





Innovative Solutions for Live Sound

The term "live sound" covers a lot of ground, and no two applications are exactly alike. You might only need to amplify a single microphone for a business meeting, or mix many sources and deliver several kilowatts of power at an outdoor concert. For both these scenarios, and everything in between, Yamaha has the gear you need to get the job done with maximum quality, efficiency and ease.

When choosing equipment for your application, you'll need to consider the following points:

1. Scale

How big is your audience? How big is the venue or area you need to cover? For larger setups, you will need to have enough speakers and power to cover the area, smaller venues with space limitations require equipment that can provide the required functionality and performance without getting in the way.

2 Sources

Do you only need one or two microphones for speech or vocals? Will you be supporting live music using numerous microphones and line-level inputs? Do you need to handle recorded sound effects or background music? All these factors will determine the size and type of mixer you'll need, as well as monitoring and output equipment (equalizers, power amplifiers, and speakers).

3 Indoors or Outdoors

The requirements for indoor and outdoor sound can be quite different. While power, coverage and sound quality are relatively easy to handle indoors, these factors often require extra care outdoors where there are no room reflections to reinforce the sound and your audience may be spread out over a large area. You also have nature to deal with when working outdoors, so you need a setup that can withstand less-than-favorable weather conditions.

Explanation of icons 96kHz compatible **♦** CUBASE AI₄ One-switch Compressor equipped Cubase AI4 bundled YAMAHA Speaker Processing MIDI format compatible **Ys**Processing Frequency Response Correction System Virtual Circuitry VCM Effects pre-installed YAMAHA EEEngine equippded Feedback Channel Locating System model Class-D Amp equippded MY16 series compatible Multi-band MAXIMIZER equipped Max non-clip level of analog output +24dB **Automatic Feedback Suppressor** =18dR SPX EFFECT equipped Max non-clip level of analog output +18dB



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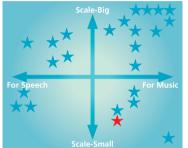
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Band Rehearsal 1

If you rehearse with real drums and other instruments at realistic levels, you'll need some vocal amplification for well-balanced, productive rehearsals. Even if the band is relatively quiet, vocalists need to practice with microphones and amplified sound if amplification will be used on stage. Compression can also help to make the vocal sound stand out. This simple system featuring the EMX312SC powered mixer, a pair of BR12 speakers, an SM10V (CM10V) for monitoring, and a couple of good microphones is ideal.

System Chart



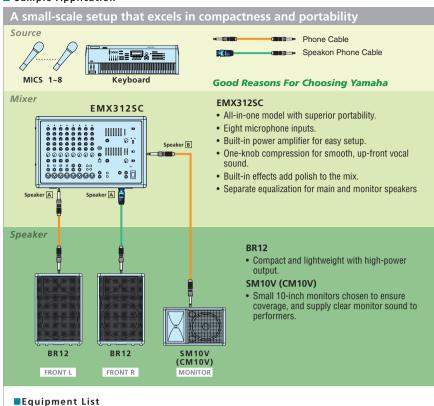
~ 8 channels Line: Four stereo

Scale: Rehearsal studio. approx. 30 square

Audience: 5 ~ 8 people

meters

■ Sample Application



■ Sample Application

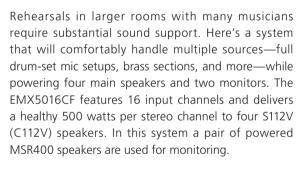
FMX312SC

SM10V (CM10V)

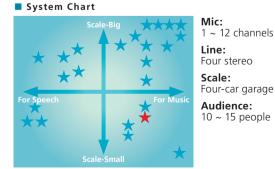
RR12

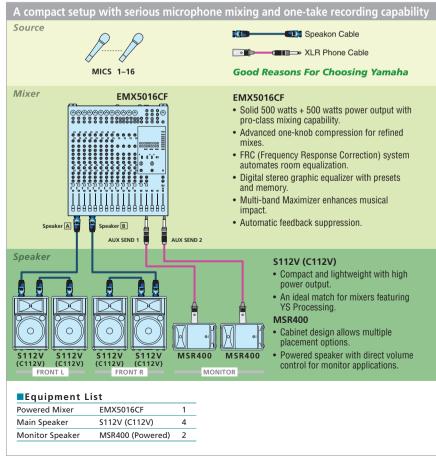
Powered Mixer

Monitor Speaker



Band Rehearsal 2

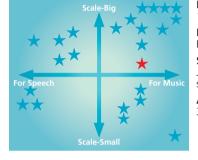




Piano Bar

Here's a system that's ideal for small venues hosting musical entertainment ranging from solo pianists to iazz trios with a vocalist. The volume may be low, but vocalists need solid monitor sound to deliver their best performance. Compression can be a real advantage, too. In this example, an EMX512SC powered mixer delivers the house sound via a pair of S112V (C112V) speakers, while a single SM12V (CM12V) provides quality monitor sound.

System Chart



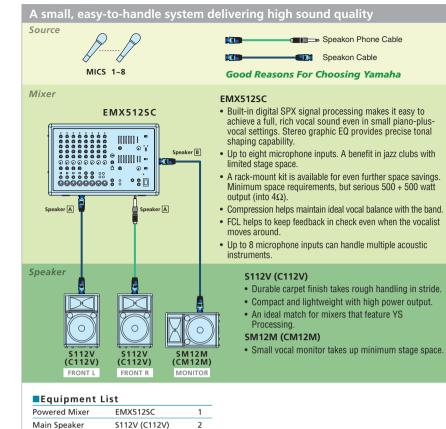
~ 8 channels Line: Four stereo

Scale: Jazz club with small stage

Audience: 30 ~ 50 people

~ 4 channels

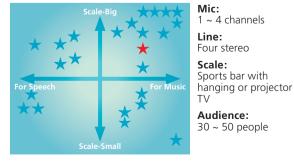
■ Sample Application



Sports Bar

There's no denying that while much of the impact of a sports event is visual, sound plays a huge role as well. Sports bars that show live or recorded sports on 50" or larger screens have the visual aspect pretty much covered, now here's a sound system that can really bring the action to life. And when it's not sports time, it serves as an outstanding BGM system as well. An MG82CX handles the required inputs with room to spare, and a pair of MSR250 powered speakers deliver big, dynamic sound.

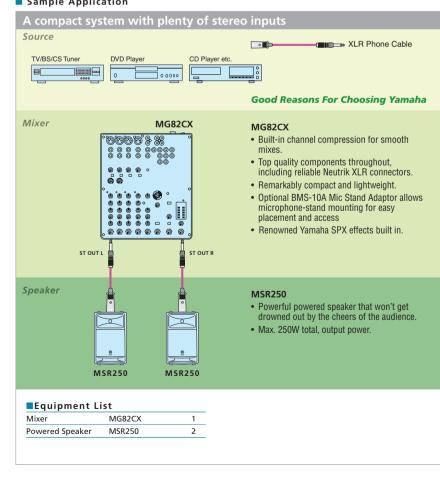
System Chart



Sample Application

SM12M (CM12M)

Monitor Speaker



Small Stage

This small-but-serious system will cover small to medium-sized live venues with thoroughly professional quality and versatility. The MG206C mixer offers plenty of input capacity. FOH sound is delivered by P5000S and P7000S power amplifiers driving S115V (C115V) full-range speakers and SW118V (CW118V) subwoofers, respectively. For monitoring, a pair of SM15Vs (CM15Vs) is powered by a P5000S amplifier with a O2031B equalizer for feedback control. An SPX2000 professional multi-effect processor could be added for additional vocal processing. The system also features a STAGEPAS 150M portable PA system for keyboard monitoring. With the main PA mixer's AUX SEND 3 connected to Input 5 (mono) of the STAGEPAS mixer, the STAGEPAS speaker is able to operate as a monitor. In addition, input from keyboards and synths can be mixed on the STAGEPAS mixer and then sent to the main PA.

System Chart

Mic: 1 ~ 16 channels

Line: 4 stereo

Scale:

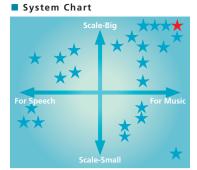
Medium-size live space with stage

Audience:

150 ~ 200 people

Large Outdoor Stage

Outdoor sound poses some unique problems, and power and speaker coverage are of prime importance. The system shown here delivers three kilowatts to FOH via three P5000S power amplifiers and a combination of S215V (C215V) full-range speakers and SW218V (CW218V) subwoofers. An additional 2.8 kilowatts is allotted for monitoring via four P3500S amps and SM15V (CM15V) monitor speakers. The MG32/14FX console handles the wide range of inputs and signal processing often required at outdoor music events.



Mic: 1 ~ 24 channels

Line:

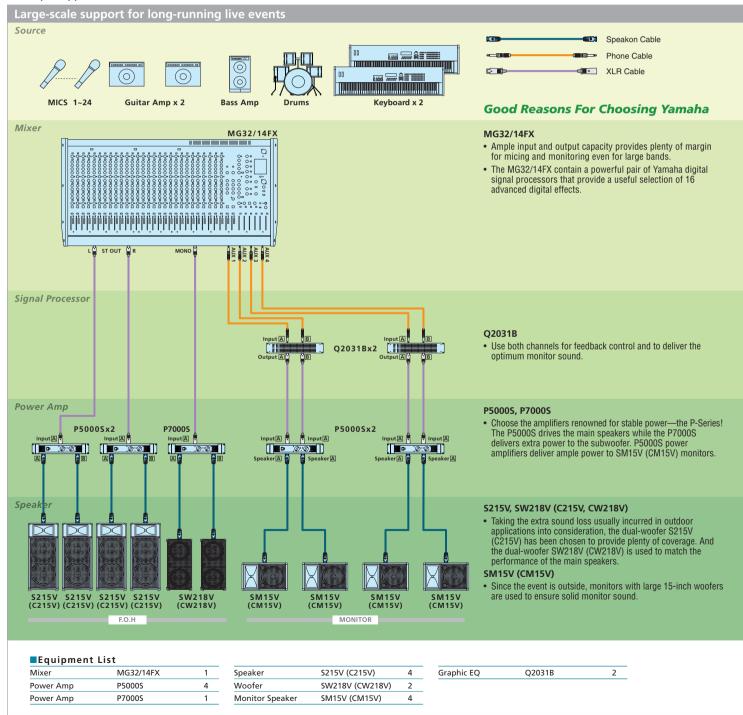
Four stereo Scale:

Large outdoor concert stage

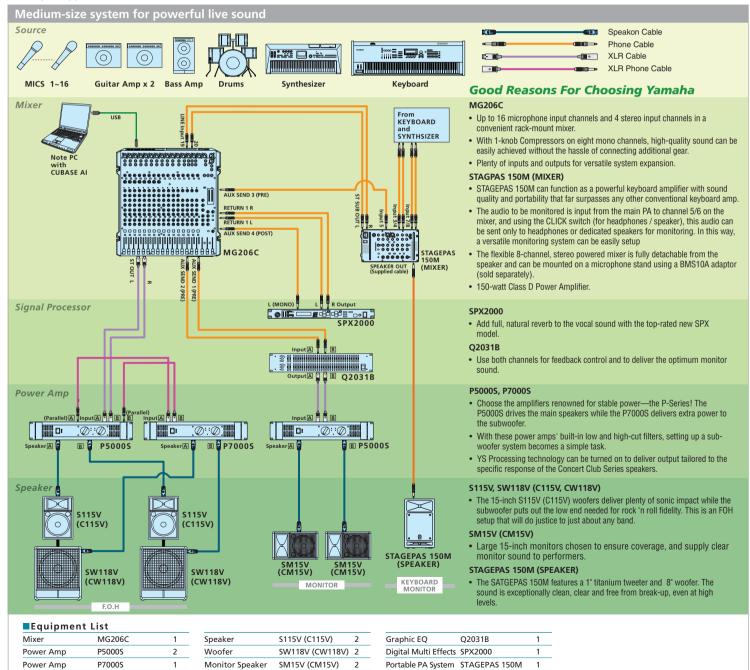
Audience:

500 ~ 1,000 people

■ Sample Application

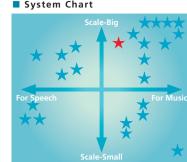


Sample Application



Compact Digital Solution for 2-Band Events

Large music events featuring two or more bands can involve a large amount of sound reinforcement gear. Here's a compact digital system that can handle up to 80 input channels in all while allowing easy switching between completely different band setups. The pair of 01V96V2 digital consoles used provide all the effects and dynamics processing you'll need built-in, so you won't even need outboard processor racks.



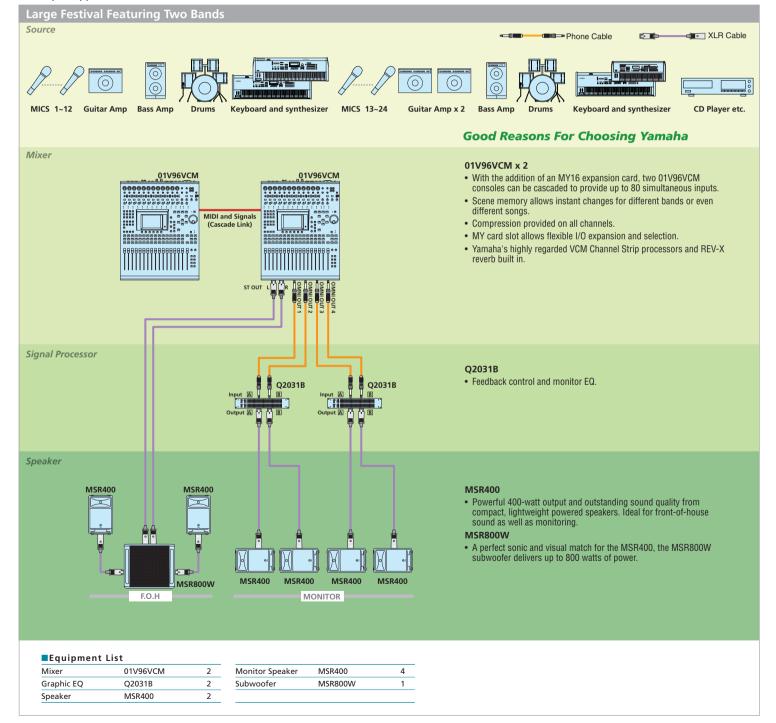
Mic and Line: Max. 80 channels

Scale:

Large events or festivals featuring two or more bands. Plenty of capacity is also provided for announcements and

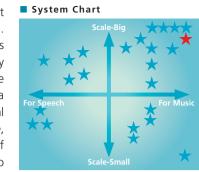
Audience: 300 ~ 500 people





Full-Fledged Live SR and Recording

This powerful sound reinforcement (SR) solution for the live performance environment uses a single IM8-24 mixing console to handle both FOH and monitor functions. Offering 24 mono and 4 stereo inputs, extensive output connectivity, best-in-class sound quality, and a highly-functional, ergonomic panel layout, the IM8-24 can easily satisfy even the most demanding of professionals. And using Steinberg's Cubase Al, which is bundled with the mixer, you can easily record live performances on a PC. Despite its compact design, the IM8-24 is ideal for a wide range of professional applications, and when combined with a full set of powered speakers, for example, provides a highly-portable solution for the live-performance environment. In houses of worship and other small venues, this type of setup is ideal when equipment needs to be moved around.



1 ~ 24 channels Line:

4 stereo

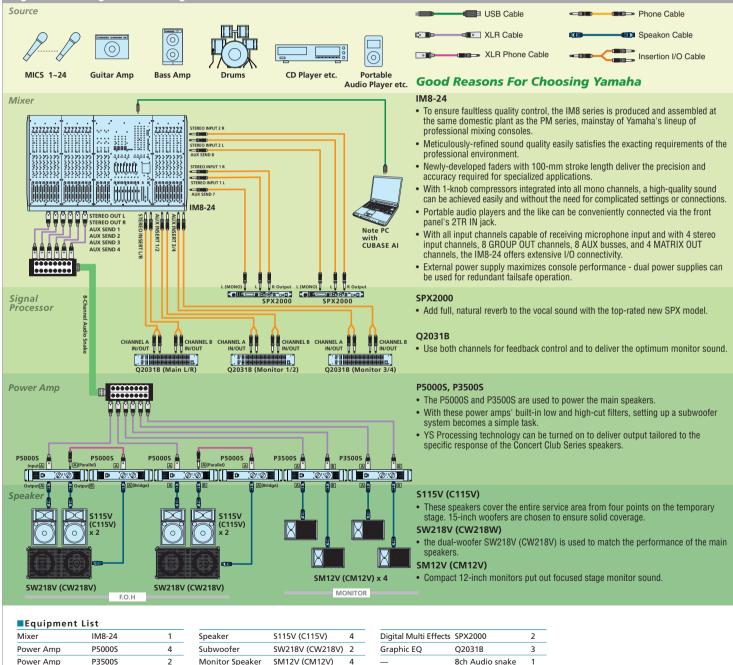
Scale:

Central outdoor stage

Audience:

100 ~ 500 people





Festival

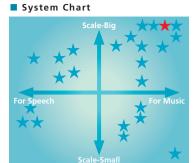
This system is designed for optimum sound coverage from a central stage surrounded by the audience. Special attention must be paid to monitoring in this type of situation, so each performer is provided with his or her own monitor speaker. The MG166CX console provides ample mixing and signal-processing capability, while dual P5000S power amplifiers driving four S115V (C115V) speakers on stands effectively cover the required area. The built-in SPX digital effects make it convenient and easy to add high-quality reverb and delay for band performances.

