



M 1 A C T I V E M k 2

REFERENCE MANUAL

ALESIS

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™

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Alesis M1 Active Mk2 Reference Manual

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ALESIS CONTACT INFORMATION

Alesis Studio Electronics, Inc.
www.alesis.com

Do not use this address for repair service.

E-Mail: info@alesis.com
Website: <http://www.alesis.com>

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INTRODUCTION

Thank you for purchasing the Alesis M1 Active Mk2 Bi-amplified Reference Monitors. To take full advantage of the M1 Active Mk2's operation, and to enjoy long and trouble-free use, please read this user's manual carefully. We value any comments you may have about this monitor system, this manual, your Alesis dealer or our factory service. Please take a minute now to fill out your warranty card and tell us what you think.

HOW TO USE THIS MANUAL

This manual is divided into the following sections describing the various features of the M1 Active Mk2. Though we recommend you take time to read through the entire manual once carefully, those having general knowledge about monitors should use the table of contents to reference specific functions.

Chapter 1: About the M1 Active Mk2. Engineering specifications and reasons why nearfield monitors have become so popular.

Chapter 2: Speaker Installation. This chapter explains how to connect the M1 Active Mk2s to a mixer or other line-level source and discusses proper speaker placement for stereo operation.

Chapter 3: Surround Sound. If you are using the M1 Active Mk2 speakers in a multichannel surround sound setup, you'll find helpful setup and operation information here.

Chapter 4: Troubleshooting. This chapter contains troubleshooting tips and service information should problems occur.



When something important appears in the manual, an icon (like the one on the left) will appear in the left margin. This symbol indicates that this information is vital when operating the M1 Active Mk2s.

ABOUT THE M1 ACTIVE Mk2

UNPACKING AND INSPECTION

Your M1 Active Mk2 Biamplified Reference Monitors were packed carefully at the factory, and the shipping carton was designed to protect the speakers during shipping. Please retain this container in the highly unlikely event that you need to return the speakers for servicing.

The shipping carton should contain the following items:

- This instruction manual
- M1 Actives with the same serial numbers as shown on the shipping cartons
- Two AC power cords (NEMA to CEE type)
- Alesis warranty card



It is important to register your purchase; if you have not already filled out your warranty card and mailed it back to Alesis, please take the time to do so now.

The M1 Active Mk2s are designed as mirror-imaged pairs. If you have received a pair where the tweeters are both on the same side, contact your dealer immediately.

INSIDE YOUR NEW SPEAKERS



If you're in a hurry to get started, skip ahead to Chapter 2, "Speaker Installation", for connection and placement tips.

ABOUT POWERED MONITORS

The M1 Active Mk2s combine a speaker and amplifiers in the same compact cabinet. Powered monitors are growing in popularity over the traditional separate amplifier and speakers for a number of reasons, convenience and ease of hookup being only one of them. You can connect a powered speaker directly to any line-level source (normally, the control room output of a mixer) simply by connecting a patch cord.

Another benefit is improved sound quality. With careful design, the speaker, amplifier and electronic crossover can be optimised for each other. The M1 Active Mk2s are *bi-amplified*, meaning that low frequencies and high frequencies are handled not only by separate speakers (the tweeter and the woofer), but by separate amplifiers. A pair of M1 Active Mk2s contain a total of four power amplifiers. Because of the increased efficiency of bi-amplification, these are much louder than a single-channel power amplifier of the same wattage feeding a passive crossover, as in other designs. New technology makes it possible to make these high-wattage amplifiers small enough to fit inside the speaker cabinet, with very little weight or size gain.



Since the M1 Active Mk2s are self-powered, **DO NOT** connect them to the speaker output of another amplifier (such as a powered mixer or hi-fi receiver). Connect them only to the line-level outputs of such devices (+4 dBu nominal, +24 dBu maximum).

TONE

Tonally, every effort has been made to create a powered monitor that is as linear as possible, representing a flat frequency response from 50Hz to 20KHz $\pm 2\text{dB}$. (see p. 21) Additionally, the use of electronic crossovers within the critical upper midrange frequencies has reduced phase and time delay anomalies often associated with passive crossovers. By including discrete woofer and tweeter amplifiers and application-specific electronic high and low pass crossover filters in the same enclosure the tonal accuracy of the M1 Active Mk2 will always remain constant.

ABOUT THE DRIVERS

The M1 Active Mk2's 6.5" woofer cone is made of proprietary non-woven carbon fiber. This material is 25% lighter than polypropylene with twice the stiffness for quicker transient response in the low-to-midbass region and vastly improved midrange intelligibility (over polypropylene at 2000 Hz). The non-woven carbon fiber cone, along with the closed cell synthetic rubber surround, are both nearly impervious to ozone, direct sunlight, heat and humidity. Therefore, after an initial break-in period, the sound should remain virtually unchanged for the life of the product.

Alesis' proprietary-design tweeter features a special wave guide baffle to improve polar response, a silk dome, and low viscosity ferrofluid formulated specifically to retain the best balance of transient response to power handling. The shielded tweeter utilizes a vented pole piece with a separate rear chamber to lower the free air resonance. The low 2000Hz crossover point of this tweeter produces a very widely dispersed, low-distortion signal within the critical midrange frequencies. It is these frequencies which are often the most problematic to get "right" during a mix. The tweeter is designed for an optimal response for non-fatiguing, long term, high level, nearfield mixdowns. This design results in a flat, linear mix when played back on home or car systems from appropriate distances.

MAGNETIC SHIELDING

Your M1 Active Mk2s are at home in recording environments wherein video is a major component. They are magnetically shielded for use in fairly close proximity (3" minimum) to a computer monitor or video screen. (Non-shielded components can make the colors of a screen "smear" or appear out of focus if the speaker is too close.) Magnetic shielding was designed into both the woofer and the tweeter from the beginning so that the system exhibits about low magnetic leakage. Shielding of both the woofer and the tweeter is accomplished by the use of a second opposing-field-oriented "bucking" magnet.

THE M1 ACTIVE MK2 BI-AMPLIFIERS & ELECTRONIC CROSSOVERS

The M1 Active Mk2 uses a 75W woofer amp and a 25W tweeter amp along with 4th order high and low pass electronic crossover filters centered at 2000Hz. Electronic crossovers, which separate the frequencies before being sent to the amplifiers, have fewer phase problems and less power loss than the traditional passive crossover used after the amplifier. Fourth-order filters are steep (24 dB per octave), which minimizes the interaction between woofer and tweeter near the crossover point. Additionally, the high pass (tweeter) section of this system employs an electronic time alignment circuit. Thus the “launch” of frequencies from both the woofer and the tweeter will originate at exactly the same time.

