

The World Leader in High Performance Signal Processing Solutions



# VisualAudio Advanced Features

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# About this Module

- ◆ **This module provides advanced training on VisualAudio. Examples and demonstrations will be based on the ADSP-BF533 EZ-KIT. You will learn about:**
  - **Advanced tool features such as high and low-level variables, the expression language, and presets.**
  - **How to use the external interface to control VisualAudio from other applications, such as MATLAB.**
  - **The basics of writing audio modules.**
  
- ◆ **Target Audience**
  - **Audio algorithm developers**
  - **Comfortable writing C code**
  - **Some familiarity with Blackfin processors and the VisualDSP++ development environment**



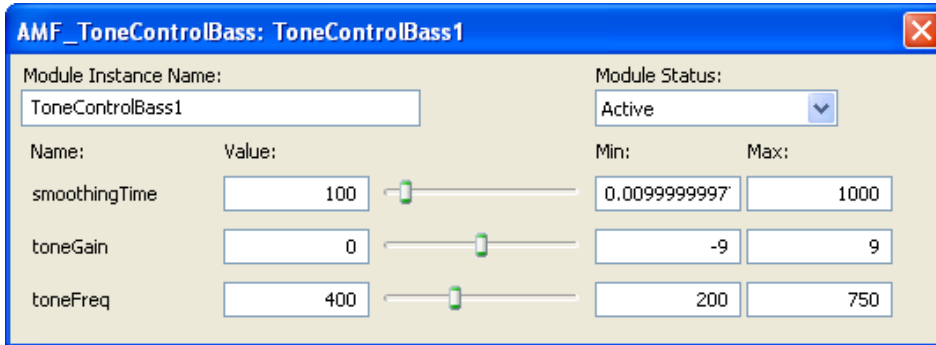
# Module Outline

- ◆ **VisualAudio Designer advanced features**
  - High and low-level parameters
  - The expression language
  - Presets
- ◆ **Using the external interface**
- ◆ **Writing custom audio modules**
- ◆ **Conclusion**



# VisualAudio Designer Advanced Features

# High and Low-Level Module Variables

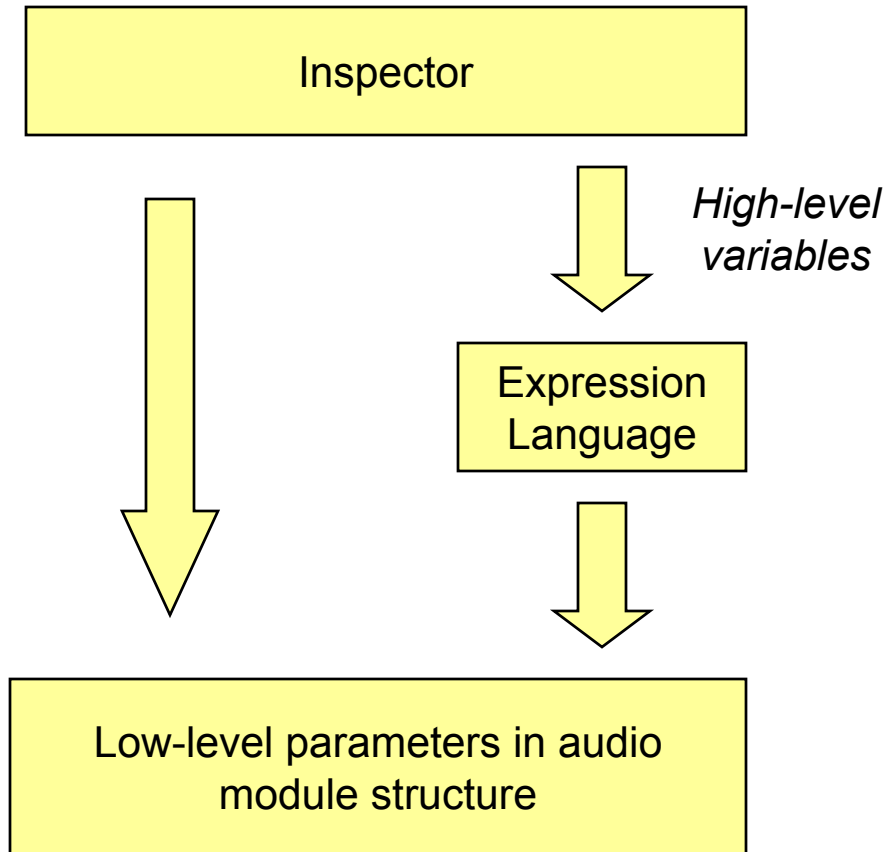


*High-level variables appear on a module's inspector*

```
typedef struct {  
    AMF_Module b;  
  
    float ampSmoothing, ampTarget, b0;  
    float amp, state;  
  
} AMF_ToneControlBass;
```

*Low-level (or render) variables appear within the module's data structure*

# Expression Language



The expression language converts between high-level and low-level parameters

Expression language examples:

1. Convert from smoothing time (msecs) to coefficient
2. Convert from dB to linear units.
3. Convert a balance control setting into 2 separate gains

