

# MMMAudio

# Documentation

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DSP library for Python and Mojo

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# 1. MMMAudio Documentation

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Welcome to the MMMAudio documentation! MMMAudio is a high-performance audio processing library that combines the ease of Python with the speed of Mojo for real-time audio applications.

## 1.1 Features

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- **High Performance:** Leverages Mojo's SIMD capabilities for optimal audio processing
- **Dual Language Support:** Write audio logic in Python, optimize critical paths in Mojo
- **Real-time Capable:** Designed for low-latency audio applications
- **Modular Design:** Composable DSP building blocks
- **ML Integration:** Support for neural network audio processing

## 1.2 Quick Start

---

```
from mmm_src.MMMAudio import MMMAudio
from mmm_dsp.Osc import SineOsc

# Create audio engine
audio = MMMAudio(sample_rate=44100, buffer_size=512)

# Create and connect oscillator
osc = SineOsc(frequency=440.0)
audio.connect(osc, audio.output)

# Start processing
audio.start()
```

## 1.3 Documentation Structure

---

- **Getting Started:** Installation and basic usage
- **API Reference:** Complete API documentation
- **Examples:** Practical usage examples
- **Development:** Contributing and development guide

## 1.4 Architecture

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MMMAudio is built around a graph-based processing model where audio flows through connected nodes. Each node can be implemented in either Python (for flexibility) or Mojo (for performance).

### 1.4.1 Core Components

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- **DSP Modules** (`mmm_dsp`): Basic audio processing building blocks
- **Framework** (`mmm_src`): Audio engine and graph management
- **Utilities** (`mmm_utils`): Mathematical and utility functions

## 1.5 Community

---

- **GitHub:** <https://github.com/tedmoore/MMMAudio>
- **Issues:** Report bugs and feature requests on GitHub

## 2. Getting Started with MMMAudio

---

MMMAudio is a high-performance audio processing library that combines Python's ease of use with Mojo's performance for real-time audio applications.

### 2.1 Installation

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#### 2.1.1 Prerequisites

---

- Python 3.9 or higher
- Mojo compiler (latest version)
- Audio drivers (ASIO on Windows, CoreAudio on macOS, ALSA on Linux)

#### 2.1.2 Install from Source

---

1. Clone the repository:

```
git clone https://github.com/tedmoore/MMMAudio.git
cd MMMAudio
```

1. Install Python dependencies:

```
pip install -r requirements.txt
```

1. Verify Mojo installation:

```
mojo --version
```

## 2.2 Quick Start

---

### 2.2.1 Basic Audio Setup

---

Create a simple audio processing chain:

```
from mmm_src.MMMAudio import MMMAudio
from mmm_dsp.Osc import SineOsc

# Initialize audio engine
audio = MMMAudio(
    sample_rate=44100,
    buffer_size=512,
    channels=2
)

# Create a sine wave oscillator
osc = SineOsc(frequency=440.0, amplitude=0.5)

# Connect oscillator to audio output
audio.connect(osc, audio.output)

# Start audio processing
audio.start()

# Let it run for 5 seconds
import time
time.sleep(5)

# Stop audio
audio.stop()
```

### 2.2.2 Using Mojo for Performance

---

For performance-critical operations, use Mojo implementations:

```

from mmm_utils.functions import linlin
from algorithm import parallelize

# Process control data with SIMD optimization
midi_velocities = [64, 80, 96, 127] # MIDI velocity values
gains = []

for velocity in midi_velocities:
    # Convert MIDI velocity to linear gain using Mojo function
    gain = linlin(float(velocity), 0.0, 127.0, 0.0, 1.0)
    gains.append(gain)

print(f"Converted gains: {gains}")

```

## 2.3 Core Concepts

---

### 2.3.1 Audio Graph

---

MMMAudio uses a graph-based processing model where audio flows through connected nodes:

```

# Create nodes
input_node = audio.input
osc1 = SineOsc(440.0)
osc2 = SineOsc(880.0)
mixer = Mixer(2) # 2-input mixer
output_node = audio.output

# Connect the graph
audio.connect(osc1, mixer.input[0])
audio.connect(osc2, mixer.input[1])
audio.connect(mixer, output_node)

```

### 2.3.2 SIMD Optimization

---

Mojo functions support SIMD operations for processing multiple values simultaneously:

```

# Process 4 frequencies at once
from mmm_utils.functions import midicps

midi_notes = SIMD[DType.float64, 4](60.0, 64.0, 67.0, 72.0) # C major chord
frequencies = midicps[4](midi_notes) # Convert to frequencies

