

Kenwood Microphone MC60A Modification - Another use for the SSM2165

The Kenwood MC60A with preamp, suffers from RF feedback, a search on the web reveals the extent of the problem. the many looking for solutions to this problem.

There exists an error on the user documentation, that shows the microphone audio wiring as reversed from one socket to the other, the Ground system is poorly implemented in the layout on the switch board, further the use of a single transistor as an amplifier is not a great idea, there is no common mode rejection, further adding to the problem.

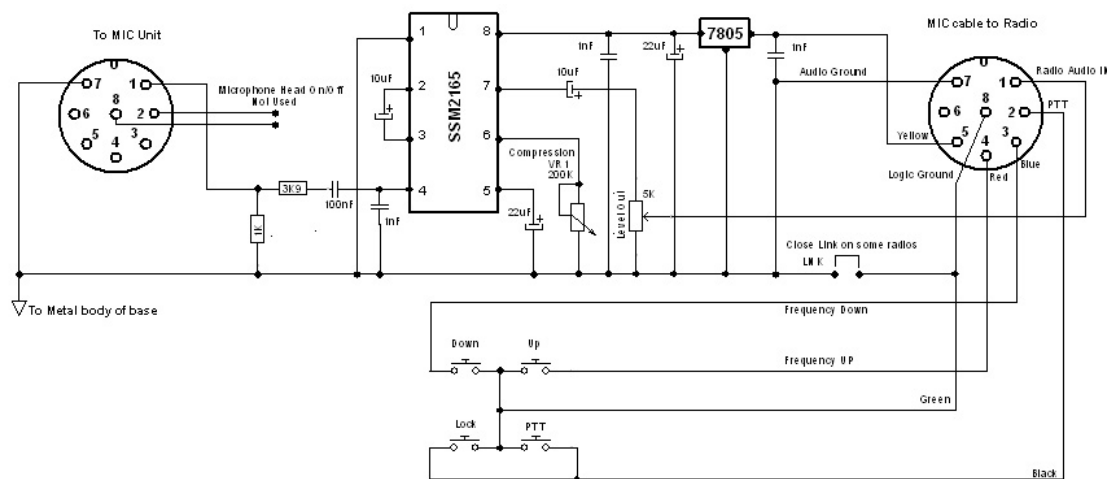
To address the above issues, one IN3ILX tackled the problem by the way of extensive modification to the Ground in both the microphone head, the base and the addition of ferrites on some of the wires.

<https://www.youtube.com/watch?v=Vv7Ak549IsA>

I chose different approach, that to use an IC, the differential input will provide common mode rejection keeping the offending RF out of the audio path, plus the ease of impedance matching, with this I also considered the use of the SSM2165 providing all the features of levels conditioning that this IC provides.

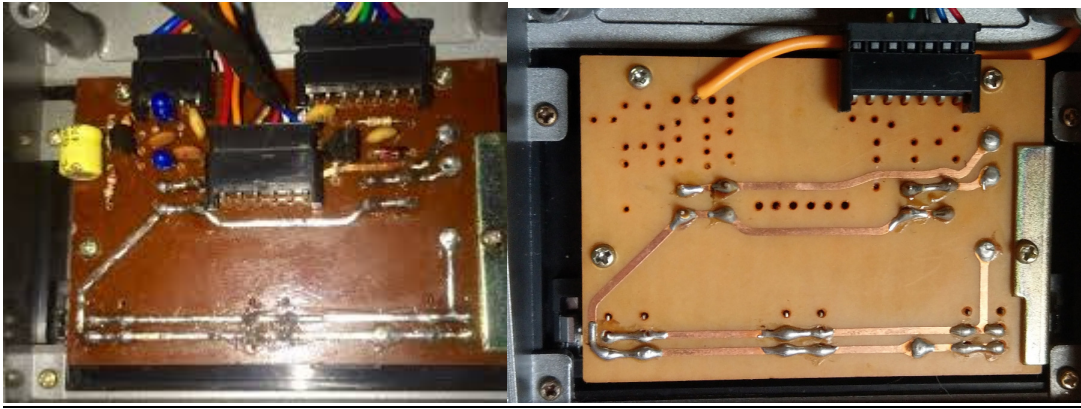
The microphone after the mod has a dynamic adjustable range of over 38dbm with variable compression, in addition of peak limiting,

Compare this diagram with that of the Original Kenwood Wiring



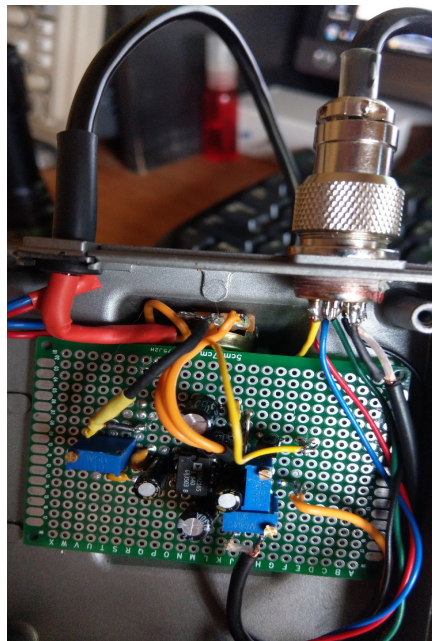
Preparation

My preparation for this mod, was to remove the 3 Volt regulator, plus the single transistor amplifier and associated components and sockets no longer required.