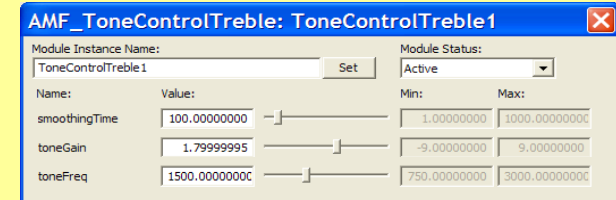
# Presets

- ◆ Convenient mechanism for managing audio module parameter sets
- ◆ Step 1 – Tune the system to a desired state
  - Inspectors
  - External interface
- ◆ Step 2 – Capture the preset
- ◆ Step 3 – Apply the preset from the Tool
- ◆ Step 4 – Optionally compile the preset with the application.
- ◆ Presets are written in Intel hex format
- ◆ Can be stored on host and downloaded to the DSP
- ◆ Typical uses
  - Dealing with multiple sample rates
  - Preserving default EQ settings
  - Making A/B comparisons to fine tune system performance

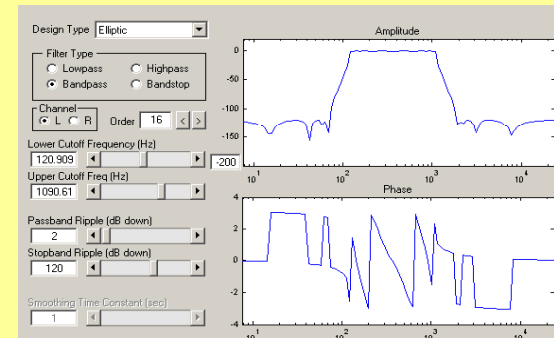
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## Inspectors

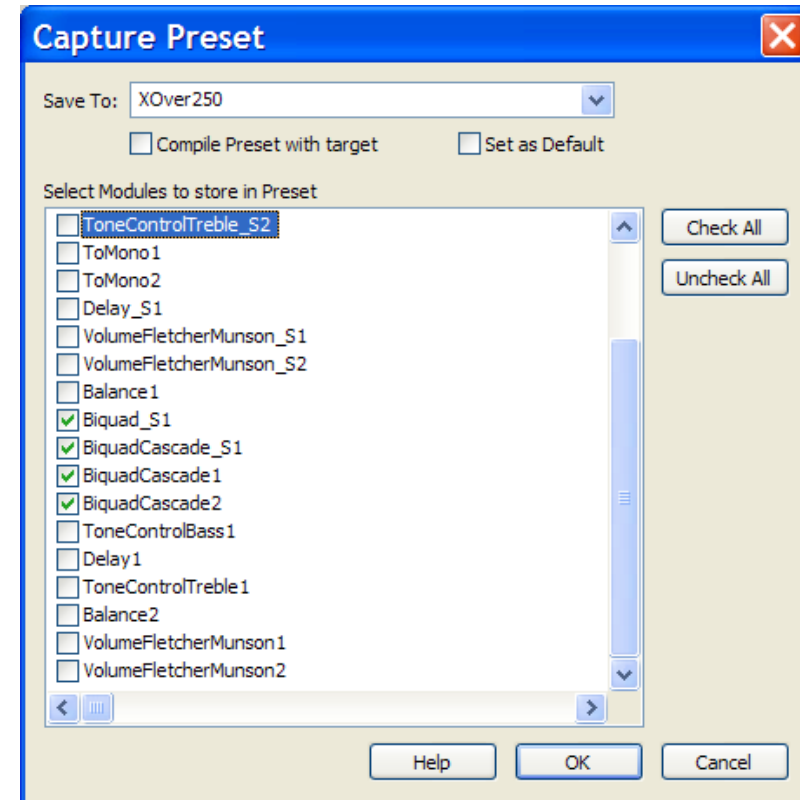


## External Interface



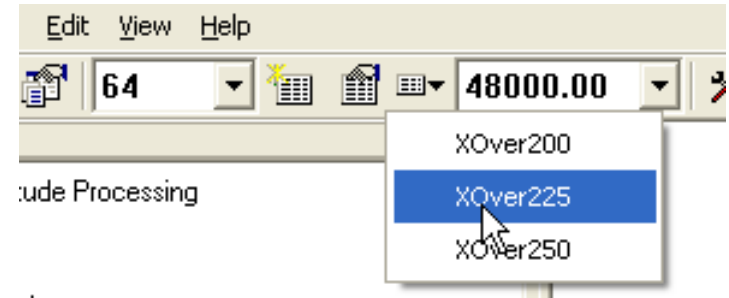
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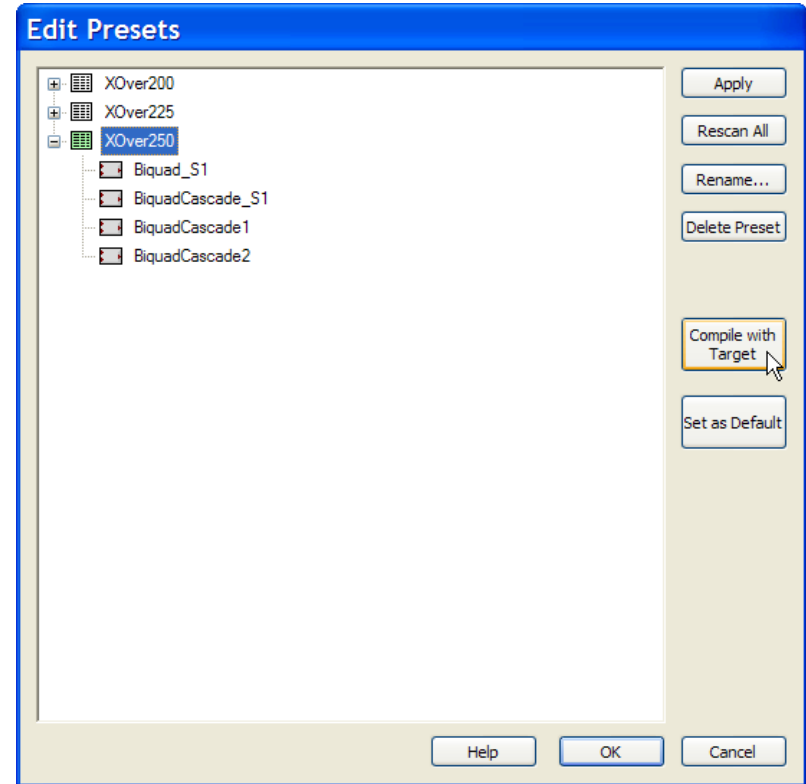
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# The External Interface