```

### 2.3.3 Real-time Processing

---

MMMAudio is designed for real-time audio with low latency:

```

# Configure for low latency
audio = MMMAudio(
    sample_rate=44100,
    buffer_size=128, # Small buffer for low latency
    channels=2
)

# Use efficient processing chains
reverb = Reverb(room_size=0.5, damping=0.7)
audio.connect(audio.input, reverb)
audio.connect(reverb, audio.output)

```

## 2.4 Examples

---

### 2.4.1 Simple Synthesizer

---

```

from mmm_src.MMMAudio import MMMAudio
from mmm_dsp.Osc import SineOsc
from mmm_dsp.Env import ADSR
from mmm_dsp.Filters import LowPass

# Create audio engine
audio = MMMAudio()

# Create synthesis components
osc = SineOsc(frequency=440.0)
envelope = ADSR(attack=0.1, decay=0.2, sustain=0.7, release=0.5)
filter = LowPass(cutoff=2000.0, resonance=0.5)

```

```
# Build signal chain
audio.connect(osc, filter)
audio.connect(envelope, filter.cutoff_mod) # Envelope modulates filter
audio.connect(filter, audio.output)

# Start synthesis
audio.start()
envelope.trigger() # Trigger note

time.sleep(2)
envelope.release() # Release note
time.sleep(1)

audio.stop()
```

## 2.4.2 Multi-channel Processing

```
# Stereo processing with different effects per channel
audio = MMMAudio(channels=2)

# Create stereo sources
osc_left = SineOsc(440.0)
osc_right = SineOsc(442.0) # Slightly detuned for stereo effect

# Different processing per channel
delay_left = Delay(time=0.3, feedback=0.3)
delay_right = Delay(time=0.4, feedback=0.2)

# Connect stereo processing
audio.connect(osc_left, delay_left)
audio.connect(osc_right, delay_right)
audio.connect(delay_left, audio.output.left)
audio.connect(delay_right, audio.output.right)

audio.start()
```

## 2.5 Performance Tips

### 2.5.1 Use SIMD When Possible

```
# Instead of processing one value at a time:
for i in range(len(values)):
    result[i] = linlin(values[i], 0.0, 1.0, 20.0, 20000.0)

# Process multiple values with SIMD:
from mmm_utils.functions import linlin

# Convert list to SIMD and process all at once
values_simd = SIMD[DType.float64, 4].from_list(values[:4])
results_simd = linlin[4](values_simd, 0.0, 1.0, 20.0, 20000.0)
```

### 2.5.2 Optimize Buffer Sizes

```
# Balance latency vs CPU usage
audio = MMMAudio(
    buffer_size=256, # Good balance for most applications
    sample_rate=44100
)

# For very low latency (may increase CPU usage):
audio_low_latency = MMMAudio(buffer_size=64)

# For maximum efficiency (higher latency):
audio_efficient = MMMAudio(buffer_size=1024)
```

### 2.5.3 Reuse Objects

```
# Create objects once and reuse
osc = SineOsc()

# Change parameters instead of creating new objects
osc.set_frequency(880.0)
osc.set_amplitude(0.5)

# This is more efficient than:
# osc = SineOsc(frequency=880.0, amplitude=0.5) # Creates new object
```

## 2.6 Troubleshooting

---

### 2.6.1 Audio Dropouts

---

If you experience audio dropouts:

1. Increase buffer size:

```
audio = MMMAudio(buffer_size=512) # or higher
```

2. Reduce processing complexity in real-time callbacks
3. Use Mojo implementations for CPU-intensive operations

### 2.6.2 Import Errors

---

Make sure the project is in your Python path:

```
import sys
sys.path.append('/path/to/MMMAudio')
```

Or install in development mode:

```
pip install -e .
```

### 2.6.3 Mojo Compilation Issues

---

Ensure you have the latest Mojo compiler:

```
mojo --version
```

Update if necessary according to Mojo documentation.

## 2.7 Next Steps

---

- Explore the [API Reference](#) for detailed function documentation
- Check out [Examples](#) for more complex usage patterns
- Read the [Development Guide](#) to contribute

## 2.8 Community and Support

---

- **GitHub Issues:** Bug reports and feature requests
- **Discussions:** Questions and community interaction
- **Documentation:** Complete API reference and guides

Happy audio processing with MMMAudio!

