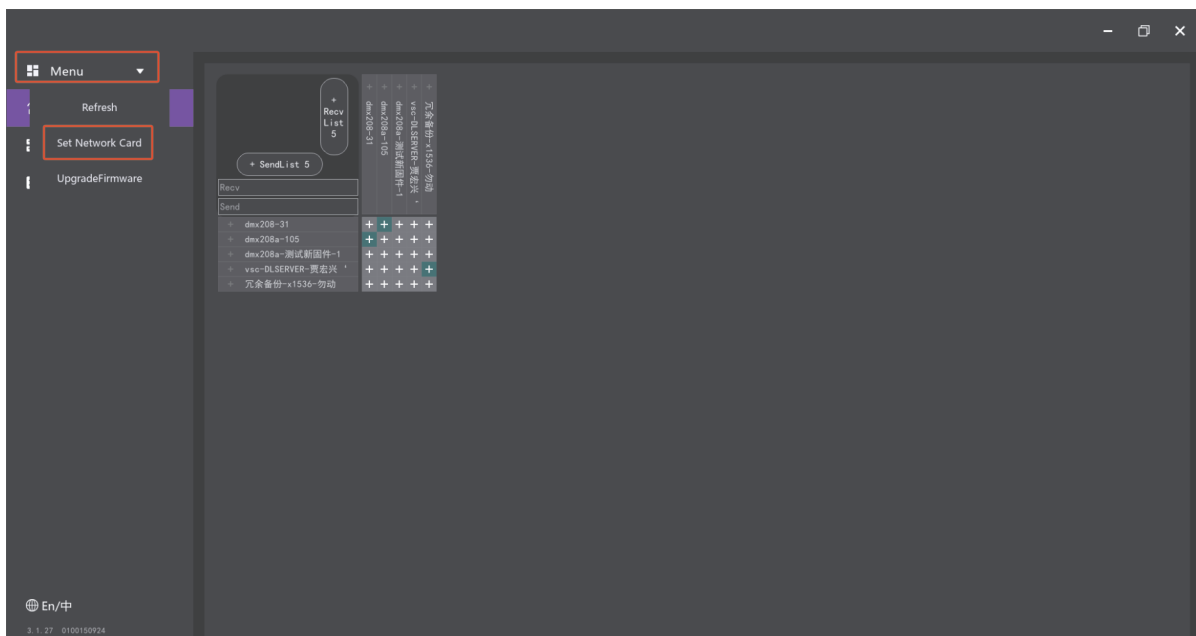
3. Discover device

The software discovers and controls devices through the local area network. When opening the software, it is necessary to specify which network card of the computer is used (**different network cards may correspond to different local area networks**). If there is only one network card, the software will select it by default. If there are multiple network cards,

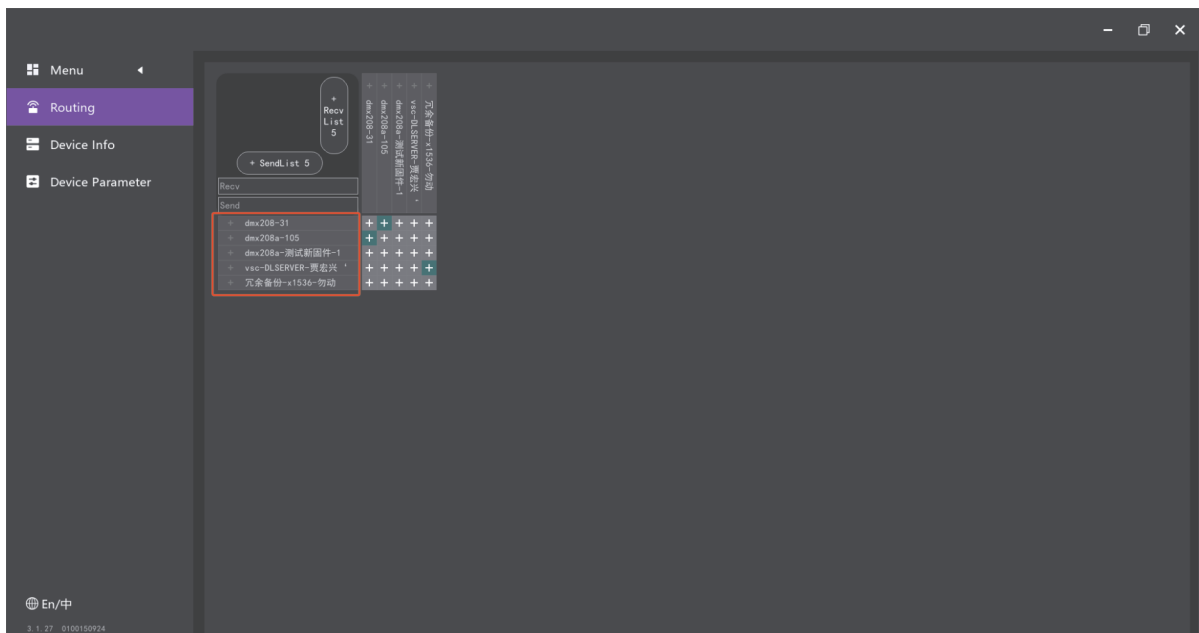
it needs to be manually specified



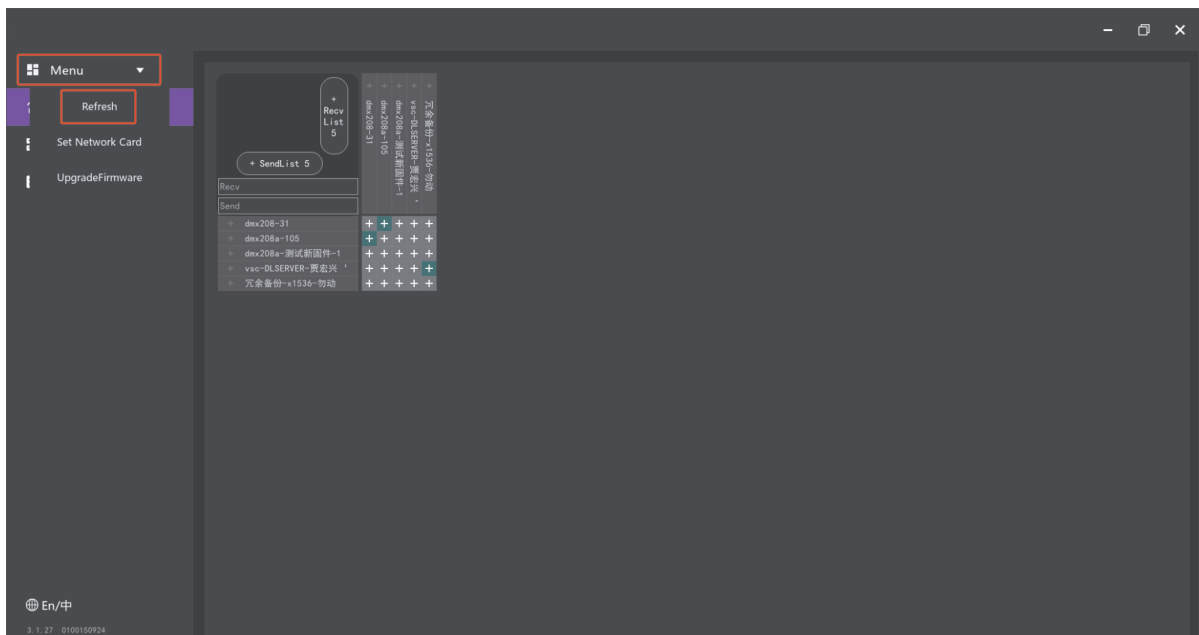
If the network cable of the computer is plugged or unplugged or the network is switched, the corresponding network card needs to be reselected. Click the menu and select it in the drop-down window :



After selecting the network card, the software will automatically discover the devices in the corresponding local area network. Wait for about **5 seconds** to discover the devices



You can also refresh the device list by **Menu - Refresh** :



4. View device information and parameters

On the left side of the software, click **Device Information** column, You can view the basic information of the device

Device Name	Device Type	Device ID	IP Address	Synchronous State	PPM	Master/Slave	Clock Priority	Version
dmx208-31	DMX208	dmx208-31	192.168.10.248	locked	-6	Slave	128	3.1.3
dmx208a-105	DMX208A	dmx208a-105	192.168.10.250	locked	0	Master	128	3.1.3
dmx208a-测试新固件-1	DMX208A	dmx208a-2303e1bb	192.168.10.254	locked	3	Slave	128	3.1.4
vsc-DL SERVER-贾宏兴	VSC	vSndCard-a6999cce	192.168.10.172	locked	0	Slave	NA	NA
冗余备份-x1536-勿动	X1536A0D	x1536a0d-db6fa90	192.168.10.244	locked	-7	Slave	128	3.1.3

- Device name** : The name displayed by the device on the interface. You can be modified in the **Device configuration** .Double-click the name of the device, click Device Configuration, find the device name in the device information column, and then click to modify it

- **Device type**: Used to distinguish device types. Different types correspond to different products, functions, and numbers of channels and cannot be modified.
- **Device ID**: Used to uniquely identify a device. It will not change, will not be repeated, and cannot be modified.
- **Synchronization status**: Judge whether it has been synchronized. This page is refreshed every 5-10 seconds. After the device is connected to the network, it will automatically synchronize with the main device. The synchronization is completed in about one minute.
- **PPM**: Description parameter for clock alignment of master and slave devices. Generally, it is within ± 30 .
- **Master/Slave**: Whether it is a master device or a slave device. There is only one master device in a local area network. If there are other non-DIGISYN LINK devices using PTP clocks, they will also participate in the selection of master and slave devices together. That is, if there is a non-DIGISYN LINK device in the same local area network and its clock priority is higher, all DIGISYN LINK devices on the software may act as slaves.
- **Clock priority**: A parameter used to select master and slave devices. The range is 0-255. The smaller the value, the higher the priority. The default is 128. The system will automatically select the device with the highest priority (smallest value) in the network as the master device. You can manually specify a certain device to be the master (increase the priority). **Virtual sound cards (VSC) and some special devices cannot be masters. The clock priority cannot be modified and is displayed as NA.**
- **Version**: Displays the firmware version number of the device. The version of the virtual sound card is viewed on the virtual sound card software. This page shows NA.

