

Complete and Customizable Communications Solutions

ADECIA



"SOUND" ACCELERATES BUSINESS.

With new and ever-changing communication requirements, one thing remains the same: the need for high-quality, worry-free audio for collaboration. ADECIA helps organizations overcome challenges with implementation, configuration and room acoustics by providing all components required for a successful installation - designed to work seamlessly together to reduce set up time, costs and complexities.





Remote Conference Processor





Complete Coverage from Sound Input to Output

ADECIA helps organizations overcome challenges with implementation, configuration and room acoustics by providing all components required for a successful installation - designed to work seamlessly together to reduce set up time, costs and complexities.

Reduce Time, Spend and Complexities

Audio/control signals can be transmitted with one network cable using audio network standard, "Dante", solving troublesome wiring and installation problems.

Integration into All Audio Environments

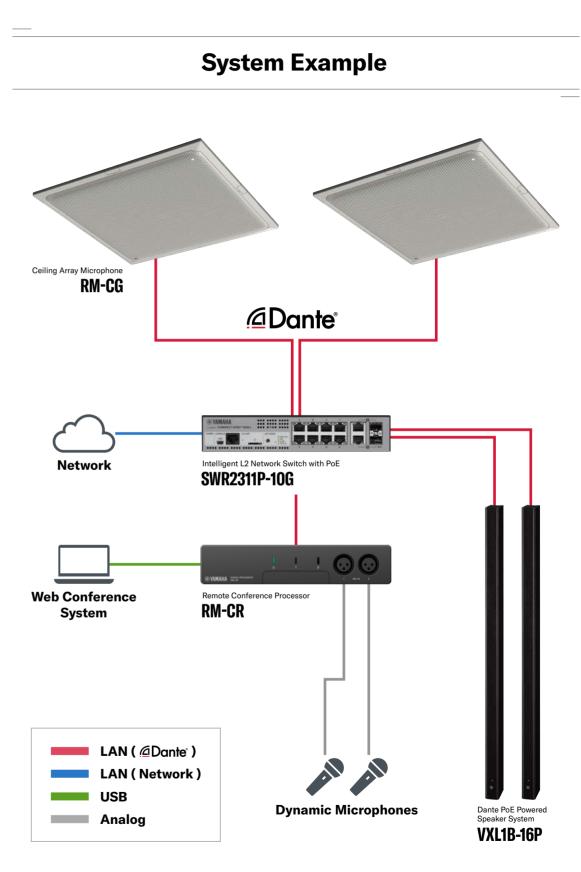
Equipped with various interfaces such as USB, Bluetooth[®], SIP, and audio input/output terminals, the system allows the flexibility to use for both web conference system and telephone line conferences. Supporting Windows, macOS, iOS and Android, as well as compliance with market leading UC applications, ADECIA gives you the flexibility to customize the solution to your specific needs.

Best Audio in Shortest Time

System automatically detects all components and configures them to be optimized for the room environment, taking into consideration location of speakers and microphones, reverberation characteristics of the room, and echo behavior. Setting up a room can be as easy as following the configurator through four steps of configuration.

Capture Every Word

Dynamic multi-beam voice tracking technology supports lively discussions with multiple participants, conveying a clear audio signal from each participant to remote parties. Its unique processing functionalities that help capture the human voice over noise help provide a smooth and stress-free remove conversation experience.





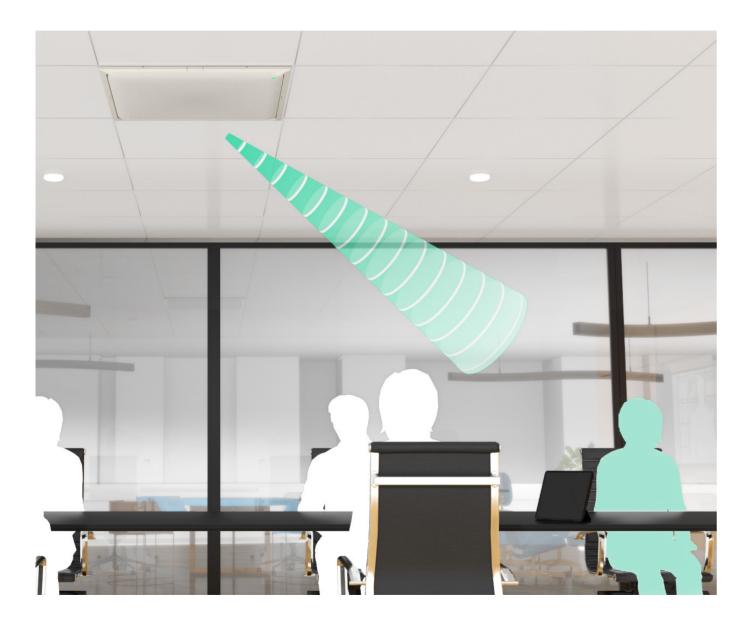
Sound Processing for High Quality Remote Conferencing Experience