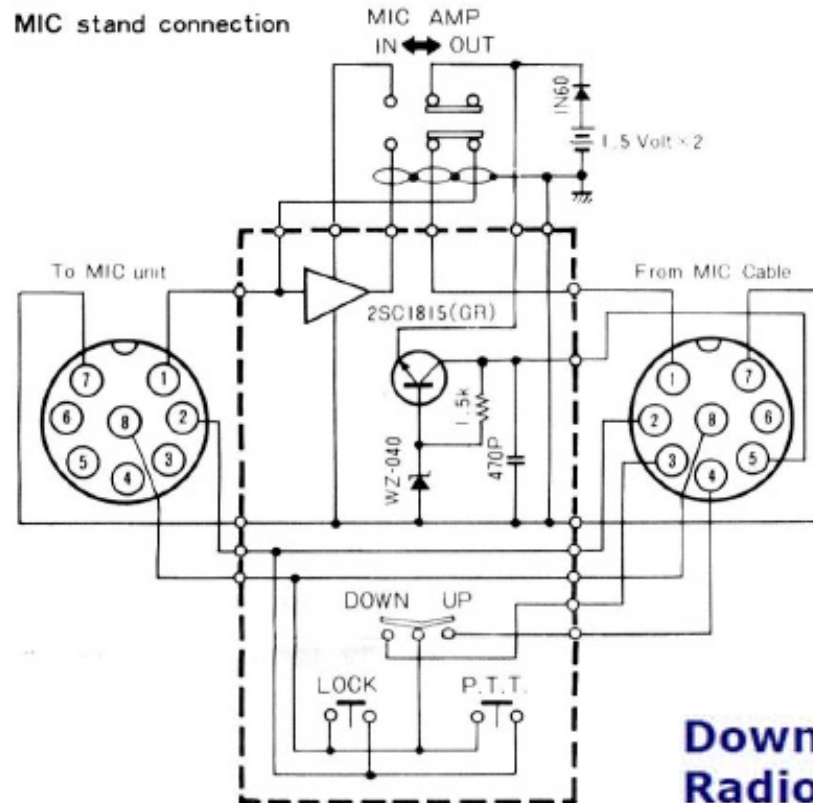
Only the socket for the Ch Up/Down and PTT is retained. The Black, Red, Blue and Green wires remain attached to plug, the rest can be cut at the plug end. Dont cut them at the MIC cable end.

The Amp conditioner, can be constructed on any prototype board available to you. This is what I did.



The image shows a trim pot to the left that is now replaced with fixed value resistors. The position of the components will depend on your construction.

Original Microphone base wiring



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The above diagram shows that the microphone has two ground systems, one on Pin 8 and one on Pin 7 of the MIC Cable Socket

Pin 7 is the Audio ground and Pin 8 is the logic ground, it so happens that the Logic ground is also the chassis ground (RF ground) on my radio a Kenwood TS590S, there is no physical ground for the microphone base, the microphone itself is insulated from the base by the plastic holder, this is a path for any RF to enter the audio path.

The first thing was to ground the base to the RF ground of the radio, it so happens that its Pin 8, also known as the logic ground, to this I combined the audio ground with a wire to ground the base and the shaft holding the microphone bracket using the link (LNK) Pin 7 and 8 are joined, both the microphone ground and Logic ground become one, With some radios the joining of these grounds may not be desirable, for instance if the audio uses balanced input.

Mounting the Compression Control

The 200K Ohm Pot is mounted as per the picture



Dress the cable coming from the Microphone head by cutting off the connector, then strip back the wires, the only ones you require are the shielded one. (Ground and Shield)

Don't cut the other two, just insulate them with heat shrink, could later be used to turn the preamp on or off, using the microphone head on off switch.

Wire the 200 K pot so that when the control is fully anti clockwise (Looking from the rear of the MIC you have 0 resistance (This is the 1:1 compression point)

The idea is that when facing the microphone, you turn the knob clockwise to increase compression.

Setting UP the levels

The best way is to use your ALC indication on your radio.

Turn the radio TX power as low as it can go.

Transmit in a dummy load

Set the compression control for 1:1

Now whistle or call out the word Five as loud as you can,

Adjust the Level out control for the allowed maximum ALC on your radio

Note that the ALC is a mechanism that reduces the audio level, so if your radio has a processor, turn it on and off and compare the obtained ALC levels.

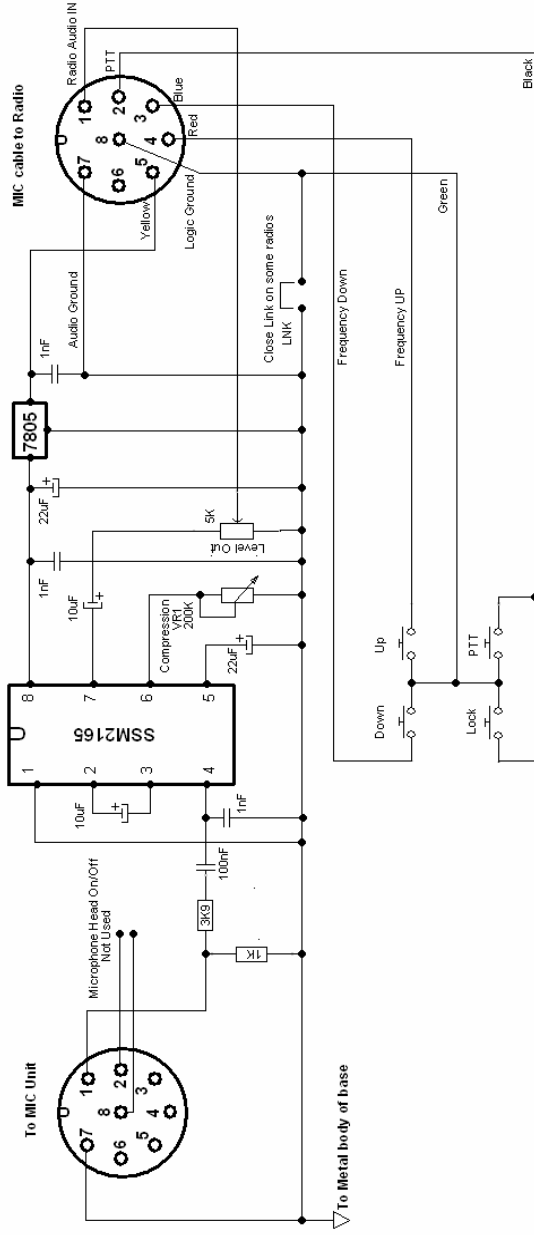
If you see that in the processor mode the ALC is reduced, this is normal, you can slightly increase the level out from the microphone. On my Radio this was 100mV RMS

Now with a normal voice look at the ALC then increase the Compressor level all the way to 15 to one, you will note that you can move back from the microphone and the ALC will remain very close to when you spoke close to the head of the microphone.

The circuit is lifting the level (Expansion) to just before the compression point, reducing compression also reduces the expansion level range.

Resist the temptation to drive more audio into the rig, all you will achieve is clipping and distortion. This mod works well, I have run 400 Watts with no hint of RF feedback, the Audio reports are good, the SSB signal produced has more punch, with no negative reports.

