

**Devices Connected/Referenced**

<b>ADMP504</b>	Ultralow Noise Analog MEMS Microphone
<b>SSM2167</b>	Microphone Preamplifier with 18 dB Fixed Gain

## Low Noise Analog MEMS Microphone and Preamplifier with Compression and Noise Gating

### EVALUATION AND DESIGN SUPPORT

#### Circuit Evaluation Boards

[SSM2167 Evaluation Board \(SSM2167Z-EVAL\)](#)

[ADMP504 Evaluation Board \(EVAL-ADMP504Z-FLEX\)](#)

#### Design and Integration Files

[Schematics, Layout Files, Bill of Materials](#)

### CIRCUIT FUNCTION AND BENEFITS

This circuit, shown in Figure 1, interfaces an analog MEMS microphone to a microphone preamp. The **ADMP504** consists of a MEMS microphone element and an output amplifier. Analog Devices' MEMS microphones have a high signal-to-noise ratio (SNR) and a flat wideband frequency response, making them an excellent choice for high performance, low power applications.

The **SSM2167** is a low voltage, low noise mono microphone preamp that is a good choice for use in low power audio signal

chains. This preamp includes built-in compression and noise gating, which gives it an advantage for this function over using just an op amp in the preamp circuit. Compressing the dynamic range of the microphone signal can reduce the peak signal levels and add additional gain to low level signals. Noise gating attenuates the level of signals below a certain threshold, so that only desired signals, such as speech, are amplified, and noise in the output signal is reduced. These features help to improve the intelligibility of the voice signal picked up by the microphone.

### CIRCUIT DESCRIPTION

The **ADMP504** analog MEMS microphone is connected to the **SSM2167** INPUT pin through a 0.1  $\mu$ F capacitor. This coupling capacitor is necessary because the microphone's output is biased at 0.8 V, and the preamp's input is biased at 0.4 V. The preamp's input and the ac coupling capacitor between the microphone and preamp input form a high-pass filter.

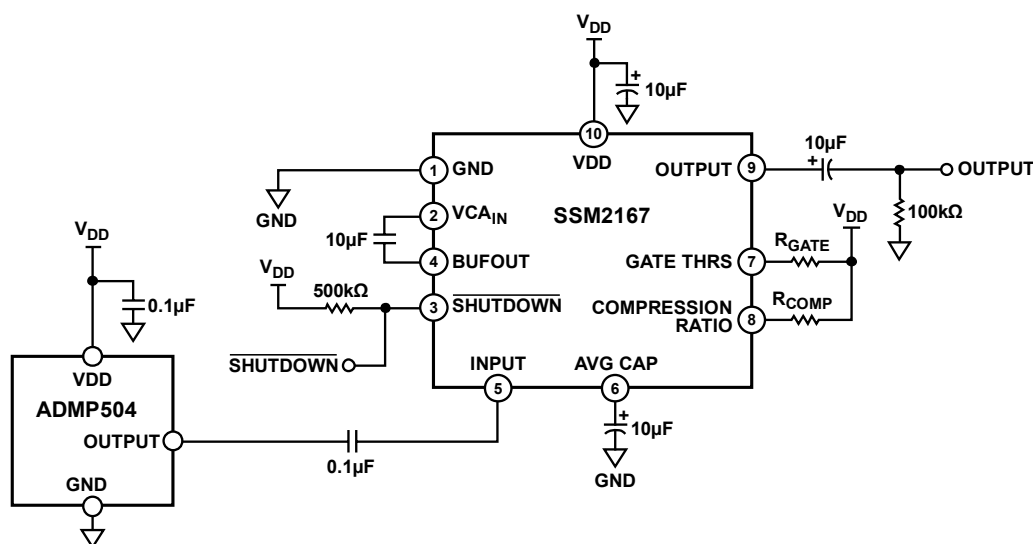


Figure 1. Analog MEMS Microphone Connection to Preamplifier

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#### Rev.0

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The  $-3$  dB corner frequency of this filter is  $1/(2\pi RC)$ , where  $C$  is the capacitor size, and  $R$  is the preamp's input impedance ( $100\text{ k}\Omega$ ). For a  $0.1\text{ }\mu\text{F}$  coupling capacitor, the high-pass filter's corner is  $16\text{ Hz}$ . Increasing the capacitor size will lower the filter corner frequency.