There are several advantages to using built-in amplifiers and active (electronic) crossovers versus a passive network system, most of which have been previously enumerated in Vance Dickason’s “Loudspeaker Design Cookbook”:

- Lower intermodulation distortion due to amplifier operation over a more narrow bandwidth. Also, clipping caused by low frequency overload is reduced, being limited to only one driver within a two driver system.
- Increased dynamic range. The M1 Active Mk2’s 75W and 25W amplifiers in their bi-amp setup will clip at about the same levels as one 200W amplifier operating into a passive crossover.
- Improved transient response.
- Better amplifier/speaker coupling for woofers.
- Better crossover performance working into a constant impedance load.
- Better subjective sound quality than high level (passive) networks.
- Easier control over driver sensitivity differences.
- Easier manipulation of phase, time delay, resonance, and various kinds of shaping, contouring and equalization.
- Specifically in the case of the M1 Active Mk2, the wide, mid-frequency polar response of the tweeter can be utilized down to a low 2000 Hz thereby mating more cohesively with the 6.5" non-woven carbon fiber (NWCF) woofer. This low crossover point is only possible through the use of a steep, *electronic* fourth-order (24 dB per octave rolloff) high pass filter.

THE M1 ACTIVE MK2 POWER SUPPLY

The M1 Active Mk2 employs a regulated switching power supply. This type of supply has previously only been available in amplifiers sold at the very high end of the Pro Audio market.

Most power amp customers are familiar with the large transformer and output capacitors in high quality amplifiers. These large components are used to keep the amplifier's supply rails as close to DC as possible. For traditional supplies, bigger is better.

But by using Pulse Width Modulation (PWM) and output voltage feedback in the M1 Active Mk2, we have the equivalent of near-infinite transformer and output capacitors, but at a fraction of the cost. PWM provides protection from line voltage surges that would otherwise destroy the amplifier. No user-replaceable fuse is required because nuisance tripping is eliminated. And finally, PWM practically eliminates hum because the line frequency (60 Hz) transformer is replaced with a very high frequency (130,000 Hz) transformer.

The regulated supply rails increase the amplifier dynamic range and reduce clipping distortion under heavy loading. The compact size of the supply does not significantly reduce the internal air volume of the speaker cabinet. And lastly, full power can be maintained even with low or "flat-topped" line voltages.



Note, however, that this switching supply is NOT a universal voltage type that may be used with any input voltage in any country simply by changing the cord. If you wish to use the M1 Active Mk2s in a country whose line voltage is other than 120 volts AC, they must be internally modified or used with a step-down transformer.

THE CABINET AND PORT DESIGN

The M1 Active Mk2 cabinet construction employs vinyl composite material laminated to a .625" MDF (medium density fiberboard) core for the four sides and the rear. The front panel is 1" thick MDF, to withstand the greater demands of the drivers mounted on it. The entire cabinet is braced extensively in random fashion to dissipate intramodal vibration.

WHY THE TWEETERS ARE OFFSET

The M1 Active Mk2 speakers have been developed as a mirror-imaged pair with dual front mounted ports and an offset tweeter. In some applications, an extremely wide dispersion tweeter, such as used in the M1 Active Mk2, will exhibit response dips if mounted exactly on center. Wave-guide style tweeters, used in other products, purposely limit dispersion and thus can be center mounted. But as hi-end audiophile speaker companies have found, a high quality dome tweeter, offset in the cabinet toward the *stereo "image center"* (*between the left and right speakers*) will have a flatter characteristic frequency and power response. This (flat power response) more closely characterizes the way humans hear.

On a horizontal polar response curve of the M1 Active Mk2, with the cabinet placed vertically, the tweeter response at one meter stays essentially symmetrical on either side of the cabinet's centerline. This is because the angle subtended between the cabinet-centerline-to-tweeter-centerline (.5"), and the nominal nearfield listening distance (1 meter or 39.375"), is only 1.27 degrees. Thus, acoustically speaking, the offset tweeter design has no negative effects, only the positive of smoothing the frequency response in the 2KHz-4KHz range.

In the tradition of the Alesis Superport design, which uses the energy from the back of the speaker to enhance bass performance, the M1 Active Mk2s use dual, front baffle mounted, long folded ports. The dual ports on the M1 Active Mk2 are also non-symmetrical with relation to their distances from the woofer. In developing the M1 Active Mk2, Alesis engineers determined that dual ports, offset to the outside of the stereo image center gave the most coherent and extended low bass augmentation. The ports are flared on the front baffle to provide smooth and quiet exit airflow while inside they are cut with a 45° angle and faced toward the amplifier's internal heat sink. This arrangement aids in keeping the internal amps working at a more constant temperature. At the same time, this 45° angle increases the port's air entry efficiency.

ABOUT NEARFIELD MONITORING

In the early days of recording, most recording studios used big monitor speakers almost exclusively. Unfortunately, they also required high powered amplifiers and expensive acoustic treatment (often poorly done) of the *entire* control room. Still, a well-constructed big monitoring system really was impressive to listen to, a fact not overlooked by the studio owners who wanted to impress the record company executives who paid for the big studio's time. These big systems had big level control knobs, and clients enjoyed "cranking-up" the volume. Fortunately, recording engineers and producers eventually learned that this was not the best way to accurately mix music because it wasn't the way most people listened to their radios, cassettes and CD players. Also, big monitor systems and the costs for the required control room acoustic treatments were going through the roof (no pun intended), particularly beyond the budget limits of smaller project and home studios which were growing in numbers. A new way of accurate monitoring was needed: nearfield monitoring.

Nearfield monitors, by their definition, are intended for mounting close to the listener. The idea here is to improve the direct acoustic path between the speaker and the listener by making it shorter, thereby giving less opportunity for the always present indirect (reflected) sounds to get back in and muddle things up. With nearfield monitoring, the surrounding acoustic environment becomes a much less significant factor in establishing the monitor system's sound character.

A good set of small monitors properly placed in a reasonably non-reverberant room and properly powered will yield surprisingly accurate results at budget prices. Carried to another studio, the same monitor should also provide *repeatable* results. In fact, some recording engineers carry their own speakers around because they know how they will sound in almost any room. Now, even the big studios use smaller speakers to augment their big monitoring systems, and nearfield monitors have become proven tools in the recording business.

SPEAKER INSTALLATION

Like any speaker system, your M1 Active Mk2 speakers will work best when properly positioned in a suitable acoustic environment. Achieving proper speaker placement is usually straightforward, but even with nearfield monitors, speaker placement and the acoustics of the listening room itself are too often overlooked and can become significant contributors to an inaccurate and uninspiring monitoring environment.

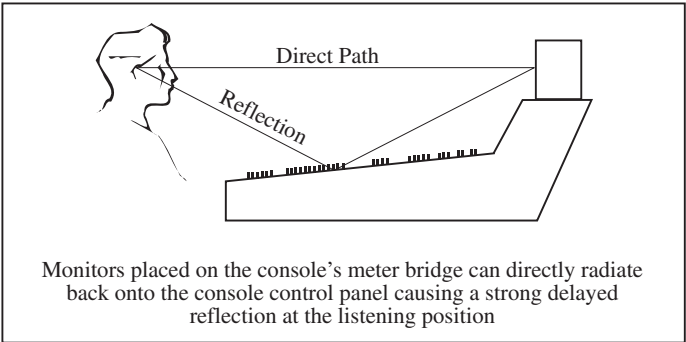
AVOIDING REFLECTIONS IN THE STUDIO

While nearfield monitors are more forgiving of the surrounding room acoustics, it is always prudent to optimize the listening environment whenever possible. First, the user should be aware of the effect that the size of the listening room can have on low frequency response. In general, the smaller the room, the stronger the bottom end will be, although placement within a larger room can also make a difference. This has to do with the way low-frequency waves travel in closed spaces. If you find your monitor system to be either light or heavy on the bottom, try moving them around within your listening room.