## 3. API Reference

---

### 3.1 API Reference

---

This section contains the complete API reference for MMMAudio, organized by module.

#### 3.1.1 Core DSP (`mmm_dsp`)

---

The core DSP modules provide the fundamental building blocks for audio processing:

- **Utilities:** Mathematical utility functions
- **Oscillators:** Sine, square, sawtooth, and noise generators
- **Filters:** Low-pass, high-pass, band-pass filters
- **Envelopes:** ADSR and other envelope generators
- **Delays:** Delay lines and echo effects
- **Buffers:** Audio buffer management
- **Effects:** Distortion and saturation

#### 3.1.2 Framework (`mmm_src`)

---

The framework modules handle audio engine management and processing:

- **Audio Engine:** Main audio processing engine
- **Graph System:** Audio graph management
- **Traits:** Core interfaces and traits
- **Scheduler:** Event scheduling and timing

#### 3.1.3 Utilities (`mmm_utils`)

---

Utility modules provide mathematical functions and helpers:

- **Functions:** Mathematical utility functions
- **FFT:** Fast Fourier Transform utilities
- **Windows:** Window functions for DSP

#### 3.1.4 Documentation Conventions

---

##### Function Signatures

Functions are documented with their complete signatures including type information:

```
fn llinlin(N: Int = 1) {
  value: SIMD[DType.float64, N],
  in_min: SIMD[DType.float64, N],
  in_max: SIMD[DType.float64, N],
  out_min: SIMD[DType.float64, N],
  out_max: SIMD[DType.float64, N]
} -> SIMD[DType.float64, N]
```

##### SIMD Support

Most functions support SIMD operations for processing multiple values simultaneously. The `N` parameter controls the SIMD width.



**Examples**

Each function includes practical examples showing typical usage patterns.

## 3.2 Core DSP

### 3.2.1 functions

#### MMM Utility Functions

This module provides essential utility functions for audio processing and mathematical operations in the MMMAudio framework. All functions are optimized for SIMD operations to achieve maximum performance on modern processors.

The functions in this module include: - Range mapping functions (linear and exponential) - Clipping and wrapping utilities - Interpolation algorithms - MIDI/frequency conversion - Audio utility functions - Random number generation

All functions support vectorized operations through SIMD types for processing multiple values simultaneously.

#### Functions

##### LINLIN

Maps values from one range to another range linearly.

This function performs linear mapping from an input range to an output range. Values outside the input range are clamped to the corresponding output boundaries. This is commonly used for scaling control values, normalizing data, and converting between different parameter ranges.

Examples:

```
# Map MIDI velocity (0-127) to gain (0.0-1.0)
velocity = SIMD[DType.float64, 1](64.0)
gain = linlin(velocity, 0.0, 127.0, 0.0, 1.0) # Returns 0.504

# Map multiple control values simultaneously
controls = SIMD[DType.float64, 4](0.25, 0.5, 0.75, 1.0)
frequencies = linlin[4](controls, 0.0, 1.0, 20.0, 20000.0)

# Invert a normalized range
normal_vals = SIMD[DType.float64, 2](0.3, 0.7)
inverted = linlin[2](normal_vals, 0.0, 1.0, 1.0, 0.0)
```

##### Signature

```
linlin[N: Int = 1](value: SIMD[DType.float64, N], in_min: SIMD[DType.float64, N], in_max: SIMD[DType.float64, N], out_min: SIMD[DType.float64, N], out_max: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

##### Parameters

- **N:** `Int` - Size of the SIMD vector (defaults to 1).

##### Arguments

- **value:** `SIMD` - The values to map.- **in\_min:** `SIMD` - The minimum of the input range.- **in\_max:** `SIMD` - The maximum of the input range.- **out\_min:** `SIMD` - The minimum of the output range.- **out\_max:** `SIMD` - The maximum of the output range.

##### Returns

**Type:** `SIMD` The linearly mapped values in the output range.

##### LINEXP

Maps values from one linear range to another exponential range.

This function performs exponential mapping from a linear input range to an exponential output range. This is essential for musical applications where frequency perception is logarithmic. Both output range values must be positive.

Examples:

```
# Map linear slider (0-1) to frequency range (20Hz-20kHz)
slider_pos = SIMD[DType.float64, 1](0.5)
```