Device Name	Device Type	Device ID	Sampling Rate	Packet Time	Recv/Send Bandwidth	Latency	Temperature	Elapsed Time	Default
dmx208-31	DMX208	dmx208-31	48000	0.25ms	4.1/3.6 Mbps	5ms	68.1°C	178:40:4	
dmx208a-105	DMX208A	dmx208a-105	48000	0.25ms	3.6/4.1 Mbps	5ms	68.5°C	178:40:4	
dmx208a-测试新固件-1	DMX208A	dmx208a-2303e1bb	48000	0.25ms	0.0/0.0 Mbps	5ms	58.6°C	178:40:0	
vsc-DLSERVER-贾宏兴	VSC	vSndCard-a6999cce	48000	1.00ms	0.0/2.0 Mbps	4ms	NA	NA	
冗余备份-x1536-勿动	X1536A0D	x1536a0d-dbf6a90	48000	0.50ms	2.0/0.0 Mbps	5ms	60.6°C	48:7:53	

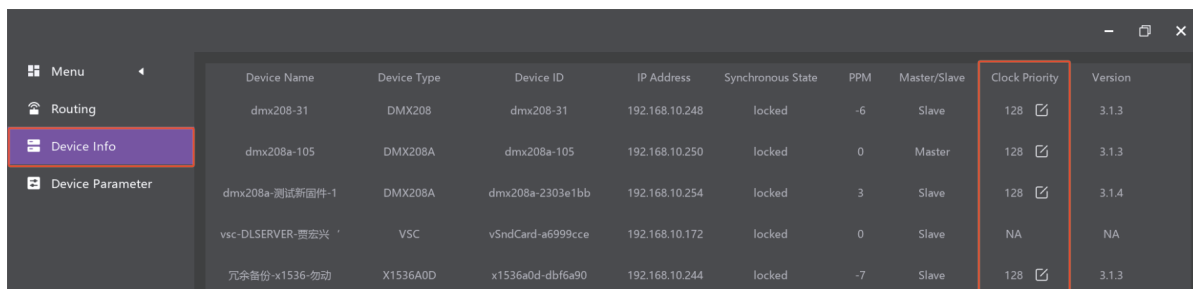
On the left side of the software, click the **device parameter** column, You can view the parameter information of the device

- **Device name**: Same as above. Device type: same as above. Device ID: same as above.
- **Sampling rate**: The audio sampling rate of the device. If audio is transmitted between different devices, make sure the sampling rates are the same. It can be modified in "Device Configuration".
- **Packet time**: A setting parameter for network audio data packets. It can be modified in "Device Configuration".
- **Running time**: The continuous running time of the device since this startup. It can be sorted from largest to smallest or from smallest to largest.

- **Network latency**: A setting parameter for network audio transmission latency. It can be modified in "Device Configuration".
- **Temperature**: The temperature of the core part of the device. During normal use, control the temperature below 70°C and do not exceed 80°C.
- **Clock priority**: It is a parameter used to select master and slave devices. The range is 0 to 255. The smaller the value, the higher the priority. The default is 128. The system will automatically select the device with the highest priority (smallest value) in the network as the master device. You can manually specify a certain device to be the master (increase the priority). Virtual sound cards and some special devices cannot be masters and cannot modify the clock priority, so "NA" is displayed.
- **Version**: It shows the firmware version number of the device. The version of the virtual sound card can be viewed on the virtual sound card software. This page shows "NA".

5. Clock priority setting

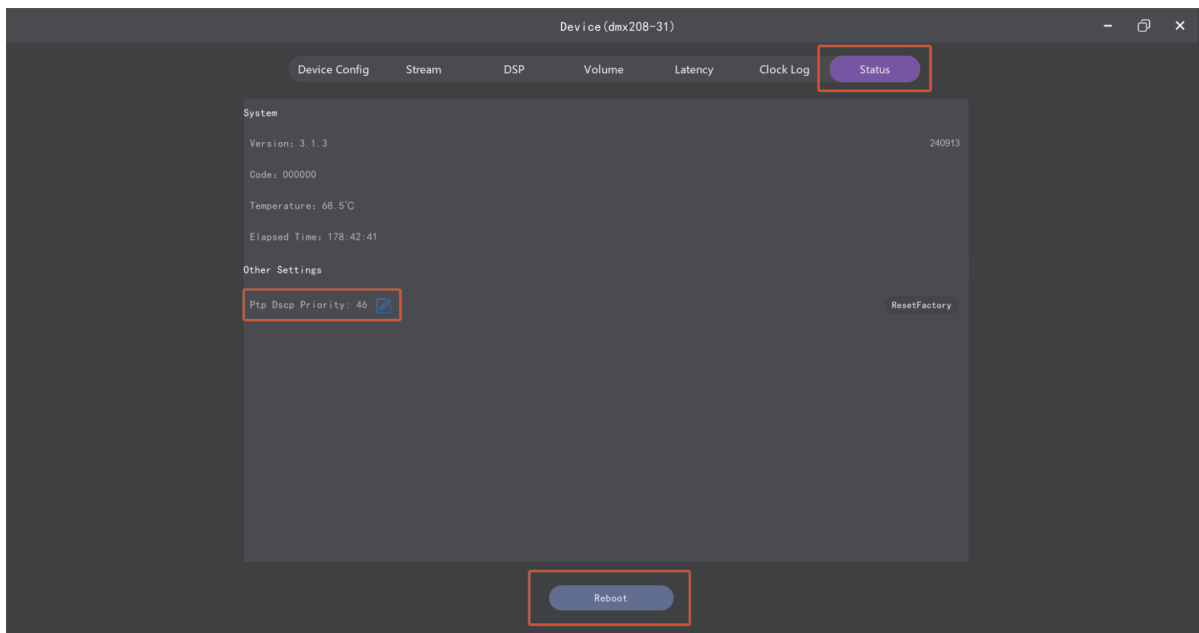
It can be modified in the "Device Information" column. Click **Device Information** to enter. Select the required device in the clock priority column for modification



Device Name	Device Type	Device ID	IP Address	Synchronous State	PPM	Master/Slave	Clock Priority	Version
dmx208-31	DMX208	dmx208-31	192.168.10.248	locked	-6	Slave	128	3.1.3
dmx208a-105	DMX208A	dmx208a-105	192.168.10.250	locked	0	Master	128	3.1.3
dmx208a-测试新固件-1	DMX208A	dmx208a-2303e1bb	192.168.10.254	locked	3	Slave	128	3.1.4
vsc-DLSERVER-贾宏兴	VSC	vSndCard-a6999cce	192.168.10.172	locked	0	Slave	NA	NA
冗余备份-x1536-勿动	X1536A0D	x1536a0d-dbf6a90	192.168.10.244	locked	-7	Slave	128	3.1.3

6. PTP DSCP Priority setting

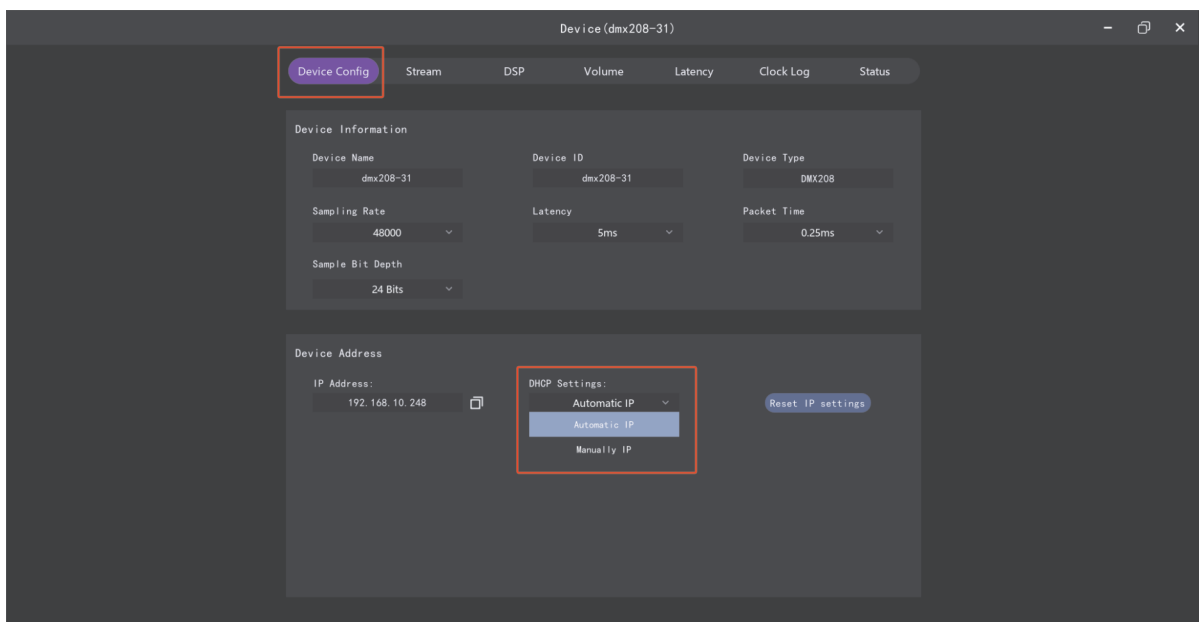
It can be modified on the **Device Details - Others** page. After modification, click **Restart** to complete:



This parameter, in switches that support QoS, will affect the priority of the audio transmission stream. The range is 0 to 63, and the default is 46. The larger the value, the higher the priority and the higher the transmission stability.