# External Interface

- ◆ **Works in both Design Mode and Tuning Mode**
  - Design mode → module data structures
  - Tuning mode → module data structures AND sent to the DSP in real-time
- ◆ **Capabilities**
  - Manipulating audio module parameters
  - Basic control of the system (loading, saving, building, capturing presets, etc.) is also supported
  - Advanced control (instantiating and wiring modules)
  - Exchanging audio data with the target processor.
    - ◆ Block-by-block
    - ◆ Non-real-time
    - ◆ Speed is determined by the speed of the tuning interface
- ◆ **Implemented as a local COM server (housed in an EXE)**
  - Accessible by any COM compliant language/application (C/C++, Excel, VisualBasic, etc.)
  - Total of 53 APIs supported
  - Prog-ID is 'VisualAudioDesigner'

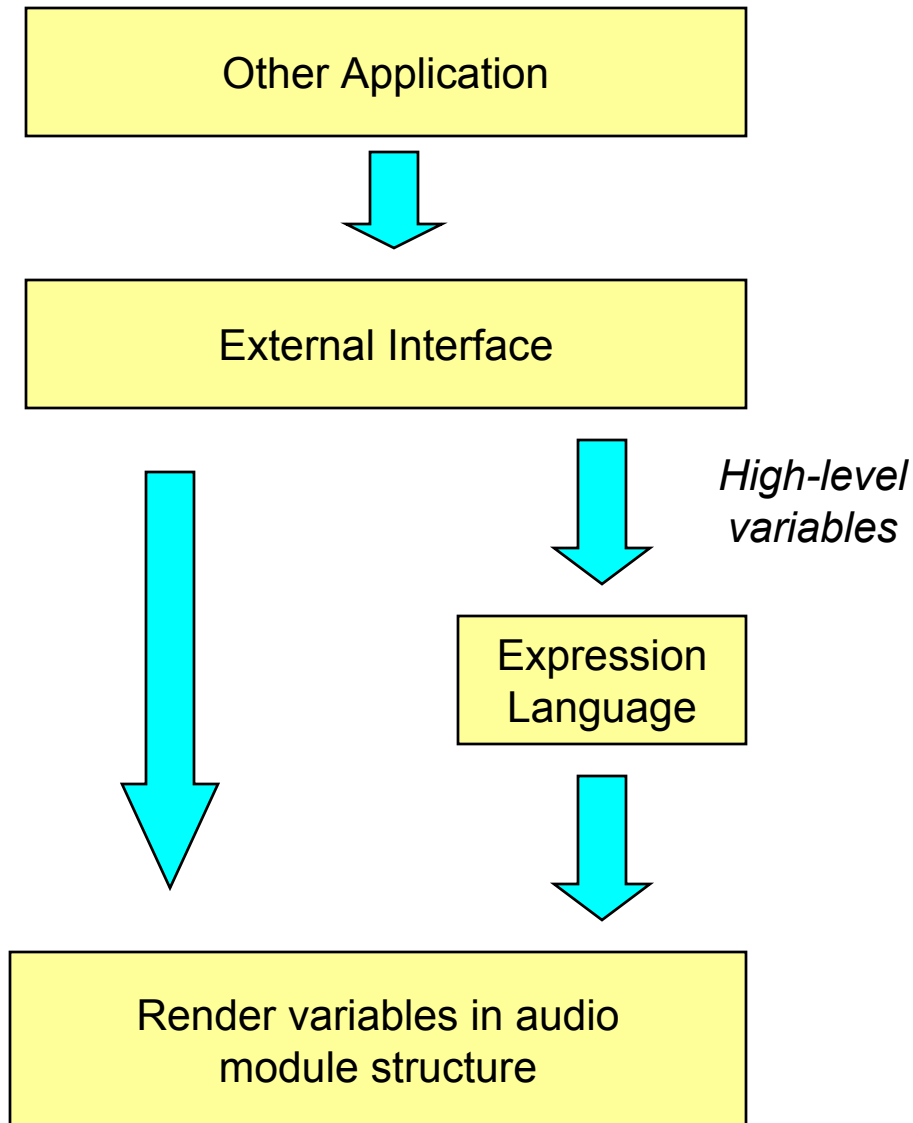


# Uses of the External Interface

- ◆ **Creating custom audio module design functions**
- ◆ **Creating custom GUIs**
  - **Control panels**
  - **Full or restricted functionality**
- ◆ **Leveraging existing design tools and methodologies**
- ◆ **Automating system design and tuning**
- ◆ **Regression testing of audio modules and systems**