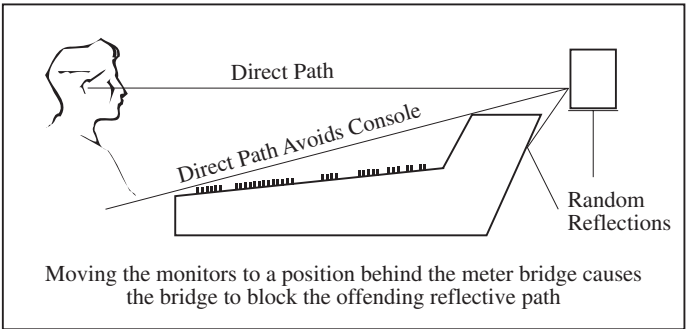
You should avoid locating your M1 Active Mk2s near reflective surfaces such as glass, tile, large open walls or table tops. Still, many rooms used for recording have these surfaces, so the best way to deal with them is to place the monitors out in the room away from reflective walls, windows and sizable objects. Even with these reflective surfaces separated from the monitoring position, typical mixing situations usually still have the top surface of the mixing board to deal with.

Unfortunately, the board itself can be a major source of reflections and the additional acoustic conduction into the board can affect your monitor's amplitude and phase response. Speaker placement on the console's meter bridge allows two different acoustic paths between the speakers and the recording engineer, which results in undesirable comb filtering effects and poor imaging. The first path is the direct one, and the second is via a reflection off the mixer main control panel:

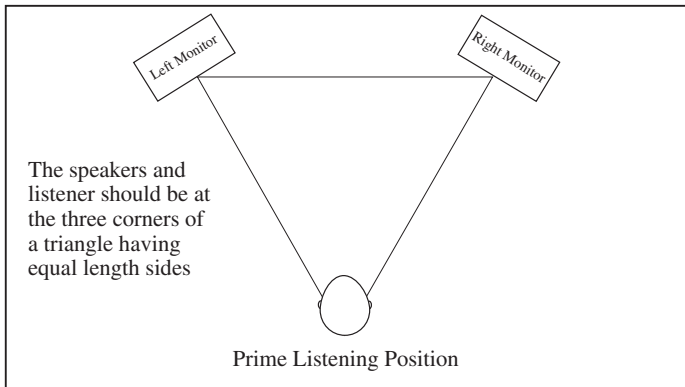
Speaker Installation



This kind of speaker placement also couples acoustic energy from the speaker's cabinet more readily into the console's chassis. Both conditions should be reduced by placing the speakers on their own stands acoustically detached from, and slightly behind, the console as shown below. In this location, the reflective path off the console's control panel is now blocked by the meter bridge.



Careful consideration should also be given to the physical spacing between the speakers. Alesis recommends that the distance between the speakers for stereo applications equal the distance between the listener and either speaker. In other words, the listener and the two speakers are at the three corners of a triangle having equal-length sides. The illustration below shows this concept. Note that both speakers are turned in somewhat, so that the prime listening position is directly in front of each speaker. Applications that require monitoring by more than one engineer are accommodated by a smaller rotation of the cabinets. This will widen the prime listening position somewhat.



STEREO NEARFIELD PLACEMENT OF THE M1 Active Mk2

NOTE: We recommend that the M1 Active Mk2 speakers be placed with the tweeters to the inside, not the outside, of the listening triangle.

The “classic” studio monitor layout used to be that the tweeters be placed to the *outside* of a horizontally-oriented speaker. In the past, this configuration was actually beneficial in time aligning the tweeter with the woofer *if* the cabinets were not toed-in toward the listener. However, this configuration is highly prone to comb filtering effects if the listener’s head is moved from side-to-side while mixing. This “comb filtering” causes the mid-to-high frequency tones to get louder, then softer, then louder again as you move your head from side-to-side, making it very confusing when trying to mix with precision.

Some people still believe that stereo separation is “improved” with tweeters to the outside, but this is advice left over from the early days (the sixties) of stereo recordings when “correct” stereo often meant a complete hard right or hard left placement of an instrument (or singer). As stereo mixing techniques changed toward placing the vocalist (for example) in the center, the old “tweeters-out” orientation would indeed widen the image *if* one’s head were kept in the *exact center* position. But this set-up proves to be very tiring, very quickly for the recording engineer. And, to others who are listening to the mix from either side of the engineer, the sound will seem to be coming primarily from the speaker closest to them.

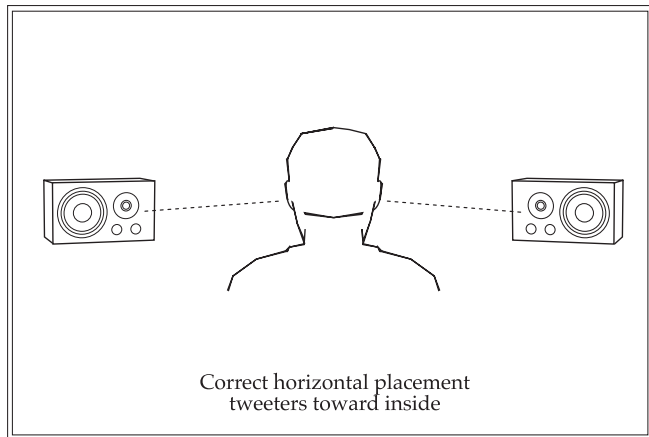
Fortunately, recording techniques have changed radically since the sixties. Engineers have learned to how to “place” an instrument or singer within the mix so that an accurate re-creation of the actual instrument and vocal positioning (left-to-right and front-to-back) is achieved. In the M1 Active Mk2, advances in crossover design technologies and improvements in the off-axis response of tweeter domes and woofer cone materials and profile have made the requirement for tweeters to be placed to the outside of the cabinet obsolete. **In fact, using a non-toed-in, tweeters-out orientation with a modern, wide dispersion design like the M1 Active Mk2 will increase the likelihood of hearing unwanted first reflections and a variety of phase anomalies in your mix.**

HORIZONTAL PLACEMENT RECOMMENDATION

Traditionally, proper horizontal placement of speaker systems slightly *behind* (not *on*) a meter bridge accomplished two purposes: it allowed both woofer and tweeter to be at ear level, and many times, it permitted the recording engineer to see over the speakers and into the studio.

Horizontal placement is a recommended arrangement with the M1 Active Mk2 because the left and right mirror-imaged pair permits a symmetrical alignment of drivers and ports. This will be conducive to a balanced mix. See Figure 2 below.

