

Using the ADAU1381 Sound Engine

INTRODUCTION

This user guide explains the signal flow and parameter settings for the ADAU1381 sound engine. The ADAU1381 is ideal for low power portable applications, such as digital camera audio. During the recording or playing back of audio, the sound engine provides many signal processing features to improve audio quality.

DIGITAL CAMERA SYSTEM OVERVIEW

Although the ADAU1381 is flexible enough to be used in several types of portable audio applications, its design specifically targets digital camera systems. The sound processing engine was, therefore, designed especially with such a system in mind. In general, digital cameras use audio processing when recording or playing back video. When recording, one or more microphones mounted in the camera or connected externally capture the audio data, which is then stored in the memory along with the video data. During playback or review mode, the audio data is retrieved from memory and played back through a speaker mounted in the camera or through a jack for headphones or other external connections.

In record mode, the source is an audio transducer (microphone) and the target is memory. In playback mode, the source is memory and the target is an audio transducer (speaker). In both modes, the sound engine is positioned between the source and target, processing the signal to improve audio quality.

Because the required audio processing differs depending on the operating mode of the camera, several audio processing modes have been implemented in the sound engine of the ADAU1381.

AUDIO PROCESSING MODES

Record Mode

Record mode takes audio input from a microphone. Wind noise reduction is applied to remove unwanted noise from the signal and improve audio clarity. The enhanced stereo capture algorithm provides improved stereo separation when microphones are spaced close together. The six-band equalizer can be programmed to augment bands of interest and filter out unwanted frequencies. The dual-band dynamics processor acts as an automatic level control, compensating for fluctuating input signal levels. The processed signal is output to a digital storage medium.

Two record modes exist: Record A (**REC A**) and Record B (**REC B**). The only differences between the two modes are the six-band equalizer and the dual-band dynamics processor settings. The two record modes allow for different audio recording profiles, such as voice or music. The recording profile can be changed by a single write to the RAM parameter.

Playback Mode

Playback mode takes audio input from the digital storage. The six-band equalizer is used for frequency compensation with the output speaker or headphones. The dual-band dynamics processor acts as a compressor, allowing for suitable playback levels even in noisy environments. The playback output includes a digital volume control for output level adjustment.

SOUND ENGINE SIGNAL FLOW BLOCK DIAGRAM

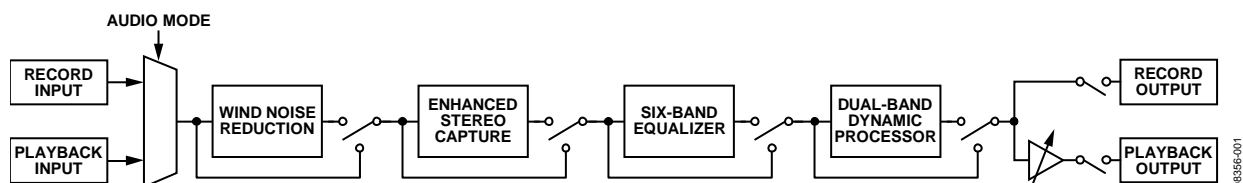


Figure 1.

TABLE OF CONTENTS

| | | | |
|--|---|---|----|
| Introduction | 1 | Outputs and Mute | 4 |
| Digital Camera System Overview | 1 | Mode Selection | 4 |
| Audio Processing Modes | 1 | Main Page | 4 |
| Sound Engine Signal Flow Block Diagram | 1 | Wind Noise Reduction Page | 6 |
| Revision History | 2 | Enhanced Stereo Capture Page..... | 8 |
| SigmaStudio Interface to the Sound Engine | 3 | Equalization Filters Page | 9 |
| SigmaStudio Interface | 3 | Dual-Band Compression Page..... | 14 |
| ADAU1381 Power-Up Sequence..... | 3 | SigmaStudio Tools | 27 |
| Connecting the ADAU1381 Evaluation Board to the Computer..... | 3 | Changing Sample Rate..... | 27 |
| Editing the Signal Flow..... | 3 | Capture Window | 27 |
| Controlling Parameters in Real Time..... | 3 | Parameter Visualization Window | 27 |
| Output File Generation..... | 3 | Sequence Window..... | 27 |
| Sound Engine Signal Processing Flow..... | 4 | Export Parameter and Register Settings..... | 28 |
| Description..... | 4 | SigmaStudio Help File | 28 |
| Inputs | 4 | Full Parameter Map..... | 29 |

REVISION HISTORY

11/09—Revision 0: Initial Version

SIGMASTUDIO INTERFACE TO THE SOUND ENGINE

SIGMASTUDIO INTERFACE

SigmaStudio™ is a software tool that allows the user to configure the registers and parameters of the ADAU1381 via a graphical user interface. SigmaStudio can communicate directly with target hardware via the EVAL-ADUSB2EBZ board, also known as the USBi, which uses the I²C® and SPI communications protocols. The ADAU1381 evaluation board is configured for use with the USBi. Prototype hardware can also be configured for a USBi connection using a 10-pin communications header.

More information on the USBi can be found in the AN-1006 application note at www.analog.com.

ADAU1381 POWER-UP SEQUENCE

When power is supplied to the ADAU1381, a boot sequence is initiated to clear the memory to a default state. When the boot sequence is complete, all of the sound engine parameters are set to 0. The parameters in the ADAU1381 memory do not match the values shown in SigmaStudio until they are overwritten.

CONNECTING THE ADAU1381 EVALUATION BOARD TO THE COMPUTER

To connect the ADAU1381 to the computer, complete the following steps:

1. Install SigmaStudio; refer to the evaluation board documentation for step-by-step instructions.
2. Set up the USBi and ADAU1381 evaluation board as described in the evaluation board documentation.
3. Connect the USBi to the PC with a USB cable and install the drivers as described in the AN-1006 application note.
4. Connect the communications ribbon cable to the target ADAU1381 board to initiate the built-in hardware self-boot function of the ADAU1381.
5. Run SigmaStudio.
6. Open the ADAU1381.dspproj file, which is located in the SigmaStudio program directory.
7. Write registers and parameters from SigmaStudio to the hardware to enable the audio signal paths. To download all parameters for the ADAU1381.dspproj file at once, click **Link-Compile-Download** in the main toolbar.

EDITING THE SIGNAL FLOW

The signal flow of the ADAU1381 is fixed function. The corresponding SigmaStudio project file is locked. Therefore, no cells can be added to or deleted from the project. Only the parameters and register settings can be modified.

CONTROLLING PARAMETERS IN REAL TIME

SigmaStudio can be used for real-time tuning of the evaluation board or a production system via the USBi control interface. The method for changing the parameters of each cell is described in the help documentation for that cell.

New parameter values should always be generated within the SigmaStudio tool. The default minimum and maximum limits for each control should be obeyed.

OUTPUT FILE GENERATION

SigmaStudio includes built-in code and header file generation tools that can greatly simplify integration in the host controller of a target system. Parameter values and register settings can easily be exported via the **Export System Files** command in SigmaStudio to C-compatible output files.

SOUND ENGINE SIGNAL PROCESSING FLOW

The sound engine processing flow of the ADAU1381 is partitioned into multiple hierarchy pages in the SigmaStudio tool. In this section, each page and its corresponding controls and parameters are described in detail.

DESCRIPTION

The main page presents an overview of the signal flow, with the processing blocks of the sound engine presented as hierarchy boards. Using the main page controls, the audio modes and output volumes can be modified.

INPUTS

There are four audio inputs to the sound engine: **Record Input 0**, **Record Input 1**, **Playback Input 0**, and **Playback Input 1**. The source of the signals on the record inputs is the ADCs or digital microphones. **Record Input 0** comes from the left ADC or **Digital Microphone Input 1** (the LMIC/LMICN/MICD1 pin), and **Record Input 1** comes from the right ADC or **Digital Microphone Input 2** (the RMIC/RMICN/MICD2 pin). The inputs to the playback path are from the digital serial data interface. **Digital Serial Input 0** (the left channel of the DAC_SDATA/GPIO0 pin) is connected to **Playback Input 0**, and **Digital Serial Input 1** (the right channel of the DAC_SDATA/GPIO0 pin) is connected to **Playback Input 1**.

These two input pairs are routed to the subsequent processing blocks based on the mode selections. In **REC A** and **REC B** modes, the record input pair is routed through the processing algorithms; in playback mode, the playback input pair is routed through the processing algorithms.

MAIN PAGE

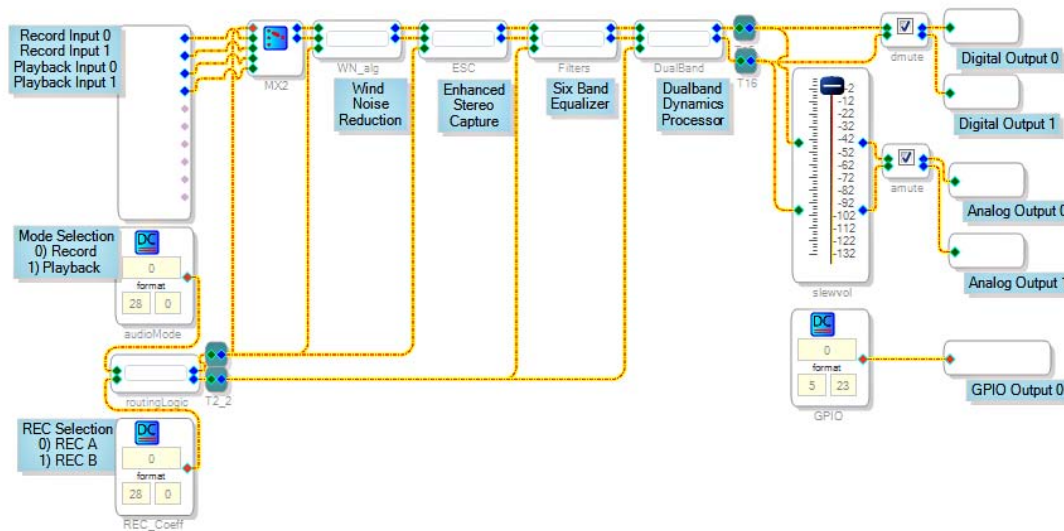


Figure 2. Main Page

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OUTPUTS AND MUTE

There are four audio outputs from the sound engine: **Record Output 0**, **Record Output 1**, **Playback Output 0**, and **Playback Output 1**. The record output signals (also labeled as **Digital Output 0** and **Digital Output 1**) are sent to the digital serial data interface, and the playback output signals (also labeled as **Analog Output 0** and **Analog Output 1**) go to the DACs of the ADAU1381. **Playback Output 0** is sent to the left DAC, and **Playback Output 1** is sent to the right DAC.

The digital and analog outputs have separate mute settings. In SigmaStudio, each of these is enabled by checking the appropriate box for the mute control.

There is a single flag in the sound engine that outputs a high or a low logic signal on the GPIO pin of the ADAU1381. This output is set by writing either a 0 or a 1 to the **GPIO** parameter.

MODE SELECTION

The sound engine can be put into three modes: **REC A** (Record A), **REC B** (Record B), or **Playback**. Using the settings on the two mode selection blocks, the routing logic properly configures the signal flow for the selected mode. The parameter settings for each mode are shown in Table 1.

Table 1. Record/Playback Modes

| Mode | Mode Selection | REC Selection |
|------------------|----------------|---------------|
| REC A (Record A) | 0 | 0 |
| REC B (Record B) | 0 | 1 |
| Playback | 1 | Don't care |

Controls

Set the audio mode by typing 0 or 1 into the **audioMode** cell in the default 28.0 format (see Figure 3). More information on 28.0 and other numeric formats can be found in the Numeric Formats section of the SigmaStudio help file.

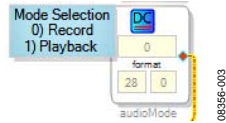


Figure 3. audioMode Control

Record Mode A (**REC A**) or Record Mode B (**REC B**) can be selected by typing 0 or 1 into the **REC_Coeff** cell in the default 28.0 format (see Figure 4).

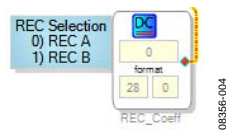


Figure 4. REC_Coeff Control

The playback (analog) output volume can be adjusted using the **slewwol** cell. Click and drag the slider to select a value (see Figure 5).

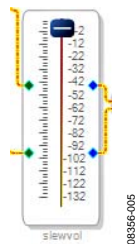


Figure 5. slewwol Control

Click on the slider to type the value in directly (see Figure 6).

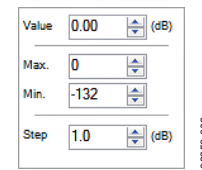


Figure 6. slewwol Control Direct Value Entry

Click the **dmute** cell to disable the record (digital) output (see Figure 7). A check corresponds to a mute setting.



Figure 7. dmute Control

Click the **amute** cell to disable the playback (analog) output (see Figure 8). A check corresponds to a mute setting.



Figure 8. amute Control

To manually toggle the GPIO output, type a value into the **GPIO** cell (see Figure 9). This value is in 5.23 format. More information on 5.23 and other numeric formats can be found in the Numeric Formats section of the SigmaStudio help file.



Figure 9. GPIO Control

Table 2. Main Page Control Settings

| Setting Name | Description | Default | Control Type |
|--------------|--|---------|----------------------|
| audioMode | Record/playback selection | 0 | Function selection |
| REC_Coeff | Selects REC A or REC B path | 0 | Function selection |
| slewwol | Analog volume control with slew | 0 dB | Processing parameter |
| dmute | Digital output mute using slew | Enabled | Processing parameter |
| amute | Analog output mute using slew | Enabled | Processing parameter |
| GPIO | Sets the GPIO pin high/low (active high) | 0 | Processing parameter |

Parameters

The main page parameters are stored in RAM, as outlined in Table 3. These addresses can be directly accessed and modified via the control port of the ADAU1381.