Dynamic Voice Tracking

The beams constantly scan, detect and follow human voice. No fixed beams allow for flexible room layout and participant positioning.

Auto Tracking Beamformimg

The four dynamic beams simultaneously track multiple people, allowing for lively conversations. With a narrow directivity of each beam, the system accurately captures speech while suppressing background noise, enabling a highly intelligible voice to the far end.



Unique Yamaha Processing Technologies

Yamaha's unique HVAD technology distringuishes the human voice vs. noise to block unneeded sound from being sent to the far-end.

ADAPTIVE ECHO CANCELLER

While other echo canceller's measure the room and set levels once, Yamaha's adaptive echo cancellation is constantly scanning the environment for changes to deliver an uninterupted, echo-free audio stream.

NOISE REDUCTION

When constant noise like a projector fan or air conditioner system enters the communication space, Yamaha's Noise Reduction technology removes these components to clearly convey the presenter's audio to the far-end.

AUTOMATIC GAIN Control

This sound technology automatically adjusts voice levels to deliver a consistent, optimal volume level to the far-end, no matter if some voices are louder vs. wearker or closer vs. further away from the microphone. This automatic feature enables everyone to hear or be heard at an ideal level.

DEREVERBERATION

Reverberation from an acousticaly-poor environment can make it difficult to distinguish a clear voice signal. Dereverberation suppresses these reverb components to deliver a highly-intelligible output.

Various Mounting Methods to Best Fit Your Usage

Ceiling Array Microphone RM-CG Installation

Ceiling Mount



Wire Mount



Pole Mount



Remote Conference Processor RM-CR Installation

Rack Mount



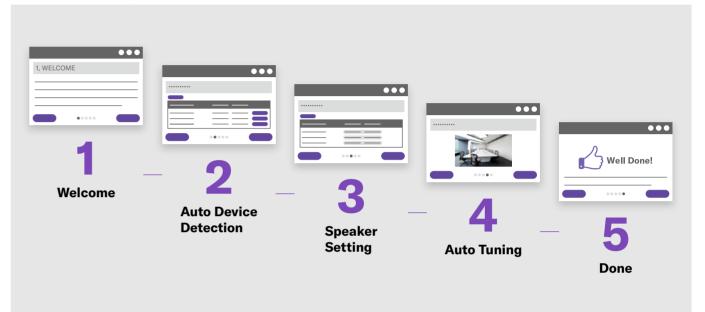
Table Mount



Setup Wizard

Follow the setup wizard on your web browser and with just a few clicks it routing and room environment.

RM-CR Quick Setup



DSP Setting List

Following items are able to customize.

DSP	Settings
	Off
	Gentle
Adaptive Echo Canceller	Medium (Default)
	Strong
Noise Reduction	Off
	Gentle
	Medium (Default)
	Strong
	Off
Dereverberation	Gentle
Dereverberation	Medium (Default)
	Strong

Follow the setup wizard on your web browser and with just a few clicks it will automatically adjust the audio according to your system's audio

DSP	Settings	
	Off	
AGC: Type	Soft (Default)	
	Hard	
ACO: Sarad	Slow (Default)	
AGC: Speed	Fast	
	1Ch Gating	
Auto Minima	2Ch Gainsharing (Default)	
Auto Mixing	4Ch Gainsharing	
	4Ch Mixing	
Burn Taulia Gaud	Fast (Default)	
Beam Tracking: Speed	Slow	
	Disable (Default)	
Beam Tracking: Area Limitation	Enable	

Flexible Application Support

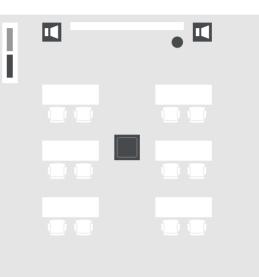


Boardroom

With a simple and sophisticated design, the unit blends into high-end spaces and enables a clean and clear table.