The time constant of the [SSM2167](#)'s true rms level detector is controlled by the size of the capacitor connected to AVG CAP (pin 6). A  $10\text{ }\mu\text{F}$  capacitor results in a time constant of about  $100\text{ ms}$ , which is a reasonable setting for speech signals. This time constant controls the rms detector's averaging, as well as the compressor's release time. A smaller capacitor used here will give a shorter time constant, and a larger capacitor will result in a longer time constant. The time constant in milliseconds is calculated by  $10 \times C_{\text{AVG}}$ , where  $C_{\text{AVG}}$  is in  $\mu\text{F}$ .

Both the microphone and the preamp can be powered from a single  $2.5\text{ V}$  to  $3.3\text{ V}$  supply.

The [SSM2167](#) preamp requires some additional external passive components for its operation:

- $10\text{ }\mu\text{F}$  capacitor between  $V_{\text{DD}}$  and ground
- $10\text{ }\mu\text{F}$  capacitor between pins  $\text{VCA}_{\text{IN}}$  and  $\text{BUF}_{\text{OUT}}$
- AC coupling capacitor on OUTPUT pin
- $500\text{ k}\Omega$  pull-up resistor on SHUTDOWN pin
- $R_{\text{GATE}}$ : sets the threshold of the noise gate
- $R_{\text{COMP}}$ : sets the compression ratio

The [ADMP504](#) has a  $-38\text{ dBV}$  sensitivity, which means that an input signal of  $94\text{ dB SPL}$  ( $1\text{ Pascal}$ ) will be output from the microphone at  $-38\text{ dBV}$ . This microphone's maximum input level is  $120\text{ dB SPL}$ , at which level its output will be  $-12\text{ dBV}$ . Its dynamic range is  $91\text{ dB}$ , so the microphone's noise floor is at  $-103\text{ dBV}$ .

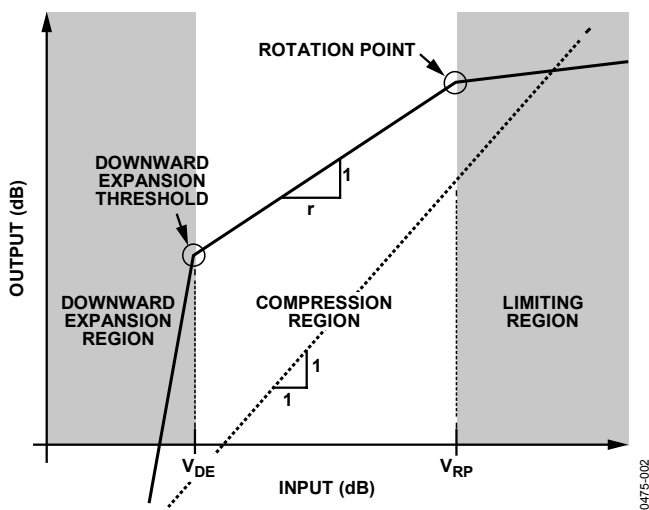


Figure 2. Gain Regions

Gain is applied to the [SSM2167](#) input signal in three different regions—downward expansion, compression, and limiting, as shown in Figure 2. The [SSM2167](#) applies a fixed  $18\text{ dB}$  of gain to input signals and can also apply an additional variable gain to signals between the downward expansion point and rotation point.

The downward expansion threshold is the boundary between the downward expansion and compression regions. This point is set by the selection of resistor  $R_{\text{GATE}}$  (Table 1). The threshold can be set at a point for input signals between  $77\text{ dB}$  and  $92\text{ dB SPL}$ , or  $-55\text{ dBV}$  to  $-40\text{ dBV}$  input to the preamp.