73's VK2YMU



Title		MC60A Modification	
Author		Robert Campiciano	
File	dsn Diagrams\Verwood 60 Compressor Mod.dsn	Document	
Revision	1.0	Date	
		Sheets	1 of 1

FEATURES

Complete Microphone Conditioner in an 8-Lead Package
Single 5 V Operation
Preset Noise Gate Threshold
Compression Ratio Set by External Resistor
Automatic Limiting Feature Prevents ADC Overload
Adjustable Release Time
Low Noise and Distortion
20 kHz Bandwidth (± 1 dB)
Low Cost

APPLICATIONS

Microphone Preamplifier/Processor
Computer Sound Cards
Public Address/Paging Systems
Communication Headsets
Telephone Conferencing
Guitar Sustain Effects Generator
Computerized Voice Recognition
Surveillance Systems
Karaoke and DJ Mixers

GENERAL DESCRIPTION

The SSM2165 is a complete and flexible solution for conditioning microphone inputs in computer audio systems. It is also excellent for improving vocal clarity in communications and public address systems. A low noise voltage controlled amplifier (VCA) provides a gain that is dynamically adjusted by a control loop to maintain a set compression characteristic. The compression ratio is set by a single resistor and can be varied from 1:1 to over 15:1 relative to the fixed rotation point. Signals above the rotation point are limited to prevent overload and to eliminate "popping." A downward expander (noise gate) prevents amplification of noise or hum. This results in optimized signal levels prior to digitization, thereby eliminating the need for additional gain or attenuation in the digital domain that could add noise or impair accuracy of speech recognition algorithms. The flexibility of setting the compression ratio and the time constant of the level detector, coupled with two values of rotation point, make the SSM2165 easy to integrate in a wide variety of microphone conditioning applications.

The SSM2165 is an ideal companion product for audio codecs used in computer systems, such as the AD1845 and AD1847. The device is available in 8-lead SOIC and P-DIP packages, and guaranteed for operation over the extended industrial temperature range of -40°C to $+85^{\circ}\text{C}$. As shown in Figure 1a, the SSM2165-1 has a rotation point of -25.7 dBu (40 mV)¹, a VCA gain of 18 dB, and gives -7.7 dBu (320 mV) before limiting. As shown in Figure 1b, the SSM2165-2 has a rotation point of -17.8 dBu (100 mV),

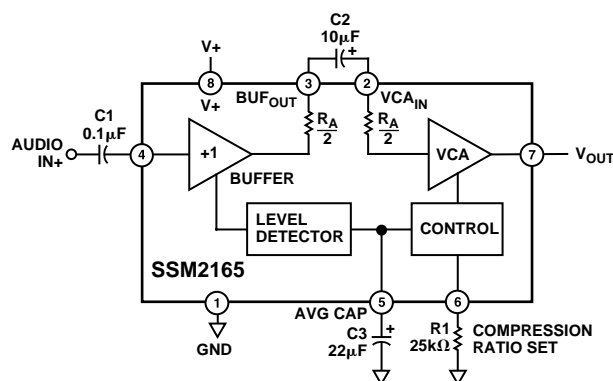
*Patents pending.

¹All signals are in rms volts or dBu (0 dBu = 0.775 V rms).

REV. B

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FUNCTIONAL BLOCK DIAGRAM



a VCA gain of 8 dB and gives -9.8 dBu (250 mV) before limiting. Both have a noise gate threshold of -64 dBu (500 μV), below which downward expansion reduces the gain with a ratio of approximately 1:3. That is, a -3 dB reduction of output signal occurs with a -1 dB reduction of input signal. For applications requiring adjustable noise gate threshold, VCA gain up to 18 dB, and adjustable rotation point, please refer to the SSM2166.

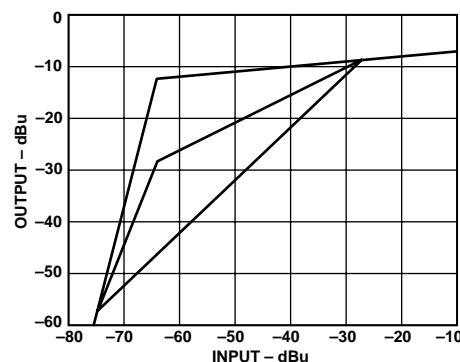


Figure 1a. SSM2165-1 Compression and Gating Characteristics

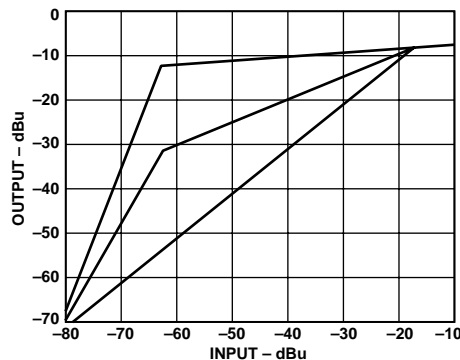


Figure 1b. SSM2165-2 Compression and Gating Characteristics

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SSM2165—SPECIFICATIONS (V₊ = 5 V, f = 1 kHz, R_L = 100 kΩ, R_{COMP} = 0 Ω, T_A = 25°C, unless otherwise noted)

Parameter	Symbol	Conditions	Min	Typ	Max	Unit
AUDIO SIGNAL PATH						
Voltage Noise Density	e _n	15:1 Compression, V _{IN} = GND		17		nV/√Hz ²
Noise		20 kHz Bandwidth, V _{IN} = GND		–109		dBu ¹
Total Harmonic Distortion	THD+N					
SSM2165-1		2nd and 3rd Harmonics, V _{IN} = –30 dBu		0.2	0.5	%
SSM2165-2		2nd and 3rd Harmonics, V _{IN} = –20 dBu		0.2	0.5	%
		22 kHz Low-Pass Filter				
Input Impedance	Z _{IN}			180		kΩ
Output Impedance	Z _{OUT}			75		Ω
Load Drive		Resistive	5			kΩ
		Capacitive			2	nF
Input Voltage Range		1% THD		1		V rms
Output Voltage Range		1% THD		1.4		V rms
Gain Bandwidth Product		1:1 Compression				
SSM2165-1		VCA G = 18 dB		300		kHz
SSM2165-2		VCA G = 8 dB		100		kHz
CONTROL SECTION						
VCA Dynamic Gain Range				40		dB
VCA Fixed Gain						
SSM2165-1				18		dB
SSM2165-2				8		dB
Rotation Point						
SSM2165-1				40		mV rms
SSM2165-2				100		mV rms
Compression Ratio, Min				1:1		
Compression Ratio, Max				15:1		
Control Feedthrough		15:1 Compression		±5		mV
POWER SUPPLY						
Supply Voltage Range	V _S		4.5		5.5	V
Supply Current	I _{SY}			7.5	10	mA
Quiescent Output Voltage Level				2.2		V
Power Supply Rejection Ratio ²	PSRR			50		dB

NOTES

¹0 dBu = 0.775 V rms.