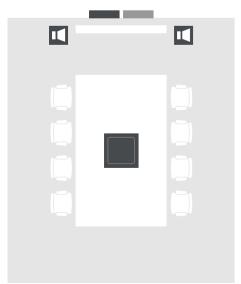
Flexible layouts are not an issue with dynamic beams. No fixed lobes can enable discussions, seminars and training areas without the need to reprogram the system.

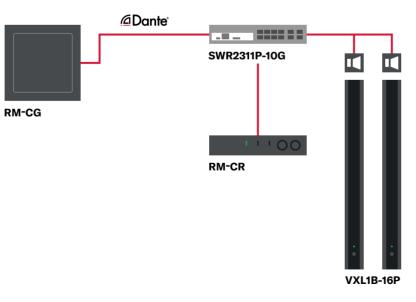


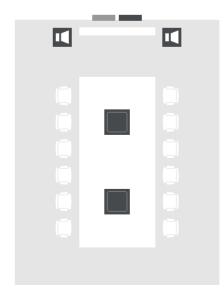
RM-CG

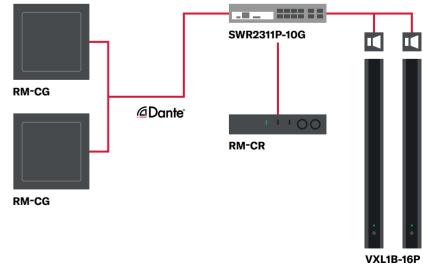
Medium-sized Conference Room

With the ability to connect two ceiling array microphones, you can expand the coverage in larger collaboration spaces to capture more people.