Table 1. Noise Gate Threshold Settings

Preamp Noise Gate Threshold (dBV)	Microphone (dB SPL)	$R_{\text{GATE}}$ (k $\Omega$ )
$-40$	92	0 (short to $V_{\text{DD}}$ )
$-48$	84	1
$-54$	78	2
$-55$	77	5

The compression region lies between the downward expansion and limiting regions. In this region, the dynamic range of the input signal can be reduced, or compressed, so that the output signal level is more smooth and constant. The [SSM2167](#) can achieve compression ratios of up to  $10:1$ . This downward compressor will increase the level of the signal below the rotation point threshold. The level of compression is controlled by resistor  $R_{\text{COMP}}$ , as detailed in Table 2.

Table 2. Compression Ratio Settings

Compression Ratio	$R_{\text{COMP}}$ Value (k $\Omega$ )
1:1	0 (short to $V_{\text{DD}}$ )
2:1	15
3:1	35
5:1	75
10:1	175

The boundary between the compression and limiting regions is fixed at a  $-24\text{ dBV}$  preamp input signal level, which corresponds to a  $108\text{ dB SPL}$  acoustic input to the microphone. Above this point, after the  $18\text{ dB}$  of fixed gain, the preamp output will be limited to a  $-6\text{ dBV}$  level.

Figure 3 shows the voltage output levels vs. acoustic input levels of the circuit for a selection of different compression ratios.

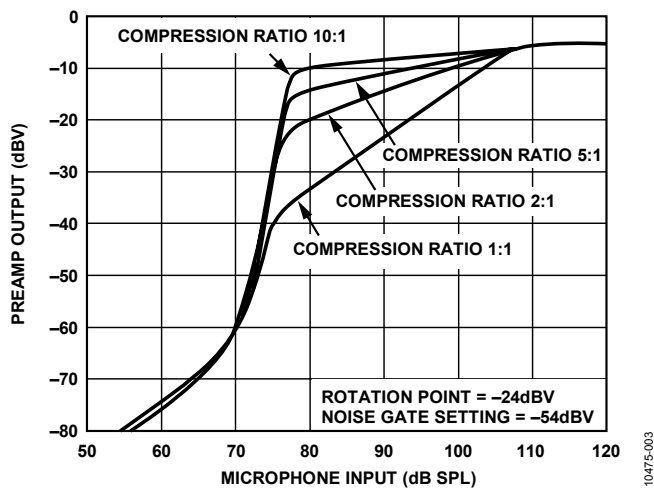


Figure 3. Circuit Output vs. Input Characteristics

The [SSM2167](#) output is biased at 1.4 V. The 10  $\mu$ F ac-coupling capacitor in series with the output signal allows the output to be connected to grounded loads. The 100 k $\Omega$  resistor references the back side of the coupling capacitor to ground to avoid pops when loads are hot-plugged. This resistor is not necessary in a design where the load is hardwired to the preamp's output.

A complete design support documentation package for this circuit can be found at this address:  
[www.analog.com/CN0262-DesignSupport](http://www.analog.com/CN0262-DesignSupport).

### Noise Performance

Both the [ADMP504](#) and [SSM2167](#) are low noise audio devices. The [ADMP504](#) SNR is 65 dB, which gives a  $-103$  dBV noise floor. The noise floor of the [SSM2167](#) with 10:1 compression (worst-case setting for noise) is  $-70$  dBV. So, the microphone signal's noise level can be increased by more than 20 dB before combining the noise floors of the two devices results in a degradation of the overall system noise floor. For example, a loud 100 dB SPL signal output picked up by the microphone with the preamp set to a 10:1 compression ratio will have about 24 dB of gain added to the system. This will put the microphone signal's noise floor at  $-79$  dBV ( $-103 + 24$ ). Combining a  $-79$  dBV noise source with the  $-70$  dBV noise floor of the preamp will result in degradation in the system of only about 0.5 dB, so the noise floor of the output signal in these conditions is about  $-69.5$  dBV.