Table 3. Main Page Parameters

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|----------------|-----------|-----------------------------|--|-------------------------------------|-------|------------------------|
| 0x0009 | audioMode | DCInpAlg1 | 0x00, 0x00, 0x00, 0x00 | Set record/playback mode | 4 | No |
| 0x000A | REC_Coeff | DCInpAlg3 | 0x00, 0x00, 0x00, 0x00 | Set record mode A or B | 4 | No |
| 0x000B | GPIO | DCInpAlg4 | 0x00, 0x00, 0x00, 0x00 | Set GPIO output flag | 4 | No |
| 0x01B8 | slewwol | GainS200AlgGrow1gain_target | 0x00, 0x80, 0x00, 0x00 | Analog output volume control | 4 | No |
| 0x07FA, 0x07FB | slewwol | GainS200AlgGrow1alpha | 0x00, 0x7F, 0xF2, 0x59, 0x00, 0x00, 0x0D, 0xA7 | Slew rate for analog volume control | 8 | Yes |
| 0x01B6 | dmute | MuteSWLinSlewAlg1mute | 0x00, 0x00, 0x00, 0x00 | Mute digital (record) output | 4 | No |
| 0x01B7 | dmute | MuteSWLinSlewAlg1step | 0x00, 0x00, 0x40, 0x00 | Slew rate for digital mute | 4 | Yes |
| 0x01BA | amute | MuteSWLinSlewAlg2mute | 0x00, 0x00, 0x00, 0x00 | Mute analog (playback) output | 4 | No |
| 0x01BB | amute | MuteSWLinSlewAlg2step | 0x00, 0x00, 0x40, 0x00 | Slew rate for analog mute | 4 | Yes |

WIND NOISE REDUCTION PAGE

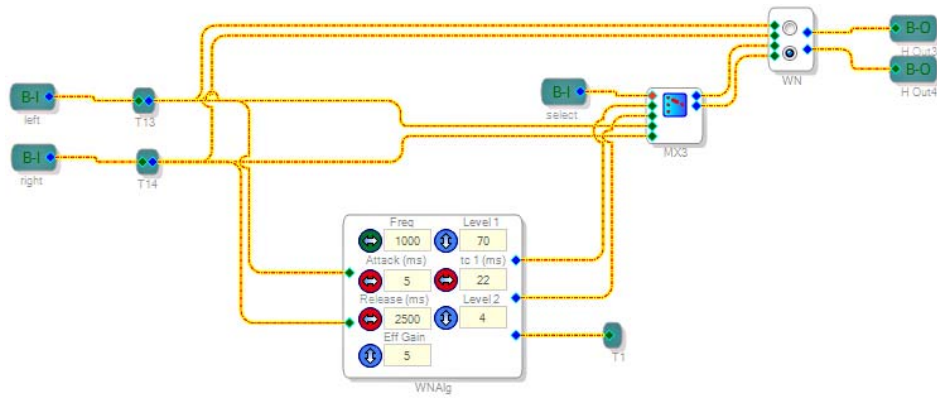


Figure 10. Wind Noise Reduction Page

Description

The wind noise reduction page houses the wind noise reduction algorithm, which uses two microphones to detect and filter wind noise from the audio signal. Wind noise can easily overwhelm an audio recording; this reduction algorithm can be used to lower the effect and increase the clarity of the signal to be recorded. The algorithm works by detecting the presence of wind noise and smoothly enabling or disabling a high-pass filter that removes the noise from the signal. Much of the wind noise that the microphones pick up is at low frequencies; therefore, the cutoff frequency of the high-pass filter should be adjusted to adequately remove the unwanted noise.

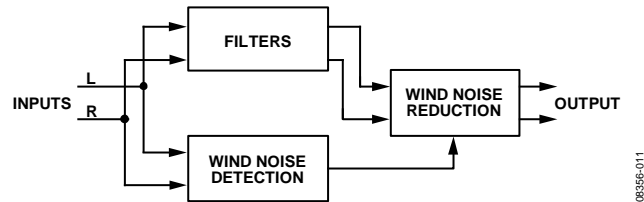


Figure 11. Wind Noise Reduction Block Diagram

Routing and Bypass

The wind noise reduction processing path is automatically enabled on the multiplexer (MX3) when the sound engine is put into either Record Mode A or Record Mode B. When in playback mode, this multiplexer bypasses the wind noise reduction algorithm. The switch on this page (WN) can be used to manually bypass the wind noise reduction, even in the record modes, if desired.

Controls

Three controls are recommended for in-system tuning: frequency (Freq), Level 1, and Level 2.

The frequency control sets the detector filters. This parameter should be tuned so that wind noise is removed, but the desired audio signal is preserved. The frequency parameter should be tuned while the system is presented with a constant wind noise,

such as from a fan blowing across, not directly onto, the microphones. The value can be entered by clicking the up/down arrows or by entering text directly in the box.

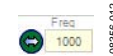


Figure 12. Freq Control

Level 1 should be tuned while turning the wind source on and off and simultaneously tuning the parameter setting between 0 and 100. The Level 1 setting is recommended to be between 60 and 90, but this varies depending on the application. The value can be entered by clicking the up/down arrows or by entering text directly in the box.



Figure 13. Level 1 Control

Level 2 should be tuned in the same way as Level 1; its settings range from 0 to 15, with 0 being for strong wind noise and 15 being for a signal with a weak wind noise component. The value can be entered by clicking the up/down arrows or by entering text directly in the box.



Figure 14. Level 2 Control

The WN switch manually enables or bypasses the algorithm independently of multiplexer MX3, which allows the algorithm to be disabled even when a record mode is active. The switch can be changed by clicking on the appropriate radio button.

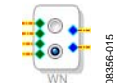


Figure 15. WN Control

Table 4. Wind Noise Reduction Page Control Settings

| Setting Name | Description | Default | Control Type |
|------------------|--|------------------|--------------------|
| Freq | High-pass filter setting | 1000 | Tune |
| Attack (ms) | Wind noise reduction effect attack time | 5 | Use default |
| Release (ms) | Wind noise reduction effect release time | 2500 | Use default |
| Eff Gain | Effect gain | 5 | Use default |
| tc 1 (ms) | Time constant | 22 | Use default |
| Level 1 | Level of wind noise reduction | 70 | Tune |
| Level 2 | Wind noise strength (0 = strong, 15 = weak) | 4 | Tune |
| WN Switch Bypass | Switch to disable algorithm | Enable algorithm | Function selection |
| MX3 Mux Bypass | Switch to bypass algorithm (via multiplexer) | Enable algorithm | Function selection |

Parameters

The wind noise reduction page parameters are stored in RAM, as outlined in Table 5. These addresses can be directly accessed and modified via the control port of the ADAU1381.

Table 5. Wind Noise Reduction Page Parameters

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-----------|----------------------|------------------------|---|-------|------------------------|
| 0x0011 | WNAIlg | WindNoiseAlg2F11 | 0x00, 0xE8, 0x5D, 0x19 | Frequency and effect gain parameters | 4 | Yes |
| 0x0012 | WNAIlg | WindNoiseAlg2F12 | 0xFF, 0x95, 0xA1, 0x9C | Frequency and effect gain parameters | 4 | Yes |
| 0x0013 | WNAIlg | WindNoiseAlg2F20 | 0x00, 0x00, 0x80, 0x53 | Frequency and effect gain parameters | 4 | Yes |
| 0x0014 | WNAIlg | WindNoiseAlg2F21 | 0x00, 0x01, 0x00, 0xA6 | Frequency and effect gain parameters | 4 | Yes |
| 0x0015 | WNAIlg | WindNoiseAlg2F30 | 0x00, 0xE8, 0xD0, 0x3A | Frequency and effect gain parameters | 4 | Yes |
| 0x0016 | WNAIlg | WindNoiseAlg2F31 | 0xFE, 0x2E, 0x5F, 0x8D | Frequency and effect gain parameters | 4 | Yes |
| 0x0017 | WNAIlg | WindNoiseAlg2F42 | 0x00, 0x80, 0x00, 0x00 | Frequency and effect gain parameters | 4 | Yes |
| 0x0018 | WNAIlg | WindNoiseAlg2tc1 | 0x00, 0x00, 0x20, 0x00 | Time constant 1 (ms) | 4 | Yes |
| 0x0019 | WNAIlg | WindNoiseAlg2tc11 | 0x00, 0x7F, 0xE0, 0x00 | Time constant 1 (ms) | 4 | Yes |
| 0x001A | WNAIlg | WindNoiseAlg2tc2 | 0x00, 0x00, 0x20, 0x00 | Time constant 2 (ms) | 4 | Yes |
| 0x001B | WNAIlg | WindNoiseAlg2tc22 | 0x00, 0x7F, 0xE0, 0x00 | Time constant 2 (ms) | 4 | Yes |
| 0x001C | WNAIlg | WindNoiseAlg2Level1 | 0x00, 0x59, 0x99, 0x9A | Level 1 | 4 | No |
| 0x001D | WNAIlg | WindNoiseAlg2Level2 | 0x00, 0x08, 0x00, 0x00 | Level 2 | 4 | No |
| 0x001E | WNAIlg | WindNoiseAlg2attack | 0x00, 0x00, 0x80, 0x00 | Attack (ms) | 4 | Yes |
| 0x001F | WNAIlg | WindNoiseAlg2release | 0x00, 0x00, 0x00, 0x40 | Release (ms) | 4 | Yes |
| 0x0020 | WN | stereomux1940ns40 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x0020 and Address 0x0021 together) | 4 | No |
| 0x0021 | WN | stereomux1940ns41 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Addresses 0x0020 and Address 0x0021 together) | 4 | No |

ENHANCED STEREO CAPTURE PAGE

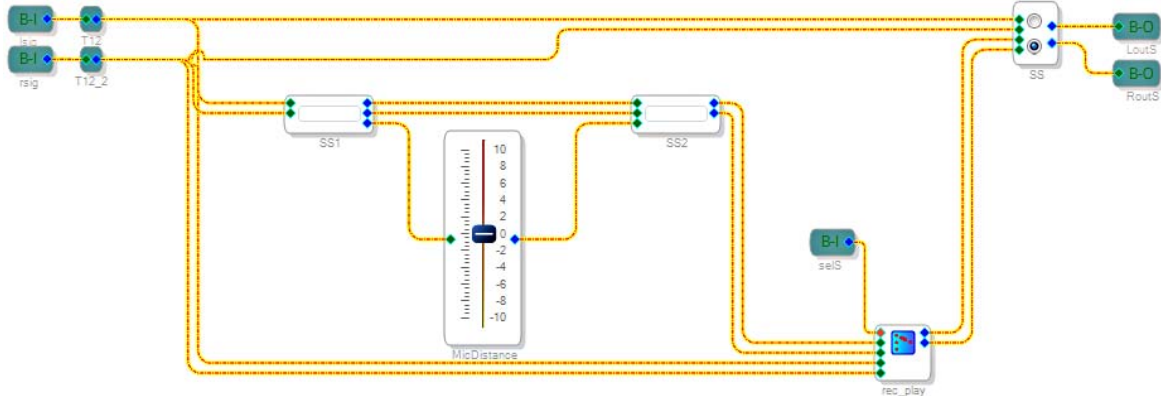


Figure 16. Enhanced Stereo Capture Page

Description

The enhanced stereo capture (ESC) algorithm takes a stereo record signal and creates a wider stereo image. ESC is used as a recording algorithm to capture an enhanced stereo image from two closely spaced microphones.

The ESC algorithm takes two input signals from two closely spaced microphones. The algorithm separates these two signals and widens the stereo image. The result is a perceived widened stereo image as if the audio was captured by microphones with greater left/right separation. ESC is based on proprietary filtering and a stereo balance gain that adjusts how much stereo effect is achieved in the algorithm.

Routing and Bypass

The enhanced stereo capture path is automatically enabled on the mux (**rec_play**) when the sound engine is put into either **REC A** or **REC B**. When in playback mode, the mux bypasses the wind noise reduction algorithm. The switch on this page (**SS**) can be used to bypass the enhanced stereo capture, even in the record modes, if desired.

Controls

The **MicDistance** control can be set from -10 to +10, with a default value of 0 (see Figure 17). This control determines the sensitivity of the ESC algorithm and directly affects the level of stereo enhancement perceived in the recorded signal. Increasing the enhancement too much may result in an unnatural quality in the recorded audio. This control may vary greatly depending on factors such as microphone selection, spacing, and housing. Therefore, it must be tuned to fit the needs of a specific design.

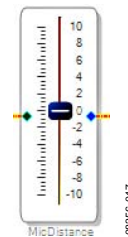


Figure 17. MicDistance Control

Right-click the slider to enter the value directly (see Figure 18).

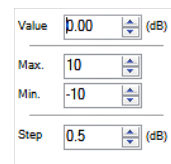


Figure 18. MicDistance Control Direct Value Entry

The **SS** switch allows the algorithm to be bypassed independently of the **rec_play** multiplexer and the active audio mode. The switch can be changed by clicking on the appropriate radio button.

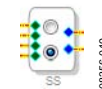


Figure 19. SS Control

Table 6. ESC Page Control Settings

| Setting Name | Description | Default | Control Type |
|---------------------|--|-------------------|--------------------|
| MicDistance | Control enhancement level | 0 | Tune |
| SS Switch Bypass | Switch to disable algorithm | Algorithm enabled | Function selection |
| rec_play Mux Bypass | Switch to bypass algorithm (via multiplexer) | Algorithm enabled | Function selection |

Parameters

The enhanced stereo capture page parameters are stored in RAM, as outlined in Table 7. These addresses can be directly accessed and modified via the control port of the ADAU1381.