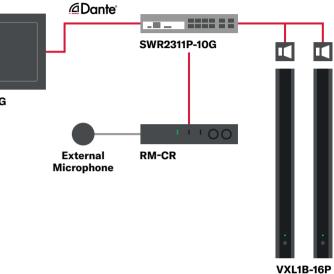








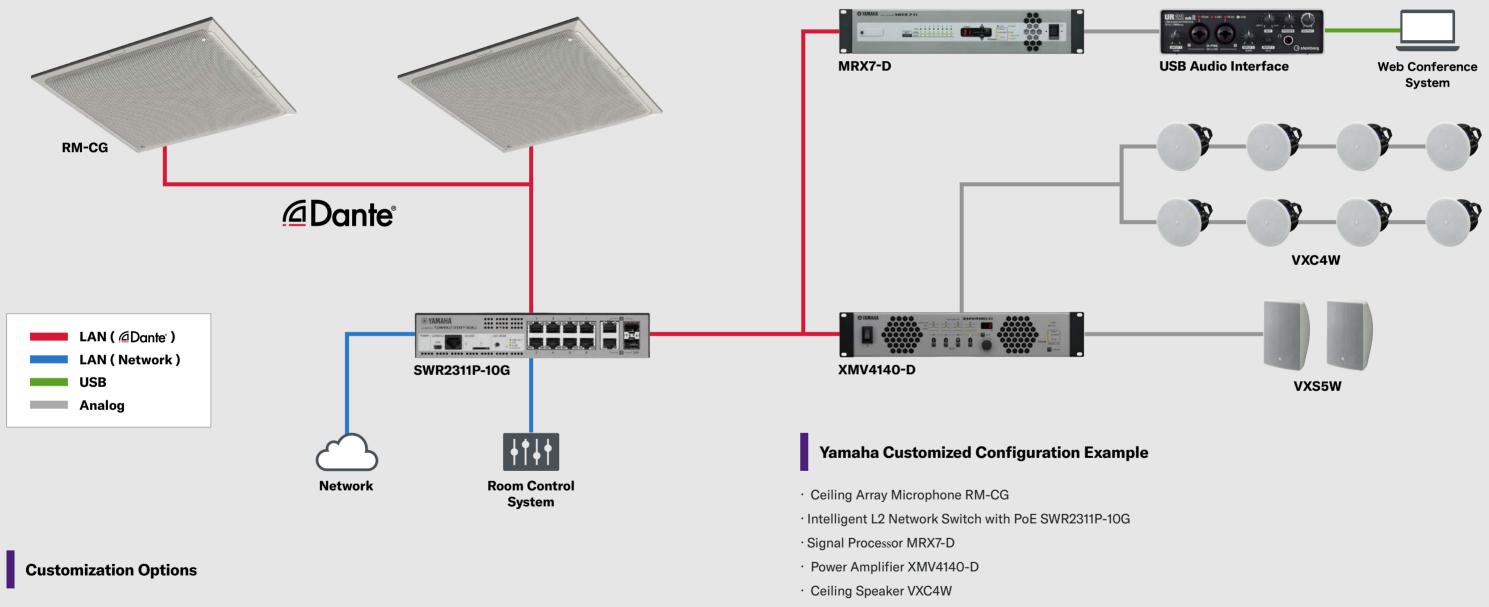




Customization

System Example

The new Yamaha components (Ceiling Array Microphone RM-CG and Remote Conference Processor RM-CR) can also be integrated with other Yamaha or third party components for required solutions with the open interface.



Speaker System VXS5W

· USB Audio Interface

- Expand the number of microphone connections
- · Expand peripheral devices/systems (amplifiers/speakers)

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Ceiling Array Microphone RM-CG



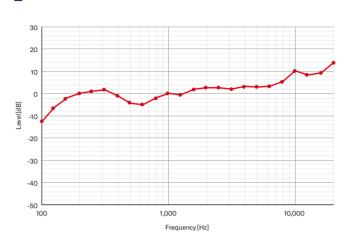
Ceiling Array Microphone

A ceiling array microphone equipped with unique audio signal processing technology that enables high-quality communication

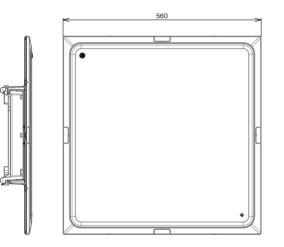
- $\cdot\,$ Standard network audio support for "Dante"
- $\cdot\,$ "Dynamic Voice Tracking" that scans for human voice
- \cdot Four "Auto Tracking Beams" track simultaneous speech allowing for lively conversations
- · Utilizes innovative Yamaha technologies like Adaptive Echo Canceller, Automatic Gain Control, Noise Reduction, and Dereverberation
- $\cdot\,$ Automatic detection and configuration with Remote Conference Processor RM-CR
- \cdot Utilizes one PoE+ network cable connection avoiding troublesome wiring and enabling audio control via Dante
- $\cdot\,$ Supports 3 types of mounting methods for use with a variety of ceiling conditions



Frequency Response

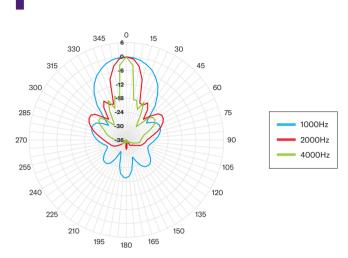


External View

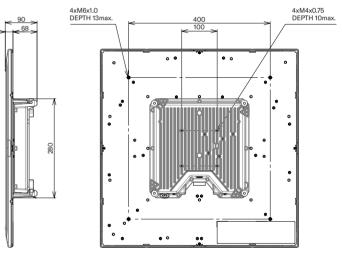


Main Specifications

Audio I/O	Dante 1in/2out
Sampling Rate	48kHz
Bit Depth	24bit
Frequency Response	160Hz - 16kHz (-10dB)
Power Requirements	PoE (IEEE 802.3af), DC 48V
Network	Dante Audio/Dante Control, Remote Control, Web UI, PoE
Latency	56ms (including DSP)
Sensitivity	-23.8dBFS/Pa



Directional Characteristics



Unit: mm

SN Ratio	94.8dB
Microphone Beam	4 (Auto Tracking)
Mount	Ceiling mount, Wire mount, Pole mount
Color	White/Black
Dimensions	W560mm x D560mm x H90mm (includes protrusion/microphone panel: 22mm)
Weight	5.6kg