# Expression Language is Included

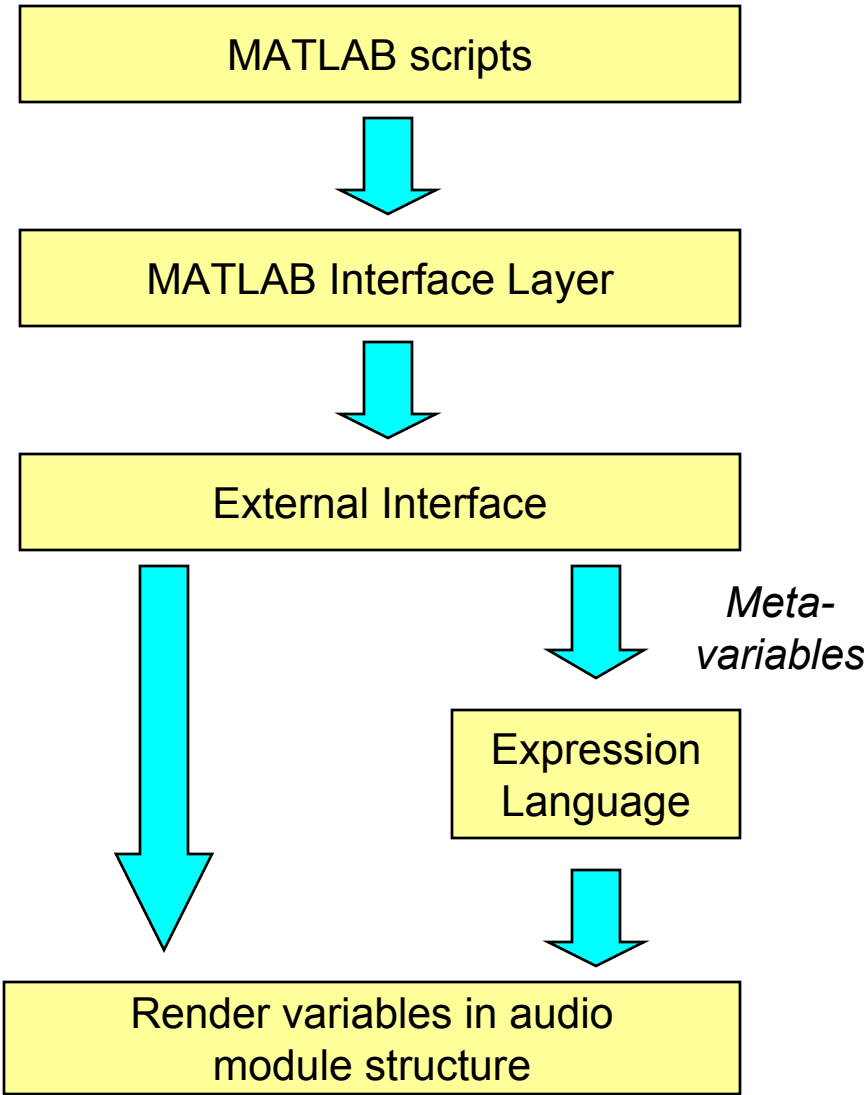


External applications can access the high-level and low-level render variables.

Changes to high-level variables invoke the expression language.

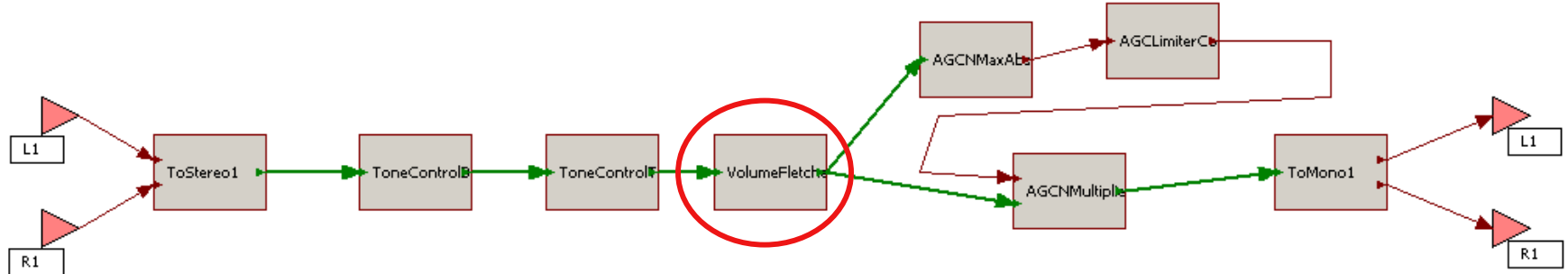
Low-level accesses bypass the expression language and manipulate DSP variables directly.