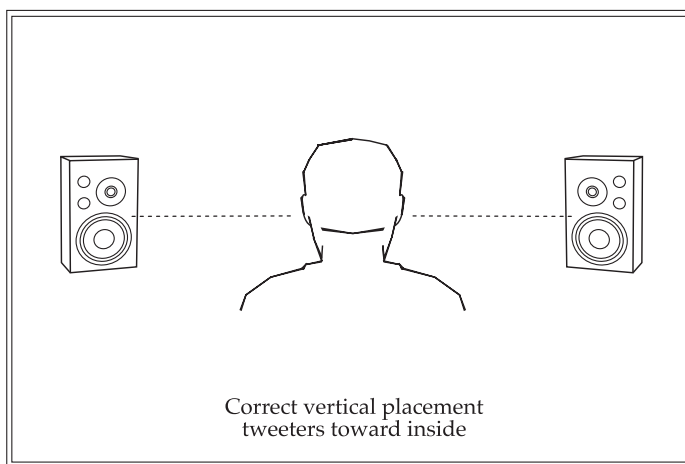
Proper tweeter orientation is toward the stereo image center (the middle) as shown. Ports are always faced toward the bottom side of the cabinet. This set-up will promote a strongly focused center image such as for the vocalist. And because the (vocal) image width will be narrower than if the speakers are placed vertically, it will be possible to place the vocalist with great precision at stage center. In this orientation there will be much less chance of first reflections from either sidewalls or the console coloring your mix.



VERTICAL PLACEMENT RECOMMENDATION

Vertical placement of the M1 Active Mk2 as shown in Figure 3 is highly recommended. This position will simulate the soundfield that will be heard in most consumers' homes and (to a great extent) their cars. For this reason, even if the M1 Active Mk2s are positioned horizontally for all of the mixing, the vertical position should always be used in the final "playback check" mode. (See next section)

There are a couple of possible drawbacks to vertical placement of a nearfield monitor relating mostly to room interaction effects. Vertical placement allows the M1 Active Mk2s to portray your mix with the widest and deepest soundstage possible, so many people in the control room have an image of the "sweet spot". However, this wide a dispersion pattern in a control room with walls in too-close proximity to the speakers can add strong reflections to the sound you hear, muddying your mix. Page 29 shows the control room placement/ distance recommendations, which should help you determine if your available recording space would work well with the M1 Active Mk2s set up vertically.



PLAYBACK CHECK MODE (VERTICAL MID-FIELD)

After you've got your mix just right it's always good practice to perform a "playback check" by standing the M1 Active Mk2s vertically and making the listening triangle larger. The purpose of the playback check is to simulate what your masterpiece will sound like in a "typical" home listening environment. It is in this configuration that the imaging specificity of your mix can best be evaluated. Generally, in the playback check mode you want to arrange the speakers in the same equilateral listening triangle that you do in a nearfield setup, but instead of each side of the triangle being about 3 feet long, the distances between stereo speakers and to your listening position should be between 7 and 12 feet.

When setting the left and right speakers vertically, place the tweeters toward the center with the ports toward the outside. In this vertical position, the M1 Active Mk2 becomes a "line source" speaker which is the most common home set up. (A 12" three-way with woofer on bottom, midrange in the center and tweeter at the top is a line source, as are most forms of tower speakers.) When the M1 Active Mk2 is set vertically in its "column" configuration, the *horizontal* dispersion (left and right) of the speaker is widest, and the vertical dispersion (up and down) is narrowest.

Unfortunately a vertical setup will also increase room effects, so there are minimum recommended setup distances from side and back walls: the speakers should be a *minimum* of 2.5 feet from the side walls and 3 feet from the rear walls. See the illustration on page 26.

After you've got the requisite distances from the side and back walls, it's often helpful to adjust the 7 to 12 foot equilateral listening triangle distances. What you want to find is the maximum distance that you can separate the two speakers while still holding a solid center image. If you get this placement correct *and* you've done a good job in the mix, you should be rewarded with a soundstage that accurately places the musicians in the width and depth perspective in which they were recorded. The sound should be very similar to what you heard at your mixing board, with possibly a little more ambience contributed by the listening room.

The "other" playback check mode, sometimes considered of greater importance than the home environment, is to listen to your mix over a car audio system. Bear in mind that an expertly designed car audio system can approach the imaging and soundstaging capabilities of an optimally setup studio or living room, but most times this is NOT the case. The inherent limitations of an always-off-axis listening position and the variables of speaker placement, bass coupling (boost) in the small interior and numerous other factors make the car environment highly suspect. So, understand the sonic capabilities (and deficiencies) of your (or other) car audio systems, and evaluate your mix accordingly.

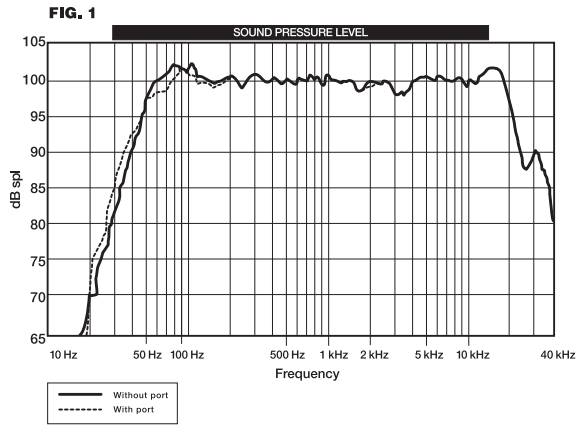
PLUGGING THE PORTS TO ADJUST FOR ROOM ACOUSTICS

The low frequency response of the M1 Active Mk2s can be custom tailored by plugging the ports. If the speakers must be placed less than one foot from a front wall, or if they are too close to a corner, then the bass may become exaggerated and inaccurate. The rules of thumb for these situations are as follows:

- a) less than 1 foot from a front wall but *away* from any corner:
plug **one** port
- b) less than 1 foot from a front wall *and* a corner:
plug **both** ports

The “plug” can be wad of cloth to completely block the port(s) or a more porous material like fiberfill which will give less bass attenuation. The bass output of the M1 Active Mk2 is thus easily variable depending on in-room positioning and/or personal preference.

If the M1 Active Mk2s are to be used in a system with a subwoofer, the proper blend of the two systems is critical if low-bass accuracy is to be maintained. Here again, your ear is your most important piece of test equipment. If the subwoofer (such as the Alesis Point One) or preamplifier has 80Hz high-pass filters built-in for connection to the other amplifiers, then no port plugging is necessary. The proper match will have already been taken care of electronically. In the absence of an 80Hz high-pass filter feeding the M1 Active Mk2s, you’ll need to experiment with various degrees of port plugging to minimize the cancellations and nodes that can occur from bass sources placed several feet apart in a room. In practice we’ve found that a single port plugged with cloth is usually the best compromise. See Figure 5 below. It shows the approximate bass attenuation that can be expected with one port plugged.



Speaker Installation

CONNECTIONS

The M1 Active Mk2 has been designed to be as simple as possible to set up and use. However, please follow the instructions to ensure that you get the best possible performance from your monitors.

Before connecting the speakers, make sure the power switches of the M1 Active Mk2s and any equipment they will be connected to are turned OFF.