```
frequency = lindex(slider_pos, 0.0, 1.0, 20.0, 20000.0) # ≈ 632 Hz

# Map MIDI controller to filter cutoff frequencies
cc_values = SIMD[DType.float64, 4](0.0, 0.33, 0.66, 1.0)
cutoffs = lindex[4](cc_values, 0.0, 1.0, 100.0, 10000.0)

# Create exponential envelope shape
linear_time = SIMD[DType.float64, 1](0.8)
exp_amplitude = lindex(linear_time, 0.0, 1.0, 0.001, 1.0)
```

**Signature**

```
lindex[N: Int = 1](value: SIMD[DType.float64, N], in_min: SIMD[DType.float64, N], in_max: SIMD[DType.float64, N], out_min: SIMD[DType.float64, N], out_max: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

**Parameters**

- **N:** `Int` - Size of the SIMD vector (defaults to 1).

**Arguments**

- **value:** `SIMD` - The values to map.- **in\_min:** `SIMD` - The minimum of the input range.- **in\_max:** `SIMD` - The maximum of the input range.- **out\_min:** `SIMD` - The minimum of the output range (must be > 0).- **out\_max:** `SIMD` - The maximum of the output range (must be > 0).

**Returns**

**Type:** `SIMD` The exponentially mapped values in the output range.

**CLIP**

Clips each element in the SIMD vector to the specified range. Parameters: N: size of the SIMD vector - defaults to 1 Args: val: The SIMD vector to clip. Each element will be clipped individually. lo: The minimum value. hi: The maximum value. Returns: The clipped SIMD vector.

**Signature**

```
clip[N: Int = 1](val: SIMD[DType.float64, N], lo: SIMD[DType.float64, N], hi: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

**Parameters**

- **N:** `Int`

**Arguments**

- **val:** `SIMD` - **lo:** `SIMD` - **hi:** `SIMD`

**Returns**

**Type:** `SIMD`

**WRAP**

Wraps a value around a specified range. Parameters: N: size of the SIMD vector - defaults to 1.

**Signature**

```
wrap[N: Int = 1](value: SIMD[DType.float64, N], min_val: SIMD[DType.float64, N], max_val: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

**Parameters**

- **N:** `Int`

**Arguments**

- **value:** `SIMD` - The value to wrap.- **min\_val:** `SIMD` - The minimum of the range.- **max\_val:** `SIMD` - The maximum of the range.

**Returns**

**Type:** `SIMD`

Wraps a value around a specified range.

#### Signature

```
wrap[N: Int = 1](value: SIMD[DType.int64, N], min_val: SIMD[DType.int64, N], max_val: SIMD[DType.int64, N]) -> SIMD[DType.int64, N]
```

#### Parameters

- **N:** `Int` - Size of the SIMD vector - defaults to 1.

#### Arguments

- **value:** `SIMD` - The value to wrap.- **min\_val:** `SIMD` - The minimum of the range.- **max\_val:** `SIMD` - The maximum of the range.

#### Returns

**Type:** `SIMD`

### QUADRATIC\_INTERP

Performs quadratic interpolation between three points.

#### Signature

```
quadratic_interp(y0: Float64, y1: Float64, y2: Float64, x: Float64) -> Float64
```

#### Arguments

- **y0:** `Float64` - The value at position 0.- **y1:** `Float64` - The value at position 1.- **y2:** `Float64` - The value at position 2.- **x:** `Float64` - The interpolation position (typically between 0 and 2).

#### Returns

**Type:** `Float64` The interpolated value at position x.

### CUBIC\_INTERP

Performs cubic interpolation between.

Cubic Intepolation equation from *The Audio Programming Book* by Richard Boulanger and Victor Lazzarini. pg. 400

#### Signature

```
cubic_interp(p0: Float64, p1: Float64, p2: Float64, p3: Float64, t: Float64) -> Float64
```

#### Arguments

- **p0:** `Float64` - Point to the left of p1.- **p1:** `Float64` - Point to the left of the float t.- **p2:** `Float64` - Point to the right of the float t.- **p3:** `Float64` - Point to the right of p2.- **t:** `Float64` - Interpolation parameter (0.0 to 1.0).