Table 7. ESC Page Parameters

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-------------|-------------------|------------------------|---|-------|------------------------|
| 0x0029 | MicDistance | Gain1940AlgNS1 | 0x00, 0x80, 0x00, 0x00 | Gain setting related to the distance between microphones that enhances the perceived effect | 4 | No |
| 0x002B | SS | stereomux1940ns30 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x002B and Address 0x002C together) | 4 | No |
| 0x002C | SS | stereomux1940ns31 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Address 0x002B and Address 0x002C together) | 4 | No |
| 0x0023 | Locked Cell | param1 | 0x00, 0xCA, 0x9A, 0x58 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0024 | Locked Cell | param2 | 0x0F, 0x35, 0x65, 0xA8 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0025 | Locked Cell | param3 | 0x00, 0x7F, 0xAA, 0xE7 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0026 | Locked Cell | param4 | 0x00, 0x08, 0x38, 0x65 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0027 | Locked Cell | param5 | 0x00, 0x00, 0x00, 0x00 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0028 | Locked Cell | param6 | 0x00, 0x7B, 0x1A, 0x7E | Locked parameter (generated by SigmaStudio) | 4 | Yes |

EQUALIZATION FILTERS PAGE

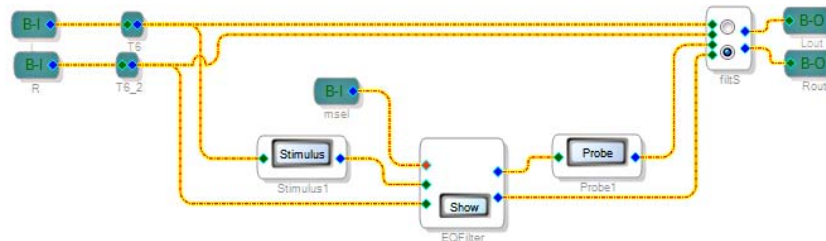


Figure 20. Equalization Filters Page

Description

Equalization (EQ) filters are used to tune the frequency response of the recorded or played back audio signal. The ADAU1381 sound engine includes three, six-band EQ paths, one for playback and the other two for different recording scenarios, such as music recording and voice.

Each EQ band is implemented as a double-precision biquad filter. These filters can be used in a wide variety of configurations, such as low-pass, high-pass, band-pass, parametric, shelving, peaking, tone control, and others.

Routing and Bypass

There are three, six-band EQ paths in the sound engine: one each for Record A (REC A), Record B (REC B), and Playback modes. Path 0 (top row) is the EQ filters for Record A (REC A), Path 1 (middle row) is the EQ filters for Record B (REC B), and Path 2 (bottom row) is the filters for the playback processing.

The appropriate path is automatically selected when the mode is selected on the main page.

The switch on this page (**filtS**) can be used to completely bypass the **EQFilter**, if desired.

Controls

Click **Show** on the **EQFilter** cell to configure the filter bands (see Figure 21).



Figure 21. EQFilter Control with Show Button

When **Show** is clicked, it displays a filter matrix with three rows and six columns (see Figure 22).



Figure 22. EQFilter Matrix

The first row represents the six bands of the Record A (**REC A**) mode, the second row represents the six bands of the Record B (**REC B**) mode, and the third row represents the six bands of the **Playback** mode.

Each button in the matrix contains a single second-order biquad filter. To individually tune a filter, click its corresponding button.

Clicking the menu at the top of the **General Filter Settings** window provides access to a large variety of filters, each with its own property pages and controls (see Figure 23 and Figure 24).

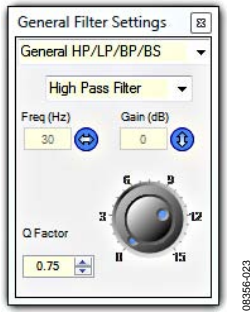


Figure 23. Individual Filter Band Settings

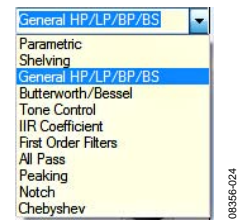


Figure 24. Filter Type Selection

More information on the various filters is available in the **Help** menu within SigmaStudio.

Click **Stimulus** and **Probe** to open the **Simulated Frequency Response** window (see Figure 25 and Figure 26).



Figure 25. Stimulus Button



Figure 26. Probe Button

The **Simulated Frequency Response** window displays a calculated frequency response for each of the filter bands. It shows only one EQ curve at a time, the one corresponding to the filter mode that was last edited.

By default, the EQ curve for Record A (**REC A**) mode is configured for voice recording (see Figure 27). The high-pass filter removes low frequencies that are not necessary for voice recording. The wide boost in the 150 Hz range amplifies the voice fundamental frequencies, and the narrow boost near 4 kHz increases vocal clarity.

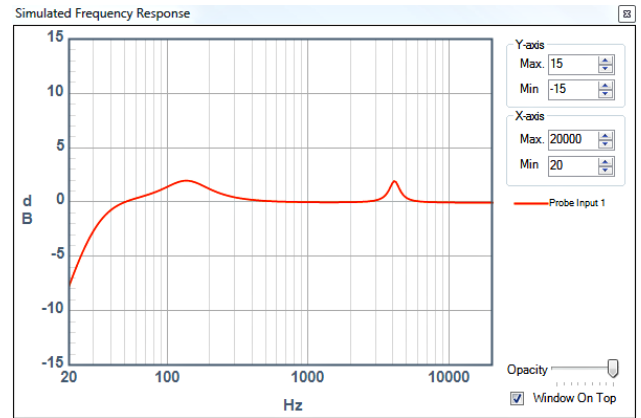


Figure 27. Default Record A (REC A) Mode EQ Curve

By default, the EQ curve for Record B (**REC B**) mode is configured for music and live concert recording (see Figure 28). The high-pass filter removes low frequency boom and rumble from a concert recording environment. The cut in the midbass range around 300 Hz helps to increase the perceived level of the bass. The low-pass filter on the high frequency range helps to reduce ringing caused by reflections in a loud concert environment.

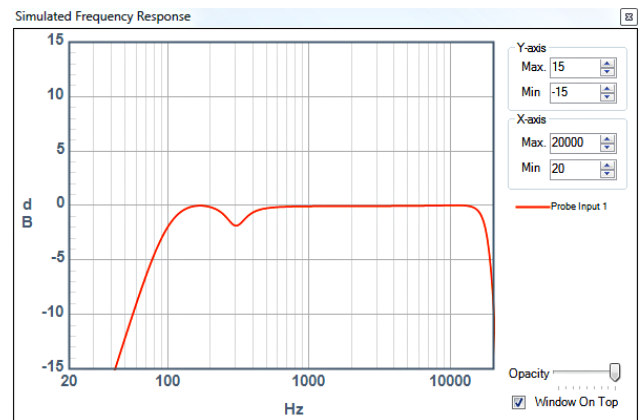


Figure 28. Default Record B (REC B) Mode EQ Curve

By default, the EQ curve for **Playback** mode is flat, which should be changed accordingly to compensate for nonlinearities due to the speaker design and housing (see Figure 29).

The default EQ curves are intended only as examples and should be specifically tuned for the target application system.

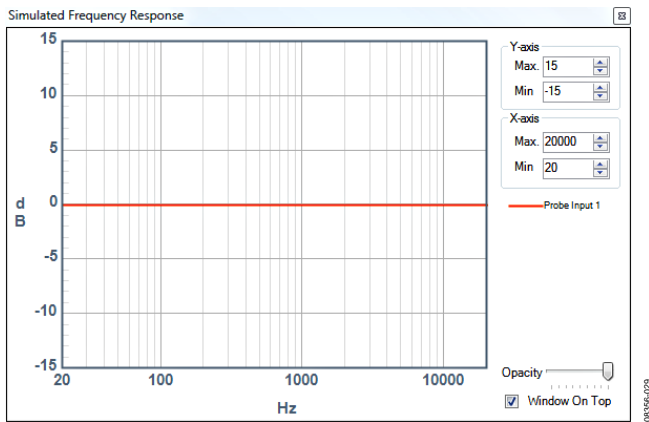


Figure 29. Default Playback Mode EQ Curve

Table 8. EQ Page Control Settings

| Setting Name | Description | Default | Control Type |
|--------------|--|--|--------------------|
| EQFilter | Three parallel six-band equalizers with independently controllable bands | Example curves for record and playback | Tune |
| filtS | Switch to disable algorithm | Algorithm enabled | Function selection |

Parameters

The equalization filters page parameters are stored in RAM, as outlined in Table 9. These addresses can be directly accessed and modified via the control port of the ADAU1381.

Table 9. EQ Page Parameters

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|-----------|--------------------------|--|--------------|-------|------------------------|
| 0x002D 0x002E 0x002F 0x0030 0x0031 | EQFilter | IndexSelMultBandAlg100b2 | 0x00, 0x7F, 0xAA, 0x50, 0xFF, 0x00, 0xAB, 0x60, 0x00, 0x7F, 0xAA, 0x50, 0xFF, 0x80, 0xAB, 0x20, 0x00, 0xFF, 0x54, 0x5F | Biquad F0, 0 | 20 | Yes |
| 0x0032 0x0033 0x0034 0x0035 0x0036 | EQFilter | IndexSelMultBandAlg101b2 | 0x00, 0x7D, 0xBD, 0xAF, 0xFF, 0x02, 0x0A, 0x2E, 0x00, 0x80, 0x42, 0x4A, 0xFF, 0x82, 0x00, 0x07, 0x00, 0xFD, 0xF5, 0xD2 | Biquad F0, 1 | 20 | Yes |
| 0x0037 0x0038 0x0039 0x003A 0x003B | EQFilter | IndexSelMultBandAlg102b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 2 | 20 | Yes |
| 0x003C 0x003D 0x003E 0x003F 0x0040 | EQFilter | IndexSelMultBandAlg103b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 3 | 20 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|-----------|--------------------------|--|--------------|-------|------------------------|
| 0x0041 0x0042 0x0043 0x0044 0x0045 | EQFilter | IndexSelMultBandAlg104b2 | 0x00, 0x71, 0xCB, 0x91, 0xFF, 0x2E, 0xCC, 0xE6, 0x00, 0x81, 0xA0, 0xD2, 0xFF, 0x8C, 0x93, 0x9D, 0x00, 0xD1, 0x33, 0x1A | Biquad F0, 4 | 20 | Yes |
| 0x0046 0x0047 0x0048 0x0049 0x004A | EQFilter | IndexSelMultBandAlg105b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 5 | 20 | Yes |
| 0x004B 0x004C 0x004D 0x004E 0x004F | EQFilter | IndexSelMultBandAlg110b2 | 0x00, 0x7E, 0xFB, 0x24, 0xFF, 0x02, 0x09, 0xB7, 0x00, 0x7E, 0xFB, 0x24, 0xFF, 0x82, 0x06, 0xEE, 0x00, 0xFD, 0xF3, 0x80 | Biquad F1, 0 | 20 | Yes |
| 0x0050 0x0051 0x0052 0x0053 0x0054 | EQFilter | IndexSelMultBandAlg111b2 | 0x00, 0x7D, 0xEB, 0x86, 0xFF, 0x02, 0x83, 0x95, 0x00, 0x7F, 0xC2, 0xF7, 0xFF, 0x82, 0x51, 0x83, 0x00, 0xFD, 0x7C, 0x6B | Biquad F1, 1 | 20 | Yes |
| 0x0055 0x0056 0x0057 0x0058 0x0059 | EQFilter | IndexSelMultBandAlg112b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 2 | 20 | Yes |
| 0x005A 0x005B 0x005C 0x005D 0x005E | EQFilter | IndexSelMultBandAlg113b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 3 | 20 | Yes |
| 0x005F 0x0060 0x0061 0x0062 0x0063 | EQFilter | IndexSelMultBandAlg114b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 4 | 20 | Yes |
| 0x0064 0x0065 0x0066 0x0067 0x0068 | EQFilter | IndexSelMultBandAlg115b2 | 0x00, 0x4A, 0x91, 0x00, 0x00, 0x95, 0x22, 0x00, 0x00, 0x4A, 0x91, 0x00, 0xFF, 0xD1, 0x47, 0xB1, 0xFF, 0x84, 0x74, 0x4F | Biquad F1, 5 | 20 | Yes |
| 0x0069 0x006A 0x006B 0x006C 0x006D | EQFilter | IndexSelMultBandAlg120b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 0 | 20 | Yes |
| 0x006E 0x006F 0x0070 0x0071 0x0072 | EQFilter | IndexSelMultBandAlg121b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 1 | 20 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|-----------|--------------------------|--|---|-------|------------------------|
| 0x0073 0x0074 0x0075 0x0076 0x0077 | EQFilter | IndexSelMultBandAlg122b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 2 | 20 | Yes |
| 0x0078 0x0079 0x007A 0x007B 0x007C | EQFilter | IndexSelMultBandAlg123b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 3 | 20 | Yes |
| 0x007D 0x007E 0x007F 0x0080 0x0081 | EQFilter | IndexSelMultBandAlg124b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 4 | 20 | Yes |
| 0x0082 0x0083 0x0084 0x0085 0x0086 | EQFilter | IndexSelMultBandAlg125b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 5 | 20 | Yes |
| 0x008E | filtS | stereomux1940ns10 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x008E and Address 0x008F together) | 4 | No |
| 0x008F | filtS | stereomux1940ns11 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Address 0x008E and Address 0x008F together) | 4 | No |

DUAL-BAND COMPRESSION PAGE

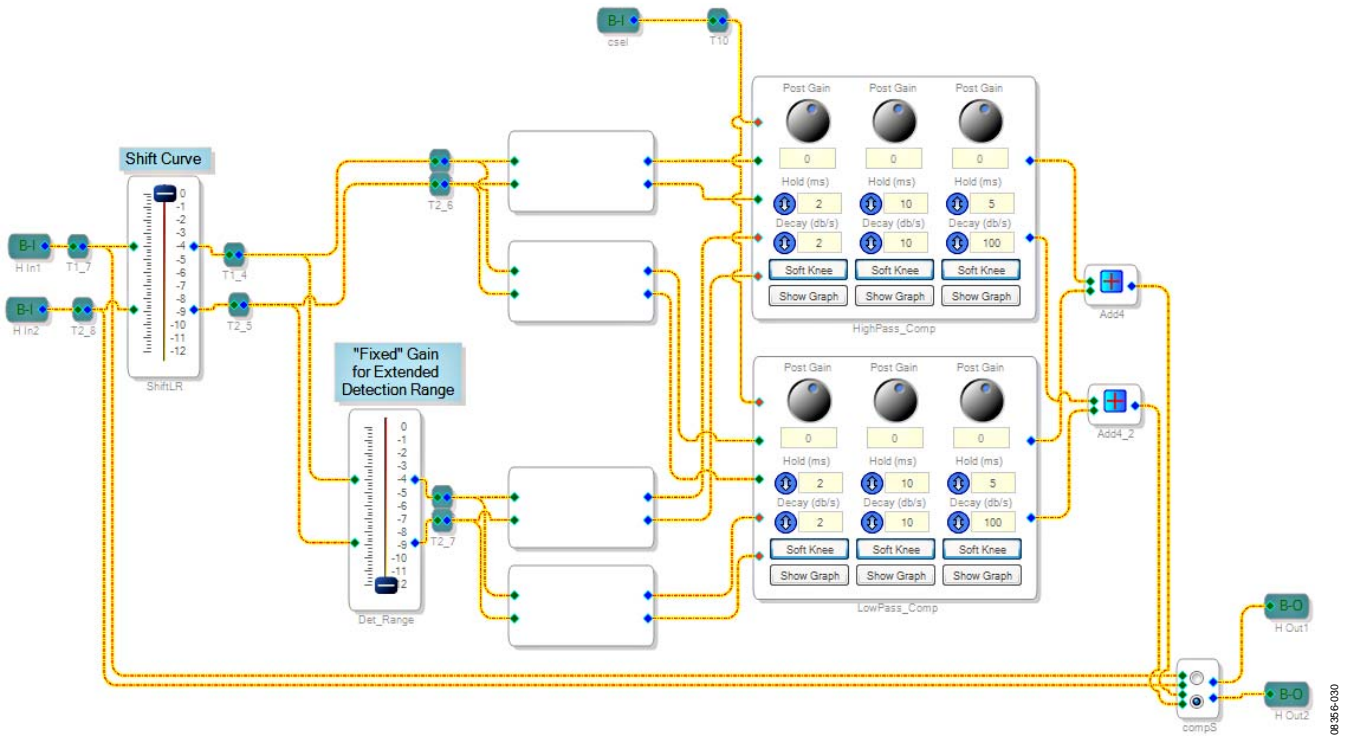


Figure 30. Dual-Band Compression Page

Description

The dual-band compression page contains dynamic processors designed to alter the dynamic range of the audio signal during record or playback. To provide high audio quality, the input signal is sent into a crossover network that divides it into high and low bands. Each band is detected and processed individually. The end result is that a sudden peak in one band (for example, a kick drum in the low band) will not cause a dip in the overall signal level.