AC POWER CABLE

READ ALL SAFETY WARNINGS IN THE PREVIOUS SECTION OF THIS MANUAL TO ENSURE SAFE OPERATION OF THESE UNITS. Connect the M1 Active Mk2s to the specified power using the AC cables supplied. The AC cables are removable. If the distance to your AC outlet is longer or shorter than the supplied cable, you may substitute an approved standard NEMA-to-CEE power cable of the correct length, available from most electronics stores.

GROUNDING



CONNECT THE M1 ACTIVES TO A PROPERLY GROUNDED OUTLET ONLY. DO NOT USE ADAPTERS WHICH REMOVE THE SAFETY GROUND PROTECTION OR CUT OFF THE GROUNDING PRONG ON THE POWER CORD. Proper grounding is essential for user safety and low noise. If you experience 60-cycle hum in your sound system as a result of different ground potentials between different units in your system, plug all units into the same AC circuit (if the total power load allows) and make sure other devices in the system are properly grounded themselves. The M1 Active features balanced inputs, so if it is properly connected to other balanced units, AC ground potentials will not affect the audio. If you cannot get rid of ground loops, consult a professional electrician familiar with sound system power designs.

USE CLEAN POWER

The M1 Active Mk2's internal power supply is designed to filter out most AC line noise. However, it is still good practice to plug your sound equipment into an AC circuit that is not shared with lighting dimmers, refrigerators, air conditioning units, or other appliances that may induce noise into the power system.

POWER SWITCH

The POWER switch is located on the back panel. A blue power indicator lamp is on the front panel.

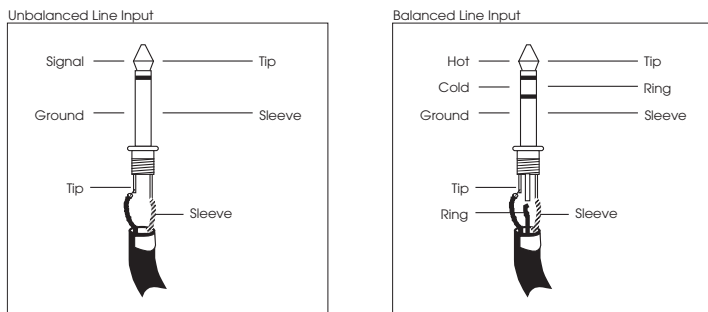
INPUT CONNECTIONS

Almost any conventional line-level source may be plugged into the M1 Active Mk2's LINE INPUT jack. The Line Input jack accepts both XLR and 1/4" input plugs wired either balanced or single-ended (unbalanced). The best connection is to use a high-quality XLR-to-XLR "mic cable" between the output of the source (usually, the CONTROL ROOM OUTPUT of the console) and the speakers. A conventional 1/4" "patch cord" mono TS-type plug will also work for unbalanced sources which automatically ground the minus input.

Polarity: Input connector wiring is printed to the left of the connector on the back panel and is as follows:

- + Pin 2: tip ("hot")
- Pin 3: ring ("cold")
- Shield Pin 1: sleeve (ground)

A positive voltage on the "hot" connector will cause the drivers to move outwards.



Direct connection of unbalanced -10 dBV line level sources: Most synthesizers, drum machines, effect devices, cassette decks and CD players operate at this level. Their average signal level is about 1/3 of a volt. They have a 2-conductor output jack that is either a 1/4" phone or "RCA phono" type. These may be plugged directly into the M1 Active Mk2's LINE IN jacks.

Speaker Installation

But do NOT plug the speaker output of an amplifier, receiver, or similar device into the LINE INPUT jack. Use only line-level sources. Plugging too loud of a signal into the LINE INPUT jack may damage the electronics.

Balanced +4 dBu line level sources: Professional recording and processing equipment typically provides a balanced, 3-conductor signal output that is a higher voltage (1.24 volts nominal level) than most synthesizers and stereo equipment. The LINE INPUT jacks are designed to handle these balanced inputs.

Balanced sources often feature XLR outputs. Connect them with an XLR-to-XLR cable. Many consoles (such as the Alesis Studio series) feature balanced outputs on 1/4" jacks; you may use a TRS-to-TRS balanced phone cable, or an XLR-to-1/4 inch phone TRS (tip-ring-sleeve) cable, as shown below. The advantage of using an XLR on the speaker side is that it is a locking connector and cannot be accidentally disconnected.



NOTE: Use high quality input connectors which conform to industry standard sleeve dimensions. Some off-brand connectors, particularly cheap 1/4" TRS plugs, do not provide reliable connections because of slight dimensional differences.

SETTING THE INPUT LEVEL CONTROL

In most cases, the INPUT LEVEL control should NOT be set at full rotation. For best system dynamic range, follow the instructions below.

The goal is to set the input level of the M1 Active Mk2s for the desired listening level, when all the rest of the system is at its nominal operating level. "Nominal" means its normal average level, expecting peaks from 8 to 15 dB above that on occasion. For the best signal-to-noise ratio, everything *before* the M1 Active Mk2 should be as loud as possible. The M1 Active Mk2s' INPUT LEVEL control, being last in the signal chain, is the worst place to make up gain lost earlier in the system.

TO SET THE LEVEL WITH A MIXER:

1. Turn on all system components (the M1 Active Mk2s last) and turn the M1 Active Mk2's INPUT LEVEL control all the way down.
2. Play program through the mixer and raise the mixer's fader levels until its meter averages around "0 VU".
3. Raise the Control Room Output to its nominal position (usually "2 o'clock" or 3/4 of the way up).
4. On one speaker, raise its INPUT LEVEL CONTROL for a comfortable level in the engineer's seat (if you have a sound pressure meter, 85 dB SPL is a common standard for listening).
5. Match the level for the other speaker. Note that the numbers on the back panel are only a rough guide; for precise left/right imaging, use pink noise or test tones and a sound pressure meter.

The Input Level control has a 28 dB range with a center detent. The detent has no particular meaning, but helps in adding a mental "reference position" when attempting to set levels from in front of the speaker. With the control at a maximum clockwise rotation, 2.0 volt input will produce full amplifier output (102 dB SPL "A" weighted from two speakers at 1 meter).

For the technically-minded, here is an approximate chart of SPL levels for various settings of the INPUT LEVEL control, measured at 1 meter from two speakers simultaneously (all settings, especially near the extremes of the range, are subject to a plus or minus 2 dB error):

Mark	-10 dBV nominal	+4 dBu nominal
0 (MIN)	-- dB SPL	-- dB SPL
1	--	--
2	57.0	69.0
3	63.5	75.5
4	68.2	80.2
5	71.0	83.0
6	73.0	85.0*
7	78.0	90.0
8	83.5	95.5
9	85.5*	97.5
10 (MAX)	86.5	98.5

* denotes recommended setting for control room use

AVOIDING DAMAGE

The best protection against speaker failure is to mix at a reasonable listening level. With a +4 dBu input and the input level control set to “6”, the M1 Active Mk2 speakers produce a sound pressure level (SPL) of 85 decibels at 1 meter, which is a good working level for mixing. At maximum output before clipping, they can generate peak levels ≥ 118 dB SPL at one meter – a level which may cause hearing loss after prolonged exposure. While the biamplified design of the M1 Active Mk2 avoids the common “blown tweeter” problem of conventional designs (when a broadband amplifier clips on a bass note, it sends unusually high energy to the tweeter, which tries to reproduce the square wave), it is not invulnerable. If you need higher power levels, you are not using the speaker in a near-field application. For extremely loud mid-field and far-field monitoring, we recommend the Alesis Monitor Two speaker with a 300-watt per channel amplifier. Many studios will check recordings on both systems, alternating to avoid ear fatigue and to gain perspective on the mix.