#### Returns

**Type:** `Float64` Interpolated value.

### LAGRANGE4

Perform Lagrange interpolation for 4th order case (from JOS Faust Model). This is extrapolated from the JOS Faust filter model.

lagrange4N -> SIMD[Float64, N]

#### Signature

```
lagrange4[N: Int = 1](sample0: SIMD[DType.float64, N], sample1: SIMD[DType.float64, N], sample2: SIMD[DType.float64, N], sample3: SIMD[DType.float64, N], sample4: SIMD[DType.float64, N], frac: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

## Parameters

- **N:** `Int` - Size of the SIMD vector - defaults to 1.

## Arguments

- **sample0:** `SIMD` - The first sample.- **sample1:** `SIMD` - The second sample.- **sample2:** `SIMD` - The third sample.- **sample3:** `SIMD` - The fourth sample.- **sample4:** `SIMD` - The fifth sample.- **frac:** `SIMD` - The fractional delay (0.0 to 1.0) which is the location between sample0 and sample1.

## Returns

**Type:** `SIMD` The interpolated value.

## LERP

Performs linear interpolation between two points.

`lerpN` -> `Float64` or `SIMD[Float64, N]`

## Signature

```
lerp(N: Int = 1)(p0: SIMD[DType.float64, N], p1: SIMD[DType.float64, N], t: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

## Parameters

- **N:** `Int` - Size of the SIMD vector - defaults to 1.

## Arguments

- **p0:** `SIMD` - The starting point.- **p1:** `SIMD` - The ending point.- **t:** `SIMD` - The interpolation parameter (0.0 to 1.0).

## Returns

**Type:** `SIMD` The interpolated value.

## MIDICPS

## Signature

```
midicps(midi_note_number: Int64, reference_midi_note: Int64 = 69, reference_frequency: Float64 = 440) -> Float64
```

## Arguments

- **midi\_note\_number:** `Int64` - **reference\_midi\_note:** `Int64` = 69 - **reference\_frequency:** `Float64` = 440

## Returns

**Type:** `Float64`

## Signature

```
midicps(midi_note_number: Float64, reference_midi_note: Float64 = 69, reference_frequency: Float64 = 440) -> Float64
```

## Arguments

- **midi\_note\_number:** `Float64` - **reference\_midi\_note:** `Float64` = 69 - **reference\_frequency:** `Float64` = 440

## Returns

**Type:** `Float64`

## CPSMIDI

## Signature

```
cpsmidi(freq: Float64, reference_midi_note: Float64 = 69, reference_frequency: Float64 = 440) -> Float64
```

#### Arguments

- **freq:** `Float64` - **reference\_midi\_note:** `Float64` = 69 - **reference\_frequency:** `Float64` = 440

#### Returns

**Type:** `Float64`

---

#### MIX

##### Signature

```
mix(mut output: List[Float64], *lists: List[Float64])
```

#### Arguments

- **output:** `List` - **\*lists:** `List`
- 

#### SANITIZE

##### Signature

```
sanitize[N: Int = 1](x: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

#### Parameters

- **N:** `Int`

#### Arguments

- **x:** `SIMD`

#### Returns

**Type:** `SIMD`

---

#### RANDOM\_EXP\_FLOAT64

Generates a random float64 value from an exponential distribution.

##### Signature

```
random_exp_float64[N: Int = 1](min: SIMD[DType.float64, N], max: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

#### Parameters

- **N:** `Int` - Size of the SIMD vector - defaults to 1.

#### Arguments

- **min:** `SIMD` - The minimum value (inclusive).- **max:** `SIMD` - The maximum value (inclusive).

#### Returns

**Type:** `SIMD`

---

Documentation generated with `mojo doc` from Mojo version 0.25.6.0.dev2025090605

## 3.2.2 Buffer

### BUFFER

Has 2 possible constructors:

1) Buffer(lists: List[List[Float64]], buf\_sample\_rate: Float64 = 48000.0). - lists: List of channels, each channel is a List of Float64 samples. - buf\_sample\_rate: Sample rate of the buffer (default is 48000.0).

2) Buffer(num\_chans: Int64 = 2, samples: Int64 = 48000, buf\_sample\_rate: Float64 = 48000.0). - num\_chans: Number of channels (default is 2 for stereo). - samples: Number of samples per channel (default is 48000 for 1 second at 48kHz). - buf\_sample\_rate: Sample rate of the buffer (default is 48000.0).

**Parent Traits:** AnyType, Buffable, Copyable, Movable, Representable, UnknownDestructibility

## 3.2.3 Functions

### fn get\_num\_frames

Return the number of frames in the buffer.

#### SIGNATURE

```
get_num_frames(self) -> Float64
```

#### RETURNS

**Type:** Float64

### fn get\_duration

Return the duration of the buffer in seconds.