²Referred to input.

Specifications subject to change without notice.

ABSOLUTE MAXIMUM RATINGS

Supply Voltage 10 V
 Audio Input Voltage Supply Voltage
 Operating Temperature Range -40°C to $+85^{\circ}\text{C}$
 Storage Temperature Range -65°C to $+150^{\circ}\text{C}$
 Junction Temperature (T_J) 150°C
 Lead Temperature (Soldering, 60 sec) 300°C

ESD RATINGS

883 (Human Body) Model 2.0 kV

THERMAL CHARACTERISTICS

Thermal Resistance

8-Lead Plastic DIP

θ_{JA} $103^{\circ}\text{C}/\text{W}$

θ_{JC} $43^{\circ}\text{C}/\text{W}$

8-Lead SOIC

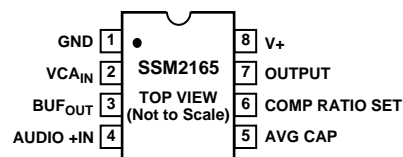
θ_{JA} $158^{\circ}\text{C}/\text{W}$

θ_{JC} $43^{\circ}\text{C}/\text{W}$

ORDERING GUIDE

Model	Temperature Range	Package Description	Package Options
SSM2165-1P*	-40°C to $+85^{\circ}\text{C}$	Plastic DIP	N-8
SSM2165-1S	-40°C to $+85^{\circ}\text{C}$	Narrow SOIC	SO-8
SSM2165-2S	-40°C to $+85^{\circ}\text{C}$	Narrow SOIC	SO-8

*Not for new design, obsolete April 2002.

PIN CONFIGURATION**PIN FUNCTION DESCRIPTIONS**

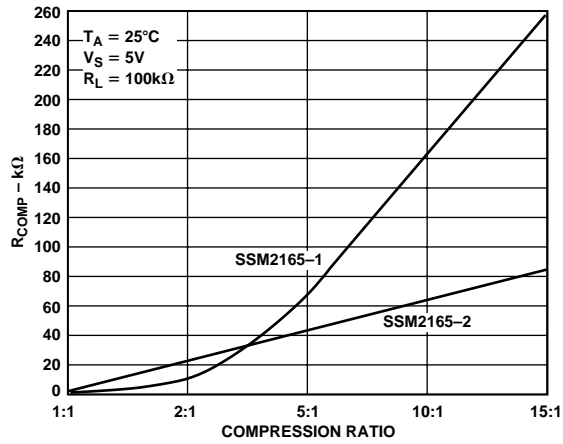
Pin #	Mnemonic	Function
1	GND	Ground
2	VCA _{IN}	VCA Input Pin. A typical connection is a 1 μF –10 μF capacitor from the buffer output pin (Pin 3) to this pin.
3	BUF _{OUT}	Input Buffer Amplifier Output Pin. Must not be loaded by capacitance to ground.
4	AUDIO +IN	Input Audio Signal. The input signal should be ac-coupled (0.1 μF typical) into this pin.
5	AVG CAP	Detector Averaging Capacitor. A capacitor, 2.2 μF –22 μF , to ground from this pin is the averaging capacitor for the detector circuit.
6	COMP RATIO SET	Compression Ratio Set Pin. A resistor to ground from this pin sets the compression ratio as shown in TPC 1.
7	OUTPUT	Output Signal.
8	V+	Positive Supply, 5 V Nominal.

CAUTION

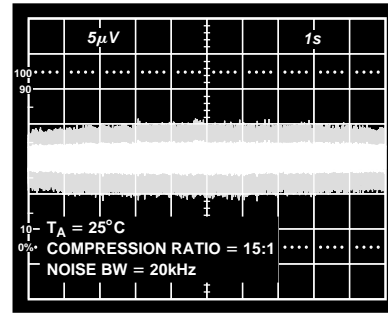
ESD (electrostatic discharge) sensitive device. Electrostatic charges as high as 4000 V readily accumulate on the human body and test equipment and can discharge without detection. Although the SSM2165 features proprietary ESD protection circuitry, permanent damage may occur on devices subjected to high-energy electrostatic discharges. Therefore, proper ESD precautions are recommended to avoid performance degradation or loss of functionality.



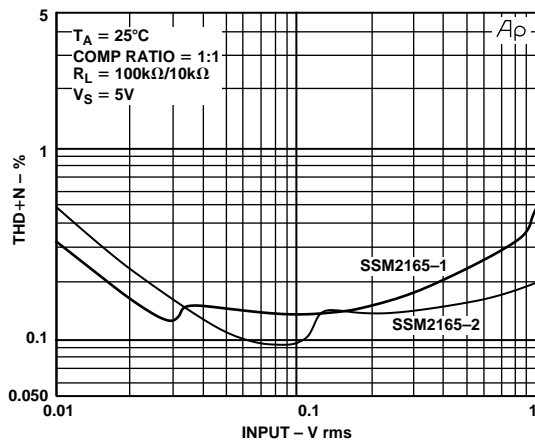
SSM2165–Typical Performance Characteristics