SURROUND SOUND

ABOUT SURROUND SOUND

Surround sound is the term used for several different systems which use multiple channels of playback in the listening room to recreate a truly three-dimensional sound experience. The practice of recording music-only in surround sound is a fairly recent innovation. Surround sound for video, however, has been around for over ten years. With modern encoding methods, multichannel audio may be delivered to consumers on standard video tape, video discs, and even through broadcast, with the proper *encoders* (devices which take multiple channels of audio and combine them into two channels) and *decoders* (used at the receiving end to split the signal back into multiple channels).

Systems are in use which use three (left, right, and surround), five (left, center, right, left surround, right surround, often supplemented by a sixth subwoofer channel), and more speakers. The original Dolby Surround® of the mid-eighties was a matrixed three-channel-only system. It had left, right and a mono surround (rear) track, which was usually played through two surround speakers. In 1987, Dolby Pro Logic®, a four-channel matrix system, was introduced. Pro Logic added a dedicated center channel to which most dialog was assigned. The rear surround channel was still in mono. In 1996, Dolby Digital® (AC-3) and DTS, fully discrete six-channel audio delivery platforms with separate left-rear and right-rear channels first became available to consumers on laserdisc and, more recently, on DVD (Digital Versatile Disc).

With so many listeners hearing music through their multimedia systems, it is important for engineers to produce music on surround systems, even if they're not working for film or video applications. Powered monitors like the Alesis M1 Active Mk2 Biampified Reference Monitor have proven to be a simple, cost-effective way to add five-channel surround to control rooms.

CENTER SPEAKERS IN MUSIC MIXES

In video applications, with left, center and right channels all set to deliver equal output, the ear tends to hear dialog as coming only from the center channel. (It's a psychoacoustic effect because all right and all left channel signals are fed to the center.)

In Dolby Pro Logic, the center channel information is derived from sum and difference information encoded in the left and right channels. The very attributes that make this matrix-type system viable for video application make it less desirable for music-only reproduction. In practice, there is a hard center dialog channel but there is also a sort of phantom mono channel developed between the left and center channel and the right and center channel. This between-channel phantom mono image exists because so much of the information contained in each channel (left and center or right and center) is identical. Hence, mono. In the few music-only recordings encoded in Dolby Pro Logic, the derived center channel could lead to some drastic changes in spatial perception.

In developing the M1 Active Mk2s for use as left, center and right front channel sources, we wanted them to be useable in both audio/video *and* audio-only "matrix" type applications. We discovered two, already available solutions to the large "wall-of-mono" effect produced by Dolby Pro Logic when playing audio-only material. The first is easy. Most home audio Dolby Pro Logic receivers will let you take the center channel out completely by putting the receiver in the "phantom" (no center channel) mode. It preserves the classic stereo effect of the front channels while adding rear surround. This method works well with most stereo-only recordings which have no surround encoding. But it's much better if the rear channels have true matrix (mono) surround (like applause) encoded in the mix.

The second method of using Dolby Pro Logic gets the center channel back into the mix. Most of these same home audio receivers have an audio-only surround mode which significantly attenuates the output of the center channel in relation to the left and right. The audible benefit here is that the mix sounds like a stereo only recording but the center image (like a vocalist) becomes a *hard* center. There is no image wander, as the frequencies change, for instance. And the classic listening "sweet spot" expands dramatically, allowing for accurate soundstage and instrument placement when listening far off-axis. (This attenuating of the center channel for audio only is also employed by advanced professional matrixed decoders such as the Circle Surround, Miles Technology and Lexicon units.)

PLACEMENT OF THE CENTER M1 ACTIVE Mk2 SPEAKER IN THE STUDIO

The left, center and right speakers should all be placed in an arc so that the distance of all drivers to the listener is identical (see page 29). The equilateral listening triangle as described earlier for stereo (two-channel) playback should be retained, with the center speaker placed *vertically* in the center, exactly between the left and right (vertical) M1 Active Mk2s.

It does not matter what type of M1 Active Mk2 is used for the center speaker (tweeter in or tweeter out). Since M1 Active Mk2s are sold as pairs, a 5-channel system requires the purchase of a single speaker, breaking up a pair. Some dealers may request that you purchase in pairs only; others may be able to arrange purchase of a single unit. For the rear channels, we recommend a mirror-image pair, although this is less critical than in the front.

As discussed above, for music-only recording the signal level to the center channel will be attenuated over the levels of the left and right speakers. As a starting point we would recommend that you set the center channel level -6dB below the left and right channels. To get these levels fairly accurate a white or pink noise source (such as from a test CD) is very helpful along with an inexpensive sound level meter (such as the Radio Shack cat. no. 33-2050).

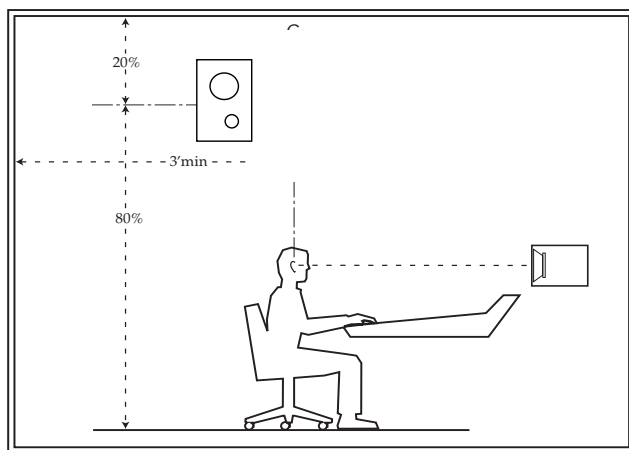
If you are recording audio-for-video, the center channel signal level should be *the same* as left and right channels. The left, center, right speaker layout configuration can remain the same as for an audio-only mix. With all front speakers operating at equal levels, the “sweet spot” listening area will become very large but the “wall-of-mono” effect mentioned above (in regards to music) will be of little consequence since most of the dialogue will be mixed specifically into the center channel.