# MATLAB Interface Layer



- ◆ Simplifies usage with MATLAB
- ◆ Each audio module appears as a MATLAB object
- ◆ Objects can be manipulated as if they were MATLAB structures

# Querying a Single Audio Module



```
>> S=va_module('VolumeFletcherMunson_S1');
```

- ◆ Queries VisualAudio for information regarding this audio module
  - Variables names
  - Data types
  - Sizes
- ◆ Generates and returns a MATLAB object

# Comparison with the Inspector

S =

```
smoothingTime: [ 100]
fmGain: [ 0]
lowFreq: [ 30]
lowQ: [ 1]
```

AMF\_VolumeFletcherMunson\_S: VolumeFletcherMunson\_S1

Module Instance Name: VolumeFletcherMunson\_S1

Module Status: Active

Name:	Value:	Min:	Max:
smoothingTime	100	0.00999999997	1000
fmGain	0	-120	0
lowFreq	30	20	40
lowQ	1	0.5	2

One-to-one correspondence between MATLAB structure members and interface variables shown on the inspector



# Manipulating Module Parameters

- ◆ **Treat them as if they were standard MATLAB structures:**

```
VolumeFletcherMunson_S1.smoothingTime=50;
```

```
VolumeFletcherMunson_S1.fmGain=-3;
```

```
VolumeFletcherMunson_S1.lowFreq=40;
```

```
VolumeFletcherMunson_S1.lowQ=1.1;
```

- ◆ **Often used to initialize parameters in a repeatable method using a script file.**

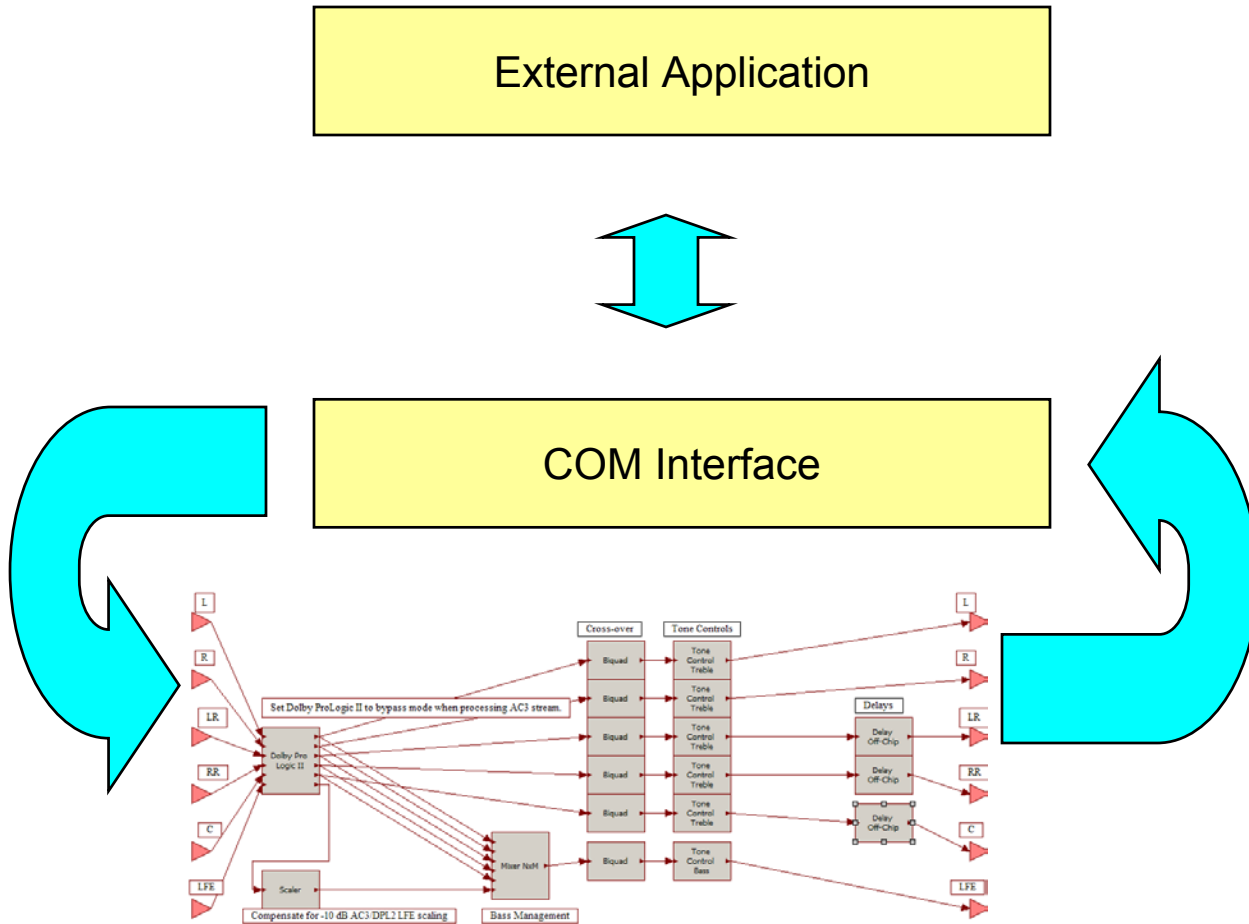
# Accessing Low-Level Render Variables

- ◆ The previous example demonstrated how to access the high-level variables shown on the inspector.
- ◆ To access low-level render variables in Tuning Mode, issue the command:

```
va_module('VolumeFletcherMunson_S1', 0)
```

```
VolumeFletcherMunson_S1 =
    ampSmoothing: [ 0.002081]
      ampTarget: [ 1.000000]
    lowAmpTarget: [ 1.000000]
          b0: [ 0.996081]
          b1: [ -1.996061]
          amp: [ 1.000000]
          lowAmp: [ 1.000000]
    aux_state1: [ 777.399841
                -438.739838]
    aux_state2: [ -21.297104
                177.371170]
```

# Regression Testing Capabilities



Platform operates in “demand render mode”

External application generates data.