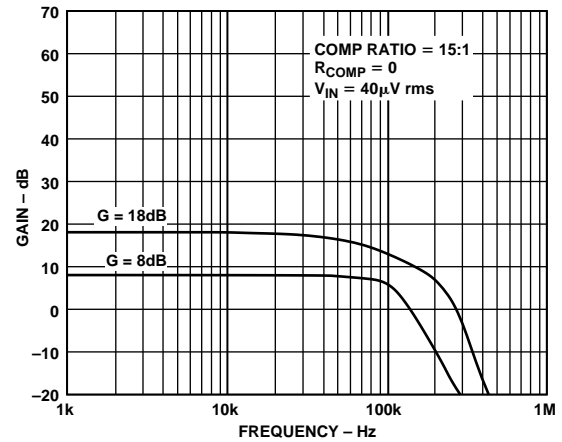
TPC 1. Compression Ratio vs. R_{COMP}



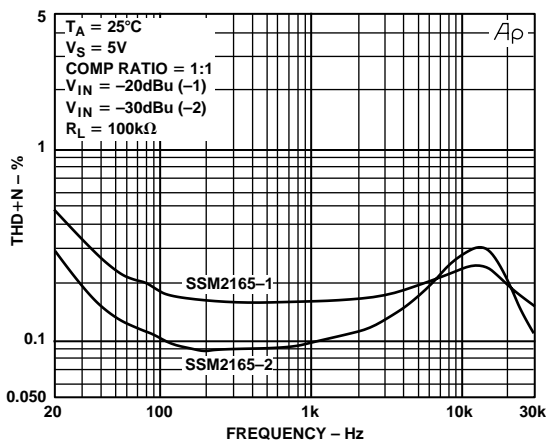
TPC 4. Wideband Output Noise



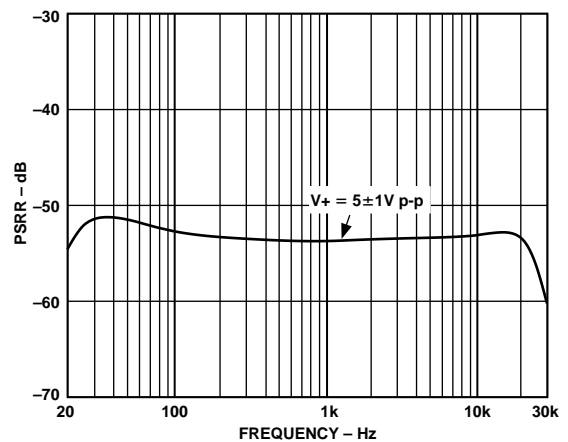
TPC 2. THD + N (%) vs. Input (V rms)



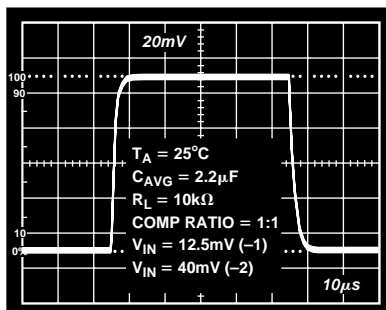
TPC 5. GBW Curves vs. VCA Gain



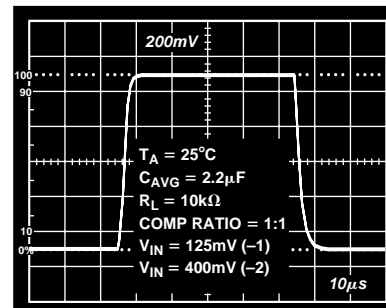
TPC 3. THD + N (%) vs. Frequency (Hz)



TPC 6. PSRR vs. Frequency, Referred to Input



TPC 7. Small Signal Transient Response



TPC 8. Large Signal Transient Response

APPLICATIONS INFORMATION

The SSM2165 is a complete microphone signal conditioning system in a single integrated circuit. Designed primarily for voiceband applications, this integrated circuit provides amplification, rms detection, limiting, variable compression, and downward expansion. The internal rms detector has a time constant set by an external capacitor. An integral voltage-controlled amplifier (VCA) provides up to 40 dB of gain in the signal path with approximately 30 kHz bandwidth. The device operates on a single 5 V supply, accepts input signals up to 1 V¹, and produces output signal levels at limiting of 320 mV and 250 mV for the SSM2165-1 and SSM2165-2 respectively, into loads > 5 kΩ.

The SSM2165 contains an input buffer and automatic gain control (AGC) circuit for audio and voice band signals. Circuit operation is optimized by providing user-adjustable compression ratio and time constant. A downward expansion (noise gating) feature reduces background and circuit noise below 500 μV. The rotation point determines the output signal levels before limiting (referred to the input), and is 40 mV for the SSM2165-1 and 100 mV for the SSM2165-2.

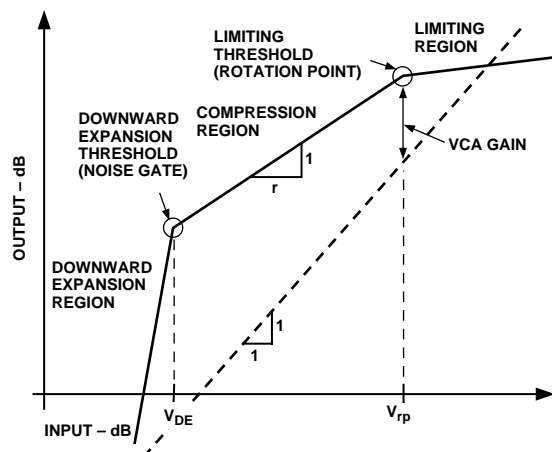


Figure 2. General Input/Output Characteristics of the SSM2165

THEORY OF OPERATION

Figure 2 illustrates the general transfer characteristic for the SSM2165 where the output level in dBu is plotted as a function of the input level in dBu (0 dBu = 0.775 V rms). For input signals in the range of V_{DE} (Downward Expansion) to V_{RP} (Rotation Point) an “r” dB change in the input level causes a 1 dB change in the output level. Here, “r” is defined as the “compression ratio.” The compression ratio may be varied from 1:1 (no compression) to over 15:1 via a single resistor, R_{COMP}. Input signals above V_{RP} are compressed with a fixed compression ratio of approximately 10:1. This region of operation is the “limiting region.” Varying the compression ratio has no effect on the limiting region. The breakpoint between the compression region and the limiting region is referred to as the “limiting threshold” or “rotation point,” and is different for the SSM2165-1 and SSM2165-2, see Table I.

Table I. Characteristics vs. Dash Number

SSM2165	Rotation Point	Gain	Output*
-1	40 mV (-25.7 dBu)	18 dB	320 mV (-6 dBu)
-2	100 mV (-17.7 dBu)	8 dB	250 mV (-8 dBu)

*At limiting.

The term “rotation point” derives from the observation that the straight line in the compression region “rotates” about this point on the input/output characteristic as the compression ratio is changed.

When the compression is set to 2:1, a -2 dB change of the input signal level in the compression region causes -1 dB change of the output level. Likewise, at 10:1 compression, a -10 dB change of the input signal level in the compression region causes a -1 dB change in the output level. The gain of the system with an input signal level of V_{RP} is fixed regardless of the compression ratio, and is different for the SSM2165-1 and SSM2165-2 (see Figures 1a and 1b). The “nominal gain” of the system is 18 dB for the SSM2165-1, and 8 dB for the SSM2165-2. System gain is measured at V_{RP} and is (V_{OUT} - V_{IN}) in dB.

Input signals below V_{DE} are downward expanded at a ratio of approximately 1:3. As a result, the gain of the system is small for very small input signal levels below V_{DE}, even though it may be quite large for input signals above V_{DE}. The downward expansion threshold, V_{DE}, is fixed at 500 μV (-64 dBu) for both dash versions.

¹All signals are in rms volts or dBu (0 dBu = 0.775 V rms).