System Chart

Mic: 1 ~ 10 channels Line: 2 ~ 4 stereo Scale: Central outdoor stage Audience: 100 ~ 300 people

County Fair

County fair type events combine announcements with competitions and music, requiring substantial sound-reinforcement capability. And since the area to be serviced can be guite large, ample power and coverage are essential. In this system, FOH power is supplied by P5000S amplifiers feeding dual-driver speaker systems for high efficiency. A solid 2.8 kilowatts is provided for monitoring via four SM12V (CM12V) monitor speakers. The MG24/14FX console can easily handle the wide range of sources this type of event can entail.



Mic: 1 ~ 16 channels

Line: Four stereo

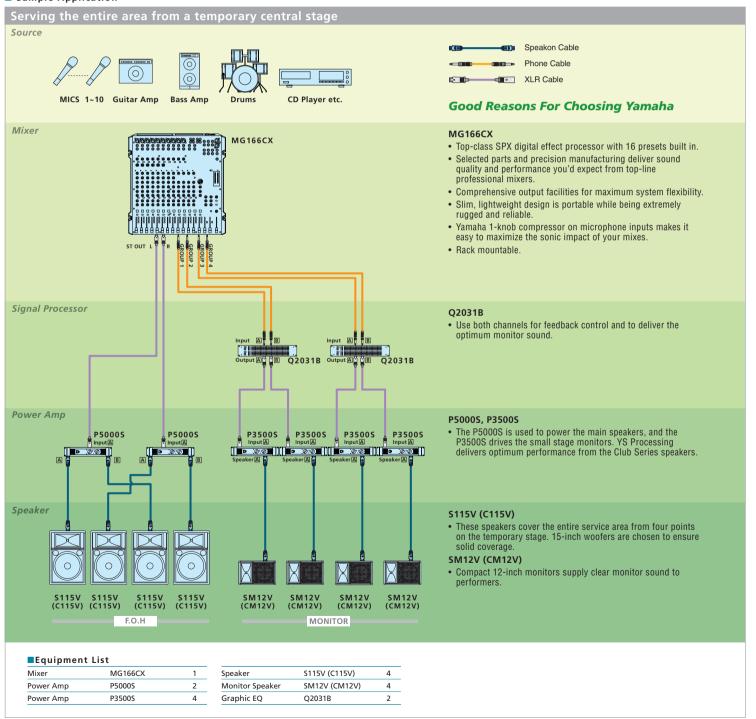
Scale:

Temporary outdoor stage. Contests being held in front of the stage with full-time announcements and commentator. Live music during breaks in the activities.

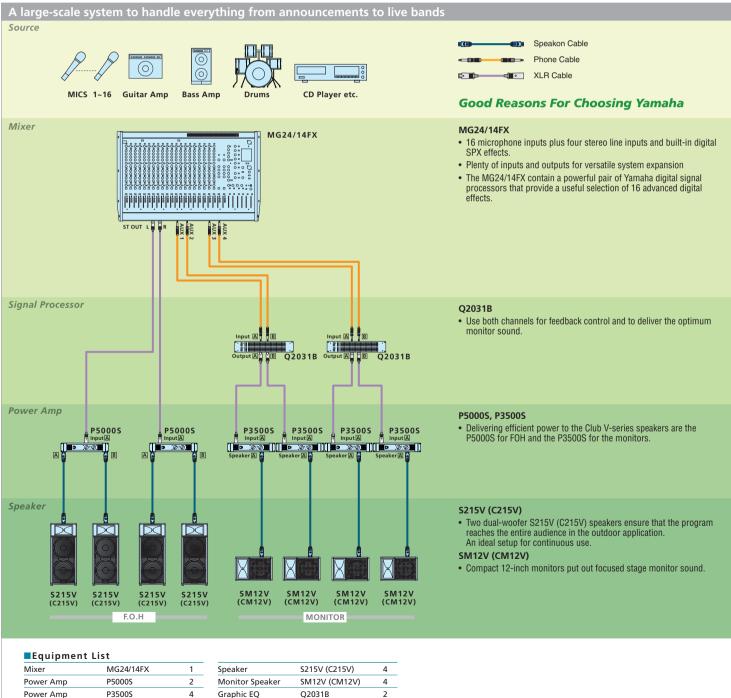
Audience:

500 ~ 1.000 people

■ Sample Application



■ Sample Application

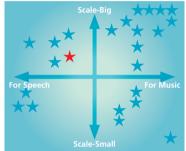


Church Church

Small Church

Designed primarily for background music and speech, this compact, easy-to-handle system is ideal for small houses of worship. It's portable and easily re-configurable, so it can easily be adapted to a variety of programs, indoors or out. An MG124C mixer offers advanced mixing potential and versatility in a space-saving package, and a P3500S power amplifier delivers more than enough ultra-clean power to make the most of the S112V (C112V) house speakers.

System Chart



Line: Four stereo

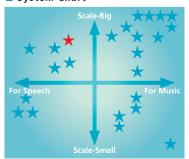
Scale: Small town church

Audience: Approximately 50 people

Mid-sized Church

A system like this is an excellent starting point for worship programs that involve live music as well as the spoken word. An MG166C console provides outstanding sonic quality and mixing flexibility. And an output chain consisting of P5000S power amps driving C115V (S115V) full-range speakers and CW118V (SW118V) subwoofers will get the musical message across without compromise. Powered monitor speakers such as the MSR100 can be added if monitoring is required.

System Chart

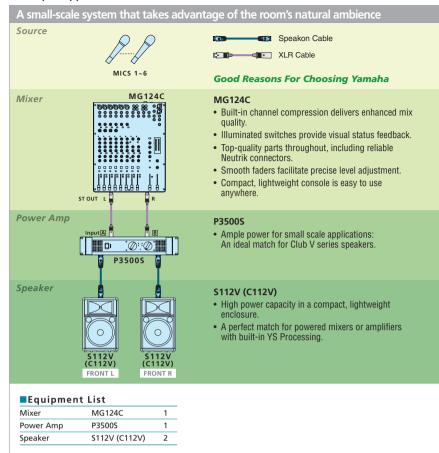


Mic: ~ 10 channels Line: Four stereo

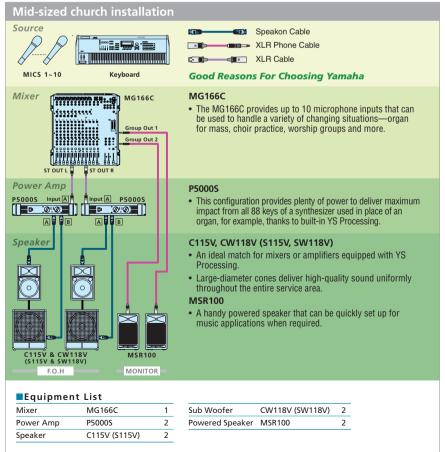
Scale: Town church

Audience: Approximately 100 people

■ Sample Application



■ Sample Application



House of Worship Installation

In addition to handling church activities, this installation is designed to deliver System Chart the best possible sound for music and theatrical presentations as well.

A 40-channel IM8-40 console ensures that plenty of I/O capacity is available for any type of event the church may be required to host. Four stage monitors are powered by a pair of P3500S amplifiers, while left and right mains with subwoofers plus flown center and side speakers deliver uniformly high quality sound throughout the house. The mono-fed center speaker is particularly important in achieving maximum clarity and intelligibility for sermons and speeches, while the subwoofers deliver maximum musical impact. The console's matrix is used to provide sound for the fover, dressing rooms, and other remote locations via a P2500S amplifier with output transformers and appropriate ceiling-mounted speakers. A computer running the supplied Cubase AI4 digital audio workstation software can be connected to the console's USB I/O port for recording and playback as required.

1 ~ 40 channels

Line: 4 stereo

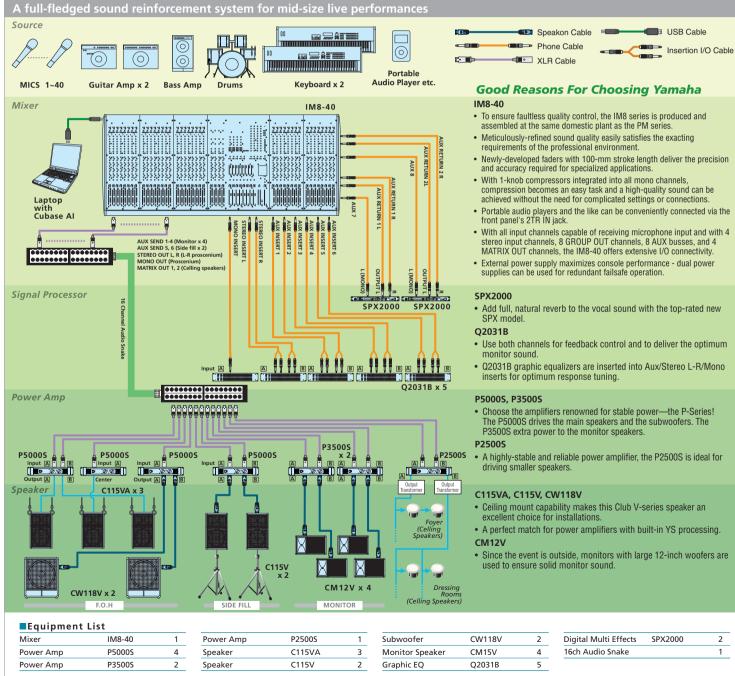
Scale:

Installation for Church (Large), House of Worship

Audience:

500 ~ 1.000 people

■ Sample Application



Portable audio players and the like can be conveniently connected via the

PA Basics

The term "PA" traditionally stands for "Public Address," and "PA System" refers to an electronic system for amplifying the voice of a speaker addressing a large crowd or people distributed throughout a large building. These days the term is often applied to the main amplification system at events and concerts as well, although such systems are sometimes called "sound reinforcement systems," or simply "sound systems."

The main functions of a PA or sound reinforcement system are summarized below:

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(0) (0) (0)

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O.

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Input

- The sound from acoustic sound sources such as speech or singing, drums, pianos, acoustic guitars, and electric guitar amplifiers is initially picked up using microphones. Microphones sources are input to the sound system's mixer via its microphone or "mic" inputs.
- The output from electronic sources such as CD players, synthesizers, and other electronic instruments are connected directly to the mixer via its line inputs.

Amplification

- The balanced program created by the mixer is sent to a power amplifier which boosts the program signal so that it can drive the system's speakers
- Although separate power amplifiers can be used in just about any application, mixers with built-in power amplifiers and powered speakers are often a better choice especially when system simplicity and portability are required.

Mixing and Processing

- As its name implies, a "mixer" is a device which is used to mix and balance
- speakers that actually deliver audible sound to the audience. The type and number of speakers required will depend on the size and configuration of the venue as well as the type of event
- In live music applications monitor speakers for the performers will also be necessary in most cases

- the signals from the various microphone and line sources to create the final program the audience will hear. The mixer usually also includes equalization
- Some mixers also include effects such as reverb or delay to allow more exten-control over the output sound.

Delivery

- •The output from the power amplifier (whether separate or built-in) drives the

Mixer Essentials

described below, using the simple MG102C mixer as an example.

OYAMAHA

SOLE MG102C

0

PHANTOM

10

These are the mixer's microphone inputs, featuring top-quality Neutrik XLR type connectors. The MG102C allows connection of up to four microphones.

Electronic keyboards and other electronic instruments, effect devices, electric-acoustic guitars and other line-level devices can be connected to the LINE inputs. Both mono and stereo line inputs are provided, so synthesizers with stereo outputs, for example, can be connected to the stereo line inputs.

Primarily used to adjust the input gain (sensitivity) of the microphone inputs. Microphone signals are much lower in level than line signals, and require extra amplification.

HIGH/MID/LOW (Equalizer) /HIGH/LOW

be used to refine the sound of each individual channel. Independent controls for the high, midrange, and low frequencies allow precise

In addition to the main stereo program outputs, the mixer has "auxiliary" outputs that can be used to send the channel signals to external signal processors or monitor amps and speakers. Independent AUX controls are provided for each channel.

These controls position the corresponding channel's signal from left to right in the stereo sound field. "Pan" is short for "panorama" or "panoramic" control.

These are the main level controls for each of the mixer's channels. The MG102C uses rotary level controls. On larger mixers these are often linear "faders".

The mixer is the core of any sound system. The basic elements of a mixer are

These are line-level inputs provided specifically for connecting to standard CD players or tape decks and thus feature pin-jacks that are directly compatible with most equipment of this type.

These inputs are primarily used to receive the stereo signal from a CD player or other source for background music

The REC OUT connectors can be connected to a stereo tape deck or other recording device to allow convenient recording of the mixer's stereo program.

STEREO OUT

These are the mixer's main stereo program outputs.

Sends the assigned signals to an external signal processing device

Receives the processed signal returned from an external signal processing device to be mixed back in with the program signal.

Allows accurate visual monitoring of the mixer's stereo signal levels.

Adjusts the level of the mixer's STEREO outputs, and thus the volume of the sound heard from the system's speaker

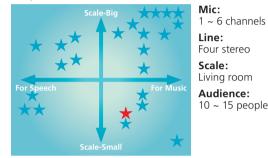
MONITOR/PHONES

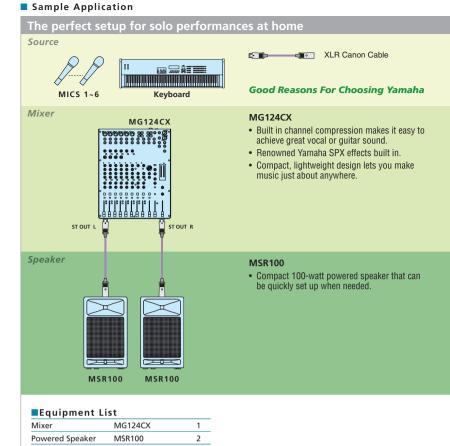
This control adjusts the signal level monitored via both headphones and the mixer's MONITOR outputs.

Live Music at Home

Here's a system that will be appreciated by individuals or families who enjoy live music at home. Although an electronic keyboard and just a couple of microphones are shown, this setup can handle guite a bit more. The MG124CX console will handle up to 12 inputs with top-quality effect processing built-in, and a pair of MSR100 powered speakers put out sound adequate for all but the most spacious living rooms.

System Chart

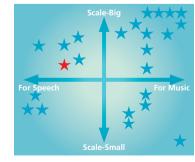




Wedding Ceremony

Two elements essential to any wedding (in addition to the bride and groom) are speech and music. This system is ideal when the joyous mood of the moment needs to be conveyed to a fairly large group of guests. In most cases, just a couple of microphones are sufficient, and an electronic keyboard such as one of Yamaha's superlative MOTIF series can function as pipe organ, piano... just about any instrumentation you need. An EMX312SC driving a pair of S115V (C115V) speakers for FOH, and an SM12V (CM12V) for monitoring should cover the sound delivery requirements.

System Chart



Mic: 1 ~ 8 channels

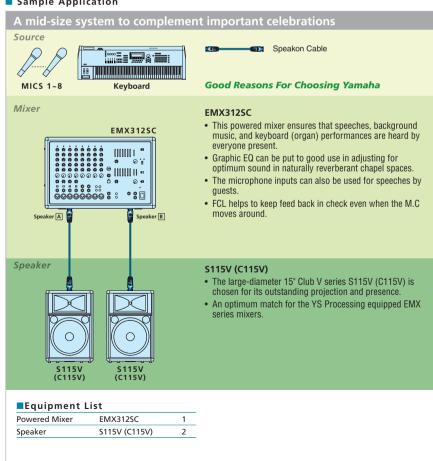
~ 6 channels

Line: Four stereo

Wedding chapel with central aisle Audience:

20 ~ 30 people

■ Sample Application



Although a PA system's mixer, amplifiers and speakers are indispensable, so are the cables and connectors that get the system's signals from one place to another. In fact, choosing the right cables for the various system connections is of the utmost importance. Here's a brief overview of the main types of cables and connectors you are likely to encounter

1. Cable Types

Microphone/Line Cables-

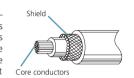
These are the cables used to connect microphones and electronic instruments to the mixer's inputs, and connect the line-level signal from the mixer's outputs to the system's power amplifier(s). These types of cables are shielded to minimize noise pickup.

* Never use this type of cable to connect the outputs of a power amplifier to speakers. Microphone/line cables are designed to handle low-level signals, and heat up and even pose a fire hazard if used for power connections.