Remote Conference Processor RM-CR

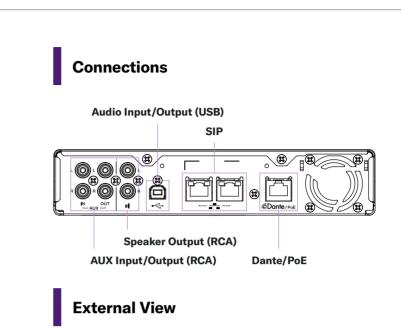


Remote Conference Processor

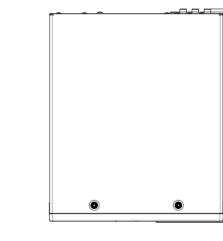
Signal processor at the heart of the RM series

- · Standard network audio support for "Dante"; all compatible devices can be connected with single network cable
- $\cdot\,$ Up to 2 RM-CG and 2 VXL1B-16P can be connected via PoE switch SWR2311P-10G
- $\cdot\,$ Automatically recognizes connected devices on the same network for audio routing
- · Equipped with the Auto Tuning functions that automatically measure and provide the optimum sound settings for the environment
- \cdot Equipped with the Auto Mixer functions that automatically mix the audio of all connected microphones
- $\cdot\,$ Equipped with various input/output connections such as USB, VoIP, analog, and Bluetooth







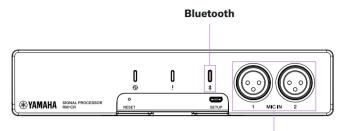




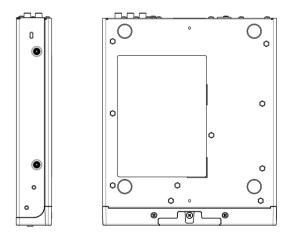
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Main Specifications

Audio I/O	Dante 16in/16out
	USB 2.0 (Type-B)
	RCA unbalanced (Line Level), Input: 2ch, Output: 2ch
	XLR balanced, Input 2ch
	RCA unbalanced (Line Level), Output: 2ch
	SIP
	Bluetooth



Microphone Input (XLR)



Unit: mm

Sampling Rate	48kHz
Bit Depth	24bit
Frequency Response	20Hz - 20kHz
Power Requirements	PoE+ (IEEE 802.3at, LLDP), DC 48V
Network	Dante Audio/Dante Control, Remote Control, Web UI, PoE+
Mount	Rack mount, Table mount (optional accessory)
Dimensions	W215mm x D260mm x H44mm
Weight	1.6kg

Specifications

RM-CG Specifications

	Dimensions (W x D x H)		560mm x 560mm x 90mm (includes protrusion/microphone panel: 22mm)		
	Weight		5.6kg		
	Power Requirements		PoE (IEEE 802.3af), DC 48V		
	Maximum Power Consumption		7.2W		
	Connector		RJ45		
	In Operation	Temperature	0°C - 40°C		
		Humidity	30% - 90% (no condensation)		
	Storage	Temperature	-20°C - 60°C		
		Humidity	20% - 90% (no condensation)		
	Indicator		Front indicator (Mute/Unmute & Status), Network port indicator		
	Mount		Ceiling mount, Wire mount, Pole mount		
	Maximum Device Number with RM-CR		2		
	Accessories * Including 1 spare		Installation manual, grill x 1, Grill fixing screw (M3 x 8mm) x5*, Screw cover x 4, Drilling pattern x 1, C ring x 1, C ring fixing screw x 2, U-shaped bracket x 1, U-shaped bracket fixing screw (M4 x 20mm) x 5*, Terminal cover x 1, Safety wire x 1, Cable tie x 1, Temporary retaining bracket x 2		
	Dust Protection		IP5X equivalent (when using junction cover)		
	Plenum Rating		UL2043 (with junction cover)		

Ether		Dante Audio/Dante Control, Remote Control, Web UI, PoE+		
Cable Requirements	;	Cat 5e or higher (STP recommended)		
Audio I/O	Dante	1in/2out		
Sampling Rate		48kHz		
Bit Depth		24bit		
Latency		56ms (including DSP)		
Frequency Response		160Hz – 16kHz (-10dB)		
Maximum Input Level of SPL (OdBFS)		117.8dB SPL		
Self Noise		-0.8dBA SPL		
SN Ratio (Ref. 94dB SPL at 1kHz)		94.8dB		
Sensitivity		-23.8dBFS/Pa		
Dynamic Range		118.6dB		
Signal Processing		Multi Beam Tracking, Adaptive Echo Canceller, Noise Reduction, Dereverberation, Auto Mixer, Automatic Gain Control, User Program EQ, Output Gain		