SSM2165

The SSM2165 Signal Path

Figure 3 illustrates the block diagram of the SSM2165. The audio input signal is processed by the unity gain input buffer and then by the VCA. The buffer presents an input impedance of approximately 180 k Ω to the source. A dc voltage of approximately 1.5 V is present at AUDIO +IN (Pin 4), requiring the use of a blocking capacitor (C1) for ground-referenced sources. A 0.1 μ F capacitor is a good choice for most audio applications. The buffer is designed to drive only the low impedance input of the VCA, and must not be loaded by capacitance to ground. The VCA is a low distortion, variable-gain amplifier whose gain is set by the internal control circuitry. The input to the VCA is a virtual ground in series with 500 Ω . An external blocking capacitor (C2) must be used between the buffer's output and the VCA input. The desired low frequency response and the total of 1 k Ω impedance between amplifiers determine the value of this capacitor. For music applications, 10 μ F will give high pass $f_c = 16$ Hz. For voice/communications applications, 1 μ F will give $f_c = 160$ Hz. An aluminum electrolytic capacitor is an economical choice. The VCA amplifies the input signal current flowing through C6 and converts this current to a voltage at the SSM2165's output (Pin 7). The net gain from input to output can be as high as 40 dB for high compression ratios and depending on the gain set by the control circuitry. The output impedance of the SSM2165 is typically less than 75 Ω , and the external load on Pin 7 should be >5 k Ω . The nominal output dc voltage of the device is approximately 2.2 V. Use a dc blocking capacitor for grounded loads.

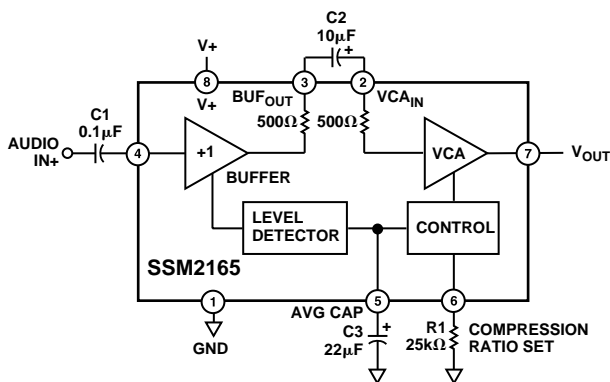


Figure 3. Functional Block Diagram and Typical Voice Application

The bandwidth of the SSM2165 is quite wide at all gain settings. The upper -3 dB point is approximately 300 kHz. The GBW plots are shown in TPC 5. While the noise of the input buffer is fixed, the input referred noise of the VCA is a function of gain. The VCA input noise is designed to be a minimum when the gain is at a maximum, thereby optimizing the usable dynamic range of the part. A photograph of the SSM2165's wideband peak-to-peak output noise is illustrated in TPC 4.

The Level Detector

The SSM2165 incorporates a full-wave rectifier and a patent-pending, true rms level detector circuit whose averaging time constant is set by an external capacitor connected to the AVG CAP pin (Pin 5). Capacitor values from 18 μ F to 22 μ F have been found to be more appropriate in voiceband applications, where capacitors on the low end of the range seem more appropriate for music program material. For optimal low frequency

operation of the level detector down to 10 Hz, the value of the capacitor should be around 22 μ F. Some experimentation with larger values for the AVG CAP may be necessary to reduce the effects of excessive low frequency ambient background noise. The value of the averaging capacitor affects sound quality: too small a value for this capacitor may cause a "pumping effect" for some signals, while too large a value can result in slow response times to signal dynamics. Electrolytic capacitors are recommended here for lowest cost.

The rms detector filter time constant is approximately given by $10 \times C_{AVG}$ milliseconds where C_{AVG} is in μ F. This time constant controls both the steady-state averaging in the rms detector as well as the release time for compression, that is, the time it takes for the system gain to react when a large input is followed by a small signal. The attack time, the time it takes for the gain to be reduced when a small signal is followed by a large signal, is mainly controlled by internal circuitry that speeds up the attack for large level changes, and controlled partly by the AVG CAP value. This limits overload time to under 1 ms in most cases.

The performance of the rms level detector is illustrated in Figure 4 for $C_{AVG} = 2.2 \mu$ F and Figure 5 for $C_{AVG} = 22 \mu$ F. In each of these photographs, the input signal to the SSM2165 (not shown) is a series of tone bursts in six successive 10 dB steps. The tone bursts range from -66 dBu (0.5 mV rms) to -6 dBu (0.5 V rms). As illustrated in the photographs, the attack time of the rms level detector is dependent only on C_{AVG} , but the release times are linear ramps whose decay times are dependent on both for C_{AVG} and the input signal step size. The rate of release is approximately 240 dB/s for a $C_{AVG} = 2.2 \mu$ F, and 12 dB/s for a C_{AVG} of 22 μ F.

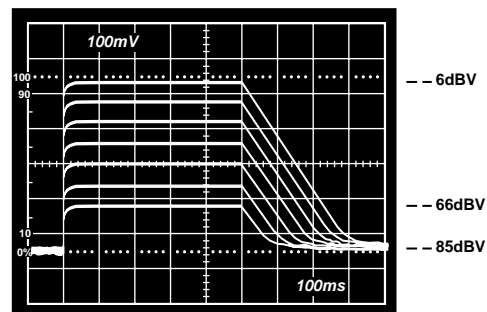


Figure 4. RMS Level Detector Performance with $C_{AVG} = 2.2 \mu$ F

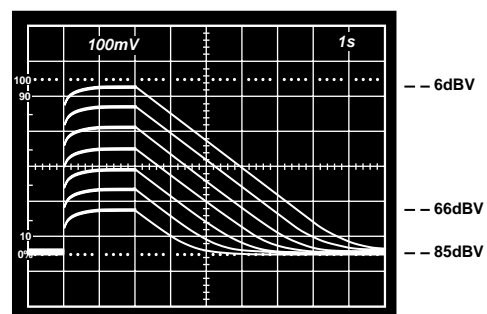


Figure 5. RMS Level Detector Performance with $C_{AVG} = 22 \mu$ F

Control Circuitry

The output of the rms level detector is a signal proportional to the log of the true rms value of the buffer output with an added dc offset. The control circuitry subtracts a dc voltage from this signal, scales it, and sends the result to the VCA to control the gain. The VCA's gain control is logarithmic: a linear change in control signal causes a dB change in gain. It is this control law that allows linear processing of the log rms signal to provide the flat compression characteristic on the input/output characteristic shown in Figure 2.