7. IP Address setting

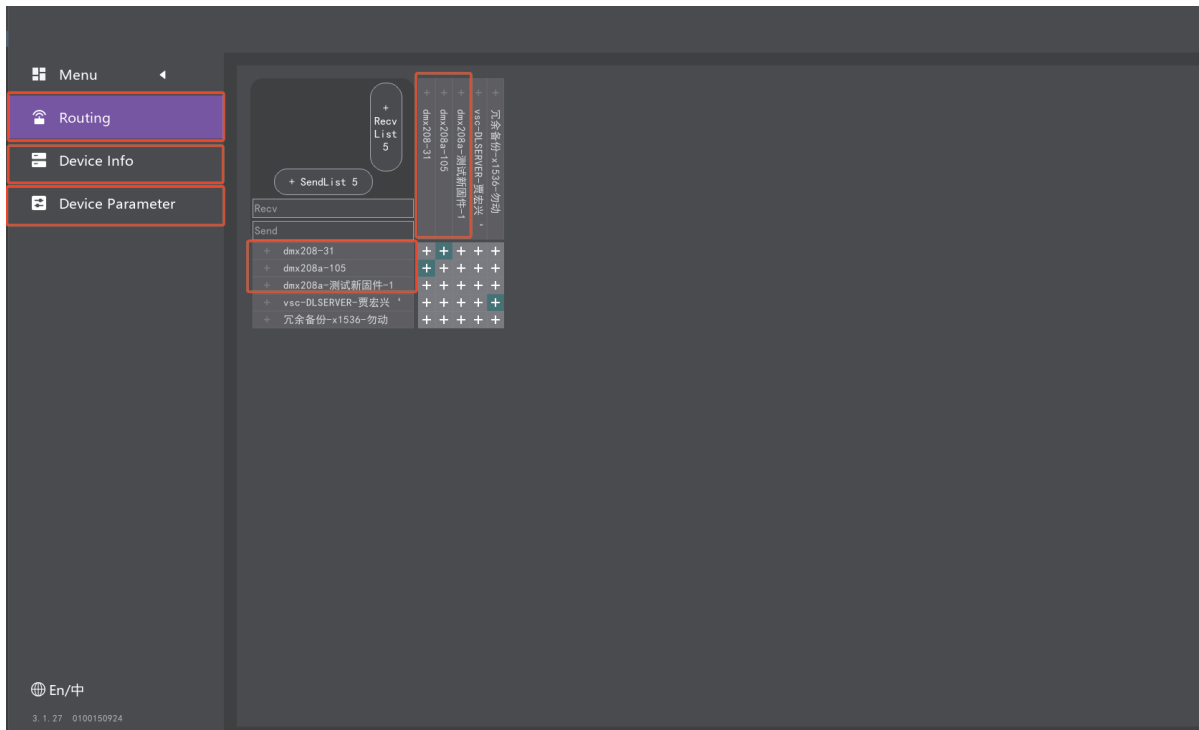
It can be modified on the [Device Details - Device Configuration](#) page of device Config. Two methods are supported: automatic acquisition and manual setting



- If it is automatic acquisition, make sure that the IP address of the computer is also obtained automatically, or manually set the same network segment.
- If it is manual setting, make sure that the IP of the computer is also manually set and set in the same network segment.

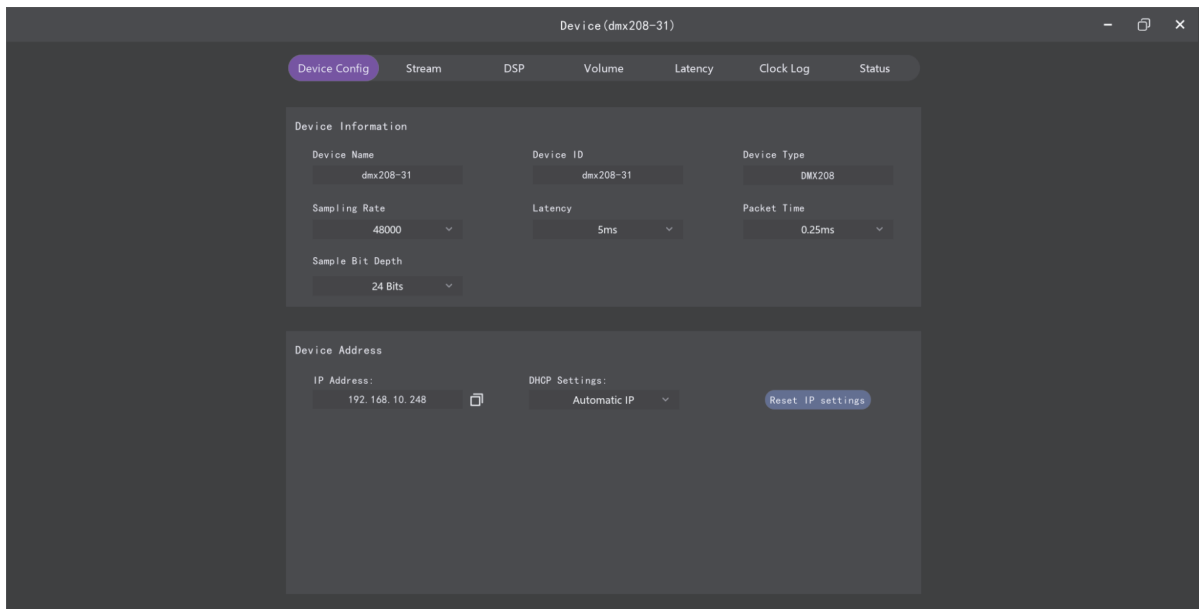
If the IP addresses of the computer or device are not in the same network segment, it will cause the device to be unable to be viewed/controlled normally (except when the switch is set to support cross-network segments).

8. Device details



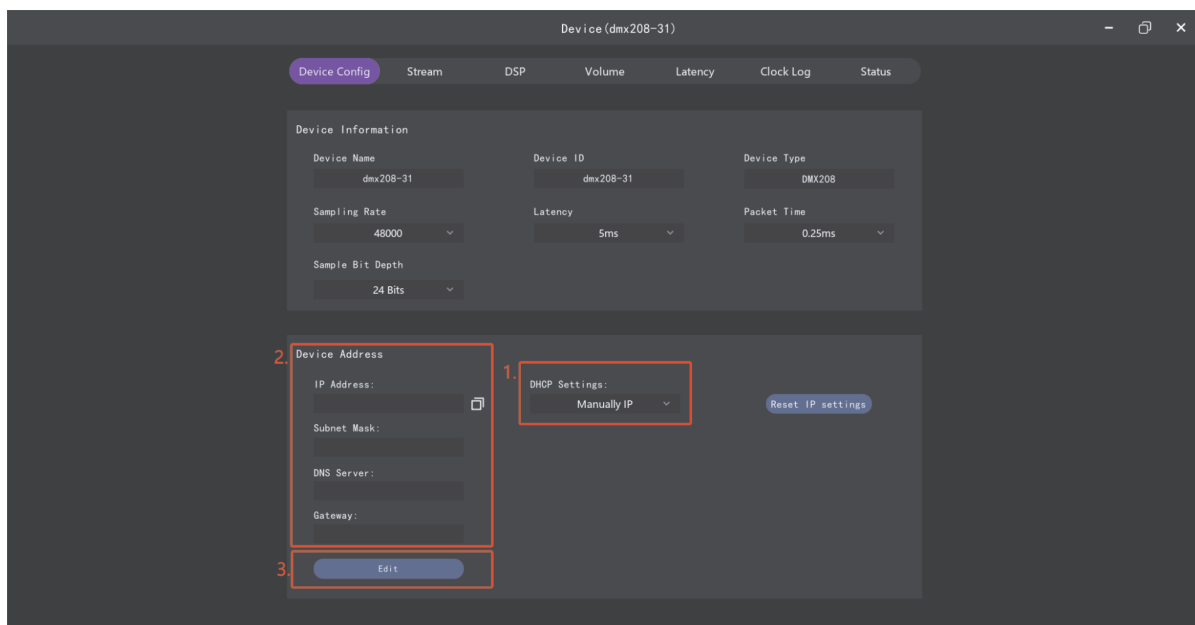
On any page of [Device Routing](#)、[Device Information](#)、[Device Parameters](#) , double-clicking the device name or device row will pop up the [Device Details](#) page:

8.1 Device configuration

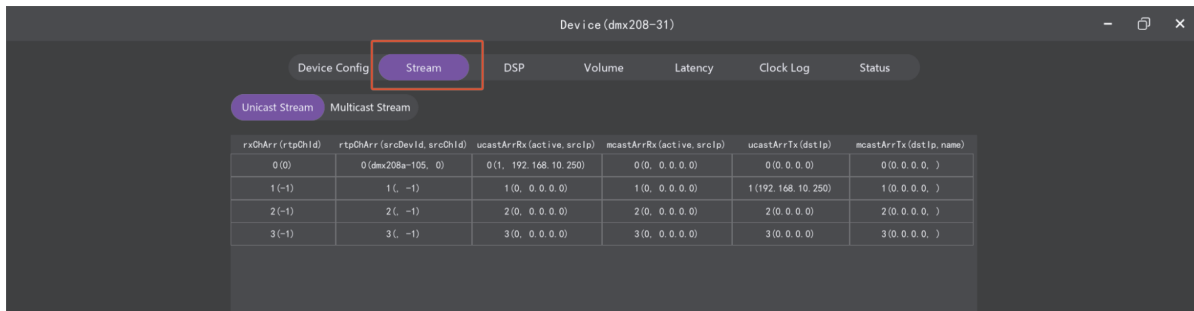


- [Device name](#) : The name of the device can be modified. It cannot exceed 32 characters. By default, it is the same as the device ID.
- [Device ID](#) : The unique ID for identifying the device cannot be modified.
- [Device type](#) : Used to distinguish the device type and cannot be modified.
- [sampling rate](#) : Audio sampling rate. It is distinguished according to the device type. For some types, it can be modified.