Options





RM-CR Specifications

_	-				-		
General	Dimensions (W x D x H)		W215mm × D260mm × H44mm			Bluetooth	Version 4.2
	Weight		1.6kg				Supported Profiles: HFP (1.6), A2DP, AVRCP
	Power Requirements		PoE+ (IEEE 802.3at, LLDP), DC 48V				Supported Codec: CVSD, SBC, mSBC
	Maximum Power Consumption		15.0W				Wireless Output: Class2
	In Operation Temperature		0°C - 40°C				Maximum Communication Distance: 10m (32.8ft)
		Humidity	30% - 90% (no condensation)				(no obstacles)
	Storage	Temperature	-20°C - 60°C				Radio Frequency (Operational Frequency):
		Humidity	20% - 90% (no condensation)				2,402MHz - 2,480MHz
	Indicator	· ·	Power indicator, Status indicator, Bluetooth indicator, Network port indicator x 3				Maximum Output Power (EIRP): 4.0dBm (2.5mW)
						SIP	Call Handling:
	Accessories		Installation manual, USB Cable (A-B) 5m (16.4ft), USB Cable (A-microB) 1m (3.3ft), Rubber foot x 4, Access panel	Audio	Audio I/O		Dial, Answer, Hold, Resume, Forwarding, Do not disturb, Redial, Call timer, Call ID,
							Voice mail notifications (switch configured)
							Call Bridging:
	Options		Rack Mount Kit RM-MTL, Table Mount Kit RM-MRK				Supports bridging SIP, USB, BT and AUX calls. Join,
Genera	Ether1		Dante Audio/Dante Control, Remote Control, Web UI, PoE+				Split, Hold, Resume, 5+1 lines: up to 2 SIP calls, 1 USB call, 1 Bluetooth call, 1 AUX, Plus the user
lera	Ether2/Ether3		Corporate Network, Remote Control, SIP, Web UI				Codecs: G.711, G.722HD, G.729ab, G.726
	USB		USB2.0, Network Class, Web UI, Fixed IP				DTMF Support:
		Dante	16in/16out				RTP event, SIP in-band, SIP info package
		USB	USB2.0, Audio Class 1.0, Input: 2ch, Output: 2ch				Security: SRTP support (RFC 1889),
A			@48kHz			IETF SIP support (RFC 3261 and companion RFCs)	
Audio	Audio I/O	Microphone Input	XLR balanced, Input: 2ch		Sampling Ra	ate	48kHz
		AUX	RCA unbalanced (Line Level), Input: 2ch, Output: 2ch		Bit Depth		24bit
		Speaker Out	RCA unbalanced (Line Level), Output: 2ch		Latency		7ms (Dante In to USB Out, includes signal processing)
					Frequency F	Response	20Hz - 20kHz

RM-CR Analog Input Characteristics

Input Terminal	Input	Connector		
input terminal	Nominal	Max. before Clip	Connector	
MIC IN 1,2	-46dBu	-26dBu	XLR-3-31 (Balanced)	
AUX IN L,R	-14dBV	+6dBV	RCA Pin	

RM-CR Analog Output Characteristics

	Source Impedance	For Use with Nominal	Output Leve	A	
Output Terminal			Nominal	Max. before Clip	Connector
AUX OUT L,R	1kΩ	10kΩ	-14dBV	+6dBV	RCA Pin
SP OUT L,R	1kΩ	10kΩ	-14dBV	+6dBV	RCA Pin



An intelligent L2 switch that has the optimal functions for Dante and can supply PoE to devices on the network.



Mount kit for mounting the Remote Conference Processor "RM-CR" in a rack.

Mount kit for mounting the Remote Conference Processor "RM-CR" on the table

Intelligent L2 Network Switch with PoE





A Dante powered speaker by PoE power supply.



YAMAHA CORPORATION

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