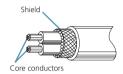
Unbalanced Cables

Unbalanced cables have just two conductors generally a core and a shield. Electric guitar cables and most types of electronic instrument cables have this type of construction. Cables of this type are convenient to use and offer good noise rejection, but are inferior to balanced cable when it Core conductors comes to noise-rejection with very low-level signals or long cable runs.



Balanced Cables

Balanced cables have three conductors—two core conductors surrounded by a shield. This type of connection was invented to overcome the noise pickup problems commonly encountered with unbalanced connections, and is the type used in most professional sound equipment. This type of cable can only be used with balanced connectors.



Speaker Cables

Speaker cable is specifically designed to transfer the amplified signal from the power amplifier's outputs to the speaker's inputs. Speaker cables feature heavy-duty conductors designed to handle the high-power signal delivered by the power amplifier, and because of the much higher signal levels no shield is





Speaker cable cannot be used in place of microphone/line cable because of its susceptibility to noise pickup.

Maximum Cable Length

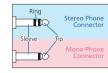
The maximum usable length of a cable will depend on the output impedance of the device feeding it. The maximum length of cable that can be used with highimpedance outputs is about 10 meters. For low-impedance outputs, the maximum is about 20 meters with unbalanced cable or 80 meters with balanced cable. Longer cables will probably cause some signal degradation, particularly a loss of the high frequencies. (The actual maximum length will also depend on the construction and quality of the cable.)

2. Connector Types

Phone Connectors

The name "phone connector" (phone plugs and phone jacks) comes from the fact that these connectors were originally used in telephone switchboards. Phone connectors come in mono and stereo types. The stereo type is also sometimes referred to as a "TRS" (Tip, Ring, Sleeve) phone connector, and these types can be used for headphones and other stereo signal connections, input/output insert patching, and balanced signals. Mono types are only used for unbalanced connections, and are commonly used for guitar and instrument cables.





RCA Pin Connectors

Most home-use audio and audio/video equipment use this type of unbalanced connector. The connectors are often color-coded according to the type of signal they carry: white for the left audio channel, and red for the right audio channel.



XLR-type Connectors

These connectors are primarily used for balanced connections. These are the connectors of choice for most professional applications because the connectors themselves are extremely durable and reliable, and some feature a locking mechanism to prevent accidental disconnection. Normally "male" connectors are used for outputs and "female" connectors are used for inputs



Speakon Connectors

This is a relatively new type of connector that is becoming widely used for speaker connections in professional applications. This type of connector features easy connection as well as high reliability.



Dinner Presentation

For formal or informal gatherings that require only a few microphones for speech and an electronic musical instrument or two, a system like the one shown here should be more than sufficient.

With the STAGEPAS 250M, you get a portable PA system comprising a stereo powered mixer, a high-quality speaker, and a speaker cable. As such, it is the ideal choice for speeches, dinner presentations, and other basic PA applications. The default mono specification of the STAGEPAS 250M is all you will need to amplify spoken word, and if you need to also handle stereo music playback or instruments, an MSR250 speaker can be added to realize a flexible stereo PA system.

System Chart



1, 2 channels

Line:

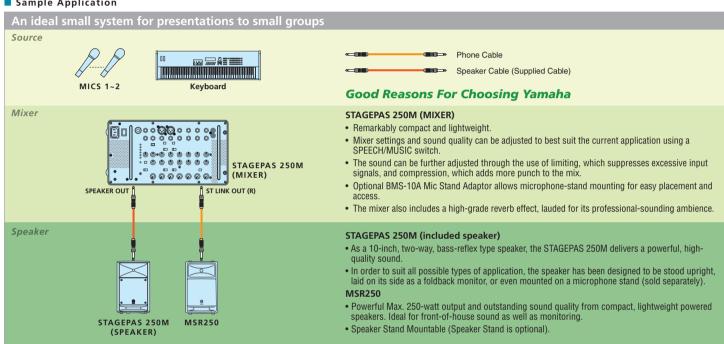
1 ~ / staran

Scale: Large room or restaurant with dinner tables

Audience:

Approximately 50 people

■ Sample Application



1 Simple Steps To Better Sound

Gain is the key to level control

The role of a sound system is basically to mix and amplify multiple sources to the required level. Learning to set appropriate levels is vitally important to achieving the best possible sound quality.

■Equipment List

Portable PA System STAGEPAS 250M

MSR250

- 1. The most important level adjustment is gain control. For the best mix and sound, the gain of individual channels should be as high as possible without distorting or clipping the signal. Some basic levels are given in the chart below.
- If your mixer has peak indicators, gain should be set so that they light only occasionally on brief signal peaks.
- 2. Set the master fader about three-quarters of the way toward it's
- **3.** Use the channel faders to set the balance between the input sources. Watch the level meters while doing this, and try to set the overall mix so

that the meters just reach peak level on the highest peaks that will be encountered during the program.

■Gain Setting Guide

Microphone:	Speech	-500
Microphone:	Soft Vocal	-500
Microphone:	Loud Vocal	-40
Cassette/VTR		-10
CD/LD		0.

Turn the amplifier's attenuator or volume control up to the required level. The important thing to remember is that you should not move the mixer's master fader at this point. Use the power amplifier controls to set the overall level. In powered mixers, however, the master fader doubles as the power amp attenuator, and can be used to set the overall volume

Party and DJ School and Business

Outdoor Dance Event

This relatively large system is designed to deliver dynamic full-spectrum sound that will keep dancers and revelers on the floor. The high-capacity MG32/14FX console comfortably handles a wide range of sources and signal-processing, while four P5000S power amplifiers driving four S215V (C215V) speakers and SW118V (CW118V) subwoofers lay down sound the dancers will delight in. Monitor sound is delivered via P3500S amps and SM12V (CM12V) monitor speakers, and graphic EQ for feedback control.

System Chart

Mic: 1 ~ 24 channels

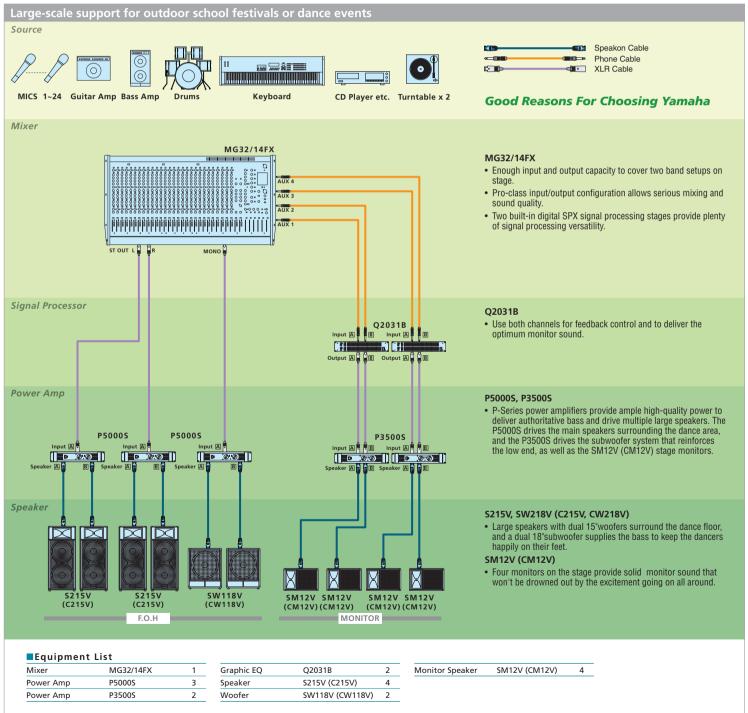
Line: Four stereo

Scale: Public area with stage set up for music and a dance floor

Audience:

500 people—audience and dancers

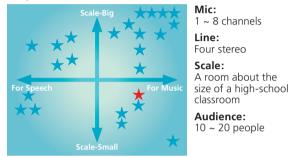
Sample Application



Acoustic Music

Acoustic instruments often need amplification to be heard and appreciated by more than a handful of people. Here's a little system that is ideal for such applications, with room to spare. Plug your electric-acoustic guitars and microphones into an EMX212S powered mixer that directly drives a pair of BR10 speakers.

■ System Chart

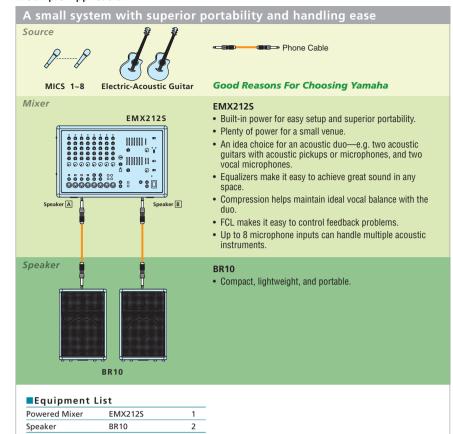


~ 8 channels Line:

Scale: A room about the

classroom Audience: 10 ~ 20 people

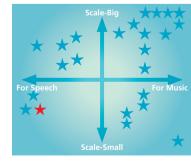
■ Sample Application



Conference Room

The system shown here can deliver both optimum speech intelligibility for meetings as well as quality music reproduction for breaks. The compact MG166C-USB mixer is a perfect choice for this type of application, with enough capacity and control features to cover just about any conceivable situation. This system allows extra audience microphones to be provided, and Q&A sessions can be easily recorded to the computer.

System Chart

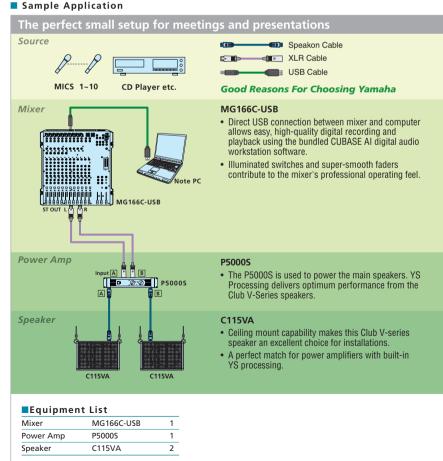


Mic: ~ 10 channels Line: 2 ~ 4 stereo Scale:

Meeting or presentation room

Audience: 20 ~ 30 people

Sample Application

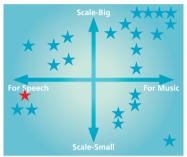


School and Business School and Business

Lecture Hall

Lecture hall sound systems require a surprising degree of functionality—in addition to microphones, there are tape and disc audio sources, audio from a video projector. and computer sound output for OHP presentations. The MG206C-USB mixer shown in this example is large and versatile enough to handle all of the above and more. And a pair of C115V (S115V) powered speakers delivers enough power and quality to project the sound to the highest tier in the room. This system allows extra audience microphones to be provided, and Q&A sessions can be recorded to the computer. Recorded material can be easily turned into podcasts for distribution.

System Chart



Gymnasium

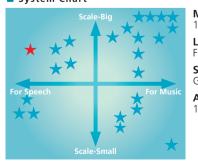
Mic: ~ 16 channels Line: 1 ~ 4 stereo

Scale: College lecture hall Audience: 50 ~ 100 people

Speaker

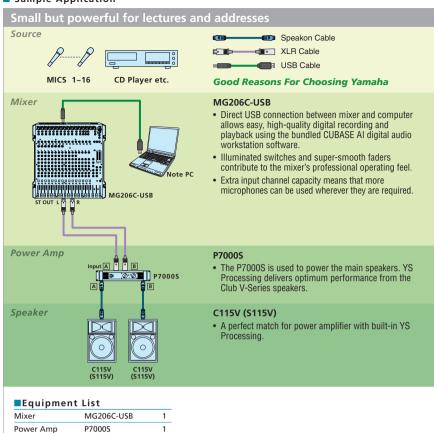
In addition to sports events, the "gym" also serves as a venue for gatherings, meetings, and even movies and concerts. Plenty of PA power and mix capacity is required to cover all possibilities. Here's a system that will do the job nicely without being excessive: an EMX5014C powered mixer driving a pair of BR15 speakers. Superb quality and useful capacity without a lot of extra equipment.

System Chart



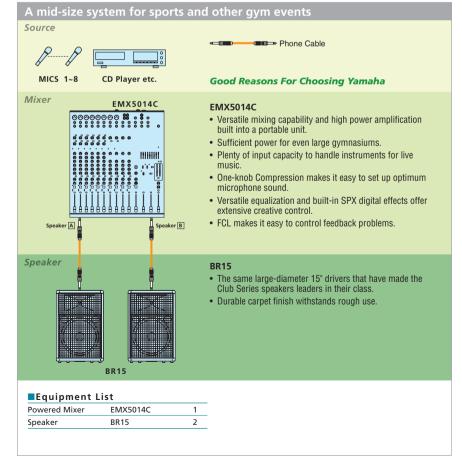
Mic: ~ 8 channels Line: Four stereo Scale: Gymnasium Audience: 100 ~ 200 people

■ Sample Application



■ Sample Application

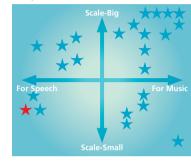
C115V (S115V)



Point of Sale

There are a multitude of uses for small PA systems in markets and malls. A small mixer, a single microphone, and either the MSR100 or MS101III powered speaker are all you need for a basic system, but you could add a CD player or other music source for BGM as required. Yamaha's 1-knob compressor can help to achieve maximum intelligibility and projection.

System Chart

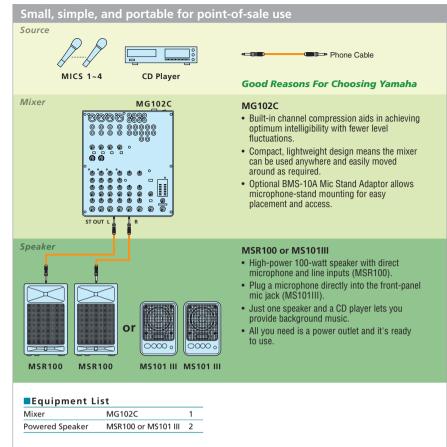


~ 4 channels Line: 2 ~ 4 stereo

Scale: Large room or restaurant with dinner tables

Audience: Approximately 50 people

■ Sample Application



2 Simple Steps To Better Sound

Connection and power switching order

Although simple, the following points will help you to keep your speakers and sound gear in top condition for as long as possible.

Connecting Cables

Always make sure that all equipment is turned off when making connections. Also make sure that all volume and level controls are turned down to minimum before turning the power on.

Power ON/OFF Switching

When turning on the power to your system, follow the procedure outlined below to protect your speakers from the power surge that occurs when sound gear is switched on or off

- **1.** Turn on electric/electronic musical instruments and sources such as CD or cassette players
- **2.** Turn on the mixer
- **3.** Turn on any graphic equalizers
- **4.** Turn on the power amp(s)

Reverse this procedure when turning the system off.
See "Simple Steps To Better Sound – 2" for information on preventing speaker overload and "Simple Steps To Better Sound – 3" for feedback

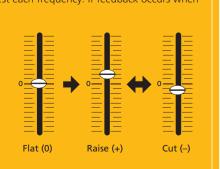
3 Simple Steps To Better Sound

Feedback control

Piercing squeals of feedback can put a damper on even the finest performance. Feed back control is a vitally important part of live sound.

- 1. Check the relative positions of microphones and speakers—feedback is caused by sound from the speakers being picked up again by the microphones and "re-circulated"
- In many cases, feedback can be eliminated simply by adjusting the positions of the speakers and microphones. Hand-held microphones require extra care so that the performer doesn't move into the "feedback zone"
- 2. Use graphic EQ. If speaker and microphone positioning doesn't solve the problem, use some graphic equalization. Begin by bringing the mixer levels up to a point just before feedback begins—raise the microphone-input channel faders and then raise the master fader until feedback begins, then back off a bit. Start with the graphic EQ controls flat (set at "0") and, starting with the lowest frequency, raise each control a few dB. If no feedback occurs, return the controls to "0" as you test each frequency. If feedback occurs when vou boost one

frequency, cut that frequency by a few dB, and continue testing the remaining frequency bands. This procedure can be effective in preventing feedback in situations in which the microphones and monitor speakers must be used in close proximity



Portable Solution Portable Solution

A complete PA system Extra Power and Capacity designed from the start for for Larger Venues portability and convenience

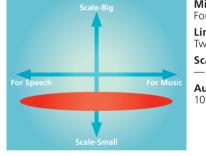
STAGEPAS 300

 Perfect for public speaking, vocalists, guitarists... just about any performance or event requiring portability and convenience combined with superior sound and versatility.



- Eight-channel powered mixer with four microphone and two stereo line inputs, 150W + 150W power, and digital reverb.
- High-performance two-way bass-reflex speakers double as convenient storage for mixer and cables... with enough room left over for microphones and other articles you might need.
- The powered mixer can be operated while mounted in the speaker cabinet, or separated for versatility and easy access.
- Optional BMS10A Mic Stand Adaptor allows the mixer to be mounted on a mic stand.

System Chart



Mic: Four inputs Line: Scale:

Audience:

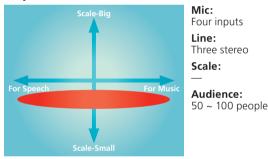
STAGEPAS 500

• Dual 250-watt power amplifier can fill surprisingly large clubs or rooms, or project your sound over a considerable distance outdoors, Class-D power delivers superior sound quality as well as reliability in a remarkably compact unit.