- **network latency** : Network latency parameter. The smaller the value, the smaller the latency. However, it requires a better sending device and network environment. If you need the smallest possible latency, it can be set to 1ms (requiring a sufficiently good sending device and network environment). If you need better compatibility with the sending device and network environment, it can be set to 10ms. The default is 5ms. If using a virtual sound card as the sending device, since the virtual sound card depends on the computer system, it cannot be set to 1ms. You can set it to 5ms or 10ms according to the actual situation. (If there is discontinuity or packet loss at 5ms, it can be changed to 10ms).
- **Packet time** : A parameter for network data packets. The network latency must be not less than 4 times the packet time. That is, if the packet time is set to 0.25ms, the network latency can be set to a minimum of 1ms. If the packet time is set to 1ms, the network latency can be set to a minimum of 5ms and cannot be set to 1ms. If you want to set the network latency to 1ms, you must first change the packet time to 0.25ms.
- **IP address** : Display the IP address of the current device.
- **DHCP settings** : The acquisition method of the device IP address supports automatic acquisition and manual setting. When setting manually, first click **Modify**, then enter the specified value in the corresponding input box, and click **Submit**. When setting manually, it is necessary to ensure that the IP address, subnet mask, DNS server address, gateway, etc. are correct. Otherwise, it may not be possible to view/obtain device information normally.
- **Reset IP address** : If the IP address is in manual setting mode and incorrect parameters are set, resulting in the software being unable to view/obtain device information, the IP can be reset to change the device to automatic acquisition mode.



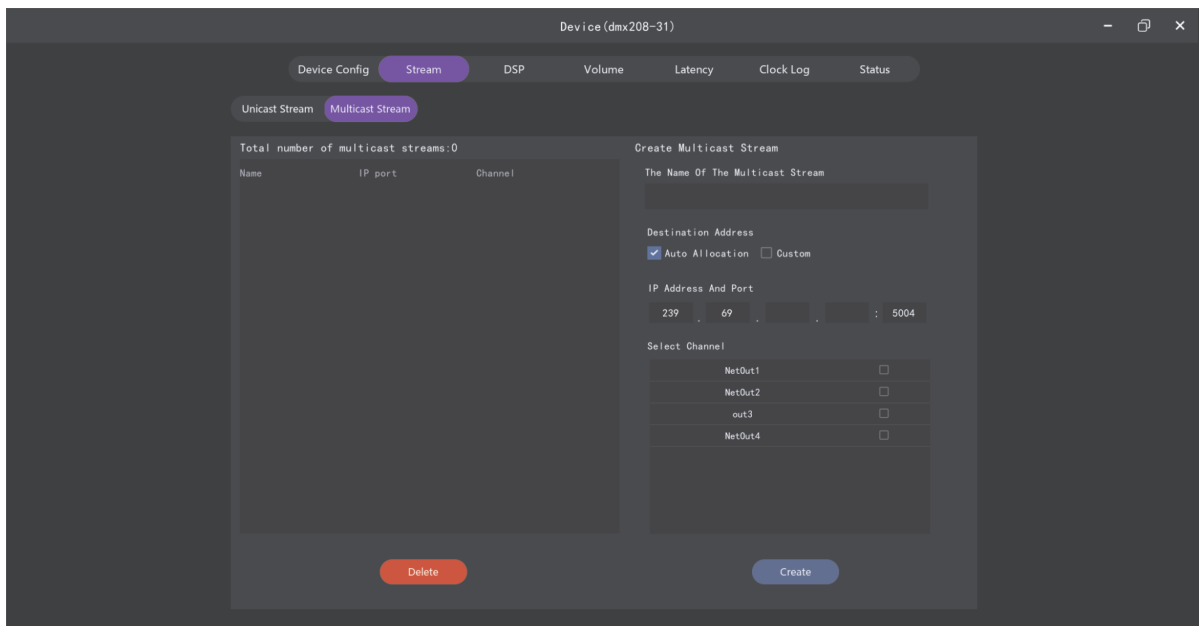
8.2 transport stream



rxChArr (rtpChId)	rtpChArr (srcDevId, srcChId)	ucastArrRx (active, srclp)	mcastArrRx (active, srclp)	ucastArrTx (dstIp)	mcastArrTx (dstIp, name)
0 (0)	0 (dmx208a-105, 0)	0 (1, 192.168.10.250)	0 (0, 0.0.0.0)	0 (0.0.0.0)	0 (0.0.0.0)
1 (-1)	1 (-1)	1 (0, 0.0.0.0)	1 (0, 0.0.0.0)	1 (192.168.10.250)	1 (0.0.0.0)
2 (-1)	2 (-1)	2 (0, 0.0.0.0)	2 (0, 0.0.0.0)	2 (0.0.0.0)	2 (0.0.0.0)
3 (-1)	3 (-1)	3 (0, 0.0.0.0)	3 (0, 0.0.0.0)	3 (0.0.0.0)	3 (0.0.0.0)

Unicast stream: This page displays the unicast receiving, sending routing, stream records and status of the device. (This data is updated every time the device details are opened.)

- **rxChArr(rtpChId)**: The routing records of the receiving channels. The first number represents the channel ID (0 represents channel 1). The content in parentheses represents the subscript of rtpChArr, corresponding to the index of the rtpChArr(srcDevId,srcChId) list. -1 means there is no routing. For example, 0(-1) means that there is no routing record for receiving channel 1. 4(4) means that there is a routing record for receiving channel 5, and the routing record corresponds to the record 4(dmxx208-5a7b066, 0) with an index of 4 in the rtpChArr(srcDevId,srcChId) list. The routing records are generated and deleted when checking the network audio routing on the routing interface. Generally, there is no need for manual maintenance. If the other device associated with the routing is offline and the routing record needs to be deleted, you can double-click the corresponding channel in the rxChArr(rtpChId) column to delete it.
- **rtpChArr(srcDevId, srcChId)**: srcDevId corresponds to the ID of the sending device. srcChId corresponds to the channel index of the sending device. 0 represents channel 1.
- **ucastArrRx(active, srclp)**: It represents the array of unicast received data streams. "active" indicates whether it is active. 0 means inactive, and 1 means active. Only in the active state can audio streams be transmitted. "srclp" represents the IP address of the sending device.
- **mcastArrRx(active, srclp)**: Represents the array of multicast received data streams. "active" indicates whether it is active. 0 means inactive, and 1 means active. Only in the active state can audio streams be transmitted. "srclp" represents the IP address of the sending device.
- **ucastArrTx(dstIp)**: Represents the array of sending channels that have currently established connections (the array index has nothing to do with the channel). "dstIp" represents the IP address of the receiving device.
- **mcastArrTx(dstIp, name)**: The array of multicast sending streams that have been created. "dstIp" represents the multicast stream address. "name" represents the multicast stream name.



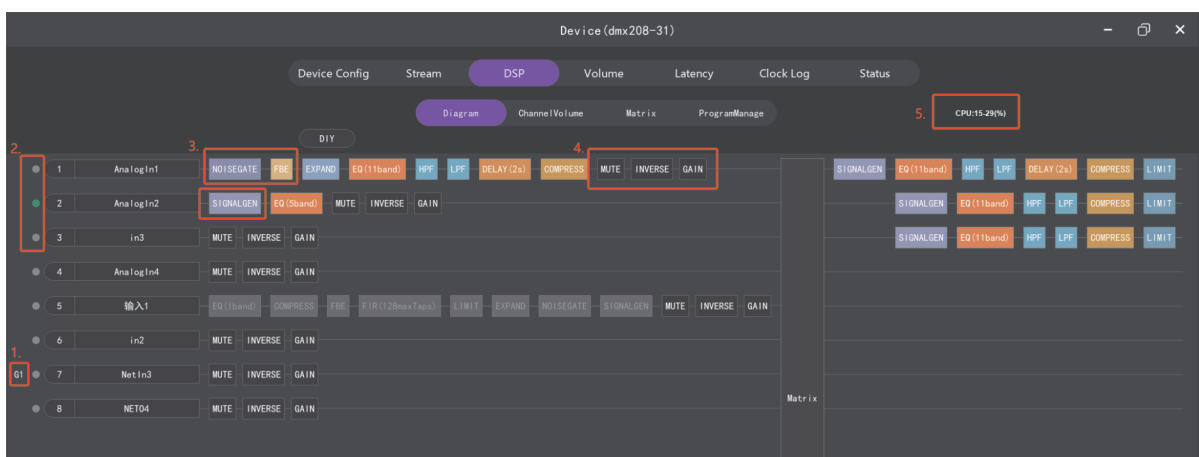
`multicast stream` : This page can create, view, and delete multicast streams.

`Multicast stream name` : The name displayed after creation. It should not exceed 32 characters.

`destination address` : The address of the multicast stream. If there are no special requirements, automatic allocation can be selected. You can also customize the address.

`Select channel` : The sending channels that the multicast stream needs to include. The selected channels can be sent out using the multicast stream.