Compression Ratio

Changing the scaling of the control signal fed to the VCA causes a change in the circuit's compression ratio, "r." This effect is shown in Figure 6. The compression ratio can be set by connecting a resistor between the COMP RATIO pin (Pin 6) and GND. Lowering RCOMP gives smaller compression ratios as indicated in TPC 1, with values of about 5 k Ω or less resulting in a compression ratio of 1:1. AGC performance is achieved with compression ratios between 2:1 and 15:1, and is dependent on the application. A 200 k Ω potentiometer may be used to allow this parameter to be adjusted.

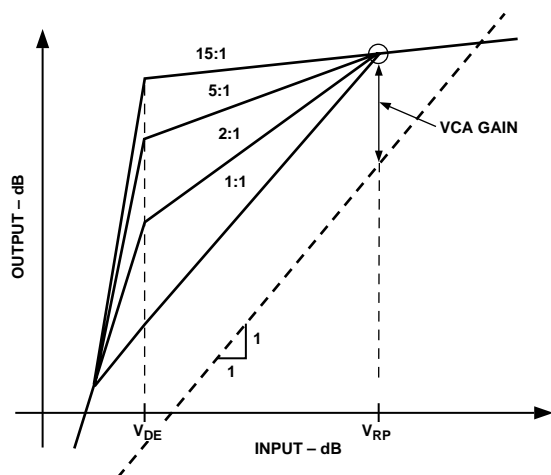


Figure 6. Effect of Varying the Compression Ratio

Rotation Point

An internal dc reference voltage in the control circuitry sets the rotation point. The rotation point determines the output level above which limiting occurs. That is, in the limiting region, a 10 dB change of input results in a 1 dB change of output. The rotation point is set to 40 mV (–26 dBu) for the SSM2165-1 and 100 mV (–18 dBu) for the SSM2165-2. In the SSM2165, limiting is compression at a fixed compression ratio of approximately 15:1. The fixed gain in the VCA is 18 dB for the SSM2165-1 and 8 dB for the SSM2165-2. The output signals at limiting are, therefore, 320 mV and 250 mV respectively. These are summarized in Table I.

Maximum Output

Since limiting occurs for signals larger than the rotation point ($V_{IN} > V_{RP}$), the rotation point effectively sets the maximum output signal level. The application will determine which version of the SSM2165 should be selected. The output level should match the maximum input allowed by the following stage. Occasional larger signal transients will then be attenuated by the action of the limiter.

Downward Expansion Threshold

The downward expansion threshold, or noise gate, is determined by a reference voltage internal to the control circuitry. The noise gate threshold is 500 μ V for both versions of the SSM2165. Users requiring some other noise gate should consider using the SSM2166. High volume users may wish to consider a custom version of the SSM2165 with other noise gate thresholds or rotation points.

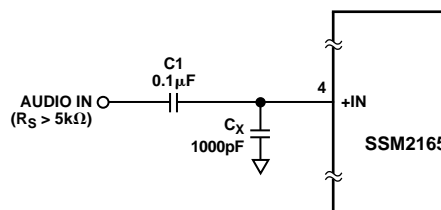
Power-On/Power-Off Settling Time

Cycling the power supply to the SSM2165 will result in quick settling times: the off-on settling time of the SSM2165 is less than 200 ms, while the on-off settling time is less than 1 ms. Note that transients may appear at the output of the device during power up and power down. A clickless mute function is available on the SSM2166 only.

PC Board Layout Considerations

Since the SSM2165 is capable of wide bandwidth operation at high gain, special care must be exercised in the layout of the PC board which contains the IC and its associated components. The following applications hints should be considered and/or followed:

1. In some high system gain applications, the shielding of input wires to minimize possible feedback from the output of the SSM2165 back to the input circuit may be necessary.
2. A single-point ("star") ground implementation is recommended in addition to maintaining short lead lengths and PC board runs. In systems where an analog ground and a digital ground are available, the SSM2165 and its surrounding circuitry should be connected to the analog ground. Wire-wrap board connections and grounding implementations are to be explicitly avoided.
3. The internal buffer of the SSM2165 was designed to drive only the input of the internal VCA and its own feedback network. Stray capacitive loading to ground from either Pin 3 or Pin 2 in excess of 5 pF to 10 pF can cause excessive phase shift and can lead to circuit instability.
4. When using high impedance sources, it can be advantageous to shunt the source with a capacitor to ground at the input pin of the IC (Pin 4) to lower the source impedance at high frequencies, as shown in Figure 7. A capacitor with a value of 1000 pF is a good starting value and sets a low pass corner at 31 kHz for 5 k Ω sources.



NOTE: ADDITIONAL CIRCUIT DETAILS OMITTED FOR CLARITY.

Figure 7. Circuit Configuration for Use with High Impedance Signal Sources

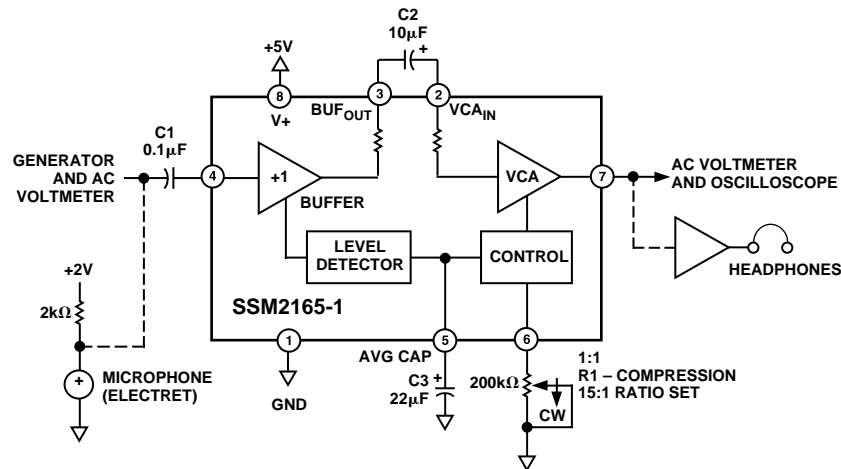


Figure 8. Electret Microphone Preamp Example

Compression Adjustment—A Practical Example

To illustrate how to set the compression ratio of the SSM2165, we will take a practical example. The SSM2165 will be used interface an electret-type microphone to a post-amplifier, as shown in Figure 8. The signal from the microphone was measured under actual conditions to vary from 2 mV to 30 mV. The post-amplifier requires no more than 350 mV at its input. We will, therefore, choose the SSM2165-1, whose “rotation” point is 40 mV and whose VCA fixed gain is 18 dB (×8), thus giving 320 mV at limiting. From prior listening experience, we will use a 2:1 compression ratio. The noise gate threshold of the SSM2165-1 will operate when the input signal falls below 500 µV. These objectives are summarized in Table II. The transfer characteristic we will implement is illustrated in Figure 10.