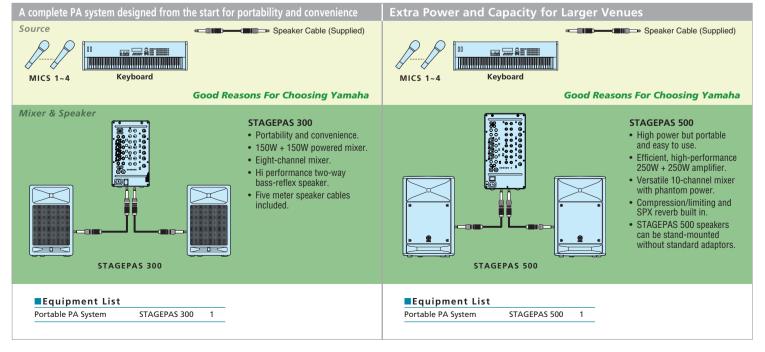


- 10 input channels in all: four mono microphone/line inputs and three stereo line inputs. Switchable phantom power is provided for high-performance phantom-powered condenser microphones, and top-quality Yamaha SPX reverb is built in.
- Channels 1 and 2 feature LIMIT/COMP switches that let you apply either limiting or compression to those channels.
- STAGEPAS 500 speakers can be stand-mounted without standard adaptors.

System Chart



■ Sample Application



An ideal choice for amplifying keyboards, guitars, and bass, Yamaha's STAGEPAS series of single-speaker systems is also perfect for small-venue PA

STAGEPAS 150M

Keyboard players and drummers can use the Click Assign switches to route a metronome or source from a main PA mixer via channel 5/6 (or channel 7/8 on the 250M) to headphones and a STAGEPAS speaker. The system's full-range speaker can be stood upright, laid on its side as a foldback monitor, or mounted on a speaker stand (using a separatelysold adapter) to suit all possible types of installation. And if necessary, an MSR100 speaker can be added to realize a stereo system.

Mic:

Line:

Scale:

Two inputs

Three stereo

Audience:

Keyboard Amp/Monitor

- Flexible 8-channel Powered Mixer
- Detachable Powered Mixer

System Chart

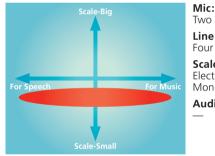
- 150 Watt Class D Power Amplifier
- Two-way Full-range Speaker System
- Built-in EO and 1-bit Modulation Reverb
- MIC/LINE switches select microphone or line level input matching for channels 1 and 2.

STAGEPAS 250M

Not only keyboard players and drummers, but also singers accompanying themselves on an electric-acoustic guitar are well served by this system. Boasting a full range extending into crystal-clear high frequencies, the STAGEPAS 250M can be stood upright, laid on its side as a foldback monitor, or mounted on a speaker stand to suit all possible types of installation. What's more, an MSR250 speaker can be added to provide support for stereo playback.

- Versatile 10-channel Powered Mixer
- Detachable Powered Mixer
- 250 Watt Class D Power Amplifier
- Two-way Full-range Speaker System
- Built-in EQ and High-quality Digital Reverb
- Channels 1 and 2 feature LIMIT/COMP switches that let you apply either limiting or compression to those channels.
- Pole mount receptacle for direct speaker stand mounting

System Chart

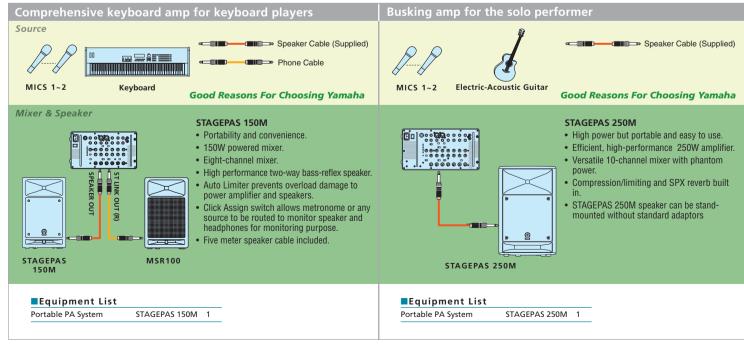


Two inputs Line: Four stereo

Scale: Electric-Acoustic Guitar Amp/

Audience:

■ Sample Application



Product Lineup

The Yamaha pro audio lineup includes everything you need to achieve professional sound in applications ranging from small events to large concerts. For serious power and system versatility, there is an excellent selection of independent components—mixers, amplifiers, equalizers, speakers. But when convenience and ease are the main criteria, there's a good range of powered mixers and speakers to choose from as well. Whatever your live sound needs. Yamaha has the solution

Recorder

Pocket Recorder **POCKETRAK C24**

Mixer

Signal

Processor

Power Amp

Speaker

Loud Speaker

S112V

S115V

S215V

C112V

Mixing Console MG102C MG124C MG166C MG166C-USB MG206C MG206C-USB

Mixing Console MG24/14FX MG32/14FX

MG82CX

MG124CX

MG166CX

Mixing Console MG166CX-USB

Powered Mixer EMX212S EMX312SC EMX512SC

Powered Mixer EMX5014C EMX5016CF

Powered Speaker

Powered Sub Woofer

Powered Speaker

MS101 III

Loud Speaker

BR10

BR12

BR15

MSR100

MSR250

MSR400

MSR800W

Professional Multi-Effect Processor SPX2000

> **Graphic Equalizer** O2031B

Mixing Console

IM8-24

IM8-32

IM8-40

Power Amplifier P7000S

P5000S P3500S P2500S

Monitor Speaker

BR12M

BR15M

STAGEPAS 250M

Portable PA System

STAGEPAS

STAGEPAS

Portable PA System

STAGEPAS

150M

300

500

Powered Monitor Speaker HS50M HS80M **Powered Sub Woofer HS10W**

> **Powered Monitor** Speaker **MSP5 STUDIO**

MSP7 STUDIO Powered Sub Woofer

SW10 STUDIO

POCKETRAK W24

Digital Mixing Console 01V96VCM

Sound Processor (Graphic Equalizer)

Power Amp

Speakers

Mixer

Feedback Control

Reverb for Vocals

Selecting PA Equipment

• Although you might think of graphic equalizers as precision sound-shaping tools (and they are), one of their main uses in live-sound applications is feedback control.

equipment for your application.

• Meetings, lectures, and other speech-based applications may only require a small mixer with a few microphone inputs. Live concerts, on the other hand, may involve microphones for vocals, chorus, and a multitude of musical

instruments. The number of microphone inputs you need is the first requirement you should consider when choosing

• If your application involves singing, you'll probably want to add some reverb and/or delay to the vocal sound.

You could use external signal processing, but it might be more convenient to have this capability built right into the mixer. The EMX series powered mixers and MG series "CX" mixers feature a range of top-quality built-in effects.

Mixers with built-in power amplification are often simply referred to as "powered mixers." The benefits of this type of
mixer include easier setup and greater portability (you don't have to carry and connect an external power amplifier).

Powered mixers are recommended for applications such as meeting rooms or other situations in which the size of the

• Graphic equalizers divide the audio spectrum into narrow frequency bands that can be individually adjusted as required. The Yamaha Q2031B, for example, is a stereo graphic equalizer that provides 31 bands of adjustment per channel. Feedback can be prevented by reducing the level of the frequencies at which it is likely to occur.

The term "PA system" can be applied to an extremely wide range

of equipment used to provide sound for an endless variety of

applications. Here's a guick guide for selecting the right

• EMX-series mixers feature built-in graphic EQ.

room and audience remains constant.

How Many Microphone Inputs?

One Watt Per Person

- The power output of a PA system is determined by the power amplifier(s) used. So how much power do you need? A basic rule of thumb is to allow about 1 watt per audience member. If your audience will be about 100 people, plan on having about 100 watts of power available
- The one-watt-per-person rule is only a guideline, however, and the actual power requirements will be affected by many factors including the natural reverberation of the venue and the efficiency of the speakers used. Outdoor applications require considerably more power than indoor events. The type of music being performed will also affect the amount of power required. In all cases, it is a good idea to have some "extra" power to spare.
- Another important power-related issue to keep in mind is speaker protection. Speakers have a maximum powerhandling capacity that, if exceeded, will result in speaker damage. Speakers with a greater power-handling capacity than the power output of the amplifiers used should always be chosen for safety.
- * Professional sound engineers will generally use larger-than-necessary power amplifiers running at considerably lower than maximum output to maximize sound quality.

 In such cases great care must be taken not exceed the speakers' maximum power-handing capacity.

Speaker Efficiency

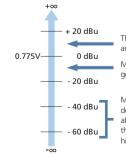
- Clearly the power output of a sound system is an important consideration, for which output "watts" is the most commonly used specification. But speaker efficiency also plays an important role in determining the "loudness" of the
- Yamaha specifies speaker efficiency by measuring sound pressure level one meter away from the cone of a speaker driven by one watt of power (dB SPL/1Wom). So for example, if 200 watts of power is supplied to the Yamaha SM10V speaker, which has an efficiency of 96 dB SPL/1Wom, the resultant sound pressure level will be 119 dB. But to achieve the same 119 dB sound pressure level with the SM15V, which has a higher efficiency of 99 dB SPL/1Wom, only 100 watts of power is required. An efficiency difference of only 3 dB means you have to either double or halve the amount of power provided to achieve the same sound pressure level (see below)

The Meaning of "dB"

If the smallest sound that can be heard by the human ear is given an arbitrary value of 1, then the loudest sound that can be heard is approximately 1,000,000 (one million) times louder. That's too many digits to deal with for practical calculations, and so the more appropriate "decibel" (dB) unit was created for sound-related measurements. In this system the difference between the softest and loudest sounds that can be heard is 120 dB. This is a non-linear scale, and a difference of 3 dB actually results in a doubling or halving of the loudness.

Ratio	0.01	0.1	0.5	1	2	4	5	8	10	100	1000
dB	-20dB	-10dB	-3dB	0dB	3dB	6dB	7dB	9dB	10dB	20dB	30dE

corresponds to a signal voltage of 0.775 V.



The standard level for inputs and outputs on mixers and most professional audio gear is +4 dBu.

Most electronic musical instruments and home audio gear have a standard line output level of -10 dBu.

Microphone levels vary over a wide range, depending on the sound source. Average speech is about -30 dBu, the chirping of a bird might be less than 50 dBu, and a drum at close range can be as

C115V CM12V **CW118V** CM15V **CW218V** C215V **C112VA C115VA**

SM₁₀V

SM12V

SM15V

CM₁₀V

Monitor Speaker Sub Woofer

SW115V

SW118V

SW218V

CW115V

* The decibel scale is a relative scale based on an arbitrarily chosen "0" value. In most audio equipment 0 db

Product Line Up—Mixers

Product Line Up—Mixers

IM8 Series

MIXING CONSOLES

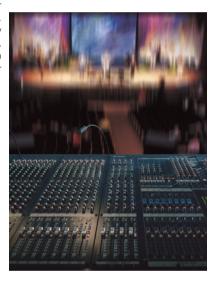
-K_{nob}

♦ CUBASE AI₄



The Yamaha IM8 series brings experience and know-how accumulated over 35 years in the production of industry-leading mixing consoles to bear in three mid-size consoles that cut no corners when it comes to overall performance and sonic quality. In addition to no-compromise design and development aimed at delivering the finest performance and most useful feature set available in this class, production and assembly are carried out at Yamaha's own domestic facilities – the same factories where the legendary Yamaha PM series consoles are produced – to ensure unrelenting quality control throughout. Every inch of these extraordinary consoles is well thought out and built for a purpose. There are no unnecessary features, and nothing is out of place. The IM8 series consoles deliver a basic but plentiful complement of features plus truly transparent, high-resolution sound with tireless reliability.

These are consoles for serious sound applications, and will provide eminently professional performance, sound, and control in permanent installations or on the road.





The IM8 Series

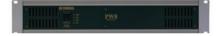
- IM8-40: 40 mono + 4 stereo inputs/8 aux + 8 group + 4 matrix + stereo + mono out buses
- IM8-32: 32 mono + 4 stereo inputs/8 aux + 8 group + 4 matrix + stereo + mono out buses
- IM8-24: 24 mono + 4 stereo inputs/8 aux + 8 group + 4 matrix + stereo + mono out buses

Main Features

- Professional build, features, and performance for serious live sound applications.
- Domestic production and assembly ensure faultless quality control.
- Unique Yamaha one-knob compressor on all mono input channels.
- Comprehensive master section provides extensive signal routing and control versatility.
- Output matrix affords extra output flexibility, particularly for installations.
- Traditional Yamaha color coded controls for easy identification and operation.
- USB audio I/O allows direct digital recording and playback with the supplied Cubase AI4 audio workstation software.
- External power supply maximizes console performance dual power supplies can be used for redundant failsafe operation.

Options

PW8 POWER SUPPLY



PSL1010
POWER SUPPLY LINK CABLE

LA1L GOOSENECK LAMP



S pecifications

	IM8-24	IM8-32	IM8-40				
Total Harmonic Distortion		Less than 0.1% (20Hz to 20kHz)					
Frequency Response		0, +1,-3dB(20Hz to 20kHz)					
Input Hum & Noise *1		Equivalent input noise; -128dBu, Residual noise; -98dBu					
Crosstalk		Less than -70dB					
Phantom Power		+48V DC (each channel)					
CH & ST High Pass Filter		80Hz, 12dB/oct					
Input	MIC: 24, LINE: 24 mono + 4 stereo, CH INSERT: 24, GROUP INSERT: 8, AUX INSERT: 8, ST INSERT: 1, MONO INSERT: 1, AUX RETURN: 4 stereo, 2TR: 1 stereo, TB IN: 1	MIC: 32, LINE: 32 mono + 4 stereo, CH INSERT: 32, GROUP INSERT: 8, AUX INSERT: 8, ST INSERT: 1, MONO INSERT: 1, AUX RETURN: 4 stereo, 2TR: 1 stereo, TB IN: 1	MIC: 24, LINE: 40 mono + 4 stereo, CH INSERT: 40, GROUP INSERT: 8, AUX INSERT: 8, ST INSERT: 1, MONO INSERT: 1, AUX RETURN: 4 stereo, 2TR: 1 stereo, TB IN: 1				
Output	STEREO OUT: 1, GROUP: 8, AUX SEND: 8, MONO: 1, MATRIX OUT: 4, CH INSERT: 24, AUX INSERT: 8, GROUP INSERT: 8, ST INSERT: 1, MONO INSERT: 1, DIRECT: 24, REC: 1 stereo, MONITOR: 1 stereo, PHONES: 1	STEREO OUT: 1, GROUP: 8, AUX SEND: 8, MONO: 1, MATRIX OUT: 4, CH INSERT: 32, AUX INSERT: 8, GROUP INSERT: 8, ST INSERT: 1, MONO INSERT: 1, DIRECT: 32, REC: 1 stereo, MONITOR: 1 stereo, PHONES: 1	STEREO OUT: 1, GROUP: 8, AUX SEND: 8, MONO: 1, MATRIX OUT: 4, CH INSERT: 40, AUX INSERT: 8, GROUP INSERT: 8, ST INSERT: 1, MONO INSERT: 1, DIRECT: 40, REC: 1 stereo, MONITOR: 1 stereo, PHONES: 1				
Compressor	24 (Mic channel Only)	32 (Mic channel Only)	40 (Mic channel Only)				
CH EQ (MONO) *2 ±15 dB (Max.)		HIGH: 10 kHz (shelving) Hi-MID: 400 Hz-8 kHz (peaking) LO-MID: 80 Hz-1.6 kHz (peaking) LOW: 100 Hz (Shelvina)					
CH EQ (STEREO) *2 ±15 dB (Max.)		HIGH: 10 kHz (shelving) Hi-MID: 3 kHz (peaking) LO-MID: 800 Hz (peaking) LOW: 100 Hz (shelving)					
Internal Digital Effect		-					
USB Audio USB IN/OUT		YES					
Dimensions (W x H x D mm)	1227 x 219 x 739	1471.5 x 219 x 739	1716 x 219 x 739				
Weight	37.5 kg	44.5 kg	51.5 kg				
Power Requirements		Use PW8 power supply unit	·				
Option	Pow	er supply unit PW8, Power supply link cable PSL1010, Gooseneck lamp	LA1L				

MG Series MIXING CONSOLES



CUBASE Al4
(USB Models Only)

12 U

An Impressive Lineup with a Common Theme

—Unmatched Quality and Performance

Whether you have a mixing application that involves only a few channels, or up to 32 inputs with substantial signal routing versatility, Yamaha's MG series offers a console that will give you the capacity, control, and performance you deserve. All models are remarkably compact and lightweight for superior handling and portability, but absolutely no compromises have been made in terms of features, performance, or durability. The MG mixers offer an extraordinary blend of technology: some inherited from professional Yamaha mixing consoles you're likely to see in distinguished halls throughout the world, and some developed specifically to deliver optimum performance in the MG-console format. In either case these performance packed mixers are founded on a world-spanning network of human and technological resources that crystallize to deliver features that really matter, and sound that is undeniably superior.