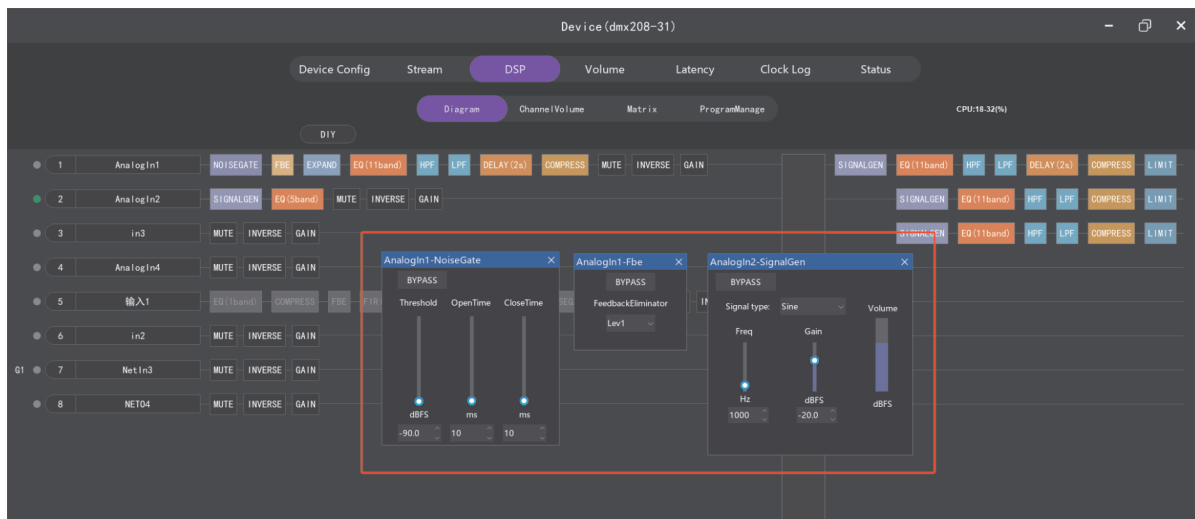
`delete` : You can delete the created multicast stream.



`flow chart` : Here you can view the functions of DSP input and output channels.

1. The area represents the groups that the channels join. After the channels join the groups, they will be displayed here.
2. The area represents the channel level indication. Green indicates there is a level, and gray indicates there is no level.
3. The area represents the DSP function of the channel. Bright color indicates that the function is enabled (non-pass-through), and gray indicates that the function is not enabled (pass-through). Double-clicking the corresponding function box will pop up the detailed parameter setting page. Multiple detailed parameter setting

pages can be opened at the same time.
The DSP functions of different channels may be different.



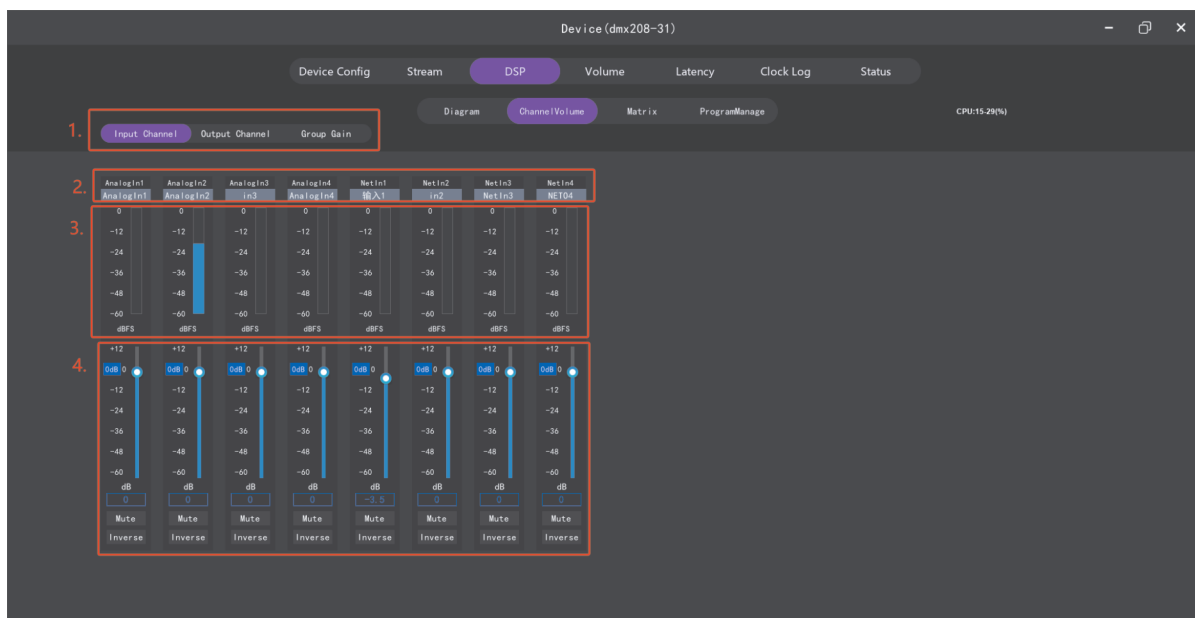
4. The area represents the mute, reverse, and gain functions. Mute and reverse can be toggled by clicking. Double-click gain to open the adjustment window.

5. The area represents the current CPU usage. Enabling DSP functions (non-pass-through) will increase CPU usage, and disabling DSP functions (pass-through) will reduce CPU usage. CPU usage should not exceed 85%.

Right-click on the channel number to pop up the function menu. You can copy and paste channel parameters and add/cancel channel grouping.

Right-click on the DSP function block to pop up the function menu. You can copy and paste function parameters.

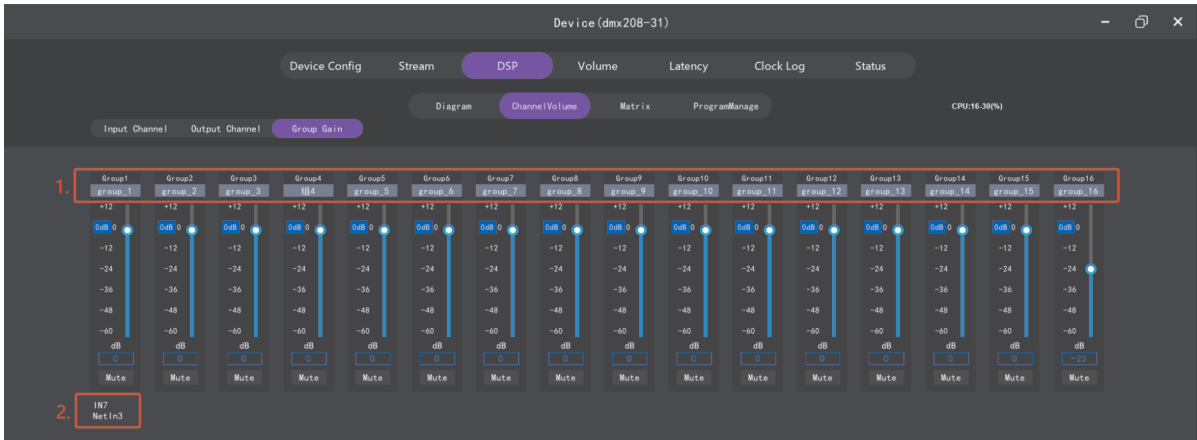
8.3 Channel volume and grouping



1. The area represents input, output channels, and group switching.

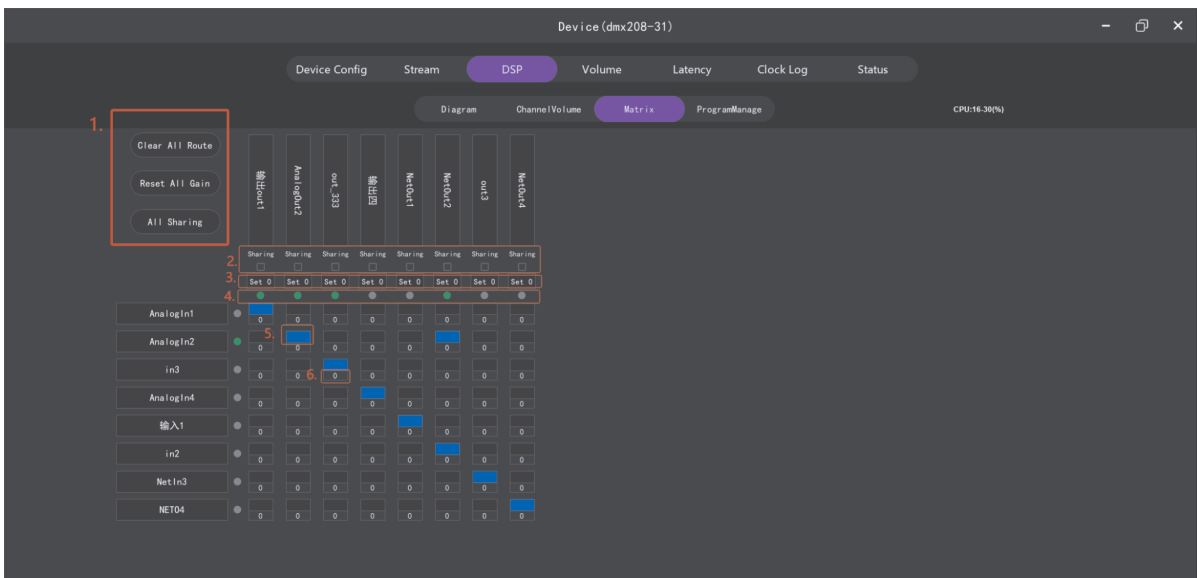
2. The area represents the channel name. The gray part cannot be modified. It indicates whether it is an analog or network channel and the channel subscript. The black name can be modified by yourself. The length should not exceed 32 characters.