Available Curves

By default, the record modes are configured with an automatic level control (ALC) curve, and the playback mode is configured with a standard compressor curve with a threshold of -6 dB and a ratio of 2:1. The Record A (REC A) mode curve is an example of hard ALC compression (see Figure 31), and the Record B (REC B) mode curve is an example of smoothed ALC compression (see Figure 32). The playback mode compressor curve has moderate compression starting at a threshold of -8 dB (see Figure 33). These default curves are intended only to be examples. The desired curve varies greatly depending on the application and other factors in the system. Therefore, unique curves should be created during the tuning process.

Figure 31, Figure 32, and Figure 33 show examples of compressor curves. The curves represent a transfer function, with the horizontal axis representing input in dB and the vertical axis representing the resulting output in dB.

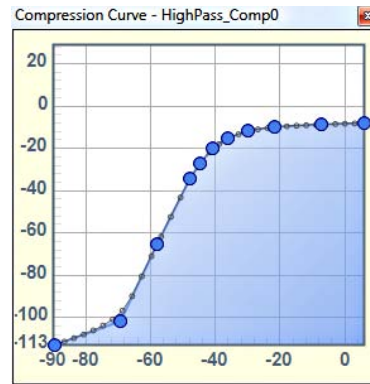


Figure 31. Default Record A (REC A) Mode Compressor Curve

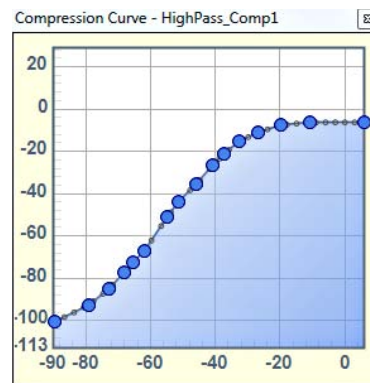


Figure 32. Default Record B (REC B) Mode Compressor Curve

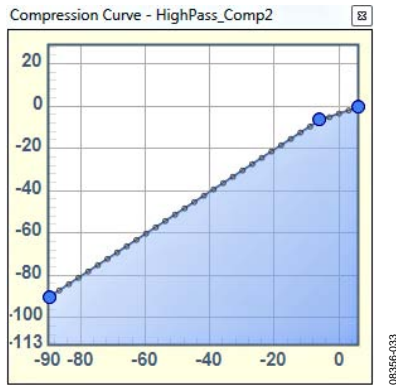


Figure 33. Default Playback Mode Compressor Curve

Curve Shift

The compressor curves can be shifted to the right using the **ShiftLR** control. This changes the input gain of the audio and detects signals routed to the compressor. The default value of 0 dB represents no shift from the original curve. Decreasing this value translates the compressor curves directly to the right by a corresponding amount. Note that there is no graphical difference shown on the compressor curve graphs, but the compressor curve points effectively shift directly to the right as the value of the slider decreases. The maximum shift allowed is 12 dB.

For a classic compression curve (linear compression ratio for low amplitudes and a ratio greater than 1 after a certain threshold), the **ShiftLR** control effectively increases the threshold value. The **ShiftLR** control allows a curve to be shifted at run time without requiring a download of new compression table parameters via the control port.

Figure 34 shows an example of curve shift being applied to an example compression curve with a gate below -80 dB, a linear transfer function between -80 dB and -28 dB, and a compression ratio of 2:1 for input amplitudes greater than -28 dB. The example curve is shown furthest to the left. Shifted curves for -3 dB, -6 dB, -9 dB, and -12 dB are shown to the right of the example curve. For all shifted curves, the compression threshold remains constant, but the gate threshold changes. The output gain for the linear section of the input range decreases as the curve shifts to the right.

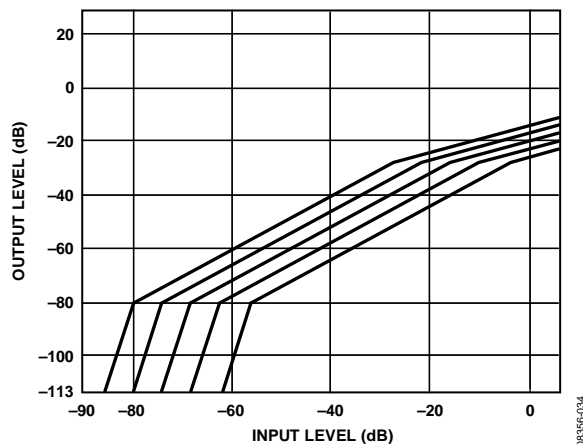


Figure 34. Curve Shift L/R Example

Detection Range Shift

The detection range control shifts the range over which the compressor operates. The algorithm typically handles inputs ranging from -90 dB to +6 dB. Any inputs outside of this range have a linear input-to-output relationship, effectively ignoring the compression curve. In applications where inputs to the compressor greater than 6 dB are expected, the detection range can be shifted to accommodate the input signal range. The default shift of -12 dB changes the detection range's lower bound to -78 dB and its upper bound to +18 dB. This curve shift must correspond to the compressors Adjust Gain Curve setting, shown in Figure 39.

The compression curve displayed in the compression curve graph represents a detection range shift of 0 dB. Decreasing the detection range effectively shifts the curve upward and to the right.

The detection range shift should be determined during system tuning and should not be altered when the system is in operation.

Figure 35 shows an example of detection range shift being applied to an example compression curve with a gate below -80 dB, a linear transfer function between -80 dB and -28 dB, and a compression ratio of 2:1 for input amplitudes greater than -28 dB. The example curve is shown furthest to the bottom and the left. Shifted curves for -3 dB, -6 dB, -9 dB, and -12 dB are shown above and to the right of the example curve. For all shifted curves, both the compression and gate thresholds increase, but the linear section of the input range remains linear.

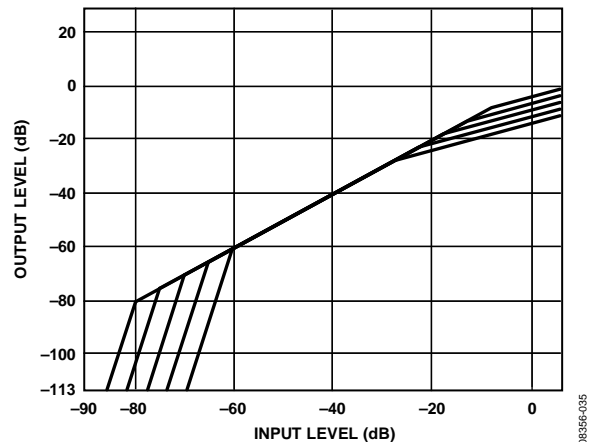


Figure 35. Detection Range Shift Example

Routing and Bypass

The dual-band compression algorithm is enabled by default in all audio modes. The **compS** switch allows the dual-band compressor to be bypassed manually.

Controls

The **ShiftLR** control shifts the compression curve horizontally. The slider can be dragged to change the value (see Figure 36). The default value of 0 dB indicates that the transfer function displayed in the compression curve editor matches the processing in the sound engine. Decreasing the value of the control shifts the curve to the right.

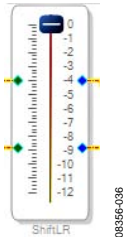


Figure 36. ShiftLR Control

Right-click the slider to type the value in directly (see Figure 37).

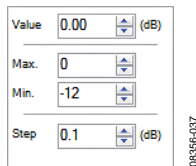


Figure 37. ShiftLR Control Direct Value Entry

The **Det_Range** control shifts the compression curve diagonally (see Figure 38). The **Det_Range** value can be controlled by dragging the slider or by entering the value manually by right-clicking the slider. The **Det_Range** control is only intended to take on the following values: 0 dB, -3 dB, -6 dB, -9 dB, and -12 dB.

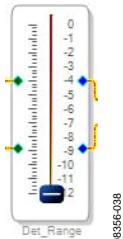


Figure 38. Det_Range Control

When the detection range is modified, the compressors must also be configured to match. By default, the compressors are configured for a detection range shift of -12 dB. To change the compressor detection range, click on a compressor cell, select the **Adjust Gain Curve** option, and select the value matching the setting of the **Det_Range** control (see Figure 39). Complete this process for both the high-band and low-band compressors.

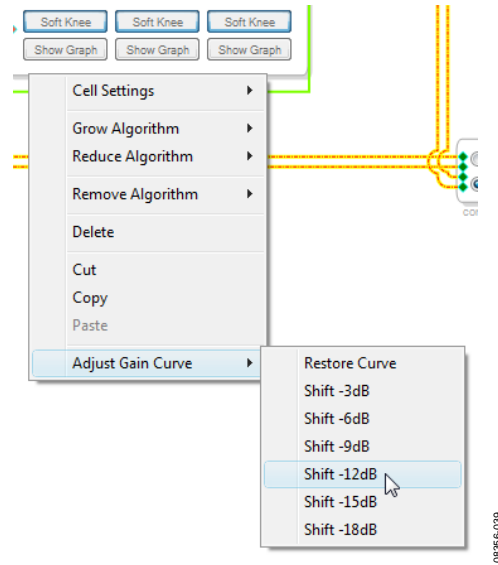


Figure 39. Changing the Detection Curve of a Compressor

Each frequency band (low and high) is fed into a stereo compressor matrix (see Figure 40). Each matrix contains three compressors, one for each audio mode. The left column corresponds to Record A (**REC A**) mode, the center column corresponds to Record B (**REC B**) mode, and the right column corresponds to **Playback** mode.

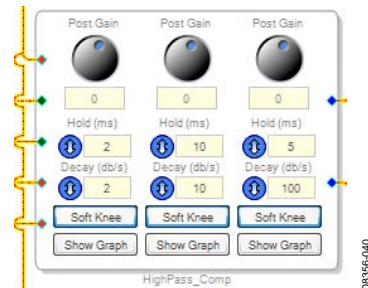


Figure 40. HighPass_Comp Control

Each compressor column contains a **Post Gain** control that adjusts the amount of gain applied to the signal at the output of the compressor (see Figure 41).



Figure 41. Post Gain Control

The **Hold (ms)** control sets the duration that the gain reduction ratio of the compressor is held after it is set by the input signal (see Figure 42).



Figure 42. Hold (ms) Control

The **Decay (dB/s)** control sets the speed by which the gain reduction ratio decays after the hold duration expires (see Figure 43).



Figure 43. Decay (dB/s) Control

Click the **Soft Knee** button to smooth the corners (also known as knees) of the **Compression Curve** (see Figure 44).



Figure 44. Soft Knee Button

Click the **Show Graph** button to display the **Compression Curve** graphical editor (see Figure 45).



Figure 45. Show Graph Button

The **Compression Curve** editor displays a graphical representation of the input/output gain transfer function, which is a curve with 33 points (see Figure 46). The horizontal axis represents the input level, and the vertical axis represents the output level. Each large point can be dragged to a new position on the graph.

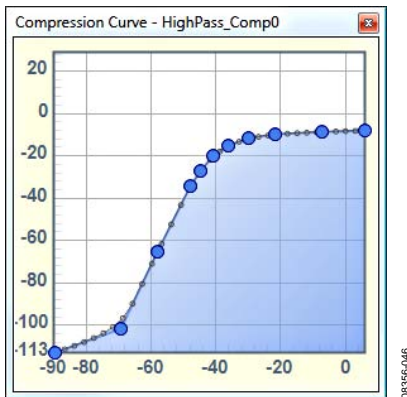


Figure 46. Compression Curve Editor

Click on a point within the graph to display the **Compression Curve Point Option** menu. This is where large points can be added, removed, or fine-tuned (see Figure 47).

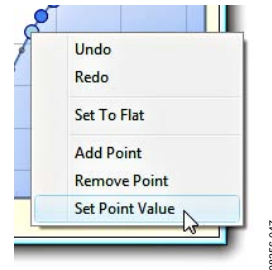


Figure 47. Compression Curve Point Option Menu

Click on a set point value to display a dialog box where the coordinates of the point can be entered manually (see Figure 48).

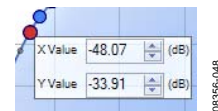


Figure 48. Compression Curve Point Direct Value Entry

The **compS** switch allows the dual-band compressors to be bypassed (see Figure 49). Click on the appropriate radio button to change the switch.