C Models with Improved Utility

10, 12, 16, and 20 channel models for a wide range of applications. They may be the most basic models in the series, but they're packed with features and performance that will let you create superior mixes with ease.

CX Models with SPX Digital Effects

With outstanding Yamaha SPX multi-effects built in, these models offer an extraordinary range of creative control while eliminating the need for bulky external effect racks.

USB Models for Live Recording

If live recording is an important aspect of your application, choose one of the USB models. They allow top-quality 2-track digital recording with the bundled CUBASE AI digital audio workstation software.

S pecifications

	MG102C	MG124C	MG166C	MG206C	MG82CX				
Total Harmonic Distortion		0.1 % (THD+N) +14	dBu, 20Hz – 20kHz, Input Gain Control a	t minimum (ST OUT)					
Frequency Response		-3, 0, +1dB 20Hz - 20kHz @ +4dBu (ST OUT)							
nput Hum & Noise *1 -128dBu Equivalent Input Noise/ -100dBu Residual Output Noise, 20Hz – 20kHz, Res 150Ω, Input Gain=Maximum, Input sensitivity=-60dB		-128dBu Equivalent Input Noise/ -98dBu Residual Output Noise, 20Hz – 20kHz, Rs=150Ω, Input Gain=Maximum, Input sensitivity=-60 dB	-128dBu Equivalent Input Noise/ -98dBu Residual Output Noise, 20Hz – 20kHz, Rs=15ΩΩ, Input Gain=Maximum, Input sensitivity=-60 dB	-128dBu Equivalent Input Noise/ -98dBu Residual Output Noise, 20Hz – 20Hz, Rs=150Ω, Input Gain=Maximum, Input sensitivity=-60 dB	-128dBu Equivalent Input Noise/ -100dBu Residual Output Noise, 20Hz – 20kHz, Rs=150Ω, Input Gain=Maximum, Input sensitivity=-60dB				
Crosstalk			-70dB @ 1kHz						
Phantom Power		+ 48V							
CH & ST High Pass Filter			80Hz 12dB/Octave						
Input	MIC: 4, LINE: 2 mono + 4 stereo, CH INSERT: 2, RETURN: 1 stereo, 2TR IN: 1 stereo		MIC: 10, LINE: 8 mono + 4 stereo, CH INSERT: 8, RETURN: 1 stereo, 2TR IN: 1 stereo	MIC: 16, LINE: 12 mono + 4 stereo, CH INSERT: 12, RETURN: 2 stereo, 2TR IN: 1 stereo	MIC: 4, LINE: 2 mono + 4 stereo, CH INSERT: 2, RETURN: 1 stereo, 2TR IN: 1 stereo				
Output	ST: 1, AUX SEND: 1 , CH INSERT: 2, REC: 1 stereo, MONITOR: 1 stereo, Phone: 1	ST: 2, AUX SEND: 2, CH INSERT: 4, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 2, Phone: 1	ST: 2, AUX SEND: 3, CH INSERT: 8, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1	ST: 2, AUX SEND: 4, CH INSERT: 12, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1	ST: 1, EFFECT SEND: 1 , CH INSERT: 2 REC: 1 stereo, MONITOR: 1 stereo, Phone: 1				
CH EQ (MONO) *2 ±15 dB (Max.)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)				
CH EQ (STEREO) *2 ±15 dB (Max.)	10 kHz (Shelving), 100 Hz (Shelving)	10 kHz (Shelving), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 kHz (Shelving), 100 Hz (Shelving)				
Internal Digital Effect	-	-	-	-	16 Programs: Parameter Control Foot Switch (Digital Effect On/Off)				
USB Audio USB IN/OUT	-	-	-	-	-				
Dimensions (W x H x D mm)	256.6 x 62.2 x 302.5	346.2 x 86.1 x 436.6	478 x 105 x 496	478 x 105 x 496	256.6 x 62.2 x 302.5				
Weight	1.5 kg	3.0 kg	5.3 kg	6.0 kg	1.6 kg				
Power Requirements	21W	30W	30W	40W	21W				
Option	BMS-10A (Mic Stand Adaptor)	_	-	-	BMS-10A (Mic Stand Adaptor), FC5 (Foot S				

	MG124CX	MG166CX	MG166C-USB	MG166CX-USB	MG206C-USB
Total Harmonic Distortion		0.1 % (THD+N) +14	4 dBu, 20Hz – 20kHz, Input Gain Control at	minimum (ST OUT)	
Frequency Response		-	3, 0, +1dB 20Hz – 20kHz @ +4dBu (ST OU	r)	
Input Hum & Noise *1 -128dBu Equivalent Input Noise/ -98dBu Residual Output Noise, 20Hz - 20kHz, Ses=150Ω, Input Gain=Maximum, Input Sensitivity-60 dB		-128dBu Equivalent Input Noise/ -98dBu Residual Output Noise/ -98dBu Residual Output Noise, -98dBu Residual Output Noise/ -98d		-128dBu Equivalent Input Noise/ -98dBu Residual Output Noise, 20Hz – 20kHz, Rs=150Ω, Input Gain=Maximum, Input sensitivity=-60 dB	
Crosstalk			-70dB @ 1kHz		
Phantom Power			+ 48V		
CH & ST High Pass Filter			80Hz 12dB/Octave		
Input	MIC: 6, LINE: 4 mono + 4 stereo, CH INSERT: 4, RETURN: 1 stereo, 2TR IN: 1 stereo	MIC: 10, LINE: 8 mono + 4 stereo, CH INSERT: 8, RETURN: 1 stereo, 2TR IN: 1 stereo	MIC: 10, LINE: 8 mono + 4 stereo, CH INSERT: 8, RETURN: 1 stereo, 2TR IN: 1 stereo	MIC: 10, LINE: 8 mono + 4 stereo, CH INSERT: 8, RETURN: 1 stereo, 2TR IN: 1 stereo	MIC: 16, LINE: 12 mono + 4 stereo, CH INSERT: 12, RETURN: 2 stereo, 2TR IN: 1 stereo
Output	ST: 2, AUX SEND: 1, EFFECT SEND: 1, CH INSERT: 4, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 2, Phone: 1	ST: 2, AUX SEND: 2, EFFECT SEND: 1, CH INSERT: 8, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1	ST: 2, AUX SEND: 3, CH INSERT: 8, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1	ST: 2, AUX SEND: 2, EFFECT SEND: 1, CH INSERT: 8, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1	ST: 2, AUX SEND: 4, CH INSERT: 12, REC: 1 stereo, MONITOR: 1 stereo, GROUP: 4, Phone: 1
CH EQ (MONO) *2 ±15 dB (Max.)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)
CH EQ (STEREO) *2 ±15 dB (Max.)	10 kHz (Shelving), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)	10 k Hz (Shelving), 2.5 kHz (Peaking), 100 Hz (Shelving)
Internal Digital Effect	16 Programs: Parameter Control Foot Switch (Digital Effect On/Off)	16 Programs: Parameter Control Foot Switch (Digital Effect On/Off)	-	16 Programs: Parameter Control Foot Switch (Digital Effect On/Off)	-
USB Audio USB IN/OUT	-	-	Sampling Frequency = 44.1kHz or 48kHz (depend on the application of PC)	Sampling Frequency = 44.1kHz or 48kHz (depend on the application of PC)	Sampling Frequency = 44.1kHz or 48kHz (depend on the application of PC)
Dimensions (W x H x D mm)	346.2 x 86.1 x 436.6	478 x 105 x 496	478 x 105 x 496	478 x 105 x 496	478 x 105 x 496
Weight	3.2 kg	5.5 kg	5.3 kg	5.5 kg	6.0 kg
Power Requirements	30W	35W	30W	35W	40W
Option	FC5 (Foot SW)	FC5 (Foot SW)	-	FC5 (Foot SW)	_

Product Line Up—Mixers

Product Line Up—Mixers

MG32/14FX, MG24/14FX

MIXING CONSOLES

Serious Capacity for Live Sound & Installations

f your application is live sound reinforcement you'll want all the channel capacity you can get just in case. With 24 and 32 input channels, respectively, the MG24/14FX and MG32/14FX are ready to handle all but the most ambitious sound-reinforcement setups. And with dual SPX digital effect systems on-board you won't need racks of outboard gear to get the sound you need. There's also a comprehensive range of group and auxiliary busses to make even complex mixes easy.



S pecifications

	MG24/14FX	MG32/14FX
Total Harmonic Distortion	Less than 0.1 % (THD+N), 20F	lz – 20kHz @ +14dBu (ST OUT)
Frequency Response	0 +1, -3dB, 20Hz - 20H	:Hz @ +4dBu (ST OUT)
Input Hum & Noise *1	-128 dBu Equivalent Input Noise 20Hz – 20kHz, Rs=150Ω, Input 0	e/-99dBu Residual Output Noise Gain=Maximum, Input Pad =OFF,
Crosstalk	-70dB	@ 1kHz
Phantom Power	+4	8 V
CH & ST High Pass Filter	80Hz 12d	B/Octave
Input	MIC: 16 + 1, LINE: 16 mono + 4 stereo, CH INSERT: 16, AUX RTN: 2 stereo, 2TR IN: 1, ST INSERT: 1, GROUP INSERT: 4	MIC: 24 + 1, LINE: 24 mono + 4 stereo, CH INSERT: 24, AUX RTN: 2 stereo, 2TR IN: 1, ST INSERT: 1, GROUP INSERT: 4
Output	ST, MONO, AUX SEND: 6, CH INSERT: 16, REC: 1 stereo, C/R: 1 stereo, GROUP: 4, FX: 2	ST, MONO, AUX SEND: 6, CH INSERT: 24, REC: 1 stereo, C/R: 1 stereo, GROUP: 4, FX: 2
CH EQ (MONO) *2 ±15 dB (Max.)	100Hz (Shelving), 0. 10kHz (S	25 – 5kHz (Peaking), Shelving)
CH EQ (STEREO) *2 ±15 dB (Max.)		800Hz (Peaking), 10kHz (Shelving)
Internal Digital Effect		Programs, Effect 2: rameter Control)
Dimensions (W x H x D mm)	819 x140 x 551	1027 x 140 x 551
Weight	18.5 kg	22 kg
Power Requirements	100W 120V / 60Hz 100W 220V / 50Hz 100W 230V / 50Hz	120W 120V / 60Hz 120W 220V / 50Hz 120W 230V / 50Hz
Option	FC5 (Foot SW)	FC5 (Foot SW)

EMX512SC, EMX312SC, EMX212S (EMX512SC) 12SC (EMX512SC) (EMX512SC)









POWERED MIXERS

here portability and convenience are important criteria, a system based on a high-performance Yamaha EMX-series powered mixer is definitely the way to go. In one integrated, portable unit you have a mixer to combine and balance your microphone, instrument, and line sources, effects to refine and polish your sound, and power to drive the main speakers and even monitor speakers as well. But that's nowhere near the whole story—Yamaha EMX-series Powered Mixers offer a range of features that let you mix, process, and deliver your sound with maximum quality and creative control... and, of course, that unrivalled Yamaha sound.

EMX512SC



EMX312SC



EMX212S



S pecifications

		EMX512SC	EMX312SC	EMX212S
Maximum Output Power @0.5% THD	at 1kHz	500W+500W/4ohms, 350W+350W/8ohms	300W+300W/4ohms, 190W+190W/8ohms	220W+220W/4ohms, 130W+130W/8ohms
Input Connectors		MIC: max.8, LINE: 4 stereo	MIC: max.8, LINE: 4 stereo	MIC: max.8, LINE: 4 stereo
Output Connectors		SPEAKER OUT: (A1, A2, B1, B2), MAIN OUT: 1 stereo, EFFECT OUT: 1, MONITOR OUT: 1, REC OUT: 1 stereo	SPEAKER OUT: (A1, A2, B1, B2), MAIN OUT: 1 stereo, EFFECT OUT: 1, MONITOR OUT: 1, REC OUT: 1 stereo	SPEAKER OUT: (A1, A2, B1, B2), MAIN OUT: 1 stereo, EFFECT OUT: 1, MONITOR OUT: 1, REC OUT: 1 stereo
EQ	HIGH MID LOW	10kHz shelving 2.5kHz peaking 100Hz shelving	10kHz shelving 2.5kHz peaking 100Hz shelving	10kHz shelving 2.5kHz peaking 100Hz shelving
Phantom Power		+15V	+15V	+15V
Graphic Equalizer		7 band (125, 250, 500, 1k, 2k, 4k, 8kHz): Main (Stereo) and Monitor	7 band (125, 250, 500, 1k, 2k, 4k, 8kHz): Main (Stereo) and Monitor	7 band (125, 250, 500, 1k, 2k, 4k, 8kHz): Main (Stereo) and Monitor
Digital Effects		SPX Digital Multi Effector (24bit AD/DA, 32bit Internal Processing): 16 programs	SPX Digital Multi Effector (24bit AD/DA, 32bit Internal Processing): 16 programs	SPX Digital Multi Effector (24bit AD/DA, 32bit Internal Processing): 16 programs
Power Amp. Mode		L/R, MAIN (L+R)/MONITOR	L/R, MAIN (L+R)/MONITOR	L/R, MAIN (L+R)/MONITOR
Yamaha Speaker Processing		Yes	Yes	Yes
Stand-by switch		Yes	Yes	Yes
Foot Switch		Effect On/Off	Effect On/Off	Effect On/Off
Dimensions (W x H x D mm)		442 x 274 x 286	442 x 274 x 286	442 x 274 x 286
Weight		8 kg	8 kg	8 kg
Power Requirements /Consumption		120V AC 60Hz 270W, 230V AC 50Hz 270W, 240V AC 50Hz 270W	120V AC 60Hz 270W, 230V AC 50Hz 270W, 240V AC 50Hz 270W	120V AC 60Hz 270W, 230V AC 50Hz 270W, 240V AC 50Hz 270W
Option		RK-512 (Rack Mount Adaptor), FC5 (Foot SW)	RK-512 (Rack Mount Adaptor), FC5 (Foot SW)	RK-512 (Rack Mount Adaptor), FC5 (Foot SW)

EMX5016CF

POWERED MIXERS

The EMX5016CF combines the convenience of an Integrated powered mixer with input capacity, flexible features, and solid sound that critical live sound applications demand. It is remarkably compact and portable for a live sound system with this much capability, but offers performance and reliability that will satisfy the discerning professional user either on the road or in installed applications. And thanks to leading Yamaha digital technology the EMX5016CF also includes a number of innovations that make it easier than ever to achieve topclass sound in just about any venue. An impressive power output of 500 watts per channel means it can handle fairly large audiences, indoors or out. The EMX5016CF goes considerably beyond the standard definition of "powered mixer," entering the realm of serious sound reinforcement.

EMX5014C

• f your sound reinforcement requirements are getting serious, but you still want the convenience and reliable performance of a Yamaha powered mixer, check out the

console-style EMX5014C. Here's an all-in-one solution that will appeal to bands and venue operators alike. The EMX5014C transports and sets up with the ease of systems

built around the smaller EMX-series powered mixers, but

will also prove it's worth in more permanent installations. It

can even be rack-mounted for vertical or angled operation, and real space savings! But of course the EMX5014C offers much more than just convenience. It provides a surprising

palette of features and versatile signal routing options

that can take your live sound to the next level. And it's a

POWERED MIXERS

SPX YSProcessins 12 U

-Knob



S pecifications

		EMX5016CF				
Maximum Output Power @0.5% THD at 1kHz		500W+500W/4ohms, 350W+350W/8ohms, 1000W/8ohms bridge				
Input Connectors		MIC: Max 12, LINE: 4 stereo, INSERT: 6				
Output Connectors		ST OUT: 1 stereo, ST SUB OUT: 1 stereo, EFFECT SEND: 2, AUX SEND: 2, REC OUT: 1 stereo, CH INSERT OUT: 8				
EQ HIGH MID LOW		10kHz shelving Peaking, 250Hz – 5kHz sweep 100Hz shelving				
Phantom	Power	+48V				
Graphic E	qualizer	9 band (63, 125, 250, 500, 1k, 2k, 4k, 8k, 16kHz)				
Digital Ef	fects	SPX Digital Multi Effector (24bit AD/DA, 32bit Internal Processing): 16 programs x 2				
Power An	np. Mode	L/R, AUX1/MONO, AUX1/2				
Power Se	lect Switch	500W / 200W / 75W				
Yamaha S	peaker Processing	Yes				
Stand-by	switch	Yes				
Foot Swit	ch	Effect On/Off				
Dimensio	ns (W x H x D mm)	444 x 155 x 493				
Weight		11 kg				
Power Re /Consum	quirements otion	120V AC 60Hz 500W, 220 – 240V AC 50Hz 500W				
Option		RK-5014 (Rack Mount Adaptor), FC5 (Foot SW)				

	EMX5014C
Maximum Output F @0.5% THD at 1kH;	500W+500W/4ohms, 350W+350W/8ohms, 1000W/8ohms bridge
Input Connectors	MIC: Max 8, LINE: 4 stereo, INSERT: 6
Output Connectors	ST OUT: 1 stereo, ST SUB OUT: 1 stereo, EFFECT SEND: 1, AUX SEND: 2, REC OUT: 1 stereo, CH INSERT OUT: 6
EQ HIC MI LO	Peaking, 250Hz – 5kHz sweep
Phantom Power	+48V
Graphic Equalizer	9 band (63, 125, 250, 500, 1k, 2k, 4k, 8k, 16kHz)
Digital Effects	SPX Digital Multi Effector (24bit AD/DA, 32bit Internal Processing): 16 programs
Power Amp. Mode	L/R, AUX1/MONO, AUX1/2
Power Select Switc	500W / 200W / 75W
Yamaha Speaker Pr	ocessing Yes
Stand-by switch	Yes
Foot Switch	Effect On/Off
Dimensions (W x H	(D): mm 444 x 155 x 493
Weight	10.5 kg
Power Requiremen /Consumption	s 120V AC 60Hz 450W, 240V AC 50Hz 450W, 230V AC 50Hz 450W
Option	RK-5014 (Rack Mount Adaptor), FC5 (Foot SW)

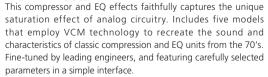
01V96VCM

Yamaha, so you know it's going to sound great.