Table II. Objective Specification of Example

Input Range	2 mV–30 mV
Output Range	To 350 mV
Limiting Level	320 mV
Compression	2:1
Gain	18 dB
Noise Gate	500 µV

Test Equipment Setup

The recommended equipment and configuration is shown in Figure 9. A low noise audio generator with a smooth output adjustment range of 100 µV to 25 mV is a suitable signal source. The output voltmeter should go up to 2 V. The oscilloscope is used to verify that the output is sinusoidal, that no clipping is occurring in the buffer, and to observe the limiting and noise gating “knees.”

Breadboard Considerations

When building a breadboard, keep the leads to Pins 2 and 3 as short as possible. Use a central analog ground and decouple power supply connections adequately.

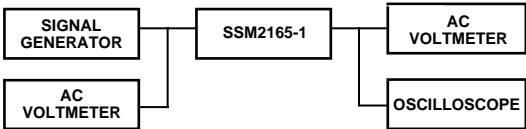


Figure 9. Test Equipment Setup

STEP 1. Initialize Potentiometer

With power off, preset R1—Compression Ratio potentiometer to zero ohms.

STEP 2. Check Setup

With power on, adjust the generator for an input level of 50 mV (–24 dBu), 1 kHz. The output meter should indicate approximately 350 mV (–6.9 dBu). If not, check the setup.

STEP 3. Find the Rotation Point

Set the input level to 50 mV (–24 dBu), and observe the output on the oscilloscope. The output will be in the limiting range of operation. Slowly reduce the input signal level until the output level just begins to stop limiting and follows the input down. Increase the input so that the output is 320 mV (–7.7 dBu). This is the knee of the rotation point.

STEP 4. Adjust the Compression Ratio

With the input set as in Step 3, note the exact value of the input signal level just below the knee (around 40 mV (–26 dBu)). Next, reduce the input to 1/4 the value noted, (around 10 mV (–38 dBu)), for a change of –12 dB. Next, increase the R_{COMP} potentiometer resistance so the output is 160 mV (–13.7 dBu) for an output change of –6 dB. You have now set the compression, which is the ratio of input change to output change, in dB, to 2:1.

STEP 5. Confirm the Noise Gate Threshold

Set the input to 1 mV, and observe the output on the oscilloscope. A 20 dB pad between generator and input may facilitate this measurement. Reduce the input gradually until the output falls off more rapidly. This point is the noise gate threshold, and should be approximately 500 µV (–64 dBu). The noise gate threshold on the SSM2165 is fixed at 500 µV, a practical value for many microphones. Should a different noise gate threshold be required, consider using the SSM2166.

STEP 6. Listen

At this time the signal generator may be replaced with a properly powered electret microphone and listen to the results through a set of headphones. The microphone's internal FET usually requires around 2 V through a 2 k Ω resistor; this varies with the manufacturer. Experiment with the compression ratio value and averaging capacitor size. More compression will keep the output steady over a wider range of microphone-to-source distance. Varying the averaging capacitor, C_{AVG} , changes the rms detector averaging time, and the decay time of the gate. Both compression ratio and decay time are usually determined by critical listening to the intended audio input.

STEP 7. Record Values

With the power removed from the test fixture, measure and record the values of the R_{COMP} and C_{AVG} .

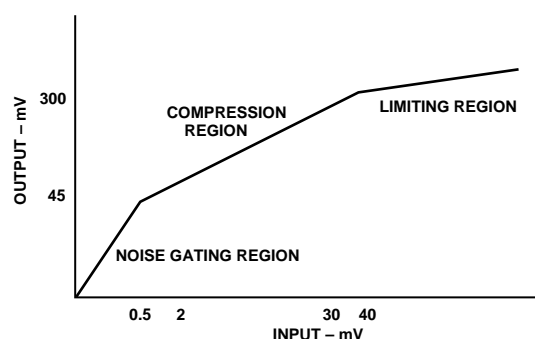


Figure 10. Transfer Characteristic

SUMMARY

We have implemented the transfer characteristic of Figure 10. For inputs below the 500 μ V noise gate threshold, circuit and background noise will be downward expanded (gain-reduced) at a ratio of approximately 1:3. That is, a -1 dB change in the noise will result in -3 dB decrease at the output. Above threshold, the signal will increase at a rate of 1 dB for each 2 dB input increase, until the rotation point is reached at an input of approximately 40 mV. In the limiting region, the compression ratio increases to approximately 15:1. That is, a 15 dB increase in input will produce a 1 dB increase at the output, so there will be little further increase for higher level inputs.

Other Versions

The SSM2165 is an 8-lead version of the 14-lead SSM2166 which is recommended for applications requiring more versatility. The SSM2166 allows selection of noise gate threshold and rotation point, and allows the buffer to provide up to 20 dB of gain. Power-down and mute functions are also built in. Customized versions of the SSM2165 are available for large volume users. The wide dynamic range of the SSM2165 makes it useful in many applications other than microphone signal conditioning such as a sustain generator for guitars. For further information, contact your Analog Devices representative.

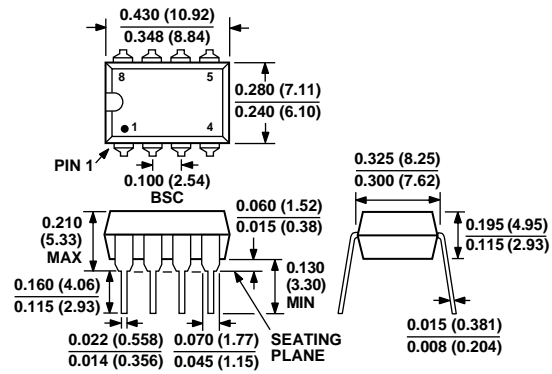
SSM2165

Revision History

Location	Page
Data Sheet changed from REV. A to REV. B.	
Edits to ORDERING GUIDE	3

OUTLINE DIMENSIONS

Dimensions shown in inches and (mm).

**8-Lead Plastic DIP
(N-8)****8-Lead Narrow-Body SOIC
(SO-8)**