When the microphone's output signal is at a level below the downward expansion threshold, a fixed 18 dB gain will be applied to the signal, which will keep the signal's noise floor well below the noise floor of the [SSM2167](#).

### COMMON VARIATIONS

This circuit can also be set up with the [SSM2166](#) instead of the [SSM2167](#). The [SSM2166](#) is a more flexible, but more expensive, preamp. The rotation point and VCA gain on the [SSM2166](#) can be adjusted with external components, while these settings are fixed on the [SSM2167](#). The [SSM2166](#) also has a lower noise floor than the [SSM2167](#) and is provided in a larger package (14-lead SOIC\_N).

The [ADMP504](#) could also be replaced with the [ADMP401](#), [ADMP404](#), or [ADMP405](#). These three MEMS microphones have a 62 dB SNR, while the SNR of the [ADMP504](#) is 65 dB. The [ADMP401](#) has a  $-42$  dBV sensitivity, while the [ADMP504](#), [ADMP404](#), and [ADMP405](#) have a  $-38$  dBV sensitivity. The [ADMP405](#) is identical to the [ADMP404](#) except that it has a low frequency cutoff at 200 Hz vs. the [ADMP404](#)'s 100 Hz cutoff. This higher frequency cutoff makes the [ADMP405](#) attractive for reducing low frequency wind noise. The [ADMP404](#) and [ADMP405](#) are also pin-compatible with the [ADMP504](#).

### CIRCUIT EVALUATION AND TEST

Evaluation boards for the [ADMP504](#) ([EVAL-ADMP504Z-FLEX](#)) and [SSM2167](#) ([SSM2167Z-EVAL](#)) are available and can easily be connected as described below.

The [EVAL-ADMP504Z-FLEX](#) has three output wires:  $-VDD$  (red), ground (black), and output (white). The  $VDD$  wire should be connected to the +3 V test point of JP3 on the [SSM2167](#) board; power is also supplied to the preamp board at this point. The ground wire can be connected to GND of JP3. The output wire of the [ADMP504](#) board can either be connected directly to the JP2 pin or to the tip of a mono 3.5 mm audio plug, with ground connected to the ring. This plug connects to the [SSM2167](#)'s evaluation board input jack J1.

Resistor R4 should be removed from the [SSM2167](#) evaluation board. This resistor is used to supply power to electret microphones, but is not needed for MEMS microphones that are powered through a separate  $VDD$  pin.

From this point, follow the documentation for the [SSM2167Z-EVAL](#) in application note [AN-583](#) regarding board setup and listening tests. The documentation for the [SSM2167](#) evaluation board ([SSM2167Z-EVAL](#)) describes the system setup and gives a complete schematic of the board. The only external connections required are the +3 V power and audio input and output on J1 and J2, respectively.

**LEARN MORE**

CN0262 Design Support Package:

[www.analog.com/CN0262-DesignSupport](http://www.analog.com/CN0262-DesignSupport)

Scarlett, Shawn, "Using the SSM2167 Evaluation Board,"

Application Note AN-583, Analog Devices.

Elko, Gary W. and Kieran P. Harney, "A History of Consumer Microphones: The Electret Condenser Microphone Meets Micro-Electro-Mechanical-Systems," *Acoustics Today*, April 2009.

Nielsen, Jannik Hammel Nielsen and Claus Fürst, "Toward More Compact Digital Microphones," *Analog Dialogue*, September 2007.

Lewis, Jerad, "Microphone Specs Explained," Application Note AN-1112, Analog Devices.

Khenkin, Alex. Application Note AN-1003, "Recommendations for Mounting and Connecting Analog Devices, Inc., Bottom-Ported MEMS Microphones," Analog Devices.

**Data Sheets and Evaluation Boards**

[SSM2167 Data Sheet](#)

[SSM2167 Evaluation Board](#)

[ADMP504 Data Sheet](#)

[ADMP504 Evaluation Board](#)

[ADMP401 Data Sheet](#)

[ADMP404 Data Sheet](#)

[ADMP405 Data Sheet](#)

**REVISION HISTORY**

1/12—Rev. 0: Initial Version

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