PLACEMENT OF REAR SURROUNDS

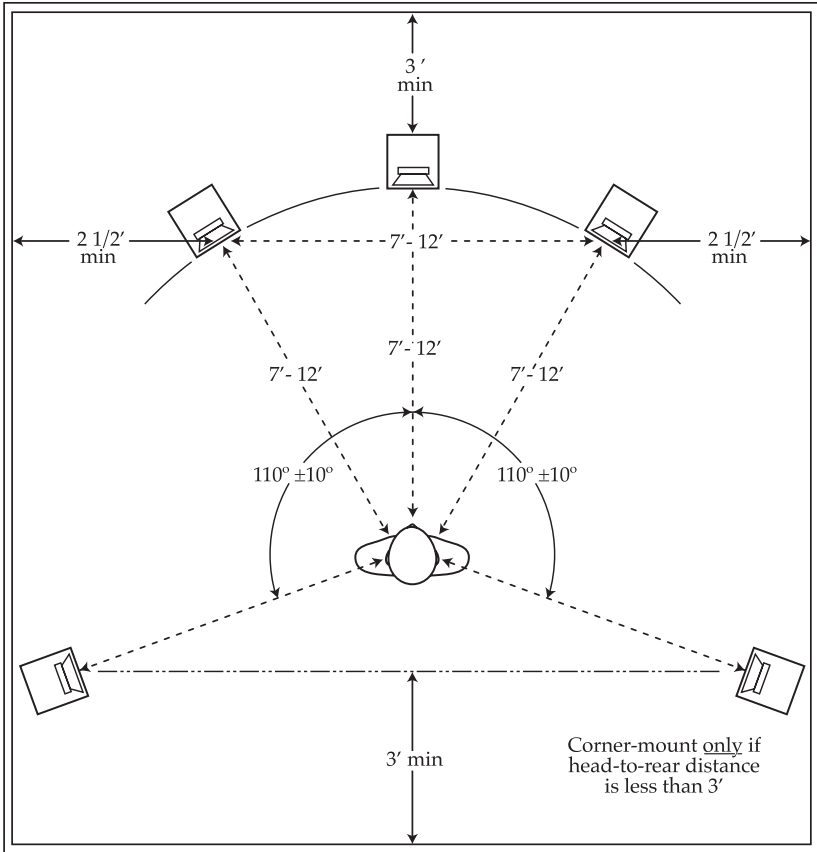
The following suggestions for optimal M1 Active Mk2 placement are derived from several sources: 1) Standard industry practice for home installations as recommended by the manufacturers of surround encoder/decoder units; 2) recommendations developed by Tom Holman for setting up THX surround theater sound systems, and 3) Alesis engineers' experience fine-tuning the performance of the SMS Monitoring System in order to find the mounting configuration which results in the most natural and accurate surround placement. With the many different playback systems in use, there is room for interpretation of what the best placement is in a control room, but the following represents a good starting point which you are free to change to your taste.

Vertically, the M1 Active Mk2s should be placed on the side wall of the control room, above your head, with the woofers on top, tweeters on the bottom. The ideal location is to have the center of the speaker (between the woofer and tweeter) at **80% of the floor-to-ceiling height**. (For a standard 8 foot ceiling height the center line would be at $96'' \times 80\% = 76.8''$ from the floor.)

Recommended positioning of surround speakers



Horizontally, place the M1 Active Mk2s slightly behind the 90° position, at 110° on each side. Make sure that at the position used, there is still at least 3 feet to the back wall of your studio. If you do not have at least 3 feet to the back of your studio, then the speakers will work better if placed diagonally in the rear corner as shown on the next page.



WALL MOUNTING

Once you have located the general area for the horizontal and vertical position, you need to find a wall stud. Attach shelving brackets or other commercial speaker-mounting device into the wall stud, then attach the M1 Active Mk2s to the mount, allowing room for the rear-panel input and AC connections.



The M1 Active Mk2s weigh 19.5 lbs., so it is vitally important that any stand they're resting on be securely mounted through the wall material (usually 3/4" sheetrock, which by itself does not have the strength to support the speakers) and into a wooden stud using hardened screws or other fastener rated to hold at least two times that weight.

MIXING FOR DISCRETE SIX CHANNEL REPRODUCTION: MATCHED VS. SPECIALIZED SPEAKERS

While most speaker systems for monitoring music in surround use the same speaker cabinet in all positions, both front and rear, keep in mind that most consumer video surround installations use a different speaker in the rear channels. These rear speakers typically do not have the same directionality or wide frequency response of the M1 Active Mk2s. When mixing for film, speakers with a broad ambient, non-directional response are used to mimic the multi-speaker situation in most movie theatres. Keep this in mind when making music mixes with full-range high-power speakers like the M1 Active Mk2s as the surround speakers.

The difference between music surround and video surround playback has a long history:

- Dolby Pro Logic, which works best with dipole (i.e., speakers splayed in different directions out of phase) surround speakers, has been *the* standard for surround sound in the home for 10 years.
- The most accepted home playback medium, VHS videotape, can only play back (mono) surround in Dolby Pro Logic. VHS will not support Dolby Digital or DTS.
- There are only two playback mediums, laser disc and DVD, that support Dolby Digital (discrete) and DTS (discrete). After 20 years in existence, laser disc has only a 2% market share. Some digital satellite broadcasters have started broadcasting in Dolby Digital, but these discrete broadcasts will be reserved for blockbuster movies initially.
- Discrete Dolby Digital, DTS and Sony's SDDS are most commonly mixed for theatrical release using surround, not monopolar (direct) monitors.
- Among the recording engineers who have recorded audio-only, the consensus seems to be that the role of the surround speaker is to give the listener a *sense of immersion* in the musical event. This is done by reserving the rear channels for audience sounds, in the case of a live recording, or perhaps by reproducing the reverberant acoustics of the recording environment itself. Other engineer/producers prefer using five identical full-range speakers and mixing primary instruments and vocals into the surround channels.

So, our basic recommendation for mixing an audio-only six channel recording is to intentionally limit the frequency response of the rear speakers using a graphic equalizer or high-pass (100 Hz rolloff)/low pass (12 kHz rolloff) filter on your mixer. This will help you get a rough idea of what a home system playback will sound like.

This arrangement will also allow your mix to sound best over a well set up car audio system. This is because professional installers have learned that a mobile system performs best when they concentrate on producing a soundstage across the *front* of the vehicle. They do this by placing high quality drivers in the front of the vehicle which produce frequencies from the

Surround Sound

mid bass (about 80Hz) on up. In the rear go the subwoofers and sometimes, small full range drivers which provide “fill”. So, in essence, today’s best car audio systems are designed to produce an enveloping environment very similar to that of a home theater.

The use of M1 Active Mk2s (or any monopolar speaker) as full-range surrounds would be useful in the following cases:

- If the surround speakers are just as important as the fronts in achieving a specialized effect, such as completely mixing the sax in the left rear only and a flute in the right rear.
- If you know that your mix will always be played back on a monopolar surround system.
- If the control room is completely damped with sound absorption on all surfaces. (This renders the reverberant-field generating properties of a bipole/dipole speaker ineffective.)

* Dolby Surround, Dolby Pro Logic, and Dolby Digital are trademarks of Dolby Laboratories, Inc.

TROUBLESHOOTING

TROUBLESHOOTING INDEX

If you experience problems while using the M1 Active/Surround systems, please use the following table to locate possible causes and solutions before contacting Alesis Technical Support for assistance.