Audio passed block-by-block through the tuning interface.

External application analyzes data for correctness.

MATLAB examples are provided.



# MATLAB Testing API

- ◆ Place platform into “demand render” mode

```
va_demandrender( 'begin' );
```

- ◆ Send and receive individual blocks of data

```
DATA_OUT=va_demandrender( 'process' , DATA_IN)
```

*DATA\_IN=TickSize x NumberOfInputs*

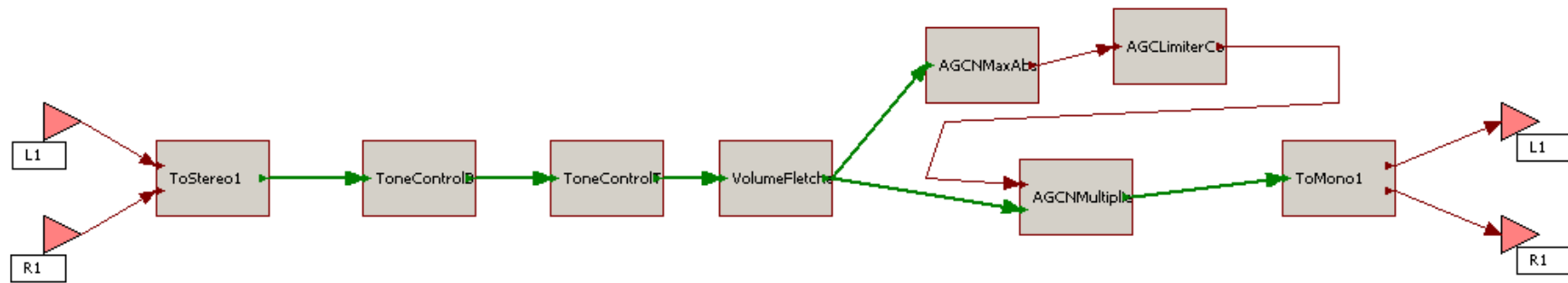
*DATA\_OUT=TickSize x NumberOfOutputs*

- ◆ Repeat for multiple blocks

- ◆ Exit demand render mode and resume real-time processing

```
va_demandrender( 'end' );
```

# Tone Control Example



◆ Place all modules into bypass mode except the treble tone control

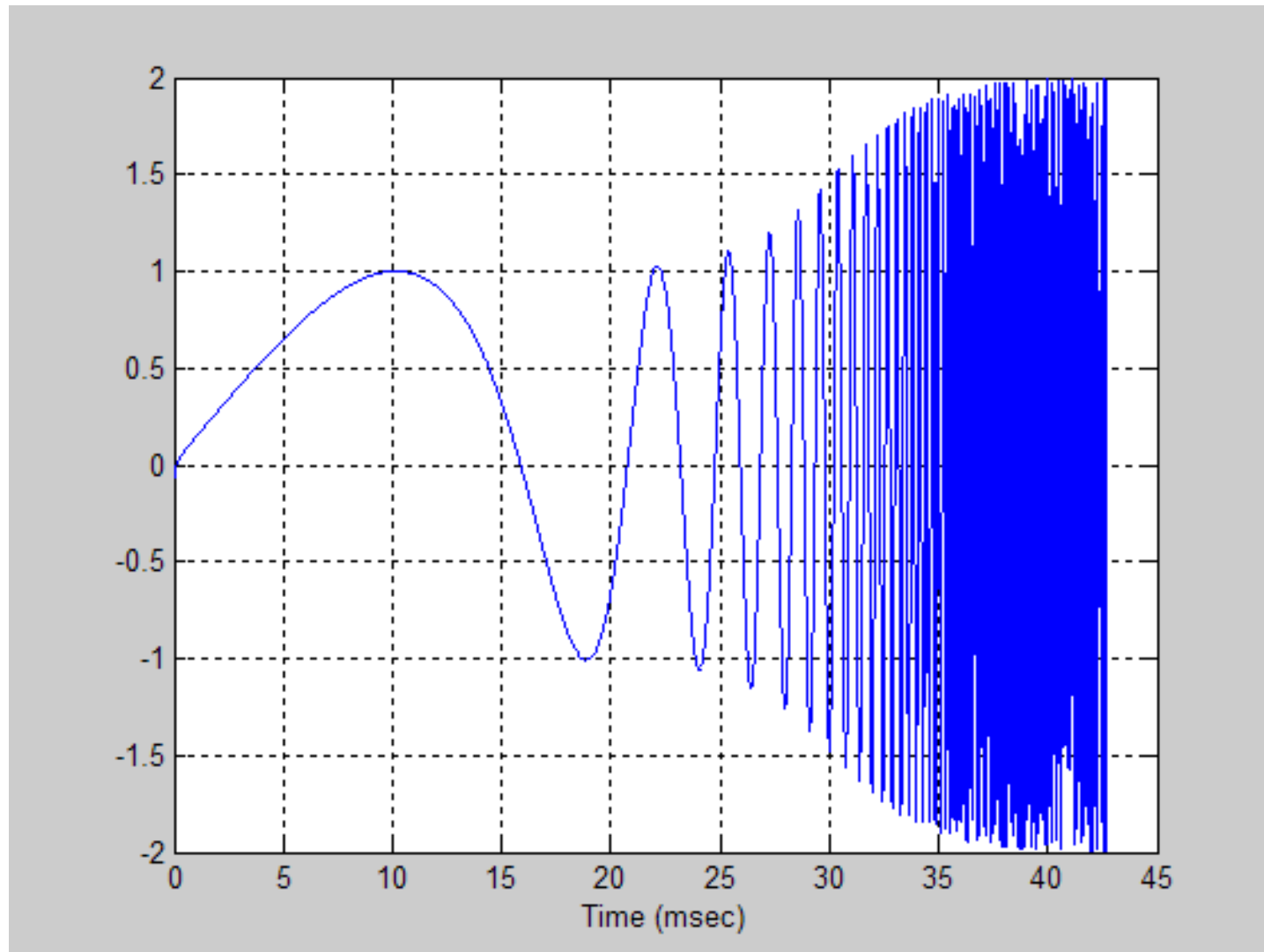
◆ Generate input data in MATLAB

```
t = ((0:2047)/48000).';
DATA_IN = chirp(t, 20, t(end), 24000, 'logarithmic', -90) * ones(1,2);
```