DIGITAL MIXING CONSOLE

The 01V96VCM delivers the performance and reliability of Yamaha's acclaimed digital live sound and production consoles in a remarkably compact design that is perfect for home and professional applications where space is limited or maximum portability is required. It may be small but it can handle up to 40 inputs, and can be cascaded if more are required. And now, in addition to the many improvements that were implemented in the 01V96 Version 2, the 01V96VCM comes with a selection of Yamaha's unsurpassed VCM effects built in. Of course the entire console - effects included - features 24 bit/96 kHz operation for ultimate resolution and sound quality that will satisfy the most demanding applications.





REVERB (**X

These reverb effects employ the latest" REV-X" algorithms first introduced in Yamaha's SPX2000 Professional Multi Effect Processor. The REV-X programs feature the richest reverberation and smoothest decay available, based on years of dedicated research and development.



















REV-X HALL/ROOM/PLATE

S pecifications

	01V96VCM				
Internal processing	32bit (Accumu	lator 58bit)			
Number of scene memories	99				
Sampling frequency	Internal	44.1kHz, 48kHz, 88.2kHz, 96kHz			
	External	Normal rate: 44.1kHz-10% -48kHz+6% Double rate: 88.2kHz-10% -96kHz+6%			
Fader resolution	100mm motor	ized x 17			
Total harmonic distortion * Input GAIN=Min.	@fs = 48kHz:	Less than 0.05%, 20Hz to 20kHz @+14dB into 600ohms Less than 0.01%, 1kHz @+24dB into 600ohms			
CH INPUT to STEREO OUT	@fs = 96kHz	Less than 0.05%, 20Hz to 40kHz @+14dB into 600ohm Less than 0.01%, 1kHz @+24dB into 600ohms			
Frequency response	@fs = 48kHz	0.5, -1.5dB, 20Hz - 20kHz @+4dB into 600ohms			
CH INPUT to STEREO OUT	@fs = 96kHz	0.5, -1.5dB, 20Hz - 40kHz @+4dB into 600ohms			
Dynamic range (maximum level to noise level)	106dB typ. AD	Converter (STEREO OUT) H+DA (to STEREO OUT) @fs=48kHz H+DA (to STEREO OUT) @fs=96kHz			
Hum & noise level ** (20Hz-20kHz) Rs=1500hms Input GAIN=Max Input PAD=0dB Input PAD=0dB Input sensitivity=-60dB	-128dB Equivalent Input Noise86dB residual output noise. STEREO OUT STEREO OUT off86dB (90dB S/N) STEREO OUT STEREO HEAD OUT STEREO TAGE at nominal level and all CH INPUT faders at minimum level64dB (68dB S/N) STEREO OUT STEREO TAGE at nominal level and one CH INPUT fader at nominal level				
Maximum voltage gain	40dB CH INPU 74dB CH INPU	T (CH1-12) to STEREO OUT/OMNI (BUS) OUT T (CH13-16) to STEREO OUT T (CH13-12) to OMNI (AUX) OUT (via pre input fader) T (CH1-12) to MONITOR OUT (via STEREO BUS)			
Crosstalk(@1kHz)	80dB adjacent	input channels (CH1-12, 13-16)			
Input GAIN=min	80dB input to	output			
Power requirements	North America	: AC120V, 60Hz, 90W			
	Other Areas:	AC220 – 240V, 50/60Hz, 90W			
Dimensions (W x H x D mm)	436 x 150 x 54	8			
Weight	15ka				





P.T. Yamaha Music manufacturing Asia

-Manufacturing the electronic instruments and PA products From the initial design to final manufacturing, all production processes for the Yamaha EMX series Powered Mixers and MG series Mixing Consoles are erformed entirely inside the company.

Moreover, every product that comes off our production line must pass strict quality controls using the sophisticated test instruments. Thus, all of this enables us to deliver the highest quality products to you.





An Interview with the EMX/MG Design Team Built-in Compression Adds Live-sound Versatility to the new EMX-series Powered Mixers

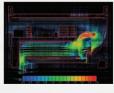
New Features

- * What is the main difference compared to previous
- The main difference is built-in compression. Compression is indispensable in almost all professional recording and live-sound applications, but we believe that this is the first time it has been built into an analog
- Most "box type" mixers have no insert connectors, so there has really been no convenient way to use compression with them. As a result, many users of this type of powered mixer have never used compression, but we wanted them to have that option in the new EMX
- Although compression is used in most pro audio applications, it has been a bit too difficult for beginners to take full advantage of. That's why we've streamlined it down to the essentials and made it very easy to use.
- Another important new feature is FCL (Feedback Channel Location). This system detects feedback and shows you which channel is causing the problem. Some mixers from other manufacturers have indicators in the graphic equalizer section that show the feedback frequency, but indicating the problem channel allows the feedback to be more effectively controlled using channel EQ.
- If you try to control feedback using the EMX graphic equalizer, for example, you end up changing the sound of the entire program. For this reason it is far more effective to control it at the input, thus avoiding degradation of the overall sound

The Battle Against Heat

- * Tell us about how you avoided heat problems in such compact enclosures.
- Heat and high power output unavoidably go hand-in-hand. In this case we were also determined to reduce weight, so the design, hardware, and mechanics teams joined forces to pursue this goal. Changing even a single component can alter the heat profile enough to require a change in heat sink design and that change can cause a change in sound quality so the design process involves a lot of trial and error.
- In this particular case, the fact that we were able to use internal heat-flow simulation and analysis was a huge advantage. We were able to define an enclosure shape on the computer, and then by analyzing the heat flow

while refining the heat sink configuration we were able to come to within 80% or 90% of the ideal final design. The final stages using physica on trial and error



• In the box-type 212C, 312SC, and 512SC, it was easy to mount the fan away from the circuit board to minimize degradation of the audio signal. But in the console-type EMX5016CF and EMX5014C finding the ideal fan location was extremely difficult. Since the fan must be located near the input circuitry, special measures have been taken to

ensuring maximum heat extraction. • The hardware team wanted to increase the size of the body by 30 millimeters, but our goals for a streamlined, compact design were important enough that we decided to find other wavs to achieve the

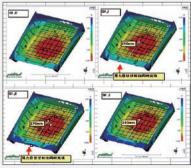
ensure that electronic and mechanical noise from the fan

do not affect sound quality, while at the same time

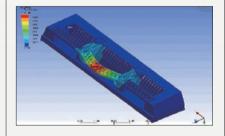
Reliability Without Compromising Performance

desired performance.

- The simplicity and aesthetic appeal of the designs are quite impressive. Tell us about the design concept
- Simplicity was the main goal, particularly in the consoletype 5016CF and 5014C. We wanted to consolidate the mixer controls, so the utility control section has been clearly separated. We didn't even want any handles to be visible.
- An important idea implemented in the box-type models is that they can be set at an angle like monitor speakers. The integral handles are also an important design feature, and achieving the required strength was a constant problem
- Achieving the ideal blend of size, weight, and durability is quite difficult. As usual, demands from the sales team continue to escalate while the hardware and mechanics teams try to turn them into reality ... without ever reducing or compromising features or performance. Computer simulation was called into play once again, providing an accurate preview of the mold-flow characteristics of the resin used for the box-type housings.
- The final strength of the molded housings depends to a large degree on how the molten resin is introduced in he mold, and how it flows within the mold.



The strength of the integral handles was also predicted using computer simulation, and as a result we have achieved strength comparable to that of



Achieving Pure Sound Quality

- What measures have been taken to ensure imum sound quality
- Of course sound quality is first and foremost in the design of any model. Achieving the lowest possible noise and hum when changing components is always. a challenge. There's influence from vibration, from the irrent flowing through the components themselves, and a simple op-amp IC change can precipitate a large change in sound. We often find ourselves using the best components we can find rather than compromise on sound quality. Even the FCL system has an effect on the sound, and we were able to achieve a dramatic improvement by simply eliminating a single component from the circuit. Once again, the final design depends on trial-and-error listening tests while changing components.
- With SPX effects in all models in this EMX series. plus compression and FCL, you can rely on a single EMX powered mixer to deliver outstanding live sound, especially in applications that use mostly
- Most compressors have at least two controls, what is the idea behind having just one?
- Simplicity. Standard compression controls can be very difficult to set quickly and accurately, but we've managed to provide well-balanced threshold and ratio settings that can be controlled by a single knob. By focusing primarily on microphone applications in which compression is applied to vocals, acoustic quitar, or similar sources, great-sounding compression can be dialed in quickly and easily.
- There's a good description of compression and its uses in the owner's manual. We hope that our users will take advantage of this very useful feature.

SPX2000 (MIDI 10

algorithm and the 24 bit/96-kHz audio DSP.



PROFESSIONAL MULTI-EFFECT PROCESSOR

The SPX2000, while inheriting the standard interface and common programs

from its predecessors, brings a new sound quality with the "REV-X" reverb

● 24-bit Liner 128-times/64-times Oversampling (@fs=44.1,48kHz/88.2,96kHz) ● PRESET BANK: 97, USER BANK: 99, CLASSIC BANK: 25 ● AC 120V, 60Hz, AC 230V, 50Hz, 25W ● 480W x 45H x 372.5D mm, 4kg







GRAPHIC EOUALIZER

Yamaha Graphic Equalizers offer features and performance that meet today's equalization needs whether they be in sound reinforcement, recording, A/V production, electronic musical instruments, broadcasting, music listening, or any other applications where precise sound tailoring is essential.



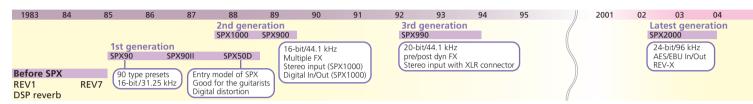
● 31 band (1/3 octave) ● Variable Range: +/- 12dB +/- 6dB | HPF: 12dB/octave (20 - 200Hz at -3dB point) ● AC 120V 60Hz, AC 230V, 50Hz

20W

480 x 93.4 x 230mm, 4.0kg

SPX History

The good reputation from users keeps our SPX effects as standard effects over 15 years...



P-Series YSProcessins ///





POWER AMPLIFIERS

💳 he P-Series power amplifiers have been designed specifically to deliver big. clean power output that will take full advantage of the quality and power handling capabilities of Yamaha speakers. The amps are equipped with YS Processing (Yamaha Speaker Processing) to deliver a signal that is optimally matched to those speakers as well. All models feature both XLR and 1/4-inch TRS inputs, and Neutrik Speakon, phone plug, and five-way binding post outputs to make connections quick and easy. Other features include Yamaha's exclusive EEEngine technology which delivers high performance with exceptional efficiency, sweepable high- and low- pass filters for optimizing output to any loudspeakers, compact and durable 2U chassis, and variable speed cooling fans.

P7000S





P5000S



P3500S

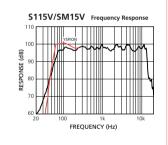


P2500S

Sound Advice

Processing) switch do, anyway? Glad you asked: it rolls off unnecessary frequencie below around 40 Hz. which helps to protect your speakers, and at the same time boosts frequencies from 60 Hz through 150 Hz by about 3 or 4 dB which, as you rocket scientists already know, is a soundpressure-level boost of about 1.5 times. And although this won't give you the boneshaking bass of a subwoofer, it will make your bass and drums sound really cool. If vou're reinforcing rock, turn it on.

So what does this YSP (Yamaha Speaker



pecifications

			P7000S	P5000S	P3500S	P2500S	
Output Power		8 ohms/STEREO	750W+750W	525W+525W	390W+390W	275W+275W	
	20Hz – 20kHz	4 ohms/STEREO	1100W+1100W	750W+750W	590W+590W	390W+390W	
	THD+N=0.1%	8 ohms/BRIDGE	2000W	1500W	1180W	780W	
THD+N	20Hz – 20kHz, half	power		≤ 0.	10%		
Intermodulation Distortion	60Hz: 7kHz, 4:1, h	alf power		≤ 0.	10%		
Frequency Response Po=1W, RL=8 ohms				0dB, +0.5dB, -1di	B f=20Hz - 50kHz		
Channel Separation	half power	RL=8 ohms		~ 704	D 11-11-		
	Att.max	input 600 ohms shunt	nt ≥ 70dB 1kHz				
Residual noise Att. min		DIN AUDIO		≥ 7	OdB		
S/N ratio		DIN AUDIO	104dB	103dB	102dB	100dB	
Voltage Gain		Att.max		32.	1dB		
Input Impedance				30KΩ (balance) 1	15kΩ (unbalance)		
Connectors		Input		XLR-3-31 type/o	h, 1/4""TRS/ch"		
		Output		SPEAKON, 5way biding	g post, 1/4""phone/ch"		
Limiter Circuit				comp. :TH	HD ≥ 0.5%		
Cooling			Dual variable-speed fan		Single variable-speed fan		
Power Requirements			20V 60Hz, 230V 50Hz, 240V 50Hz				
Idle Power Consumption			35W	35W	30W	25W	
Maximum Power Consumption	on (4 ohms)		4000W	3000W	2000W	1600W	
Dimensions (W x H x D): mm				480 x 8	88 x 456		
Weight			12kg	12kg	15kg	14kg	

Product Line Up— SR Speakers

Club V Loudspeakers S-Series & C-Series



f you're tired of breaking your back for mediocre speaker performance, it's time to look at the Yamaha Club V S-Series. Although they easily load into and out of your compact sedan, they also deliver power and performance that makes them ideal for up to mid-sized sound reinforcement systems. And in the world of the touring PA, there's virtually no way you can cart gear around without bumping into the occasional immovable object, so you'll appreciate the durable carpet covering – which also protects the interior of your vehicle from the speakers. Heavy-gauge steel grilles and steel protectors help protect your investment, too. In the Club V S-Series the refinements of generation V are taken to the next level, with larger enclosures for improved low-frequency performance, improved drivers for higher power handling, re-designed crossovers, stronger grilles, and dual Speakon® and 1/4" connectors

Superior Sound on the Ground or Overhead



The Club V C-Series includes two "VA" flyable models that are ideal for installations. In fact, the entire series features foam-backed full-face perforated steel grilles and a sprayed finish that makes for elegant yet unobtrusive installations. Performance-wise, the C-Series speakers offer the same specs as the S-Series, with large enclosures for improved low-frequency performance, improved drivers for higher power handling, re-designed crossovers, stronger grilles, and dual Speakon® and 1/4" connectors

Specifications *S-series are carpet finish, C-series are sprayed finish

Club V S- & C- series

	S112V, C112V	S115V, C115V	S215V, C215V	SM10V, CM10V	SM12V, CM12V	SM15V, CM15V
Туре	12" 2 way bass reflex	15" 2 way bass reflex	15" x 2 2 way bass reflex	10" 2 way bass reflex	12" 2 way bass reflex	15" 2 way bass reflex
Frequency Range	60Hz - 16kHz (-10dB)	55Hz – 16kHz (-10dB)	42Hz - 16kHz (-10dB)	70Hz – 20kHz (-10dB)	60Hz - 16kHz (-10dB)	55Hz – 16kHz (-10dB)
Power Capacity	175 watts (NOISE)* 350 watts (PGM) 700 watts (MAX)	250 watts (NOISE)* 500 watts (PGM) 1000 watts (MAX)	500 watts (NOISE)* 1000 watts (PGM) 2000 watts (MAX)	125 watts (NOISE)* 250 watts (PGM) 500 watts (MAX)	175 watts (NOISE)* 350 watts (PGM) 700 watts (MAX)	250 watts (NOISE)* 500 watts (PGM) 1000 watts (MAX)
Nominal Impedance	8 ohms	8 ohms	4 ohms	8 ohms	8 ohms	8 ohms
Sensitivity	97dB	99dB	99dB	96dB	97dB	99dB
LF Driver	12" cone	15" cone	15" cone x 2	10" cone	12" cone	15" cone
HF Driver	2" vc, Titanium Horn	2" vc, Titanium Horn	2" vc, Titanium Horn	1" vc, Titanium Horn	2" vc, Titanium Horn	2" vc, Titanium Horn
Crossover Frequency	2kHz	1.7kHz	1.5kHz	1.8kHz	2kHz	1.7kHz
Finish	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed
Input Connectors	1/4" Phone x 2, Speakon x 2	1/4" Phone x 2, Speakon x 2	1/4" Phone x 2, Speakon x 2	1/4" Phone x 2, Speakon x 2	1/4" Phone x 2, Speakon x 2	1/4" Phone x 2, Speakon x 2
Dimensions (W x H x D mm)	S: 420 x 632 x 333 C: 416 x 628 x 329	S: 489 x 719 x 377 C: 485 x 715 x 373	S: 495 x 1167 x 597 C: 491 x 1163 x 593	S: 560 x 353 x 277 C: 556 x 349 x 273	S: 632 x 414 x 351 C: 628 x 410 x 339	S: 719 x 483 x 343 C: 715 x 479 x 339
Net Weight	S: 20.8 kg, C:21.3 kg	S: 29.4 kg, C: 30.3 kg	S: 47.2 kg, C: 47.5 kg	S: 13.4 kg, C: 13.3 kg	S: 21.4 kg, C: 21.8 kg	S: 28.0 kg, C: 28.8 kg