3. The area refreshes the level of the channel in real time.
4. The area represents channel gain, mute, and reverse.

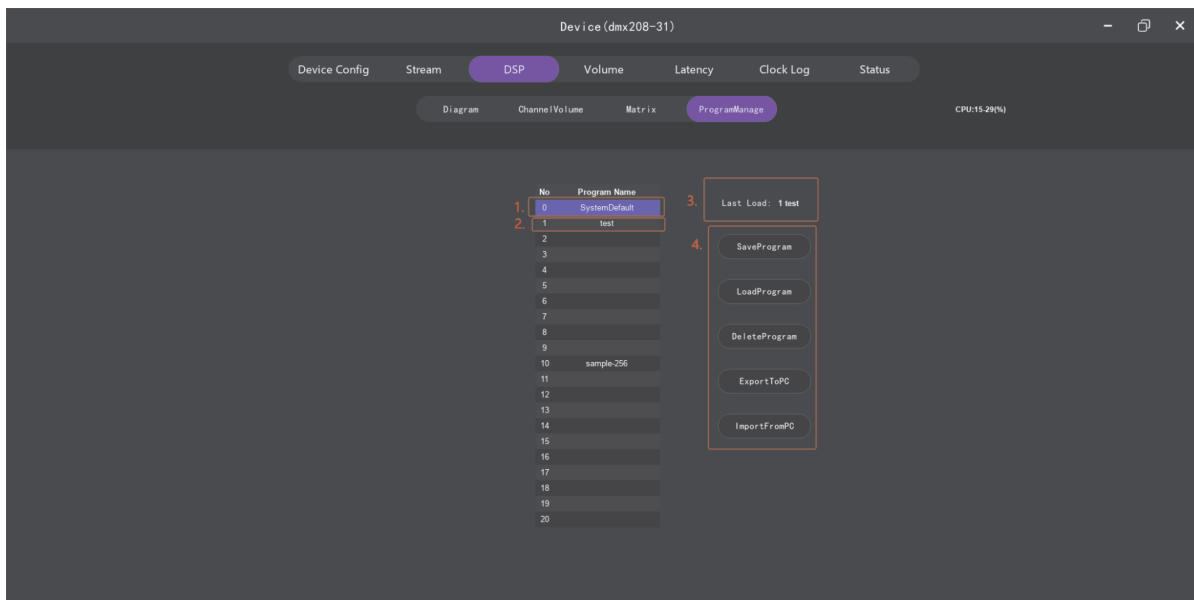


1. The area represents the group name.
2. The area indicates which channels are under the corresponding group. It is divided into two parts. The upper part is the input/output channel and subscript, and the lower part is the channel name.

Matrix control :

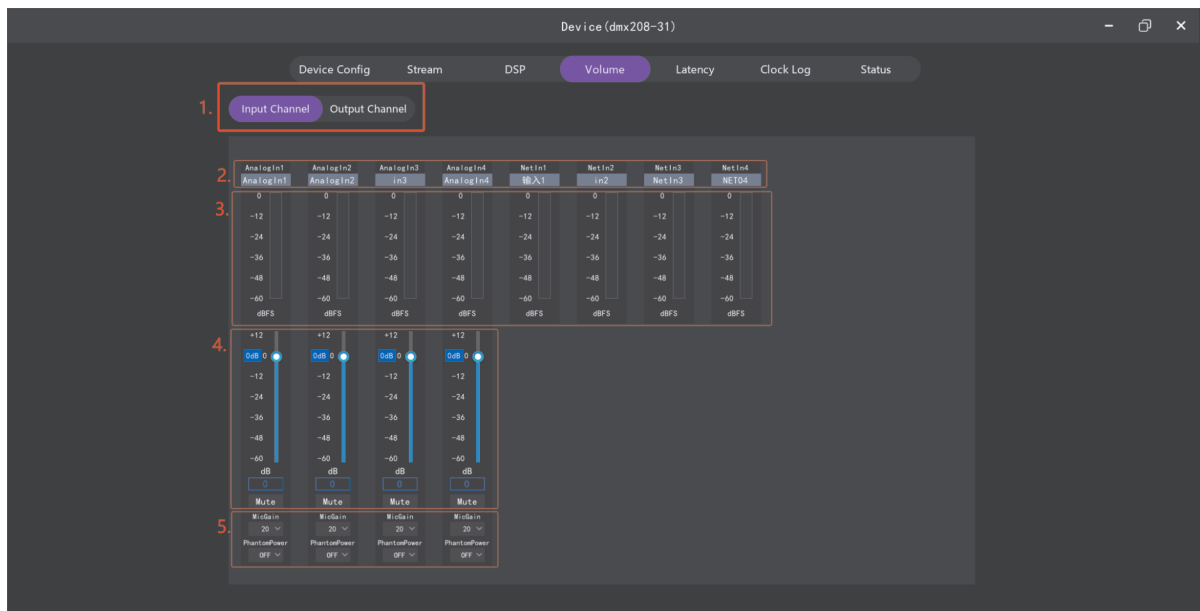


Program management :



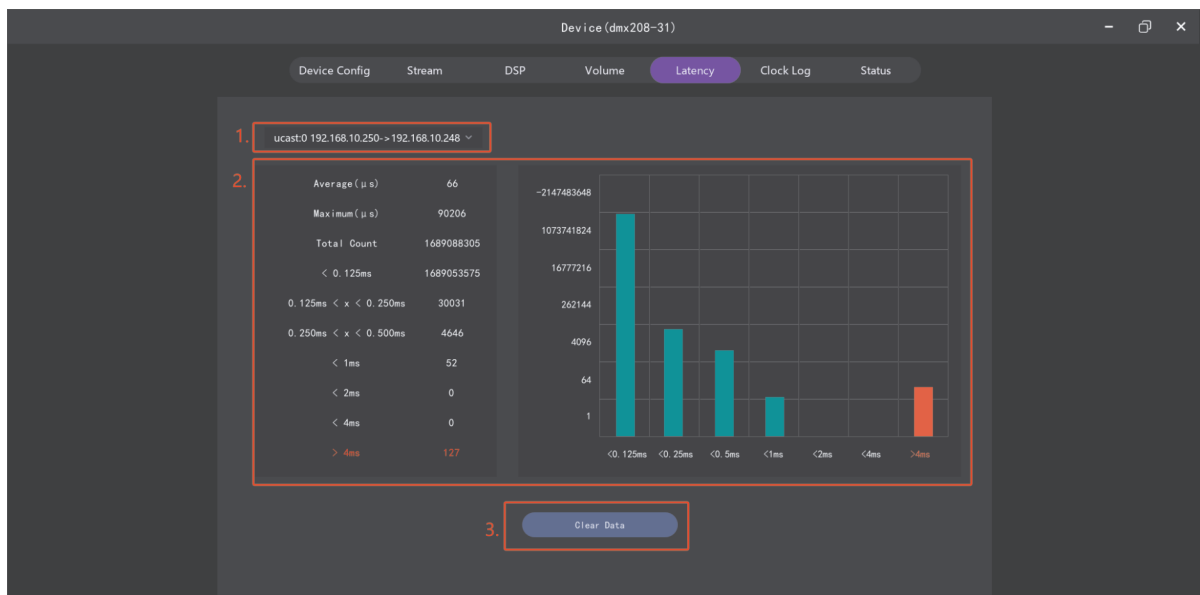
1. The area represents the system's default program. This program cannot be modified, deleted, saved, etc.
2. The area represents the programs saved by yourself. 1-20 are customizations, supporting 20 custom programs.
3. The area is the program manually loaded last time. If after loading, the parameters are manually adjusted, then the parameters will be saved in the state before shutdown and will not be automatically saved to the corresponding program. That is, if program A is manually loaded and then the parameters are modified, then after shutdown and restart, the modified parameters will still be effective. However, if program A is reloaded, the modified parameters will be lost. If you want to save the modified parameters to program A, you need to manually click "Save Program" to save them.
4. The area is the management function area.
 - Save Program: Select the corresponding serial number in the list to save the current machine parameters to the program.
 - Load Program: Select the corresponding program in the list to load the program parameters into the machine.
 - Delete Program: Select the corresponding program in the list to delete the program from the machine.
 - Export to PC: Select the corresponding program in the list to export the program to a computer file.
 - Import from PC: Select the corresponding serial number in the list to import a computer file into the machine.

8.4 Channel volume



1. The area represents the switching between input and output channels.
2. The area represents the channel name. The gray part cannot be modified and distinguishes between analog input, network input and channel number. The black part can be modified and the length should not exceed 32 characters.
3. The area represents the channel level.
4. The area represents channel gain and mute. Only analog channels have this function, while network channels do not.
5. The area has functions for some specific models, but general models do not.

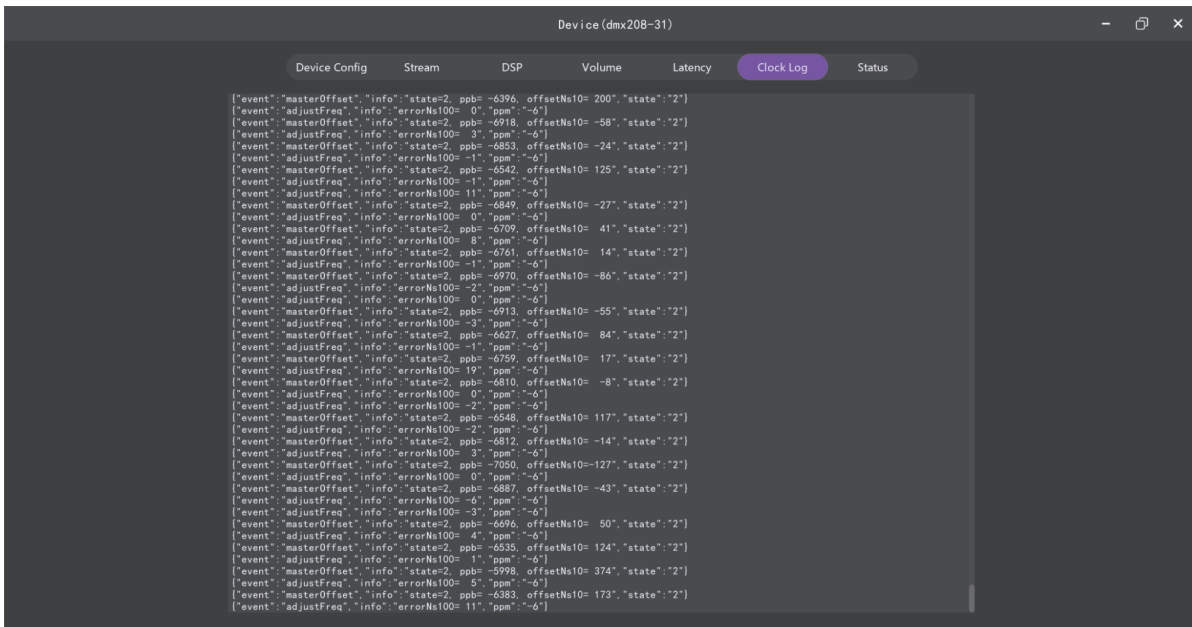
8.5 Clock log



1. The area represents the data stream. If data is received from multiple devices or the number of receiving channels exceeds the number of channels that a stream can accommodate, there will be multiple data streams at this time. Switching can view the statistical data of different streams.