Symptom	Cause	Solution
Hum or buzz	Ground loop between source and speaker	Use balanced cables. Put M1 Actives on the same AC power circuit as the mixer.
No sound	Speaker disconnected	Check input cable connections
	INPUT LEVEL down	Turn up input control on back panel
	AC power not on	Blue LED on front should be on. Check switch and power input cable.
Unfocused sound, bass frequencies muddy or missing	Inputs out of phase	Check + and - connections from mixer to input. Verify "pin 2 is hot".
	Subwoofer being used improperly	Install port plugs when using a subwoofer
	Poor mix	Compare to commercial CD of similar style: time to remix your track....
Distorted output	Power amp overloading speakers	Turn down level
	Damaged speaker components	Swap channels to see if the problem follows the speaker; if so contact Technical Support

MAINTENANCE

CLEANING

The cabinet surfaces of the M1 Active Mk2 are covered with a vinyl laminate. Clean these surfaces when necessary with a lint-free cloth dampened in warm soapy water. **USE ONLY A DRY CLOTH ON THE BACK PANEL OF THE POWER AMPLIFIER.** Do not attempt to clean the cabinet with a brush (which may damage the surface) or a sponge (which may leave small crumbs in the texture).

Do not attempt to clean either of the drivers.

The M1 Active Mk2/Surround speakers require no periodic maintenance.

REFER ALL SERVICING TO ALESIS



We believe that the M1 Active Mk2 is one of the most reliable powered monitors that can be made using current technology, and should provide years of trouble-free use. However, should problems occur, **DO NOT** attempt to service the unit yourself. Service on this product should be performed only by qualified technicians. **THERE ARE NO USER-SERVICEABLE PARTS INSIDE.**

SERVICE

Before sending the M1 Active Mk2s in for repair, make sure they are faulty and that the problem isn't being caused by something else in the system. Distortion or noise may be caused by a defective preamp, cable, equalizer etc., or a loose connection in the system. Connect the speakers to a system that is known to be working properly to check whether the monitors have malfunctioned.

OBTAINING REPAIR SERVICE

Before contacting Alesis, check all your connections, and make sure you've read the manual.

Customers in the USA and Canada: If the problem persists, call Alesis USA at 1-800-5-ALESIS and request the Technical Support department. Talk the problem over with one of our technicians; if necessary, you will be given a return order (RO) number and instructions on how to return the unit. All units must be shipped prepaid and COD shipments will not be accepted.

For prompt service, indicate the RO number on the shipping label. Units without an RO will not be accepted. If you do not have the original packing, ship the M1 Active Mk2 in a sturdy carton, with shock-absorbing materials such as styrofoam blocks or "bubble-pack" surrounding the unit. Make sure

the drivers will not be pressed on by the packing in shipment. Shipping damage caused by inadequate packing is not covered by the Alesis warranty.

Tape a note to the top of the unit describing the problem, include your name and a phone number where Alesis can contact you if necessary, as well as instructions on where you want the product returned. Alesis will pay for standard one-way shipping back to you on any repair covered under the terms of this warranty. Next day service is available for a surcharge.

Field repairs are not normally authorized during the warranty period, and repair attempts by unqualified personnel may invalidate the warranty.

Correspondence address for customers in the USA:

Alesis
200 Scenic View Drive
Cumberland, RI 02864

THIS IS NOT A SERVICE LOCATION. DO NOT SHIP UNITS TO THIS ADDRESS.

Customers outside the USA and Canada:

Contact your local Alesis distributor for any warranty assistance. The Alesis Limited Warranty applies only to products sold to users in the USA and Canada. Customers outside of the USA and Canada are not covered by this Limited Warranty and may or may not be covered by an independent distributor warranty in the country of sale. Do not return products to the factory unless you have been given specific instructions to do so.

SPECIFICATIONS

ENCLOSURE

MATERIALS AND CONSTRUCTION

Front baffle painted in metallic charcoal-gray, one inch (1") thick MDF with radiused edges to minimize diffraction. Sides and rear baffle covered in metallic charcoal-gray vinyl laminate, over five-eighths inch (0.625") thick MDF.

Dual ports with 45° inlet and flared front baffle outlet allow precise, phase accurate woofer-to-port frequency transition. Ports also aid in cooling internally mounted primary amplifier heat sink.

Flush mount, rubber woofer trim ring allows consistent power response characteristic at wide off-axis angles. Flush-mount tweeter plate assures smooth off-axis response.

UL-approved fiberglass acoustical damping material damps internal mid-frequency reflections from rear of woofer cone before entering port, lessening IM distortion.

TRANSDUCERS

LOW FREQUENCY DRIVER:

6.5" (165mm) diameter dynamic driver with proprietary non-woven carbon fiber (NWCF) cone and 1.5" diameter, long-wind voice coil.

Dampened linear rubber surround with moderate internal damping and constant linear displacement Nomex spider. ± 5 mm X-max excursion capability.

Dual magnet structure uses top magnet in opposed ("bucking") configuration for magnetic shielding.

HIGH FREQUENCY DRIVER:

1" silk soft-dome tweeter with special wave guide baffle.

Low-viscosity ferrofluid cooling extends power handling while retaining impulse response capability.

Dual magnet structure uses top magnet in opposed ("bucking") configuration for magnetic shielding.

CROSSOVER SECTION

Crossover type:	4 th -order, 24dB/octave Linkwitz-Riley @ 2000Hz
Input Impedance:	20k Ω balanced 10k Ω unbalanced
Low Frequency Filter:	4 th order optimal Q high pass filter at 48Hz with +3 dB equalization

AMPLIFIER SECTION

LOW FREQUENCY AMPLIFIER

Rated power output:	75 watts, 6 Ω load
Distortion:	<0.02% THD @ 30 watts/8 Ω 20Hz-20kHz
Slew rate:	19V/ μ sec
Signal-to-Noise ratio:	>110dB referenced to 60 watts @ 8 Ω , "A" weighted, 1kHz

HIGH FREQUENCY AMPLIFIER

Rated Power Output:	25 watts, 4 Ω load
Distortion:	0.025% THD @ rated power
Slew rate:	9V/ μ sec
Signal-to-Noise ratio:	>112dB @ rated output

ACOUSTIC SECTION

Free-Field Frequency Response:	$\pm$ 2dB, 50Hz to 20KHz
Lower Cutoff Frequency:	40Hz -10dB
Upper Cutoff Frequency:	23.5KHz -10dB
Maximum Peak SPL per pair:	$\geq$ 118dB SPL @ 1m
Maximum short term SPL:	80Hz to 3.0KHz $\geq$ 105dB SPL @ 1m

GENERAL

Power Consumption:	120 watts with musical program, loud mix 12 watts quiescent (idle)
AC Dropout Voltage:	120V AC version: 80V AC 240V AC versions: 160V AC
Weight:	19.5 lbs. (8.9 kg) each
Dimensions (H x W x D):	15" (38.1 cm) x 8.5" (21.6 cm) x 10" (25.3 cm)

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