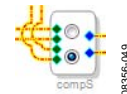


Figure 49. compS Control

Table 10. Dual-Band Compression Page Control Settings

| Setting Name | Description | Default | Control Type |
|-----------------------------|--|----------------------------|--------------|
| ShiftLR | Shift curve left/right | 0 | Tune |
| Det_Range | Fixed gain for extended detection range | -12 | Tune |
| CrossHi | Crossover for high frequencies | N/A | Locked |
| CrossLo | Crossover for low frequencies | N/A | Locked |
| HiDet Filter | Filter for high frequency detector | N/A | Locked |
| LoDet Filter | Filter for low frequency detector | N/A | Locked |
| HIGH-PASS COMPRESSOR | | | |
| Post Gain (dB) | Gain applied to the output of the compressor | 0 | Tune |
| Hold (ms) | Duration that the gain reduction ratio of the compressor is held after it is set by the input signal | 2 | Tune |
| Decay (dB/s) | Speed that the gain reduction ratio of the compressor decreases after the hold time expires | 2 | Tune |
| Soft Knee | Smooths the compression curve | Active | Tune |
| Graph Editor | Graphical entry of compression curve (input/output gain transfer function) | Default compression curves | Tune |

| Setting Name | Description | Default | Control Type |
|---------------------|--|----------------------------|--------------------|
| LOW-PASS COMPRESSOR | | | |
| Post Gain (dB) | Gain applied to the output of the compressor | 0 | Tune |
| Hold (ms) | Duration that the gain reduction ratio of the compressor is held after it is set by the input signal | 2 | Tune |
| Decay (dB/s) | Speed that the gain reduction ratio of the compressor decreases after the hold time expires | 2 | Tune |
| Soft Knee | Smooths the compression curve | Active | Tune |
| Graph Editor | Graphical entry of compression curve (input/output gain transfer function) | Default compression curves | Tune |
| compS | Switch/mux bypass | Algorithm enabled | Function selection |

Parameters

The dual-band compression page parameters are stored in RAM, as outlined in Table 11. These addresses can be directly accessed and modified via the control port of the ADAU1381.

Table 11. Dual-Band Compression Page Parameters

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-----------|------------------|------------------------|---|-------|------------------------|
| 0x0090 | ShiftLR | Gain1940AlgNS4 | 0x00, 0x80, 0x00, 0x00 | Shift curve left/right; Address 0x0090 and Address 0x0091 must contain the same value | 4 | No |
| 0x0091 | ShiftLR | Gain1940AlgNS5 | 0x00, 0x80, 0x00, 0x00 | Shift curve left/right; Address 0x0090 and Address 0x0091 must contain the same value | 4 | No |
| 0x0092 | Det_Range | Gain1940AlgNS3 | 0x00, 0x20, 0x26, 0xF3 | On/off (burst write Address 0x0092 and Address 0x0093 together) | 4 | No |
| 0x0093 | Det_Range | Gain1940AlgNS2 | 0x00, 0x20, 0x26, 0xF3 | On/off (burst write Address 0x0092 and Address 0x0093 together) | 4 | No |
| 0x00A2 | CrossHi | EQwSubDualDP32B1 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A3 | CrossHi | EQwSubDualDP31B1 | 0x0F, 0x08, 0x29, 0x18 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A4 | CrossHi | EQwSubDualDP30B1 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A5 | CrossHi | EQwSubDualDP32A1 | 0x0F, 0x88, 0x07, 0xC9 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A6 | CrossHi | EQwSubDualDP31A1 | 0x00, 0xF7, 0xB5, 0x9A | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A7 | CrossHi | EQwSubDualDP32B2 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A8 | CrossHi | EQwSubDualDP31B2 | 0x0F, 0x08, 0x29, 0x18 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A9 | CrossHi | EQwSubDualDP30B2 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00AA | CrossHi | EQwSubDualDP32A2 | 0x0F, 0x88, 0x07, 0xC9 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00AB | CrossHi | EQwSubDualDP31A2 | 0x00, 0xF7, 0xB5, 0x9A | Crossover HPF filter coefficient | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|--------------|------------------|------------------------|---|-------|------------------------|
| 0x0094 | CrossLo | EQwSubDualDP42B1 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0095 | CrossLo | EQwSubDualDP41B1 | 0x00, 0x00, 0x21, 0x4E | Crossover LPF filter coefficient | 4 | Yes |
| 0x0096 | CrossLo | EQwSubDualDP40B1 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0097 | CrossLo | EQwSubDualDP42A1 | 0x0F, 0x88, 0x07, 0xC9 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0098 | CrossLo | EQwSubDualDP41A1 | 0x00, 0xF7, 0xB5, 0x9A | Crossover LPF filter coefficient | 4 | Yes |
| 0x0099 | CrossLo | EQwSubDualDP42B2 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009A | CrossLo | EQwSubDualDP41B2 | 0x00, 0x00, 0x21, 0x4E | Crossover LPF filter coefficient | 4 | Yes |
| 0x009B | CrossLo | EQwSubDualDP40B2 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009C | CrossLo | EQwSubDualDP42A2 | 0x0F, 0x88, 0x07, 0xC9 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009D | CrossLo | EQwSubDualDP41A2 | 0x00, 0xF7, 0xB5, 0x9A | Crossover LPF filter coefficient | 4 | Yes |
| 0x00BE | HiDet_Filter | EQwSubDualDP52B1 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00BF | HiDet_Filter | EQwSubDualDP51B1 | 0x0F, 0x05, 0x76, 0x40 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C0 | HiDet_Filter | EQwSubDualDP50B1 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C1 | HiDet_Filter | EQwSubDualDP52A1 | 0x0F, 0x85, 0x67, 0x55 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C2 | HiDet_Filter | EQwSubDualDP51A1 | 0x00, 0xFA, 0x7A, 0xD5 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C3 | HiDet_Filter | EQwSubDualDP52B2 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C4 | HiDet_Filter | EQwSubDualDP51B2 | 0x0F, 0x05, 0x76, 0x40 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C5 | HiDet_Filter | EQwSubDualDP50B2 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C6 | HiDet_Filter | EQwSubDualDP52A2 | 0x0F, 0x85, 0x67, 0x55 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C7 | HiDet_Filter | EQwSubDualDP51A2 | 0x00, 0xFA, 0x7A, 0xD5 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00B0 | LoDet_Filter | EQwSubDualDP62B1 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B1 | LoDet_Filter | EQwSubDualDP61B1 | 0x00, 0x00, 0x49, 0xC4 | Crossover LPF detection path filter coefficient | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|---------------|---------------------------|------------------------|---|-------|------------------------|
| 0x00B2 | LoDet_Filter | EQwSubDualDP60B1 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B3 | LoDet_Filter | EQwSubDualDP62A1 | 0x0F, 0x8B, 0xDA, 0xCC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B4 | LoDet_Filter | EQwSubDualDP61A1 | 0x00, 0xF3, 0x91, 0xAC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B5 | LoDet_Filter | EQwSubDualDP62B2 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B6 | LoDet_Filter | EQwSubDualDP61B2 | 0x00, 0x00, 0x49, 0xC4 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B7 | LoDet_Filter | EQwSubDualDP60B2 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B8 | LoDet_Filter | EQwSubDualDP62A2 | 0x0F, 0x8B, 0xDA, 0xCC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B9 | LoDet_Filter | EQwSubDualDP61A2 | 0x00, 0xF3, 0x91, 0xAC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00CC | HighPass_Comp | PeakDBCompLUTAlgPG30decay | 0x00, 0x00, 0x00, 0x04 | REC_Auto: decay | 4 | Yes |
| 0x00CD | HighPass_Comp | PeakDBCompLUTAlgPG30hold | 0x00, 0x00, 0x00, 0x60 | REC_Auto: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|---------------|-------------------------------|--|---|-------|------------------------|
| 0x00CE 0x00CF 0x00D0 0x00D1 0x00D2 0x00D3 0x00D4 0x00D5 0x00D6 0x00D7 0x00D8 0x00D9 0x00DA 0x00DB 0x00DC 0x00DD 0x00DE 0x00DF 0x00E0 0x00E1 0x00E2 0x00E3 0x00E4 0x00E5 0x00E6 0x00E7 0x00E8 0x00E9 0x00EA 0x00EB 0x00EC 0x00ED 0x00EE | HighPass_Comp | PeakDBCompLUTAlgPG30tab | 0x00, 0x05, 0x1A, 0x96, 0x00, 0x04, 0xB6, 0x9B, 0x00, 0x04, 0xAD, 0x54, 0x00, 0x05, 0x55, 0x55, 0x00, 0x08, 0x55, 0xC1, 0x00, 0x11, 0x6C, 0xA5, 0x00, 0x25, 0x0E, 0xD2, 0x00, 0x4D, 0x9C, 0x93, 0x00, 0xA0, 0x10, 0xF5, 0x01, 0x46, 0xB2, 0x33, 0x02, 0x80, 0x9F, 0x0A, 0x03, 0xF0, 0xCE, 0x62, 0x05, 0x54, 0x59, 0xF2, 0x05, 0xEF, 0xF5, 0xFC, 0x05, 0xBA, 0x65, 0x98, 0x05, 0x17, 0xAD, 0xEF, 0x04, 0x3A, 0x77, 0xC5, 0x03, 0x56, 0xDC, 0x88, 0x02, 0x8B, 0x2A, 0xC5, 0x01, 0xE7, 0x60, 0xA5, 0x01, 0x67, 0x06, 0x39, 0x01, 0x05, 0xD8, 0x28, 0x00, 0xBE, 0x7E, 0xE5, 0x00, 0x8A, 0x46, 0xF6, 0x00, 0x64, 0x17, 0x5B, 0x00, 0x48, 0x3E, 0xCA, 0x00, 0x34, 0x06, 0x48, 0x00, 0x25, 0x6A, 0x11, 0x00, 0x1A, 0xE6, 0xE6, 0x00, 0x13, 0x05, 0xC6, 0x00, 0x0D, 0x73, 0x73, 0x00, 0x09, 0x82, 0xE3, 0x00, 0x06, 0xB9, 0xBA | REC_Auto: compressor curve points | 132 | No |
| 0x0138 | HighPass_Comp | PeakDBCompLUTAlgPG30post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Auto: post gain | 4 | No |
| 0x00EF | HighPass_Comp | PeakDBCompLUTAlgPG31decay | 0x00, 0x00, 0x00, 0x12 | REC_Manual: decay | 4 | Yes |
| 0x00F0 | HighPass_Comp | PeakDBCompLUTAlgPG31hold | 0x00, 0x00, 0x01, 0xE0 | REC_Manual: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|---------------|-------------------------------|--|---|-------|------------------------|
| 0x00F1 0x00F2 0x00F3 0x00F4 0x00F5 0x00F6 0x00F7 0x00F8 0x00F9 0x00FA 0x00FB 0x00FC 0x00FD 0x00FE 0x00FF 0x0100 0x0101 0x0102 0x0103 0x0104 0x0105 0x0106 0x0107 0x0108 0x0109 0x010A 0x010B 0x010C 0x010D 0x010E 0x010F 0x0110 0x0111 | HighPass_Comp | PeakDBCompLUTAlgPG31tab | 0x00, 0x1D, 0xC7, 0xFA, 0x00, 0x1F, 0xA6, 0x78, 0x00, 0x24, 0xCD, 0xBF, 0x00, 0x2D, 0xB0, 0xDE, 0x00, 0x3A, 0x4E, 0x7F, 0x00, 0x47, 0xE3, 0x23, 0x00, 0x67, 0x41, 0xE8, 0x00, 0xA1, 0x64, 0x8B, 0x00, 0xF3, 0x71, 0x72, 0x01, 0x4B, 0x5C, 0x80, 0x01, 0x96, 0x84, 0x19, 0x01, 0xF9, 0x0D, 0x62, 0x02, 0x8A, 0xA7, 0xCA, 0x03, 0x24, 0x67, 0xA2, 0x03, 0x9A, 0x7A, 0x83, 0x03, 0xD1, 0x34, 0x62, 0x03, 0xA2, 0xBD, 0x15, 0x03, 0x3D, 0x07, 0x65, 0x02, 0xC9, 0xE3, 0xC3, 0x02, 0x4D, 0x52, 0x70, 0x01, 0xCE, 0xBA, 0xD0, 0x01, 0x5C, 0x48, 0xE1, 0x00, 0xFF, 0x34, 0xC9, 0x00, 0xB8, 0x8A, 0x49, 0x00, 0x83, 0xA5, 0x3C, 0x00, 0x5D, 0x40, 0x76, 0x00, 0x42, 0x04, 0x69, 0x00, 0x2E, 0xBC, 0x93, 0x00, 0x21, 0x16, 0x45, 0x00, 0x17, 0x65, 0x63, 0x00, 0x10, 0x8B, 0x23, 0x00, 0x0B, 0xB2, 0xB1, 0x00, 0x08, 0x45, 0x91 | REC_Manual: compressor curve points | 132 | No |
| 0x0139 | HighPass_Comp | PeakDBCompLUTAlgPG31post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Manual: post gain | 4 | No |
| 0x0112 | HighPass_Comp | PeakDBCompLUTAlgPG32decay | 0x00, 0x00, 0x00, 0xB6 | Speaker: decay | 4 | Yes |
| 0x0113 | HighPass_Comp | PeakDBCompLUTAlgPG32hold | 0x00, 0x00, 0x00, 0xF0 | Speaker: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-------------------------|-------------------------------|-------------------------|--|-------|------------------------|
| 0x0114 | HighPass_Comp | PeakDBCompLUTAlgPG32tab | 0x00, 0x80, 0x00, 0x00, | Speaker: compressor curve points | 132 | No |
| 0x0115 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0116 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0117 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0118 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0119 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011A | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011B | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011C | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011D | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011E | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011F | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0120 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0121 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0122 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0123 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0124 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0125 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0126 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0127 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0128 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0129 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x012A | | | 0x00, 0x7F, 0xF0, 0x61, | | | |
| 0x012B | | | 0x00, 0x7D, 0x6C, 0x21, | | | |
| 0x012C | | | 0x00, 0x75, 0x7A, 0x64, | | | |
| 0x012D | | | 0x00, 0x68, 0xDB, 0x84, | | | |
| 0x012E | | | 0x00, 0x5A, 0x4C, 0x32, | | | |
| 0x012F | | | 0x00, 0x4C, 0x3E, 0xA8, | | | |
| 0x0130 | 0x00, 0x40, 0x26, 0xE7, | | | | | |
| 0x0131 | 0x00, 0x2D, 0x5C, 0xBF, | | | | | |
| 0x0132 | 0x00, 0x20, 0x13, 0x74, | | | | | |
| 0x0133 | 0x00, 0x16, 0xAE, 0x60, | | | | | |
| 0x0134 | 0x00, 0x10, 0x09, 0xBA | | | | | |
| 0x013A | HighPass_Comp | PeakDBCompLUTAlgPG32post_gain | 0x00, 0x80, 0x00, 0x00 | Speaker: post gain | 4 | No |
| 0x0140 | LowPass_Comp | PeakDBCompLUTAlgPG40decay | 0x00, 0x00, 0x00, 0x04 | REC_Auto: decay | 4 | Yes |
| 0x0141 | LowPass_Comp | PeakDBCompLUTAlgPG40hold | 0x00, 0x00, 0x00, 0x60 | REC_Auto: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|--------------|-------------------------------|--|---|-------|------------------------|
| 0x0142 0x0143 0x0144 0x0145 0x0146 0x0147 0x0148 0x0149 0x014A 0x014B 0x014C 0x014D 0x014E 0x014F 0x0150 0x0151 0x0152 0x0153 0x0154 0x0155 0x0156 0x0157 0x0158 0x0159 0x015A 0x015B 0x015C 0x015D 0x015E 0x015F 0x0160 0x0161 0x0162 | LowPass_Comp | PeakDBCompLUTAlgPG40tab | 0x00, 0x05, 0x1A, 0x96, 0x00, 0x04, 0xB6, 0x9B, 0x00, 0x04, 0xAD, 0x54, 0x00, 0x05, 0x55, 0x55, 0x00, 0x08, 0x55, 0xC1, 0x00, 0x11, 0x6C, 0xA5, 0x00, 0x25, 0x0E, 0xD2, 0x00, 0x4D, 0x9C, 0x93, 0x00, 0xA0, 0x10, 0xF5, 0x01, 0x46, 0xB2, 0x33, 0x02, 0x80, 0x9F, 0x0A, 0x03, 0xF0, 0xCE, 0x62, 0x05, 0x54, 0x59, 0xF2, 0x05, 0xEF, 0xF5, 0xFC, 0x05, 0xBA, 0x65, 0x98, 0x05, 0x17, 0xAD, 0xEF, 0x04, 0x3A, 0x77, 0xC5, 0x03, 0x56, 0xDC, 0x88, 0x02, 0x8B, 0x2A, 0xC5, 0x01, 0xE7, 0x60, 0xA5, 0x01, 0x67, 0x06, 0x39, 0x01, 0x05, 0xD8, 0x28, 0x00, 0xBE, 0x7E, 0xE5, 0x00, 0x8A, 0x46, 0xF6, 0x00, 0x64, 0x17, 0x5B, 0x00, 0x48, 0x3E, 0xCA, 0x00, 0x34, 0x06, 0x48, 0x00, 0x25, 0x6A, 0x11, 0x00, 0x1A, 0xE6, 0xE6, 0x00, 0x13, 0x05, 0xC6, 0x00, 0x0D, 0x73, 0x73, 0x00, 0x09, 0x82, 0xE3, 0x00, 0x06, 0xB9, 0xBA | REC_Auto: compressor curve points | 132 | No |
| 0x01AC | LowPass_Comp | PeakDBCompLUTAlgPG40post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Auto: post gain | 4 | No |
| 0x0163 | LowPass_Comp | PeakDBCompLUTAlgPG41decay | 0x00, 0x00, 0x00, 0x12 | REC_Manual: decay | 4 | Yes |
| 0x0164 | LowPass_Comp | PeakDBCompLUTAlgPG41hold | 0x00, 0x00, 0x01, 0xE0 | REC_Manual: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|--------------|-------------------------------|--|---|-------|------------------------|
| 0x0165 0x0166 0x0167 0x0168 0x0169 0x016A 0x016B 0x016C 0x016D 0x016E 0x016F 0x0170 0x0171 0x0172 0x0173 0x0174 0x0175 0x0176 0x0177 0x0178 0x0179 0x017A 0x017B 0x017C 0x017D 0x017E 0x017F 0x0180 0x0181 0x0182 0x0183 0x0184 0x0185 | LowPass_Comp | PeakDBCompLUTAlgPG41tab | 0x00, 0x1D, 0xC7, 0xFA, 0x00, 0x1F, 0xA6, 0x78, 0x00, 0x24, 0xCD, 0xBF, 0x00, 0x2D, 0xB0, 0xDE, 0x00, 0x3A, 0x4E, 0x7F, 0x00, 0x47, 0xE3, 0x23, 0x00, 0x67, 0x41, 0xE8, 0x00, 0xA1, 0x64, 0x8B, 0x00, 0xF3, 0x71, 0x72, 0x01, 0x4B, 0x5C, 0x80, 0x01, 0x96, 0x84, 0x19, 0x01, 0xF9, 0x0D, 0x62, 0x02, 0x8A, 0xA7, 0xCA, 0x03, 0x24, 0x67, 0xA2, 0x03, 0x9A, 0x7A, 0x83, 0x03, 0xD1, 0x34, 0x62, 0x03, 0xA2, 0xBD, 0x15, 0x03, 0x3D, 0x07, 0x65, 0x02, 0xC9, 0xE3, 0xC3, 0x02, 0x4D, 0x52, 0x70, 0x01, 0xCE, 0xBA, 0xD0, 0x01, 0x5C, 0x48, 0xE1, 0x00, 0xFF, 0x34, 0xC9, 0x00, 0xB8, 0x8A, 0x49, 0x00, 0x83, 0xA5, 0x3C, 0x00, 0x5D, 0x40, 0x76, 0x00, 0x42, 0x04, 0x69, 0x00, 0x2E, 0xBC, 0x93, 0x00, 0x21, 0x16, 0x45, 0x00, 0x17, 0x65, 0x63, 0x00, 0x10, 0x8B, 0x23, 0x00, 0x0B, 0xB2, 0xB1, 0x00, 0x08, 0x45, 0x91 | REC_Manual: compressor curve points | 132 | No |
| 0x01AD | LowPass_Comp | PeakDBCompLUTAlgPG41post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Manual: post gain | 4 | No |
| 0x0186 | LowPass_Comp | PeakDBCompLUTAlgPG42decay | 0x00, 0x00, 0x00, 0xB6 | Speaker: decay | 4 | Yes |
| 0x0187 | LowPass_Comp | PeakDBCompLUTAlgPG42hold | 0x00, 0x00, 0x00, 0xF0 | Speaker: hold | 4 | Yes |