2. The area represents statistical data. Before the machine is powered on and synchronized, the data will appear as a red bar. This is normal. At the beginning of power-on, the clock is not synchronized. After the clock is synchronized, the red color will no longer increase.
3. The area can clear the data and perform statistics again. Generally, if an occasional individual red bar appears, it is due to fluctuations in the network environment and does not affect sound transmission. If there is a continuous increase in red data at intervals, it indicates abnormal network or device transmission, and troubleshooting is required.

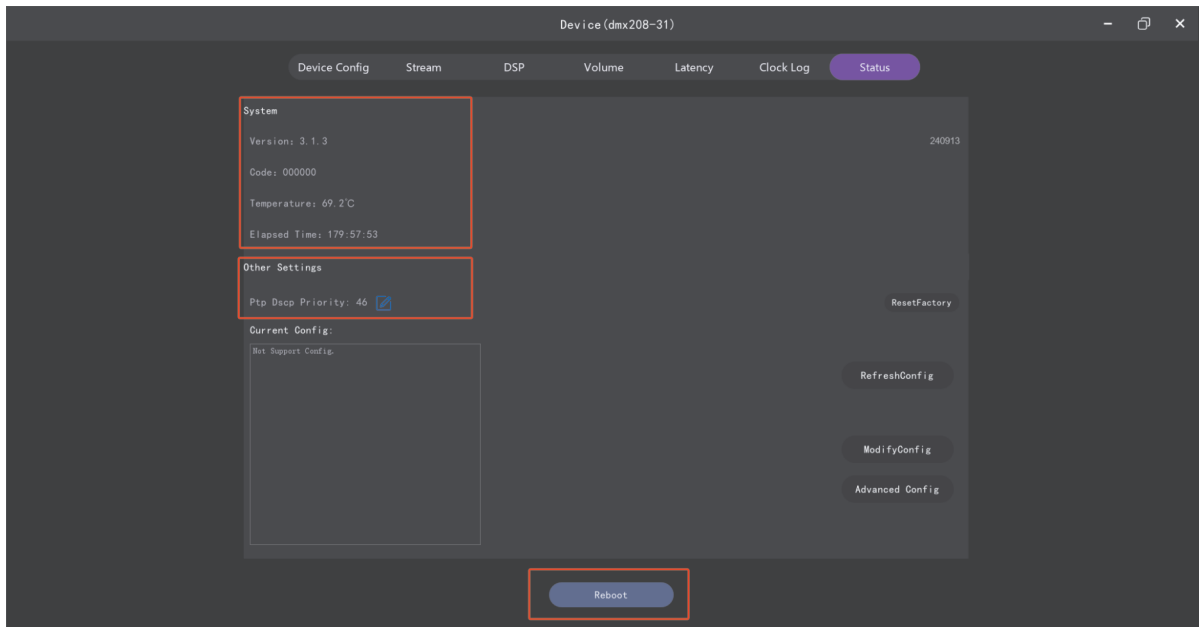
8.6 Clock log



```
Device (dmx208-31)
Device Config Stream DSP Volume Latency Clock Log Status
[{"event": "masterOffset", "info": "state=2, ppb=-6396, offsetNs10= 200", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 0", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6918, offsetNs10= -58", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 3", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6553, offsetNs10= -24", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -1", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6542, offsetNs10= 125", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -1", "ppm": "-6"}]
[{"event": "adjustFreq", "info": "errorNs100= 11", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6549, offsetNs10= -27", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 0", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6709, offsetNs10= 41", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 8", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6761, offsetNs10= 14", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -1", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6970, offsetNs10= -86", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -2", "ppm": "-6"}]
[{"event": "adjustFreq", "info": "errorNs100= 0", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6913, offsetNs10= -55", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -3", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6627, offsetNs10= 84", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -1", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6759, offsetNs10= 17", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 19", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6810, offsetNs10= -8", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 0", "ppm": "-6"}]
[{"event": "adjustFreq", "info": "errorNs100= -2", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6548, offsetNs10= 117", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -2", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6812, offsetNs10= -14", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 3", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-7050, offsetNs10= -127", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -1", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6877, offsetNs10= -43", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= -6", "ppm": "-6"}]
[{"event": "adjustFreq", "info": "errorNs100= -3", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6696, offsetNs10= 50", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 0", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6535, offsetNs10= 124", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 1", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-5998, offsetNs10= 374", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 5", "ppm": "-6"}]
[{"event": "masterOffset", "info": "state=2, ppb=-6383, offsetNs10= 173", "state": "2"}]
[{"event": "adjustFreq", "info": "errorNs100= 11", "ppm": "-6"}]
```

When the clock log page is opened, it will monitor the log information of the device. This information is used for device debugging and can be used to check whether the clock synchronization is normal.

8.7 Others



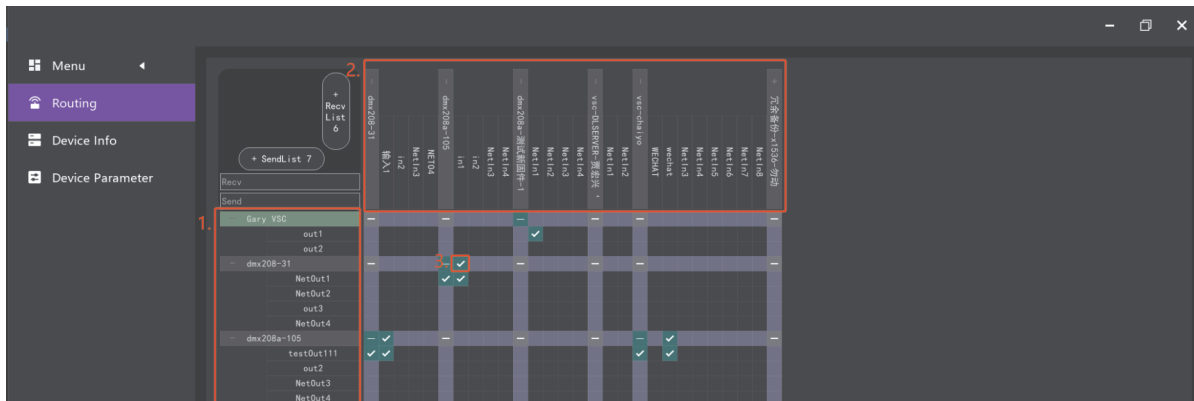
This page displays the version information, device code, temperature, running time, etc. of the device.

This page can set the Ptp DSCP priority.

This page can restart the device (some models support this function. For some devices, a full power-off restart is required).

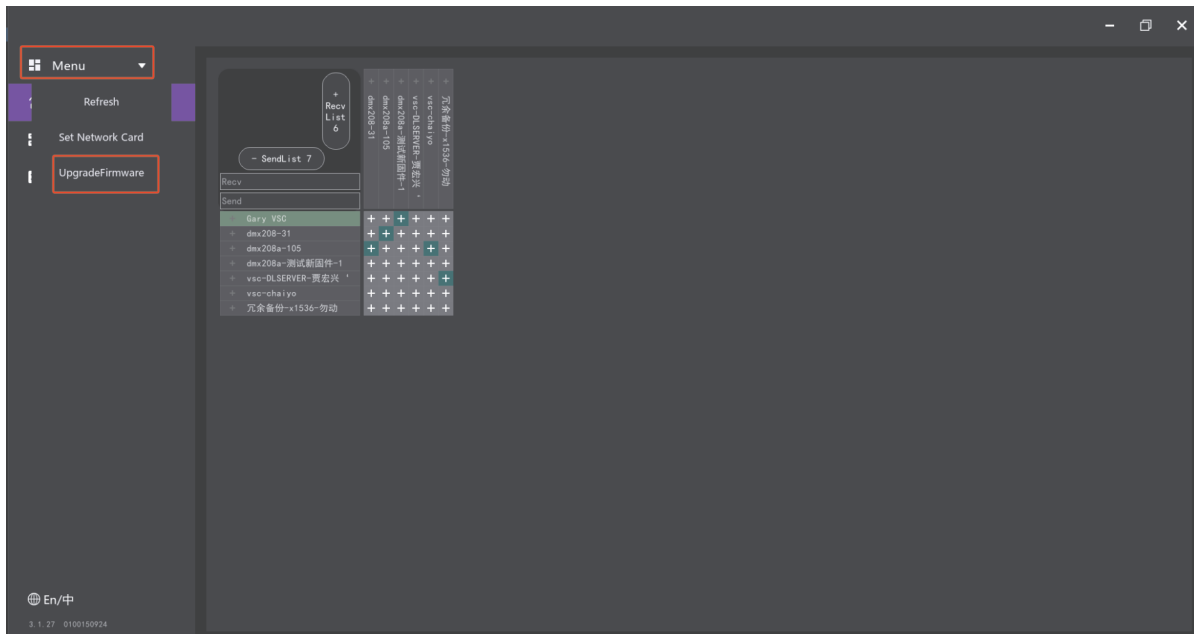
9. Device network routing settings

The device routing configuration is all configured for network channels. The channels in the following figure are all network input and output channels.