Club V S- & C- series

	SW115V, CW115V	SW118V, CW118V	SW218V, CW218V	C112VA	C115VA
Туре	15" Bass reflex	18" bass reflex	18" x 2 Bass reflex	12" 2 way bass reflex	15" bass reflex
Frequency Range	35Hz – 2kHz (-10dB)	30Hz – 2kHz (-10dB)	30Hz – 2kHz (-10dB)	60Hz – 16kHz (-10dB)	55Hz – 16kHz (-10dB)
Power Capacity	250 watts (NOISE)* 500 watts (PGM) 1000 watts (MAX)	300 watts (NOISE)* 600 watts (PGM) 1200 watts (MAX)	600 watts (NOISE)* 1200 watts (PGM) 2400 watts (MAX)	175 watts (NOISE)* 350 watts (PGM) 700 watts (MAX)	250 watts (NOISE)* 500 watts (PGM) 1000 watts (MAX)
Nominal Impedance	8 ohms	8 ohms	4 ohms	8 ohms	8 ohms
Sensitivity	95dB	96dB	98dB	97dB	99dB
LF Driver	15" cone	18" cone	18" cone x 2	12" cone	15" cone
HF Driver	-	-	-	2" vc, Titanium Horn	2" vc, Titanium Horn
Crossover Frequency	90Hz, 12dB/Oct.	90Hz, 12dB/Oct.	90Hz, 12dB/Oct.	2kHz	1.7kHz
Finish	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	S: Carpet, C: Sprayed	Sprayed	Sprayed
Input Connectors	1/4" Phone x 2, Speakon x2	1/4" Phone x 2, Speakon x2	1/4" Phone x 2, Speakon x2	Barrier strip	Barrier strip
Dimensions (W x H x D mm)	S: 506 x 611 x 532 C: 500 x 607 x 528	S: 610 x 728 x 641 C: 605 x 720 x 637	S: 1221 x 578 x 659 C: 1217 x 574 x 655	416 x 620 x 329	485 x 715 x 373
Net Weight	S: 28.2 kg, C: 28.0 kg	S: 39.0 kg, C: 37.2 kg	S: 65.4 kg, C: 64.7 kg	21.8 kg	29.9 kg

BR-Series Speakers



Great Sound and Easy handling

Whether you're playing to a rock'n roll, jazz, or classic crowd, or delivering an important spoken message, Yamaha BR-Series speakers will ensure that you're heard clearly. With the right powered mixer, or standard mixer and power amplifiers, these units can pack a heck of a wallop... make that a "high-quality wallop." But when the show is done and it's time to tear down the system and go home, you'll appreciate these speakers a second time—they're compact, remarkably light for their power and performance, and are designed for easy handling. All models, even the monitors, offer integral pole sockets for easy stand mounting.

S pecifications

BR-Series

	BR10	BR12	BR15	BR12M	BR15M
Туре	10" 2 way bass reflex	12" 2 way bass reflex	15" 2 way bass reflex	12" 2 way bass reflex	15" 2 way bass reflex
Frequency Range	65Hz – 20kHz (-10dB)	60Hz - 20kHz (-10dB)			
Power Capacity	125 watts (NOISE)* 250 watts (PGM) 500 watts (MAX)	150 watts (NOISE)* 300 watts (PGM) 600 watts (MAX)	200 watts (NOISE)* 400 watts (PGM) 800 watts (MAX)	150 watts (NOISE)* 300 watts (PGM) 600 watts (MAX)	200 watts (NOISE)* 400 watts (PGM) 800 watts (MAX)
Nominal Impedance	8 ohms				
Sensitivity	96dB	97dB	98dB	97dB	98dB
LF Driver	10" cone	12" cone	15" cone	12" cone	15" cone
HF Driver	1" vc, Titanium Horn				
Finish	Carpet	Carpet	Carpet	Carpet	Carpet
Input Connectors	1/4" Phone x 2				
Dimensions (W x H x D mm)	375 x 537 x 326	403 x 569 x 335	485 x 656 x 365	569 x 407 x 334	652 x 487 x 365
Net Weight	14.4 kg	16 kg	21.3 kg	15.6 kg	20.9 kg

4 Simple Steps To Better Sound

Why use monitor speakers?

While the purpose of the main or "FOH" (Front Of House) speakers is to deliver the sound to the audience, monitor speakers are necessary to provide a sound reference to the speakers or musicians performing on stage. This was sometimes also known as "foldback". A monitor system allows performers to clearly hear what they are saying, singing, or playing so that they can perform with confidence and provide performances of the highest quality. It may not be an exaggeration to say that a good monitor system is the key to a successful concert or event.



MSR Series Speakers POWERED SPEAKERS LEEFINGE MSR400 / 800W





owered speakers offer many advantages over their passive counterparts, whether used on the road or in permanent installations. You don't need separate amplifiers or the extra cables and connections required to a series of the extra cables and connections are cables and the extra cables are cables are cables and the extra cables are cables and the extra or the extra cables and connections required, so setup is fast and easy while reliability is significantly improved. The ability to plug directly into the speaker inputs can eliminate an entire rack of amplifiers in a modest-size system. But the benefits go beyond mere convenience. Optimum matching of amplifiers and speakers involves much more than simple impedance figures, and can be a formidable engineering task given the vast number of choices available. The Yamaha MSR-series power amplifiers have been ideally mated to their respective speakers in every way possible, delivering reproduction quality that only the most stringent matching can achieve. There are also a significant number of Yamaha refinements that put these extraordinary powered speakers in a class of their own, including an original cabinet design that eliminates internal standing waves*1 and ferrofluid-cooled voice coils for superior stability and linearity*2.

The MSR250, MSR400 or MSR100 powered speakers can be used alone or in combination with the MSR800W powered subwoofer to create sound reinforcement systems from modest to massive that will sound superb while delivering the power and coverage you need for success in just about any venue or at any event. They look great too, and can be used as-is in many permanent installations without requiring any cosmetic cover-up. They're more than rugged enough for demanding tour applications as well.

*1 MSR250 and MSR400 *2 MSR100 and MSR250

Portable, Compact **Solutions for Superior Sound and Convenience**







MSR800W

S pecifications

MSR100

	MSR100	MSR250	MSR400	MSR800W
Туре	2-way Powered Speaker	2-way Powered Speaker	Biamp. 2-way Powered Speaker	Powered Subwoofer
Frequency Range	55Hz – 20 kHz (-10 dB)	55Hz – 20kHz (-10dB)	50Hz – 20kHz (-10dB)	40Hz – 120Hz
Maximum Output Level (SPL)	112dB (1 m on Axis)	116dB (1m, on axis)	121dB (1m)	122dB (1m)
LF Driver	8" Cone	10" Cone	12" Cone	15" Cone
HF Driver	1" V.C. Compression Driver	1" V.C. Compression Driver	1.75" V.C. Compression Driver	_
Maximum Output Power	100 W at 1 kHz. THD=1 %, RL=6 Ω	200 W at 1 kHz. THD=1 %, RL=4 Ω	Biamplified system LF: 225 W/4 Ω HF: 75 W/16 Ω Maximum Power: 400 W Total	500 W at 100 Hz. THD=1 %, RL=8 Ω Maximum Power: 800 W
Input sensitivity	INPUT 1: -50 dB (Mic)/+4 dB (Line), $10 \text{ k}\Omega$ (Line), (with select SW) INPUT 2 & 3: -10 dB, $10 \text{ k}\Omega$ (Line)	INPUT 1 (XLR): -60 dBu (Mic)/-30 dBu (Line), (PHONE): -50 dBu (Mic)/-20 dBu (Line), INPUT 2/3: -20 dBu	Mic/Line: -36 dB/+4 dB, 10 kΩ	+4dB/30 kΩ
Controls	Level Control INPUT 1, 2 & 3, Master Level, EQ. LOW: ± 3 dB at 60 Hz, HIGH: ± 6 dB at 10 kHz, POWER Switch (ON/OFF)	LEVEL CONTROL: INPUT 1, 2/3, EQ. LOW: ± 3 dB at 60 Hz, HIGH: ± 6 dB at 10 kHz, POWER Switch (ON/OFF)	LEVEL CONTROL, EQ. LOW: ± 3 dB at 55 Hz, HIGH: ± 3 dB at 1.6 kHz, POWER Switch (ON/OFF)	MASTER LEVEL CONTROL, CUTOFF FREQ. Control: 80 – 100 Hz (Variable), PHASE Switch: (NORM/REV), POWER Switch: ON/OFF
Connectors	INPUT 1: XLR-3-31 (balanced) INPUT 2 & 3: Phone (unbalanced) LINK OUT: Phone (unbalanced)	INPUT 1: XLR-3-31/PHONE (balanced) INPUT 2/3: PHONE/RCA-PIN (unbalanced) LINK OUT: PHONE	1: XLR-3-31 (balanced) 2: XLR-3-32 (balanced) 3: PHONE (balanced) (All parallel)	Input (ch-A & B): XLR-3-31 (balanced) THRU Out (ch-A & B): XLR-3-32 (balanced) High Pass Out (ch-A & B): XLR-3-32 (balanced) (100 Hz, 18 dB/oct, 44 dB)
Power Requirement	AC120V/60Hz, AC230V/50Hz, AC240V/50Hz	AC100V, 110V, 120V, 220V, 230V or 240V; 50/60Hz	AC120V 60Hz, AC230V 50Hz, AC240V 50Hz, AC220V 50Hz/60Hz, AC110V 60Hz	AC120V 60Hz, AC230V 50Hz, AC240V 50Hz, AC220V 50Hz
Power Consumption	70 W	40W	110W	200W
Finish	Polypropylene Black	Polypropylene Black	Polypropylene Black	Birch Plywood, Black sprayed
Dimensions (W x H x D mm)	275 x 455.5 x 255	342 x 544.5 x 298	406 x 667 x 351	600 x 521 x 590
Net Weight	11 kg	14.1 kg	23 kg	45 kg

5 Simple Steps To Better Sound

Boosting system power

If the people in the back row aren't hearing the performance with the intended impact, you can add speakers to boost the system's output. The first thing to try would be adding an extra set of speakers parallel-connected with the main front speakers. In this case, it is important to add the same type of speakers as the originals. If you use speakers with different

sensitivity, only the louder speakers will be heard. Another option is to add a subwoofer to beef up the low end. By reinforcing the lows you effectively boost overall output. This is a good strategy for improving the sound in outdoor setups. If you're using Yamaha Club Series speakers, you can easily add a subwoofer adding an electronic crossover, or use the filters provided in the P-Series amplifiers.

MS101III

POWERED MONITOR SPEAKER SYSTEM

N othing beats these small powered monitors for performance and utility. The 10-watt MS101III is an equally fine choice for your desktop studio or point-of-sale sound in a retail outlet. In the former application, the superior sound of these remarkably accurate monitors can help you make better mixes, while in the latter you can broadcast line-fed program material, or plug a microphone directly into the front-panel mic jack when you want to grab your customers' attention. MS101III can be wall- or ceiling- mounted, or conveniently mounted on a microphone stand.

Ideal For Monitoring or Point-Of-Sale Sound



MS101III

S pecifications MS101II Frequency Range 75Hz to 18kH Mic: -45dB Line1 2:-10dB VOLUME, Tone (High, Low), Power switch Line Inputs RCA pin-jack x 1 (rear panel), Phone jack x 1 (front panel), Microphone Input Phone jack x 1 (front panel), Line Output Phone jack x 1 (front panel) Power Requiremen 120V 60Hz, 230/240V 50/60Hz nsions (W x H x D mm 147 x 214 x 19 Net Weight 2 2 kg

POCKETRAK Superior Sound Quality from the

POCKET RECORDERS

@YAMAHA

Amazing Combination of Quality and Convenience

ooking for the best way to record your music lessons, practices, band rehearsals? You've found it. The C24 delivers 24bit/96kHz sound quality in the world's smallest size, and is packed with helpful features.

POCKETRAK C24 Features

- World's smallest and lightest recorder (2oz./57g with battery).
- 24bit/96kHz recording capability.
- 26-hour MP3 recording time (PCM: 16 hours).
- High capacity 2GB internal memory
- Attachment clip included

*With 24bit/96kHz capability. Based on Yamaha data

W24 PAUSE ESC SPEED REC STOP PLAY

WZ4

Large Stereo Mic and Ultra-handy Wireless Remote Control

mompact and lightweight, the W24 features a large dual microphone assembly for outstanding stereo sound quality. What's more, recording is more convenient than ever thanks to the supplied wireless remote control. It's the perfect choice for effortless recording in the studio, on the stage or out in the field.

POCKETRAK W24 Features

• Extremely sensitive, top quality X-Y mic

Company that Knows Music Best

- 24bit/96kHz recording capability.
- Pocket size, weighs only 3.25oz./92grams
- High quality peak limiter built in.
- 38-hour PCM recording time (MP3: 56 hours).
- Wireless Remote Control includ



S pecifications

	POCKETRAK C24	POCKETRAK W24		
Memory Capacity	2 GB (External memory: microSD/microSDHC)	2 GB (External memory: microSD/microSDHC)		
Battery Available Battery	AAA alkaline battery x 1	AA alkaline battery x 1		
Recording Battery Life	AAA Alkaline : MP3: 26 h, PCM: 16 h	AA Alkaline : MP3: 56 h, PCM: 38 h		
Playback Battery Life	AAA Alkaline: MP3: 34 h, PCM: 18 h (Earphone)	AA Alkaline: MP3: 70 h, PCM: 45 h (Earphone)		
LCD Type	Dot matrix, Backlit display	Dot matrix, Backlit display		
Others	Internal Stereo Microphone, Internal Speaker, Headphone Jack (Stereo mini: ø3.5mm x 1), Ext. Stereo Mic In Jack (Stereo mini: ø3.5mm, Line In Jack switchable x 1), microSD card slot x 1	Internal Stereo Microphone, Internal Speaker, Headphone Jack (Stereo mini: ø3.5mm x 1), Ex Stereo Mic In Jack (Stereo mini: ø3.5mm, Line In Jack switchable x 1), microSD card slot x 1		
Recording Format	Linear PCM, MP3	Linear PCM, MP3		
Recording Time (using 2GB microSD)/	PCM stereo (96 kHz) 55m, 24 bit	PCM stereo (96 kHz) 55m, 24 bit		
Frequency Response/	PCM stereo (44.1 kHz) 3h 00m, 16 bit	PCM stereo (44.1 kHz) 3h 00m, 16 bit		
Sampling Frequency/	MP3 stereo (320 kbps) 13h 30m	MP3 stereo (320 kbps) 13h 30m		
Bit Rate	MP3 stereo (128 kbps) 34h 00m	MP3 stereo (128 kbps) 34h 00m		
	MP3 stereo (64 kbps) 68h 00m	MP3 stereo (64 kbps) 68h 00m		
Recording Other Functions	Voice Automatic Recording System, ALC (Auto Level Control) ON/OFF, Mic Sensitivity Select (ALC ON: 2 steps, ALC OFF: 80 steps), Recording Peak Limiter, 5-band Graphic Equalizer (8 presets/User), High Pass Filter, Timer rec, Self Timer, Rec monitor, File delete confirmation	Voice Automatic Recording System, ALC (Auto Level Control) ON/OFF, Mic Sensitivity Select (ALC ON: 2 steps, ALC OFF: 80 steps), Recording Peak Limiter, 5-band Graphic Equalizer (8 presets/User), High Pass Filter, Timer rec, Self Timer, Rec monitor, File delete confirmation, Scene Memory		
Playback Format	MP3, WMA, WAV (WAV file playback is own recording file only)	MP3, WMA, WAV (WAV file playback is own recording file only)		
Musical Function	5-band EQ (Flat, Bass1, Bass2, Pop, Rock, Jazz, User)	5-band EQ (Presets: Flat, Bass1, Bass2, Pop, Rock, Jazz, User)		
Playback Speed (Not available on PCM)	Slow PB (50 - 100 %), Normal (100 %), Fast PB (100 - 200 %)	Slow PB (50 - 100 %), Normal (100 %), Fast PB (100 - 200 %)		
Playback Other Functions	Play list Playback, Skip Search, File edit (divide, fade in/ out), Tuner, Metronome	Play list Playback, Skip Search, File edit (divide, fade in/ out), Tuner, Metronome		
Dimensions (W x H x D mm)	37.6 x 113.0 x 21.1	46.5 x 129.5 x 17.5		
Weight	57 g including battery	92 g including battery		
Accessories	Alkaline AAA battery, Attachment clip, Cubase AI5 DVD-ROM, Owner's Manual	Alkaline AA battery, Wireless remote controller, Windscreen, Microphone stand adaptor, US		
		cable, Cubase AI5 DVD-ROM, Owner's Manual		

[•] Continuous recording time will depend on the type of battery used as well as the battery's manufacturer, the conditions under which the battery is being used, and the condition under which it has been stored. Proper operation cannot be guaranteed when a battery other than the supplied eneloop battery or an alkaline battery from a reliable manufacturer is used.