SIGMASTUDIO TOOLS

CHANGING SAMPLE RATE

To change the sampling rate of the system, complete the following steps:

1. Select a new sample rate from the menu in the toolbar (see Figure 50).

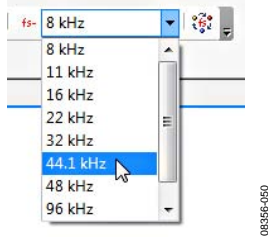


Figure 50. Sample Rate Menu

2. Click the **Set System Sampling Rate** button (see Figure 51) and click **Yes** when prompted to confirm (see Figure 52).

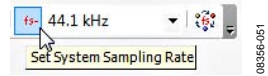


Figure 51. Set System Sampling Rate Button

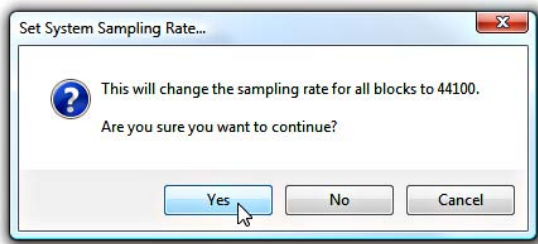


Figure 52. Set System Sampling Rate... Confirmation Window

3. If target hardware is connected to the PC, click the **Link-Compile-Download** button (see Figure 53) to download the new parameters to the hardware.



Figure 53. Link-Compile-Download Button

CAPTURE WINDOW

The **Capture** window is visible by default when SigmaStudio is executed for the first time. Its visibility can be toggled in the **View** menu.

The **Capture** window displays all communications between the PC and the control port of the ADAU1381 (see Figure 54). It is a useful tool for debugging and monitoring communications.

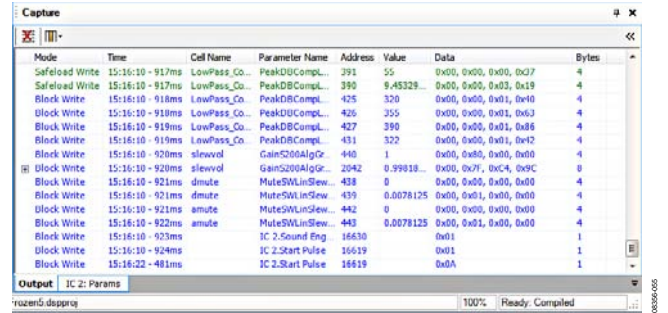


Figure 54. Capture Window

PARAMETER VISUALIZATION WINDOW

Click the **IC 2: Params** tab at the bottom of the **Capture** window to open the **Parameter Visualization** window (see Figure 55). This window shows all parameter RAM values for the project in real-time. It does not display the values of the parameters stored in addresses higher than 0x01FF (Decimal 511).

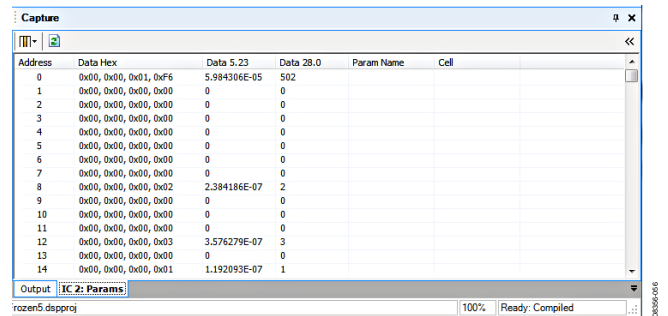


Figure 55. Parameter Visualization Window

SEQUENCE WINDOW

Click << at the top right of the **Capture** window to open the **Display Sequence Window** (see Figure 56). In this window, sequences of data writes can be created by dragging rows from the **Capture** window or by manually entering target addresses and data values.

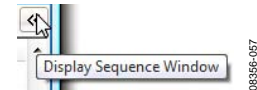


Figure 56. Display Sequence Window Button

Click the **Download Mode to Hardware** button to then initiate the sequence (see Figure 57).

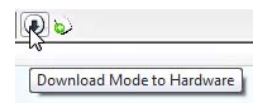


Figure 57. Download Mode to Hardware Button

Figure 58 shows an example sequence that changes the playback volume control (slewvol cell) gain value to -52 dB.

| Mode | Address | Bytes | Data | Cell Name | Parameter Name |
|-------|---------|-------|------------------------|-----------|-----------------------------|
| Write | 440 | 4 | 0x00, 0x00, 0x52, 0x4F | slewvol | GainS200AlgGrow1gain_target |

Figure 58. Example Sequence Changing the Playback Volume Control

EXPORT PARAMETER AND REGISTER SETTINGS

Click **Export System Files** in the toolbar to export the system files, such as parameter and register values (see Figure 59).

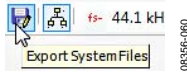


Figure 59. Export System Files Button

SIGMASTUDIO HELP FILE

SigmaStudio includes a **Help** file that further describes many of the algorithms and design functions described in this user guide. Access the **Help** file through the toolbar or by highlighting a block in the **Schematic** window and pressing **F1**, which brings you to the **Help** page for that block.