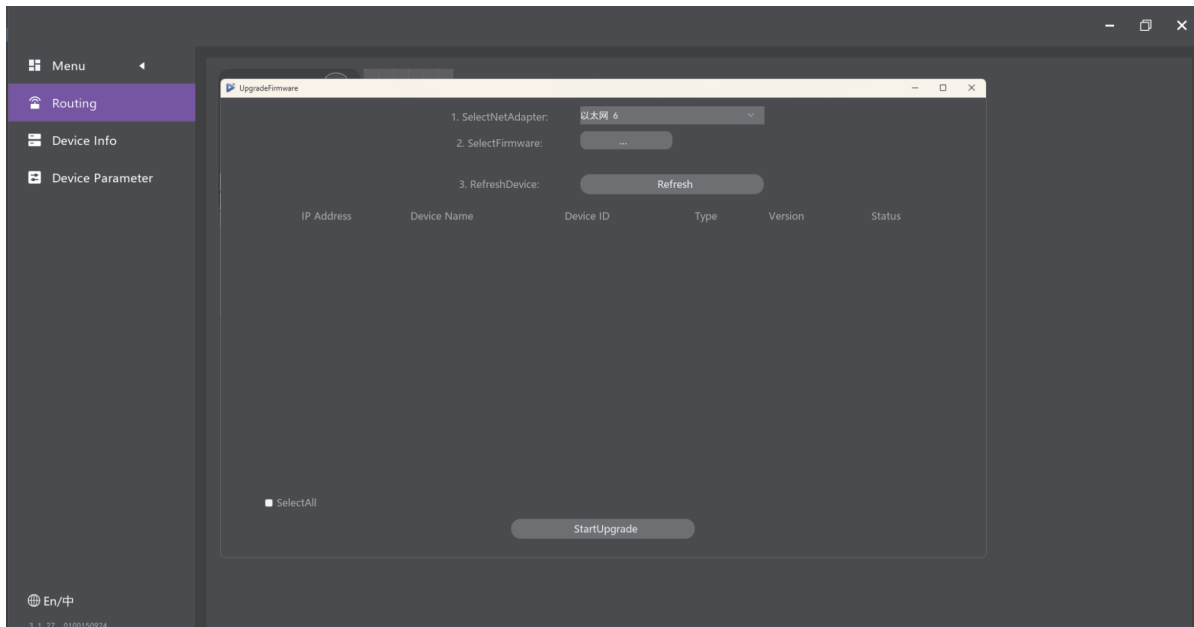


1. The area is the sending device list. After expanding it, you can see the network sending channels of the device.
2. The area is the receiving device list. After expanding it, you can see the network receiving channels of the device.
3. The area is whether to route. The selected state means that the network sending channel signal of the device corresponding to area 1 is sent to the network receiving channel of the device corresponding to area 2.

10. Device firmware upgrade



Click `Menu - Firmware Upgrade` , and the firmware upgrade interface will pop up:



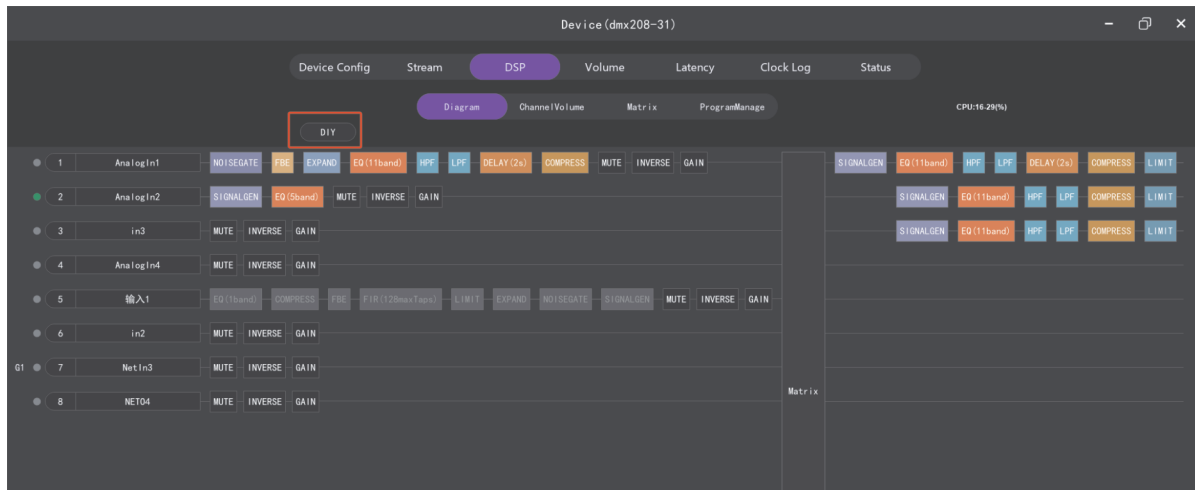
- `Select network card` : For a computer with only one network card, the default selection is fine. For a computer with multiple network cards, select the network card corresponding to the device local area network. If they are not in the same local area network, the device cannot be found.
- `Select firmware` : Select the firmware upgrade package used for machine upgrade. Do not unzip the upgrade package. Just select it directly. The upgrade package model must be corresponding, otherwise the upgrade will fail. The version of the device and the upgrade package must be corresponding, otherwise the upgrade will fail and the machine will be abnormal. For devices with version 3.x.x, the latest firmware version can be directly upgraded. For devices with version 2.x.x, a specific firmware upgrade package needs to be used to upgrade to 3.0.7 first, and then upgrade to the latest version. Click "Refresh Device", and all devices in the network will be displayed. If there are

any that do not appear, click "Refresh" twice more.

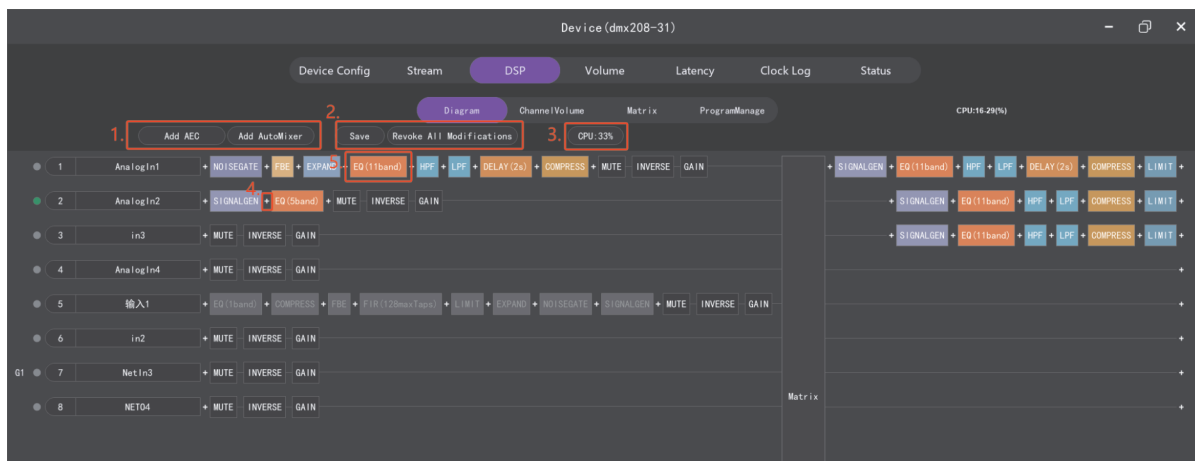
Select the corresponding device. Be sure to check the type and distinguish the device to be upgraded through IP address, device name, and device ID. Click "Start Upgrade" and observe the "Status" column until the upgrade is completed.

11. Advanced functions

11.1 Custom DSP function



For device types that support the DSP function, enter the "Audio Processing" page. Press and hold Ctrl+8 at the same time, and enter the control code 6689. A "Customize" button will be displayed in the upper left corner. Click "Customize" to enter the custom DSP page.



1. The area is for devices that support automatic mixing and echo cancellation. Functions can be added here.
 2. The area is for canceling customizations and saving custom functions.
 3. The area is for CPU usage assessment. The custom DSP function cannot exceed the maximum CPU occupancy rate.
 4. The area is the button for adding DSP functions. Clicking it can freely add function blocks.
- For equalization, high pass, and low pass, only one can be set for each channel. For the delay function, all channels share a delay upper limit.

5. In the area, for the corresponding DSP function, right-click to delete the function block.

Right-click on the channel ID to copy the function block of the entire channel and paste it to other channels.

12. Troubleshooting

12.1 When the software starts, it prompts that the xxx.dll library is missing.

The prompt for the lack of a dll library indicates that the computer system lacks Microsoft runtime libraries. Downloading [Microsoft VC++ runtime library collection](#) and installing it can solve the problem.

12.2 The software cannot detect the device

Check whether the network card selection is normal. Be sure to select the network card in the same local area network as the device. Check whether the firewall has blocked the software from connecting to the Internet.

Through the firmware upgrade program, check whether the device can be searched. If the firmware upgrade program can search but the software cannot detect it, first check the above two items.

Check whether the network port light of the device is flashing. If the network port light is not flashing, first check the hardware.

12.3 The software can detect the device but cannot control it

Check whether the network card selection is correct.

Check whether the IP address of the computer and the IP address of the device are in the same local area network. Check whether the computer has set a fixed IP, resulting in the computer and the device not being in the same local area network.

Check whether the network is normal. Check "Device Parameters" to see if the running time can be obtained. If the running time cannot be obtained, check the computer network and switch status (you can check for packet loss through the ping command). Sometimes, when a laptop uses a USB network port adapter, there may be problems with poor network connectivity.

12.4 Firmware upgrade fails

Check whether the network card selection is correct. Be sure to select the network card in the same local area network as the device. Check whether the firewall has blocked the software. Check whether the firmware type is consistent with the device type. Check whether the firmware version is correct.

If the firewall is enabled, the following programs need to be set to allow network access:

- **AES67Controller3.exe**
- **AES67Upgrade.exe**
- **gohttpserver.exe**