#### SIGNATURE

```
get_duration(self) -> Float64
```

#### RETURNS

**Type:** Float64

### fn get\_buf\_sample\_rate

Return the sample rate of the buffer.

#### SIGNATURE

```
get_buf_sample_rate(self) -> Float64
```

#### RETURNS

**Type:** Float64

### fn quadratic\_interp\_loc

#### SIGNATURE

```
quadratic_interp_loc(self, idx: Int64, idx1: Int64, idx2: Int64, frac: Float64, chan: Int64) -> Float64
```

#### ARGUMENTS

- **idx:** `Int64`
- **idx1:** `Int64`
- **idx2:** `Int64`
- **frac:** `Float64`
- **chan:** `Int64`

#### RETURNS

**Type:** `Float64`

---

#### `fn linear_interp_loc`

#### SIGNATURE

```
linear_interp_loc(self, idx: Int64, idx1: Int64, frac: Float64, chan: Int64) -> Float64
```

#### ARGUMENTS

- **idx:** `Int64`
- **idx1:** `Int64`
- **frac:** `Float64`
- **chan:** `Int64`

#### RETURNS

**Type:** `Float64`

---

#### `fn read_sinc`

#### SIGNATURE

```
read_sinc(mut self, chan: Int64, phase: Float64, last_phase: Float64) -> Float64
```

#### ARGUMENTS

- **chan:** `Int64`
- **phase:** `Float64`
- **last\_phase:** `Float64`

#### RETURNS

**Type:** `Float64`

---

#### `fn read`

A read operation on the buffer that reads a multichannel buffer and returns a SIMD vector of size N. It will start reading from the channel specified by `start_chan` and read N channels from there. `read(start_chan, phase, interp=0)`

#### SIGNATURE

```
read[N: Int = 1](mut self, start_chan: Int64, phase: Float64, interp: Int64 = 0) -> SIMD[DType.float64, N]
```

#### PARAMETERS

- **N:** `Int` - The number of channels to read (default is 1). The SIMD vector returned will have this size as well.



ARGUMENTS

- **start\_chan:** `Int64` - The starting channel index to read from (0-based).
- **phase:** `Float64` - The phase position to read from, where 0.0 is the start of the buffer and 1.0 is the end.
- **interp:** `Int64` = 0 - The interpolation method to use (0 = linear, 1 = quadratic).

RETURNS

**Type:** `SIMD`

`fn write`

SIGNATURE

```
write[N: Int = 1](mut self, value: SIMD[DType.float64, N], index: Int64, start_channel: Int64 = 0)
```

PARAMETERS

- **N:** `Int`

ARGUMENTS

- **value:** `SIMD`
- **index:** `Int64`
- **start\_channel:** `Int64` = 0

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### 3.2.4 Osc

PHASOR

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.5 Functions

**fn increment\_phase**

SIGNATURE

```
increment_phase(mut self: Phasor[N], freq: SIMD[DType.float64, N], os_index: Int = 0)
```

ARGUMENTS

- freq: SIMD
- os\_index: Int = 0

**fn next**

SIGNATURE

```
next(mut self: Phasor[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- freq: SIMD = 100
- phase\_offset: SIMD = 0
- trig: Float64 = 0
- os\_index: Int = 0

RETURNS

Type: SIMD

---### Osc

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.6 Functions

**fn next**

SIGNATURE

```
next(mut self: Osc[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, osc_type: SIMD[DType.int64, N] = 0, in terp: SIMD[DType.int64, N] = 1, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- freq: SIMD = 100

- **phase\_offset:** SIMD = 0
- **trig:** Float64 = 0
- **osc\_type:** SIMD = 0
- **interp:** SIMD = 1
- **os\_index:** Int = 0

RETURNS

Type: SIMD

fn next\_interp

SIGNATURE

```
next_interp(mut self: Osc[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, osc_types: List[Int64] = List[Int64, False](0, 4, 5, 6, Tuple[]()), osc_frac: SIMD[DType.float64, N] = 0, interp: Int64 = 1, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **freq:** SIMD = 100
- **phase\_offset:** SIMD = 0
- **trig:** Float64 = 0
- **osc\_types:** List = List[Int64, False](0, 4, 5, 6, Tuple[]())
- **osc\_frac:** SIMD = 0
- **interp:** Int64 = 1
- **os\_index:** Int = 0

RETURNS

Type: SIMD

---### SinOsc

A sine wave oscillator.

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.7 Functions

fn next

SIGNATURE

```
next(mut self: SinOsc[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, interp: Int64 = 0, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **freq:** SIMD = 100
- **phase\_offset:** SIMD = 0
- **trig:** Float64 = 0
- **interp:** Int64 = 0
- **os\_index:** Int = 0

RETURNS

Type: SIMD

---### LFSaw

A low-frequency sawtooth oscillator.