◆ Process the data

```
va_demandrender('begin');
DATA_OUT = va_demandrender('process', DATA_IN);
va_demandrender('end');
```

# Tone Control Results

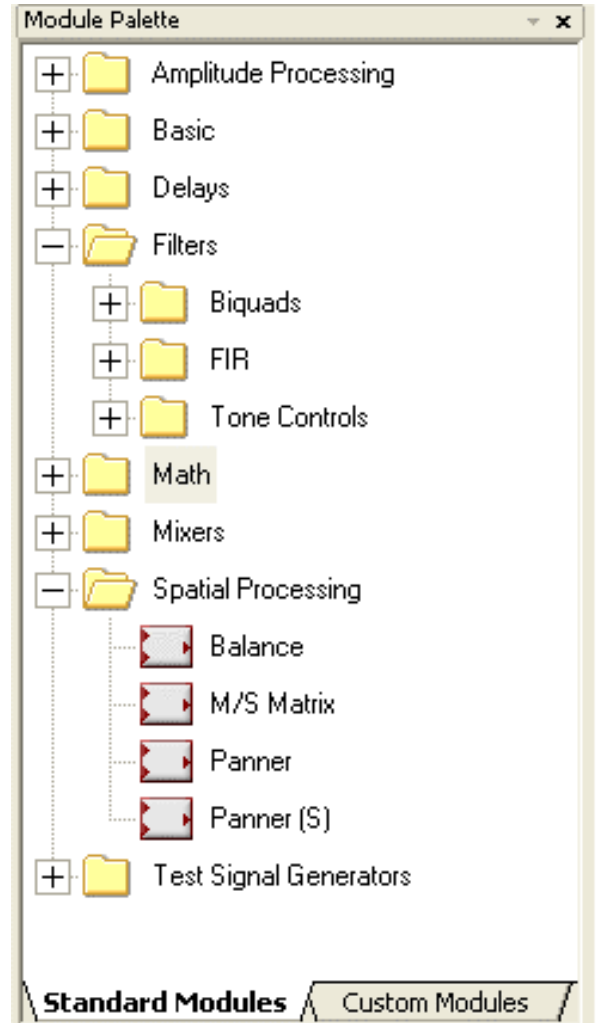




# Writing Custom Audio Modules

# Standard vs. Custom Modules

- ◆ **Standard modules are supplied with VisualAudio**
- ◆ **Custom modules are written by the user**
- ◆ **Standard and custom modules appear on separate tabs within VisualAudio Designer. This is the only distinction between standard and custom modules – no limitations or cost overhead associated with custom modules**
- ◆ **Source code is provided for all standard modules. This serves as a starting point for creating custom modules**



## 3 Components of an Audio Module

- ◆ **A header file which contains the module's run-time interface and a description of the associated data structure**
- ◆ **The module's run-time function ("render function"). This can be:**
  - **C code**
  - **ASM code**
  - **Object or library**
- ◆ **An XML file that describes the module in detail to VisualAudio**
  - **Elements of its data structure**
  - **Inspector interface**
  - **Memory allocation rules**



# Instance Data Structure

- ◆ **Each instance of an audio module has an associated C data structure**
- ◆ **All data structures start with the same set of fields**
  - **These contain elements common to all audio modules**
  - **Describe the “base class” of the “object”**
- ◆ **This is followed by module specific fields**

# AMF\_ScalerSmoothed.h

## ◆ Defines the data structure associated with the audio module

```
#include "AudioProcessing.h"
```

```
typedef struct  
{
```

```
    AMF_Module b;
```

```
    // parameters  
    fract32 ampSmoothing;  
    fract32 oneOverTickSize;
```

```
    // state  
    fract32 amp;
```

```
    // parameters  
    fract16 ampTarget;
```

```
}  
AMF_ScalerSmoothed;
```