FULL PARAMETER MAP

Table 12. Full Parameter Map with Default Values for $f_s = 48$ kHz

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-------------|----------------------|------------------------|---|-------|------------------------|
| 0x0009 | audioMode | DCInpAlg1 | 0x00, 0x00, 0x00, 0x00 | Set record/playback mode | 4 | No |
| 0x000A | REC_Coeff | DCInpAlg3 | 0x00, 0x00, 0x00, 0x00 | Set Record Mode A or Record Mode B | 4 | No |
| 0x000B | GPIO | DCInpAlg4 | 0x00, 0x00, 0x00, 0x00 | Set GPIO output flag | 4 | No |
| 0x0011 | WNAIlg | WindNoiseAlg2F11 | 0x00, 0xE8, 0x5D, 0x19 | Frequency and effect gain parameters | 4 | Yes |
| 0x0012 | WNAIlg | WindNoiseAlg2F12 | 0xFF, 0x95, 0xA1, 0x9C | Frequency and effect gain parameters | 4 | Yes |
| 0x0013 | WNAIlg | WindNoiseAlg2F20 | 0x00, 0x00, 0x80, 0x53 | Frequency and effect gain parameters | 4 | Yes |
| 0x0014 | WNAIlg | WindNoiseAlg2F21 | 0x00, 0x01, 0x00, 0xA6 | Frequency and effect gain parameters | 4 | Yes |
| 0x0015 | WNAIlg | WindNoiseAlg2F30 | 0x00, 0xE8, 0xD0, 0x3A | Frequency and effect gain parameters | 4 | Yes |
| 0x0016 | WNAIlg | WindNoiseAlg2F31 | 0xFE, 0x2E, 0x5F, 0x8D | Frequency and effect gain parameters | 4 | Yes |
| 0x0017 | WNAIlg | WindNoiseAlg2F42 | 0x00, 0x80, 0x00, 0x00 | Frequency and effect gain parameters | 4 | Yes |
| 0x0018 | WNAIlg | WindNoiseAlg2tc1 | 0x00, 0x00, 0x20, 0x00 | Time constant 1 (ms) | 4 | Yes |
| 0x0019 | WNAIlg | WindNoiseAlg2tc11 | 0x00, 0x7F, 0xE0, 0x00 | Time constant 1 (ms) | 4 | Yes |
| 0x001A | WNAIlg | WindNoiseAlg2tc2 | 0x00, 0x00, 0x20, 0x00 | Time constant 2 (ms) | 4 | Yes |
| 0x001B | WNAIlg | WindNoiseAlg2tc22 | 0x00, 0x7F, 0xE0, 0x00 | Time constant 2 (ms) | 4 | Yes |
| 0x001C | WNAIlg | WindNoiseAlg2Level1 | 0x00, 0x59, 0x99, 0x9A | Level 1 | 4 | No |
| 0x001D | WNAIlg | WindNoiseAlg2Level2 | 0x00, 0x08, 0x00, 0x00 | Level 2 | 4 | No |
| 0x001E | WNAIlg | WindNoiseAlg2attack | 0x00, 0x00, 0x80, 0x00 | Attack (ms) | 4 | Yes |
| 0x001F | WNAIlg | WindNoiseAlg2release | 0x00, 0x00, 0x00, 0x40 | Release (ms) | 4 | Yes |
| 0x0020 | WN | stereomux1940ns40 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x0020 and Address 0x0021 together) | 4 | No |
| 0x0021 | WN | stereomux1940ns41 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Address 0x0020 and Address 0x0021 together) | 4 | No |
| 0x0023 | Locked Cell | param1 | 0x00, 0xCA, 0x9A, 0x58 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0024 | Locked Cell | param2 | 0x0F, 0x35, 0x65, 0xA8 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0025 | Locked Cell | param3 | 0x00, 0x7F, 0xAA, 0xE7 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0026 | Locked Cell | param4 | 0x00, 0x08, 0x38, 0x65 | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0027 | Locked Cell | param5 | 0x00, 0x00, 0x00, 0x00 | Locked parameter (generated by SigmaStudio) | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|-------------|--------------------------|--|---|-------|------------------------|
| 0x0028 | Locked Cell | param6 | 0x00, 0x7B, 0x1A, 0x7E | Locked parameter (generated by SigmaStudio) | 4 | Yes |
| 0x0029 | MicDistance | Gain1940AlgNS1 | 0x00, 0x80, 0x00, 0x00 | Gain setting related to the distance between microphones that enhances the perceived effect | 4 | No |
| 0x002B | SS | stereomux1940ns30 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x002B and Address 0x002C together) | 4 | No |
| 0x002C | SS | stereomux1940ns31 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Address 0x002B and Address 0x002C together) | 4 | No |
| 0x002D 0x002E 0x002F 0x0030 0x0031 | EQFilter | IndexSelMultBandAlg100b2 | 0x00, 0x7F, 0xAA, 0x50, 0xFF, 0x00, 0xAB, 0x60, 0x00, 0x7F, 0xAA, 0x50, 0xFF, 0x80, 0xAB, 0x20, 0x00, 0xFF, 0x54, 0x5F | Biquad F0, 0 | 20 | Yes |
| 0x0032 0x0033 0x0034 0x0035 0x0036 | EQFilter | IndexSelMultBandAlg101b2 | 0x00, 0x7D, 0xBD, 0xAF, 0xFF, 0x02, 0x0A, 0x2E, 0x00, 0x80, 0x42, 0x4A, 0xFF, 0x82, 0x00, 0x07, 0x00, 0xFD, 0xF5, 0xD2 | Biquad F0, 1 | 20 | Yes |
| 0x0037 0x0038 0x0039 0x003A 0x003B | EQFilter | IndexSelMultBandAlg102b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 2 | 20 | Yes |
| 0x003C 0x003D 0x003E 0x003F 0x0040 | EQFilter | IndexSelMultBandAlg103b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 3 | 20 | Yes |
| 0x0041 0x0042 0x0043 0x0044 0x0045 | EQFilter | IndexSelMultBandAlg104b2 | 0x00, 0x71, 0xCB, 0x91, 0xFF, 0x2E, 0xCC, 0xE6, 0x00, 0x81, 0xA0, 0xD2, 0xFF, 0x8C, 0x93, 0x9D, 0x00, 0xD1, 0x33, 0x1A | Biquad F0, 4 | 20 | Yes |
| 0x0046 0x0047 0x0048 0x0049 0x004A | EQFilter | IndexSelMultBandAlg105b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F0, 5 | 20 | Yes |
| 0x004B 0x004C 0x004D 0x004E 0x004F | EQFilter | IndexSelMultBandAlg110b2 | 0x00, 0x7E, 0xFB, 0x24, 0xFF, 0x02, 0x09, 0xB7, 0x00, 0x7E, 0xFB, 0x24, 0xFF, 0x82, 0x06, 0xEE, 0x00, 0xFD, 0xF3, 0x80 | Biquad F1, 0 | 20 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|-----------|--------------------------|--|--------------|-------|------------------------|
| 0x0050 0x0051 0x0052 0x0053 0x0054 | EQFilter | IndexSelMultBandAlg111b2 | 0x00, 0x7D, 0xEB, 0x86, 0xFF, 0x02, 0x83, 0x95, 0x00, 0x7F, 0xC2, 0xF7, 0xFF, 0x82, 0x51, 0x83, 0x00, 0xFD, 0x7C, 0x6B | Biquad F1, 1 | 20 | Yes |
| 0x0055 0x0056 0x0057 0x0058 0x0059 | EQFilter | IndexSelMultBandAlg112b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 2 | 20 | Yes |
| 0x005A 0x005B 0x005C 0x005D 0x005E | EQFilter | IndexSelMultBandAlg113b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 3 | 20 | Yes |
| 0x005F 0x0060 0x0061 0x0062 0x0063 | EQFilter | IndexSelMultBandAlg114b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F1, 4 | 20 | Yes |
| 0x0064 0x0065 0x0066 0x0067 0x0068 | EQFilter | IndexSelMultBandAlg115b2 | 0x00, 0x4A, 0x91, 0x00, 0x00, 0x95, 0x22, 0x00, 0x00, 0x4A, 0x91, 0x00, 0xFF, 0xD1, 0x47, 0xB1, 0xFF, 0x84, 0x74, 0x4F | Biquad F1, 5 | 20 | Yes |
| 0x0069 0x006A 0x006B 0x006C 0x006D | EQFilter | IndexSelMultBandAlg120b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 0 | 20 | Yes |
| 0x006E 0x006F 0x0070 0x0071 0x0072 | EQFilter | IndexSelMultBandAlg121b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 1 | 20 | Yes |
| 0x0073 0x0074 0x0075 0x0076 0x0077 | EQFilter | IndexSelMultBandAlg122b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 2 | 20 | Yes |
| 0x0078 0x0079 0x007A 0x007B 0x007C | EQFilter | IndexSelMultBandAlg123b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 3 | 20 | Yes |
| 0x007D 0x007E 0x007F 0x0080 0x0081 | EQFilter | IndexSelMultBandAlg124b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 4 | 20 | Yes |
| 0x0082 0x0083 0x0084 0x0085 0x0086 | EQFilter | IndexSelMultBandAlg125b2 | 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x80, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00, 0x00 | Biquad F2, 5 | 20 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-----------|-------------------|------------------------|---|-------|------------------------|
| 0x008E | filtS | stereomux1940ns10 | 0x00, 0x00, 0x00, 0x00 | On/off (burst write Address 0x008E and Address 0x008F together) | 4 | No |
| 0x008F | filtS | stereomux1940ns11 | 0x00, 0x80, 0x00, 0x00 | On/off (burst write Address 0x008E and Address 0x008F together) | 4 | No |
| 0x0090 | ShiftLR | Gain1940AlgNS4 | 0x00, 0x80, 0x00, 0x00 | Shift curve left/right; Address 0x0090 and Address 0x0091 must contain the same value | 4 | No |
| 0x0091 | ShiftLR | Gain1940AlgNS5 | 0x00, 0x80, 0x00, 0x00 | Shift curve left/right; Address 0x0090 and Address 0x0091 must contain the same value | 4 | No |
| 0x0092 | Det_Range | Gain1940AlgNS3 | 0x00, 0x20, 0x26, 0xF3 | On/off (burst write Address 0x0092 and Address 0x0093 together) | 4 | No |
| 0x0093 | Det_Range | Gain1940AlgNS2 | 0x00, 0x20, 0x26, 0xF3 | On/off (burst write Address 0x0092 and Address 0x0093 together) | 4 | No |
| 0x0094 | CrossLo | EQwSubDualDP42B1 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0095 | CrossLo | EQwSubDualDP41B1 | 0x00, 0x00, 0x21, 0x4E | Crossover LPF filter coefficient | 4 | Yes |
| 0x0096 | CrossLo | EQwSubDualDP40B1 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0097 | CrossLo | EQwSubDualDP42A1 | 0x0F, 0x88, 0x07, 0xC9 | Crossover LPF filter coefficient | 4 | Yes |
| 0x0098 | CrossLo | EQwSubDualDP41A1 | 0x00, 0xF7, 0xB5, 0x9A | Crossover LPF filter coefficient | 4 | Yes |
| 0x0099 | CrossLo | EQwSubDualDP42B2 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009A | CrossLo | EQwSubDualDP41B2 | 0x00, 0x00, 0x21, 0x4E | Crossover LPF filter coefficient | 4 | Yes |
| 0x009B | CrossLo | EQwSubDualDP40B2 | 0x00, 0x00, 0x10, 0xA7 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009C | CrossLo | EQwSubDualDP42A2 | 0x0F, 0x88, 0x07, 0xC9 | Crossover LPF filter coefficient | 4 | Yes |
| 0x009D | CrossLo | EQwSubDualDP41A2 | 0x00, 0xF7, 0xB5, 0x9A | Crossover LPF filter coefficient | 4 | Yes |
| 0x00A2 | CrossHi | EQwSubDualDP32B1 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A3 | CrossHi | EQwSubDualDP31B1 | 0x0F, 0x08, 0x29, 0x18 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A4 | CrossHi | EQwSubDualDP30B1 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A5 | CrossHi | EQwSubDualDP32A1 | 0x0F, 0x88, 0x07, 0xC9 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A6 | CrossHi | EQwSubDualDP31A1 | 0x00, 0xF7, 0xB5, 0x9A | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A7 | CrossHi | EQwSubDualDP32B2 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|--------------|------------------|------------------------|---|-------|------------------------|
| 0x00A8 | CrossHi | EQwSubDualDP31B2 | 0x0F, 0x08, 0x29, 0x18 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00A9 | CrossHi | EQwSubDualDP30B2 | 0x00, 0x7B, 0xEB, 0x74 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00AA | CrossHi | EQwSubDualDP32A2 | 0x0F, 0x88, 0x07, 0xC9 | Crossover HPF filter coefficient | 4 | Yes |
| 0x00AB | CrossHi | EQwSubDualDP31A2 | 0x00, 0xF7, 0xB5, 0x9A | Crossover HPF filter coefficient | 4 | Yes |
| 0x00B0 | LoDet_Filter | EQwSubDualDP62B1 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B1 | LoDet_Filter | EQwSubDualDP61B1 | 0x00, 0x00, 0x49, 0xC4 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B2 | LoDet_Filter | EQwSubDualDP60B1 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B3 | LoDet_Filter | EQwSubDualDP62A1 | 0x0F, 0x8B, 0xDA, 0xCC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B4 | LoDet_Filter | EQwSubDualDP61A1 | 0x00, 0xF3, 0x91, 0xAC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B5 | LoDet_Filter | EQwSubDualDP62B2 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B6 | LoDet_Filter | EQwSubDualDP61B2 | 0x00, 0x00, 0x49, 0xC4 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B7 | LoDet_Filter | EQwSubDualDP60B2 | 0x00, 0x00, 0x24, 0xE2 | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B8 | LoDet_Filter | EQwSubDualDP62A2 | 0x0F, 0x8B, 0xDA, 0xCC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00B9 | LoDet_Filter | EQwSubDualDP61A2 | 0x00, 0xF3, 0x91, 0xAC | Crossover LPF detection path filter coefficient | 4 | Yes |
| 0x00BE | HiDet_Filter | EQwSubDualDP52B1 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00BF | HiDet_Filter | EQwSubDualDP51B1 | 0x0F, 0x05, 0x76, 0x40 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C0 | HiDet_Filter | EQwSubDualDP50B1 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C1 | HiDet_Filter | EQwSubDualDP52A1 | 0x0F, 0x85, 0x67, 0x55 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C2 | HiDet_Filter | EQwSubDualDP51A1 | 0x00, 0xFA, 0x7A, 0xD5 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C3 | HiDet_Filter | EQwSubDualDP52B2 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|---------------|---------------------------|--|---|-------|------------------------|
| 0x00C4 | HiDet_Filter | EQwSubDualDP51B2 | 0x0F, 0x05, 0x76, 0x40 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C5 | HiDet_Filter | EQwSubDualDP50B2 | 0x00, 0x7D, 0x44, 0xE0 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C6 | HiDet_Filter | EQwSubDualDP52A2 | 0x0F, 0x85, 0x67, 0x55 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00C7 | HiDet_Filter | EQwSubDualDP51A2 | 0x00, 0xFA, 0x7A, 0xD5 | Crossover HPF detection path filter coefficient | 4 | Yes |
| 0x00CC | HighPass_Comp | PeakDBCompLUTAlgPG30decay | 0x00, 0x00, 0x00, 0x04 | REC_Auto: decay | 4 | Yes |
| 0x00CD | HighPass_Comp | PeakDBCompLUTAlgPG30hold | 0x00, 0x00, 0x00, 0x60 | REC_Auto: hold | 4 | Yes |
| 0x00CE 0x00CF 0x00D0 0x00D1 0x00D2 0x00D3 0x00D4 0x00D5 0x00D6 0x00D7 0x00D8 0x00D9 0x00DA 0x00DB 0x00DC 0x00DD 0x00DE 0x00DF 0x00E0 0x00E1 0x00E2 0x00E3 0x00E4 0x00E5 0x00E6 0x00E7 0x00E8 0x00E9 0x00EA 0x00EB 0x00EC 0x00ED 0x00EE | HighPass_Comp | PeakDBCompLUTAlgPG30tab | 0x00, 0x05, 0x1A, 0x96, 0x00, 0x04, 0xB6, 0x9B, 0x00, 0x04, 0xAD, 0x54, 0x00, 0x05, 0x55, 0x55, 0x00, 0x08, 0x55, 0xC1, 0x00, 0x11, 0x6C, 0xA5, 0x00, 0x25, 0x0E, 0xD2, 0x00, 0x4D, 0x9C, 0x93, 0x00, 0xA0, 0x10, 0xF5, 0x01, 0x46, 0xB2, 0x33, 0x02, 0x80, 0x9F, 0x0A, 0x03, 0xF0, 0xCE, 0x62, 0x05, 0x54, 0x59, 0xF2, 0x05, 0xEF, 0xF5, 0xFC, 0x05, 0xBA, 0x65, 0x98, 0x05, 0x17, 0xAD, 0xEF, 0x04, 0x3A, 0x77, 0xC5, 0x03, 0x56, 0xDC, 0x88, 0x02, 0x8B, 0x2A, 0xC5, 0x01, 0xE7, 0x60, 0xA5, 0x01, 0x67, 0x06, 0x39, 0x01, 0x05, 0xD8, 0x28, 0x00, 0xBE, 0x7E, 0xE5, 0x00, 0x8A, 0x46, 0xF6, 0x00, 0x64, 0x17, 0x5B, 0x00, 0x48, 0x3E, 0xCA, 0x00, 0x34, 0x06, 0x48, 0x00, 0x25, 0x6A, 0x11, 0x00, 0x1A, 0xE6, 0xE6, 0x00, 0x13, 0x05, 0xC6, 0x00, 0x0D, 0x73, 0x73, 0x00, 0x09, 0x82, 0xE3, 0x00, 0x06, 0xB9, 0xBA | REC_Auto: compressor curve points | 132 | No |
| 0x00EF | HighPass_Comp | PeakDBCompLUTAlgPG31decay | 0x00, 0x00, 0x00, 0x12 | REC_Manual: decay | 4 | Yes |
| 0x00F0 | HighPass_Comp | PeakDBCompLUTAlgPG31hold | 0x00, 0x00, 0x01, 0xE0 | REC_Manual: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-------------------------|---------------------------|-------------------------|---|-------|------------------------|
| 0x00F1 | HighPass_Comp | PeakDBCompLUTAlgPG31tab | 0x00, 0x1D, 0xC7, 0xFA, | REC_Manual: compressor curve points | 132 | No |
| 0x00F2 | | | 0x00, 0x1F, 0xA6, 0x78, | | | |
| 0x00F3 | | | 0x00, 0x24, 0xCD, 0xBF, | | | |
| 0x00F4 | | | 0x00, 0x2D, 0xB0, 0xDE, | | | |
| 0x00F5 | | | 0x00, 0x3A, 0x4E, 0x7F, | | | |
| 0x00F6 | | | 0x00, 0x47, 0xE3, 0x23, | | | |
| 0x00F7 | | | 0x00, 0x67, 0x41, 0xE8, | | | |
| 0x00F8 | | | 0x00, 0xA1, 0x64, 0x8B, | | | |
| 0x00F9 | | | 0x00, 0xF3, 0x71, 0x72, | | | |
| 0x00FA | | | 0x01, 0x4B, 0x5C, 0x80, | | | |
| 0x00FB | | | 0x01, 0x96, 0x84, 0x19, | | | |
| 0x00FC | | | 0x01, 0xF9, 0x0D, 0x62, | | | |
| 0x00FD | | | 0x02, 0x8A, 0xA7, 0xCA, | | | |
| 0x00FE | | | 0x03, 0x24, 0x67, 0xA2, | | | |
| 0x00FF | | | 0x03, 0x9A, 0x7A, 0x83, | | | |
| 0x0100 | | | 0x03, 0xD1, 0x34, 0x62, | | | |
| 0x0101 | | | 0x03, 0xA2, 0xBD, 0x15, | | | |
| 0x0102 | | | 0x03, 0x3D, 0x07, 0x65, | | | |
| 0x0103 | | | 0x02, 0xC9, 0xE3, 0xC3, | | | |
| 0x0104 | | | 0x02, 0x4D, 0x52, 0x70, | | | |
| 0x0105 | | | 0x01, 0xCE, 0xBA, 0xD0, | | | |
| 0x0106 | 0x01, 0x5C, 0x48, 0xE1, | | | | | |
| 0x0107 | 0x00, 0xFF, 0x34, 0xC9, | | | | | |
| 0x0108 | 0x00, 0xB8, 0x8A, 0x49, | | | | | |
| 0x0109 | 0x00, 0x83, 0xA5, 0x3C, | | | | | |
| 0x010A | 0x00, 0x5D, 0x40, 0x76, | | | | | |
| 0x010B | 0x00, 0x42, 0x04, 0x69, | | | | | |
| 0x010C | 0x00, 0x2E, 0xBC, 0x93, | | | | | |
| 0x010D | 0x00, 0x21, 0x16, 0x45, | | | | | |
| 0x010E | 0x00, 0x17, 0x65, 0x63, | | | | | |
| 0x010F | 0x00, 0x10, 0x8B, 0x23, | | | | | |
| 0x0110 | 0x00, 0x0B, 0xB2, 0xB1, | | | | | |
| 0x0111 | 0x00, 0x08, 0x45, 0x91 | | | | | |
| 0x0112 | HighPass_Comp | PeakDBCompLUTAlgPG32decay | 0x00, 0x00, 0x00, 0xB6 | Speaker: decay | 4 | Yes |
| 0x0113 | HighPass_Comp | PeakDBCompLUTAlgPG32hold | 0x00, 0x00, 0x00, 0xF0 | Speaker: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|---------|-------------------------|-------------------------------|-------------------------|--|-------|------------------------|
| 0x0114 | HighPass_Comp | PeakDBCompLUTAlgPG32tab | 0x00, 0x80, 0x00, 0x00, | Speaker: compressor curve points | 132 | No |
| 0x0115 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0116 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0117 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0118 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0119 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011A | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011B | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011C | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011D | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011E | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x011F | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0120 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0121 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0122 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0123 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0124 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0125 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0126 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0127 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0128 | | | 0x00, 0x80, 0x00, 0x00, | | | |
| 0x0129 | 0x00, 0x80, 0x00, 0x00, | | | | | |
| 0x012A | 0x00, 0x7F, 0xF0, 0x61, | | | | | |
| 0x012B | 0x00, 0x7D, 0x6C, 0x21, | | | | | |
| 0x012C | 0x00, 0x75, 0x7A, 0x64, | | | | | |
| 0x012D | 0x00, 0x68, 0xDB, 0x84, | | | | | |
| 0x012E | 0x00, 0x5A, 0x4C, 0x32, | | | | | |
| 0x012F | 0x00, 0x4C, 0x3E, 0xA8, | | | | | |
| 0x0130 | 0x00, 0x40, 0x26, 0xE7, | | | | | |
| 0x0131 | 0x00, 0x2D, 0x5C, 0xBF, | | | | | |
| 0x0132 | 0x00, 0x20, 0x13, 0x74, | | | | | |
| 0x0133 | 0x00, 0x16, 0xAE, 0x60, | | | | | |
| 0x0134 | 0x00, 0x10, 0x09, 0xBA | | | | | |
| 0x0138 | HighPass_Comp | PeakDBCompLUTAlgPG30post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Auto: post gain | 4 | No |
| 0x0139 | HighPass_Comp | PeakDBCompLUTAlgPG31post_gain | 0x00, 0x80, 0x00, 0x00 | REC_Manual: post gain | 4 | No |
| 0x013A | HighPass_Comp | PeakDBCompLUTAlgPG32post_gain | 0x00, 0x80, 0x00, 0x00 | Speaker: post gain | 4 | No |
| 0x0140 | LowPass_Comp | PeakDBCompLUTAlgPG40decay | 0x00, 0x00, 0x00, 0x04 | REC_Auto: decay | 4 | Yes |
| 0x0141 | LowPass_Comp | PeakDBCompLUTAlgPG40hold | 0x00, 0x00, 0x00, 0x60 | REC_Auto: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|--------------|---------------------------|--|---|-------|------------------------|
| 0x0142 0x0143 0x0144 0x0145 0x0146 0x0147 0x0148 0x0149 0x014A 0x014B 0x014C 0x014D 0x014E 0x014F 0x0150 0x0151 0x0152 0x0153 0x0154 0x0155 0x0156 0x0157 0x0158 0x0159 0x015A 0x015B 0x015C 0x015D 0x015E 0x015F 0x0160 0x0161 0x0162 | LowPass_Comp | PeakDBCompLUTAlgPG40tab | 0x00, 0x05, 0x1A, 0x96, 0x00, 0x04, 0xB6, 0x9B, 0x00, 0x04, 0xAD, 0x54, 0x00, 0x05, 0x55, 0x55, 0x00, 0x08, 0x55, 0xC1, 0x00, 0x11, 0x6C, 0xA5, 0x00, 0x25, 0x0E, 0xD2, 0x00, 0x4D, 0x9C, 0x93, 0x00, 0xA0, 0x10, 0xF5, 0x01, 0x46, 0xB2, 0x33, 0x02, 0x80, 0x9F, 0x0A, 0x03, 0xF0, 0xCE, 0x62, 0x05, 0x54, 0x59, 0xF2, 0x05, 0xEF, 0xF5, 0xFC, 0x05, 0xBA, 0x65, 0x98, 0x05, 0x17, 0xAD, 0xEF, 0x04, 0x3A, 0x77, 0xC5, 0x03, 0x56, 0xDC, 0x88, 0x02, 0x8B, 0x2A, 0xC5, 0x01, 0xE7, 0x60, 0xA5, 0x01, 0x67, 0x06, 0x39, 0x01, 0x05, 0xD8, 0x28, 0x00, 0xBE, 0x7E, 0xE5, 0x00, 0x8A, 0x46, 0xF6, 0x00, 0x64, 0x17, 0x5B, 0x00, 0x48, 0x3E, 0xCA, 0x00, 0x34, 0x06, 0x48, 0x00, 0x25, 0x6A, 0x11, 0x00, 0x1A, 0xE6, 0xE6, 0x00, 0x13, 0x05, 0xC6, 0x00, 0x0D, 0x73, 0x73, 0x00, 0x09, 0x82, 0xE3, 0x00, 0x06, 0xB9, 0xBA | REC_Auto: compressor curve points | 132 | No |
| 0x0163 | LowPass_Comp | PeakDBCompLUTAlgPG41decay | 0x00, 0x00, 0x00, 0x12 | REC_Manual: decay | 4 | Yes |
| 0x0164 | LowPass_Comp | PeakDBCompLUTAlgPG41hold | 0x00, 0x00, 0x01, 0xE0 | REC_Manual: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|--|--------------|---------------------------|--|---|-------|------------------------|
| 0x0165 0x0166 0x0167 0x0168 0x0169 0x016A 0x016B 0x016C 0x016D 0x016E 0x016F 0x0170 0x0171 0x0172 0x0173 0x0174 0x0175 0x0176 0x0177 0x0178 0x0179 0x017A 0x017B 0x017C 0x017D 0x017E 0x017F 0x0180 0x0181 0x0182 0x0183 0x0184 0x0185 | LowPass_Comp | PeakDBCompLUTAlgPG41tab | 0x00, 0x1D, 0xC7, 0xFA, 0x00, 0x1F, 0xA6, 0x78, 0x00, 0x24, 0xCD, 0xBF, 0x00, 0x2D, 0xB0, 0xDE, 0x00, 0x3A, 0x4E, 0x7F, 0x00, 0x47, 0xE3, 0x23, 0x00, 0x67, 0x41, 0xE8, 0x00, 0xA1, 0x64, 0x8B, 0x00, 0xF3, 0x71, 0x72, 0x01, 0x4B, 0x5C, 0x80, 0x01, 0x96, 0x84, 0x19, 0x01, 0xF9, 0x0D, 0x62, 0x02, 0x8A, 0xA7, 0xCA, 0x03, 0x24, 0x67, 0xA2, 0x03, 0x9A, 0x7A, 0x83, 0x03, 0xD1, 0x34, 0x62, 0x03, 0xA2, 0xBD, 0x15, 0x03, 0x3D, 0x07, 0x65, 0x02, 0xC9, 0xE3, 0xC3, 0x02, 0x4D, 0x52, 0x70, 0x01, 0xCE, 0xBA, 0xD0, 0x01, 0x5C, 0x48, 0xE1, 0x00, 0xFF, 0x34, 0xC9, 0x00, 0xB8, 0x8A, 0x49, 0x00, 0x83, 0xA5, 0x3C, 0x00, 0x5D, 0x40, 0x76, 0x00, 0x42, 0x04, 0x69, 0x00, 0x2E, 0xBC, 0x93, 0x00, 0x21, 0x16, 0x45, 0x00, 0x17, 0x65, 0x63, 0x00, 0x10, 0x8B, 0x23, 0x00, 0x0B, 0xB2, 0xB1, 0x00, 0x08, 0x45, 0x91 | REC_Manual: compressor curve points | 132 | No |
| 0x0186 | LowPass_Comp | PeakDBCompLUTAlgPG42decay | 0x00, 0x00, 0x00, 0xB6 | Speaker: decay | 4 | Yes |
| 0x0187 | LowPass_Comp | PeakDBCompLUTAlgPG42hold | 0x00, 0x00, 0x00, 0xF0 | Speaker: hold | 4 | Yes |