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.8 Functions

fn next

SIGNATURE

```
next(mut self: LFSaw[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, interp: Int64 = 0, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- freq: SIMD = 100
- phase\_offset: SIMD = 0
- trig: Float64 = 0
- interp: Int64 = 0
- os\_index: Int = 0

RETURNS

Type: SIMD

---### LFSquare

A low-frequency square wave oscillator.

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.9 Functions

fn next

SIGNATURE

```
next(mut self: LFSquare[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, interp: Int64 = 0, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- freq: SIMD = 100
- phase\_offset: SIMD = 0
- trig: Float64 = 0
- interp: Int64 = 0

• **os\_index:** `Int` = 0

RETURNS

**Type:** `SIMD`

---### LFTri

A low-frequency triangle wave oscillator.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. **N:** `Int`

### 3.2.10 Functions

**fn next**

SIGNATURE

```
next(mut self: LFTri[N], freq: SIMD[DType.float64, N] = 100, phase_offset: SIMD[DType.float64, N] = 0, trig: Float64 = 0, interp: Int64 = 0, os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **freq:** `SIMD` = 100
- **phase\_offset:** `SIMD` = 0
- **trig:** `Float64` = 0
- **interp:** `Int64` = 0
- **os\_index:** `Int` = 0

RETURNS

**Type:** `SIMD`

---### Impulse

An oscillator that generates an impulse signal. Arguments: world\_ptr: Pointer to the MMMWorld instance.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. **N:** `Int`

### 3.2.11 Functions

**fn next**

Generate the next impulse sample.

SIGNATURE

```
next(mut self: Impulse[N], freq: SIMD[DType.float64, N] = 100, trig: Float64 = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **freq:** `SIMD` = 100
- **trig:** `Float64` = 0

## RETURNS

**Type:** SIMD**fn get\_phase**

## SIGNATURE

```
get_phase(mut self: Impulse[N]) -> SIMD[DType.float64, N]
```

## RETURNS

**Type:** SIMD**---### Dust**

A low-frequency dust noise oscillator.

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

## Parameters

1. N: Int

## 3.2.12 Functions

**fn next**

Generate the next dust noise sample.

## SIGNATURE

```
next(mut self: Dust[N], freq: SIMD[DType.float64, N] = 100, trig: Float64 = 1) -> SIMD[DType.float64, N]
```

## ARGUMENTS

- **freq:** SIMD = 100
- **trig:** Float64 = 1

## RETURNS

**Type:** SIMD**fn next\_range**

Generate the next dust noise sample.

## SIGNATURE

```
next_range(mut self: Dust[N], low: SIMD[DType.float64, N] = 100, high: SIMD[DType.float64, N] = 2000, trig: Float64 = 1) -> SIMD[DType.float64, N]
```

## ARGUMENTS

- **low:** SIMD = 100
- **high:** SIMD = 2000
- **trig:** Float64 = 1

## RETURNS

**Type:** SIMD

**fn get\_phase**

SIGNATURE

```
get_phase(mut self: Dust[N]) -> SIMD[DType.float64, N]
```

RETURNS

**Type:** SIMD

---### LFNoise

Low-frequency noise oscillator.

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

3.2.13 Functions

**fn next**

Generate the next low-frequency noise sample.

SIGNATURE

```
next(mut self, freq: Float64 = 100, interp: Int64 = 0) -> Float64
```

ARGUMENTS

- **freq:** Float64 = 100
- **interp:** Int64 = 0

RETURNS

**Type:** Float64

---### Sweep

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

3.2.14 Functions

**fn increment\_phase**

SIGNATURE

```
increment_phase(mut self, freq: Float64)
```

ARGUMENTS

- **freq:** Float64

**fn next**

SIGNATURE

```
next(mut self, freq: Float64 = 100, trig: Float64 = 0) -> Float64
```

ARGUMENTS

- **freq:** Float64 = 100
- **trig:** Float64 = 0

RETURNS

Type: `Float64`

.....