Common header

Module specific variables

typedef name



# AMF\_ScalerSmoothed.c – Render Function (C example. Actual code is in ASM.)

```
SEG_MOD_FAST_CODE void AMF_ScalerSmoothed_Render(AMF_ScalerSmoothed * instance,
        AMF_Signal ** buffers, int tickSize) {
    int i;
    fract32 amp = instance->amp;
    fract16 ampTarget = instance->ampTarget;
    fract32 ampSmoothing = instance->ampSmoothing;
    AMF_Signal *in = buffers[0];
    AMF_Signal *out = buffers[1];
    fract32 diff, slew;

    /* compute smoothing filter only once per tick, and derive a linear
     * slew for the per-sample update */

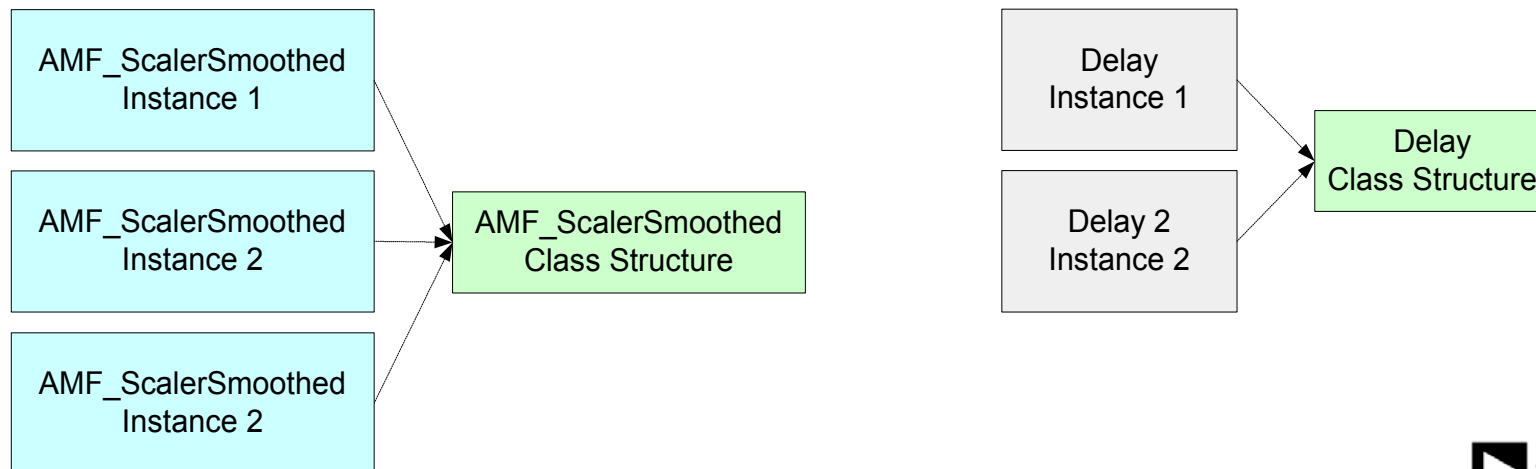
    diff = sub_fr1x32(mult_fr1x32x32NS(ampSmoothing, L_deposit_h(ampTarget)),
        mult_fr1x32x32NS(ampSmoothing, amp));
    instance->amp = add_fr1x32(amp, diff);
    slew = mult_fr1x32x32NS(diff, instance->oneOverTickSize);

    for (i=0; i<tickSize; i++) {
        out[i] = mult_fr1x32x32NS(in[i], amp);
        amp = add_fr1x32(amp, slew);
    }
}
```

- ◆ **Same arguments passed to all render functions:**
  - **Pointer to instance structure**
  - **Array of buffer pointers ordered as inputs, outputs, and scratch**
  - **tickSize = block size**

# Class Structure Declaration

- ◆ All audio modules of the same type share a single “Class Structure”
- ◆ Describes the behavior of the module to VisualAudio’s run-time interface
  - Number of inputs and outputs
  - Mono input and mono output
  - Name of render function
  - Bypass behavior
- ◆ Typically declared within the module’s C file
  - If the module’s render function is in assembly, there will be two files: an .ASM file containing the render function, and a C file with the class structure declaration





# AMF\_ScalerSmoothed – Class Structure

```
SEG_MOD_SLOW_CONST const AMF_ModuleClass AMFClassScalerSmoothed =
{
    /** Flags. */
    0,

    /** Reference to render function. */
    (AMF_RenderFunction) AMF_ScalerSmoothed_Render,

    /* Default bypass */
    (void *)0,

    /* Input descriptor - 1 input, and it is mono. */
    1, 0,

    /* Output descriptor - 1 output, and it is mono. */
    1, 0,
};
```

# Audio Module XML

- ◆ **Describes the audio module to VisualAudio Designer**
  - **Module name and palette location**
  - **Input and output pins**
  - **Compatible processors**
  - **Instance data structure**
  - **High-level variables and expressions**
  - **Memory allocation rules**
  - **Other usage rules**



# Conclusion

- ◆ **VisualAudio's design features simplify the development of advanced audio features**
  - **High and Low-level variables**
  - **Expression language**
  - **Presets**
- ◆ **Open API's allow VisualAudio's capabilities to be extended by**
  - **Interfacing to external COM compliant applications**
  - **Interfacing to MATLAB**
  - **Writing custom audio modules**



## For Additional Information

- ◆ **A free download is available at the VisualAudio product page**
  - <http://www.analog.com/en/prod/0,2877,VISUALAUDIO,00.html>
  
- ◆ **Additional examples and tutorials can be found at the VisualAudio Developer's Web site:**
  - [www.visualaudiodeveloper.com](http://www.visualaudiodeveloper.com)
  
- ◆ **Specific technical questions can be sent to:**
  - [visualaudio.support@analog.com](mailto:visualaudio.support@analog.com)
  
- ◆ **Click the “Ask A Question” button**