STAGEPAS series with



mbining high-performance power and accurate sound reproduction with remarkable versatility in a portable package, Yamaha's STAGEPAS Series Portable PA Systems are fast becoming the standard for musicians and other PA users on the go.

These combination mixer/amplifier/speaker systems are exceptionally lightweight, portable and full-featured. Since virtually everything you need is in one, easy-to-use package, you can be set up and playing within a matter of minutes.

The series now includes the new single-speaker STAGEPAS 150M and 250M, which can be expanded to a stereo system by simply adding a second speaker. They can also function either as an ultra-portable PA system, or as a powerful keyboard amplifier with sound quality and portability that far surpasses any other conventional keyboard amp. No matter what system you choose, you have a high-power, high performance sound system ready for a wide variety of venues and events, both indoors and out.

Built-in Powered Mixer

Built-in mixer provides easy operation and convenient portability. Can be used inside the

Optional BMS-10A for Convenient Mic Stand Mounting

The detachable mixer can be conveniently mounted on a conventional microphone stand for easy access, using the optional BMS-10A Mic Stand





Versatility and Power —in Four Portable Systems



STAGEPAS 250M

STAGEPAS 150M

S pecifications

Mixer

IVIIXEI				
	STAGEPAS 500	STAGEPAS 300	STAGEPAS 250M	STAGEPAS 150M
Maximum Output Power	250 W + 250 W (\pm 10 %) 4 Ω @ 10 % THD at 1 kHz (SPEAKER L/R)	150 W+150 W (±10 %) 6 ± @10 % THD at 1 kHz (SPEAKER L/R)	250 W (±10 %) 4 ± @ 10 %THD at 1 kHz (SPEAKER)	150 W (±10 %) 6± @ 10 % THD at 1 kHz (SPEAKER)
Output Power (RMS)	200 W + 200 W 4Ω @ 1 % THD at 1 kHz (SPEAKER L/R)	100 W+100 W 6 Ω @1 % THD at 1 kHz (SPEAKER L/R)	200 W 4 Ω @ 1 % THD at 1 kHz (SPEAKER)	100 W 4 Ω@ 1 % THD at 1 kHz (SPEAKER)
Maximum Output Level	116 dB (1 m)	112 dB (1 m)	116 dB (1 m)	112 dB (1 m)
Frequency Response (Nomimal output level @ 1 kHz)			ut (MUSIC/SPEECH=MUSIC, without Speaker EQ) () (STAGEPAS 250M/150M: SPEAKER)	
Total Harmonic Distortion			kHz +14 dBu GAIN= nominal STAGEPAS 250M/150M: ST SUB OUT, ST LINK OUT)	
Hum & Noise (Equivalent Input Noise, Rs = 150 Ω , MIC/LINE = MIC)	≤ -106 dBu (CH1/2) ≤ -112 dBu (CH3/4) ≤ -65 dBu Residual output noise (SPEAKER L/R)	≤ -65 dBu Residual output noise (SPEAKER L/R)	≤ -106 dBu (CH1/2) ≤ -65 dBu Residual output noise (SPEAKER)	≤ -112 dBu (CH3/4) ≤ -65 dBu Residual output noise (SPEAKER)
Crosstalk (1kHz)		±-70dB between	en input channels	
Power Consumption	65 W	70 W	35 W	20 W
Weight	24 kg (52-7/8" lbs.) (Speaker x 2 + Powered Mixer)	18 kg (39-5/8" lbs.) (Speaker x 2 + Powered Mixer)	13 kg (28-5/8" lbs.) (Speaker + Powered Mixer)	9.6 kg (21-1/8" lbs.) (Speaker + Powered Mixer
Input Channel Equalization		±15 dB, HIGH 10 kHz she	lving, LOW 100 Hz shelving	

Speaker

	STAGEPAS 500	STAGEPAS 300	STAGEPAS 250M	STAGEPAS 150M		
Enclosure	2-way bass-reflex type, Polypropylene, Black					
Crossover Frequency	4.0 kHz (LF: 12 dB/oct, HF: 12 dB/oct)					
Frequency Range	55 Hz-20 kHz (-10 dB)					
Speaker Unit	LF: 10" (25 cm) Cone HF: 1" (2.54 cm) Compression Driver	LF: 8" (20 cm) Cone HF: 1" (2.54 cm) Compression driver	LF: 10" (25 cm) Cone HF: 1" (2.54 cm) Compression Driver	LF: 8" (20 cm) Cone HF: 1" (2.54 cm) Compression Driver		

Multi-purpose Portable PA

Whether your Portable PA needs are Pro Audio (for music performance) or Public Address for school, church and meeting assemblies, the STAGEPAS systems are lightweight, compact and simple to setup—and fit a wide variety of applications.



For Bands

Multiple microphone inputs make the STAGEPAS series an ideal choice for band rehearsals and performances. The STAGEPAS 500 in particular is well suited for many sources and larger venues, while the optional MSR100 powered speakers can be used for



For Singers

Vocalists will appreciate the built-in, highquality reverb on all STAGEPAS systems. Plus, the 250M and 500 models feature a built-in limiter/compressor to let you dial in a smooth, punchy vocal sound.



For Instrumentalists

The versatile STAGEPAS mixer has more than enough input capability to directly handle many different instruments—from electricacoustic guitars to electronic keyboards.



For Keyboard Players

Players needing a simple, yet exceptionally high-quality keyboard amplifier will find ideal solutions in the STAGEPAS 150M and 250M. These compact, highly portable and easy-touse systems can serve as your main system for small gigs, or as a sub-mixer for direct connection to the main mixer at large events.



For Guitarists

The STAGEPAS 150M and 250M are excellent systems for guitarists playing in small venues, since they deliver outstanding sound for the house as well as great monitor sound for the



For Public Speakers

If you're a speaker or giving presentations, you can instantly optimize the system settings with the special Speech mode—giving you maximum clarity for speech applications. Naturally, you can mix background music from a CD player or other source with the mic



MSP STUDIO series speakers

POWERED MONITOR SPEAKER

Y amaha's "STUDIO" series monitors have been designed without compromise for serious monitoring. Years of experience and development have been applied to achieve reference-quality reproduction precision that lets you hear sonic details, rather than flattering sound. These studio-class speakers carry on in the tradition of the venerable NS10M STUDIO, which was the definitive near-field monitor in an overwhelming majority of professional studios throughout the world for many years from the 80s onward. But technology has evolved dramatically right throughout the audio chain, and speakers must follow suit. The new top-of-the-line MSP7 STUDIO Powered Monitor Speaker is capable of delivering consistent quality and performance that you can rely on in modern production environments that handle any combination of digital and analog sources as well as stereo and surround formats, while the more compact dimensions of the MSP5 STUDIO make it an ideal choice for smaller project studios and DAW-based production systems. The SW10 STUDIO subwoofer has been designed specifically for optimum matching with the MSP series speakers, and when added to either a stereo or surround system it can provide seamless extended low end for accurate ultra-wide-range monitoring.





Refined Monitoring Precision

S pecifications

		MSP5 STUDIO	MSP7 STUDIO	SW10 STUDIO
General Specifications	Туре	Biamp 2-way Powered speaker	Biamp 2-way Powered Speaker	Powered Subwoofer
	Crossover Frequency	2.5kHz, LF: 24dB/oct, HF: 24dB/oct	2.5kHz, LF: 30dB/oct, HF: 30dB/oct	-
	Overall Frequency Response	50Hz – 40kHz (-10dB)	45Hz – 40kHz (-10dB)	25Hz – 150Hz (-10dB)
	Dimensions (W x H x D mm)	179 x 208 x 279	218 x 235 x 330	328 x 476 x 459
	Weight	7.9 kg	12.2 kg	26.5 kg
Speaker Components	Speaker Components	LF: 5" cone, HF: 1.0" Titanium dome	LF: 6.5" cone, HF: 1.0" Titanium dome	LF: 10" cone
	Enclosure Type	Bass-Reflex Type	Bass-Reflex Type	Bass-Reflex Type
	Material	PP	PP	MDF
	Magnetic shielding	Yes	Yes	Yes
Amp. Unit	Output Power	LF: 40W, THD = 0.02 %, RL = 4 Ω , HF: 27W, THD = 0.02 %, RL = 6 Ω	LF: 80W, THD = 0.05 %, RL = 4Ω , HF: 50W, THD = 0.05 %, RL = 6Ω	LF: 180W, f = 100Hz, THD = 1 %, RL = 8Ω
	Input Sensitivity	XLR-3-31: +4dBu, LEVEL = Center, -6dBu, LEVEL = Max PHONE: -10dBu, LEVEL = Center, -20dBu, LEVEL = Max	XLR-3-31: +4dBu, LEVEL = Center, -6dBu, PHONE: -10dBu, LEVEL = Center, -20dBu,	XLR-3-31: +4dBu, LEVEL = Center, -6dBu, PHONE: -10dBu, LEVEL = Center, -20dBu,
	Input Impedance	10kΩ	10 kΩ	10 kΩ
	Input Connectors	1: XLR-3-31 type (balanced) 2: PHONE (unbalanced)	XLR-3-31 type (balanced)	XLR-3-31 type x 3 (balanced)
	Output Connectors	-	-	XLR-3-32 type x 3 (balanced) (parallel)
	Controls	LEVEL control: 31 Positions Detent type VR (Min = →∞ Attenuation), LOW TRIM: +1.5/0/-1.5/-3 dB at 60 Hz, HIGH TRIM: +1.5/0/-1.5 dB at 15 kHz,	LEVEL control: 31 Positions Detent type VR (Min = -∞ Attenuation), LOW CUT switch: FLAT/80/100Hz (12dB/oct), LOW TRIM: +1.5/0/-1.5/-3dB at 45Hz, HIGH TRIM: +1.5/0/-1.5dB at 15kHz,	LEVEL control: Center Click VR (Min = -∞ Attenuation) HIGH CUT control: 40 – 120Hz, 80Hz at Center Click, PHASE switch: NORM./REV.
	Power Consumption	60W	100W	160W

HS series speakers **POWERED MONITOR SPEAKER**

When choosing reference monitors for mixing and music production, what you really need is an honest reference for your mix rather than sound that has been tweaked or colored to sound impressive at the expense of accuracy. Unlike speakers that have exaggerated bass and treble that make a good first impression but can't be relied on for accuracy, Yamaha HS series reference monitors have been painstakingly crafted by our studio monitor engineering team to deliver exceptionally flat, accurate response that you can trust. The HS series speakers are true studio reference monitors in the tradition of the legendary Yamaha NS10M. Whether you're mixing for stereo or 5.1 surround, mixes that sound good on Yamaha HS series reference monitors will translate accurately to the widest possible range of reproduction systems... which is engineer-speak that means they'll sound good on anything. And that is the ultimate goal of any reference monitor. We should also mention that the HS-series monitors not only sound great, they look great, too.

High-performance Speakers and Mounting System







Specifications

		HS50M	HS80M	HS10W
General Specifications	Туре	Biamp 2-way Powered speaker	Biamp 2-way Powered speaker	Powered Subwoofer
	Crossover Frequency	3kHz	2kHz	-
	Overall Frequency Response	55Hz – 20kHz (-10dB)	42Hz – 20kHz (-10dB)	30Hz - 180Hz (-10dB)
	Dimensions (W x H x D mm)	165 x 268 x 222	250 x 390 x 332	300 x 350 x 386
	Weight	5.8 kg	11.3 kg	12.5 kg
Speaker Components	Speaker Components	LF: 5" cone, HF: 0.75" Dome	LF: 8" cone, HF: 1" Dome	LF: 8" cone
	Enclosure Type	Bass-Reflex Type	Bass-Reflex Type	Bass-Reflex Type
	Material	MDF	MDF	MDF
	Magnetic shielding	Yes	Yes	Yes
Amp. Unit	Output Power	LF: 45W, 4Ω, HF: 25W, 8Ω	LF: 75W, 4Ω, HF: 45W, 8Ω	LF: 150W, 4Ω
	Input Sensitivity	XLR-3-31: -10 dBu, PHONE: (parallel)	XLR-3-31: -10dBu, PHONE: (parallel)	XLR-3-31: -10dBu, PHONE: (parallel)
	Input Impedance	10kΩ	10kΩ	10kΩ
	Input Connectors	1: XLR-3-31 type (balanced) 2: PHONE (balanced) (parallel)	1: XLR-3-31 type (balanced) 2: PHONE (balanced) (parallel)	1: XLR-3-31 type (balanced) 2: PHONE (balanced) (parallel)
	Output Connectors	-	-	1: XLR-3-32 type (balanced) x 1 (EXT SUB) 2: XLR-3-32 type (balanced) x 2 (L&R)
	Controls	LEVEL control: +4dB, center click, LOW CUT switch: FLAT/80/100Hz, 12dB/octave, HIGH TRIM: +2/0/-2dB at 3kHz (HF), EQ: MID: +2/0/- 2dB at 2kHz, ROOM CONTROL: 0/-2/-4dB under 500Hz	LEVEL control: +4dB, center click, LOW CUT switch: FLAT/80/100Hz, 12dB/octave, HIGH TRIM: +270/-2dB at 3kHz (HF), EQ: MID: +270/- 2dB at 2kHz, ROOM CONTROL: 07-2f-4 dB under 500Hz	LEVEL control: VR, LOW CUT switch: ON/OFF, LOW CUT control: 80 – 120Hz at Center Click, HIGH CUT control: 80 – 120Hz at Center Click, PHASE switch: NORM./REV.
	Power Consumption	45 W	60W	70W

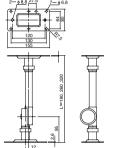
Speaker Brackets

Wall Brackets

For MSR100/STAGEPAS300

- BWS50-190
- BWS50-260



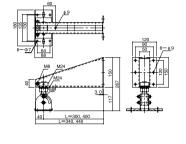


Wall Brackets

For MSR400/STAGEPAS500

- BWS251-300
- BWS251-400



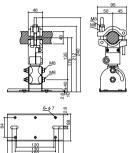


Baton Brackets

For MSR100/MSR400/ STAGEPAS300/STAGEPAS500





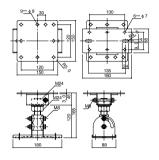


Ceiling Brackets

For MSR100/MSR400/ STAGEPAS300/STAGEPAS500

• BCS251





Rack Mount Adaptors

RK512









Mic Stand Adaptor



For MG82CX/MG102C/ STAGEPAS 300 (Mixer only) STAGEPAS 500 (Mixer only) STAGEPAS 150M (Mixer only) STAGEPAS 250M (Mixer only)

Foot Switch

Headphones RH-5MA

Not all peripherals are available in all areas. Contact the local Yamaha subsidiary for availability

Mini-YGDAI Compatible Cards For 01V96VCM

Mini-YGDA	AI Card Name	Function	Input	Output
5000	MY4-AD	Analog In	4 In	_
2777	MY8-AD24	Analog In	8 In	_
	MY8-AD96	Analog In	8 In	_
	MY4-DA	Analog Out	_	4 Out
	MY8-DA96	Analog Out	_	8 Out
	MY8-ADDA96	Analog In/Out	8 In	8 Out
A CONTRACTOR OF THE PARTY OF TH	MY8-AE	AES/EBU	8 In	8 Out
	MY8-AE96	AES/EBU	8 In	8 Out

Mini-YGDA	AI Card Name	Function	Input	Output	
	MY8-AE96S	AES/EBU	8 In	8 Out	
a second	MY16-AE	AES/EBU	16 In	16Out	
EIII)	MY8-AT	ADAT	8 In	8 Out	
	MY16-AT	ADAT	16 In	16 Out	
S. T. S.	MY8-TD	TASCAM	8 In	8 Out	
	MY16-TD	TASCAM	16 In	16 Out	
	MY16-CII	CobraNet	16 In	16 Out	
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For more options and information, please visit our website at http://www.yamahaproaudio.com



For details please contact:

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