| Address | Cell Name | Parameter Name | Default Value | Function | Bytes | Sample Rate Dependent? |
|------------------|-----------|-----------------------------|---|-------------------------------------|-------|------------------------|
| 0x01B6 | dmute | MuteSWLinSlewAlg1mute | 0x00, 0x00, 0x00, 0x00 | Mute digital (record) output | 4 | No |
| 0x01B7 | dmute | MuteSWLinSlewAlg1step | 0x00, 0x00, 0x40, 0x00 | Slew rate for digital mute | 4 | Yes |
| 0x01B8 | slewwol | GainS200AlgGrow1gain_target | 0x00, 0x80, 0x00, 0x00 | Analog output volume control | 4 | No |
| 0x01BA | amute | MuteSWLinSlewAlg2mute | 0x00, 0x00, 0x00, 0x00 | Mute analog (playback) output | 4 | No |
| 0x01BB | amute | MuteSWLinSlewAlg2step | 0x00, 0x00, 0x40, 0x00 | Slew rate for analog mute | 4 | Yes |
| 0x07FA 0x07FB | slewwol | GainS200AlgGrow1alpha | 0x00, 0x7F, 0xF2, 0x59, 0x00, 0x00, 0x0D, 0xA7 | Slew rate for analog volume control | 8 | Yes |

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