.....

.....

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## 3.2.15 Filters

### Functions

#### TF2S

##### Signature

```
tf2s(N: Int = 1)(coeffs: List[SIMD[DType.float64, N]], mut coeffs_out: List[SIMD[DType.float64, N]], sample_rate: Float64)
```

##### Parameters

- **N:** Int

##### Arguments

- **coeffs:** List - **coeffs\_out:** List - **sample\_rate:** Float64

### LAG

A lag processor that smooths input values over time based on a specified lag time in seconds.

#### Arguments:

```
**N:** Number of channels Lag will process. (This creates SIMD parallel processing.)
```

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

##### Parameters

1. **N:** Int

## 3.2.16 Functions

### fn get\_small\_simd

#### SIGNATURE

```
get_small_simd(mut self, in_samp: SIMD[DType.float64, N], j: Int)
```

#### ARGUMENTS

- **in\_samp:** SIMD
- **j:** Int

### fn put\_small\_simd

#### SIGNATURE

```
put_small_simd(mut self, j: Int)
```

#### ARGUMENTS

- **j:** Int

### fn next

#### SIGNATURE

```
next(mut self: Lag[N], var in_samp: SIMD[DType.float64, N], lag: SIMD[DType.float64, N] = 0.050000000000000003, num_lags: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **in\_samp:** SIMD
- **lag:** SIMD = 0.050000000000000003
- **num\_lags:** Int = \$0

RETURNS

**Type:** SIMD

---### SVF

State Variable Filter implementation translated from Oleg Nesterov's Faust implementation

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. **N:** Int

### 3.2.17 Functions

**fn reset**

Reset internal state

SIGNATURE

```
reset(mut self)
```

**fn next**

next a single sample through the SVF

SIGNATURE

```
next(mut self, input: SIMD[DType.float64, N], filter_type: SIMD[DType.int32, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain_db: SIMD[DType.float64, N] = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **filter\_type:** SIMD
- **frequency:** SIMD
- **q:** SIMD
- **gain\_db:** SIMD = 0

RETURNS

**Type:** SIMD

**fn lpf**

Lowpass filter

SIGNATURE

```
lpf(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD

RETURNS

**Type:** SIMD

---

**fn bpf**

Bandpass filter

SIGNATURE

```
bpf(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD

RETURNS

**Type:** SIMD

---

**fn hpf**

Highpass filter

SIGNATURE

```
hpf(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD

RETURNS

**Type:** SIMD

---

**fn notch**

Notch filter

SIGNATURE

```
notch(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD

## RETURNS

Type: `SIMD`

---

**`fn peak`**

Peak filter

## SIGNATURE

```
peak(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

## ARGUMENTS

- **input:** `SIMD`
- **frequency:** `SIMD`
- **q:** `SIMD`

## RETURNS

Type: `SIMD`

---

**`fn allpass`**

Allpass filter

## SIGNATURE

```
allpass(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

## ARGUMENTS

- **input:** `SIMD`
- **frequency:** `SIMD`
- **q:** `SIMD`

## RETURNS

Type: `SIMD`

---

**`fn bell`**

Bell filter (parametric EQ)

## SIGNATURE

```
bell(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain_db: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

## ARGUMENTS

- **input:** `SIMD`
- **frequency:** `SIMD`
- **q:** `SIMD`
- **gain\_db:** `SIMD`

## RETURNS

Type: `SIMD`

**fn lowshelf**

Low shelf filter

SIGNATURE

```
lowshelf(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain_db: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD
- **gain\_db:** SIMD

RETURNS

Type: SIMD

**fn highshelf**

High shelf filter

SIGNATURE

```
highshelf(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain_db: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD
- **q:** SIMD
- **gain\_db:** SIMD

RETURNS

Type: SIMD

---### lpf\_LR4

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

3.2.18 Functions

**fn set\_sample\_rate**

SIGNATURE

```
set_sample_rate(mut self, sample_rate: Float64)
```

ARGUMENTS

- **sample\_rate:** Float64

**fn next**

a single sample through the 4th order lowpass filter.

SIGNATURE

```
next(mut self, input: SIMD[DType.float64, N], frequency: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD
- **frequency:** SIMD

RETURNS

**Type:** SIMD

---### OnePole

Simple one-pole IIR filter that can be configured as lowpass or highpass

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

3.2.19 Functions

**fn next**

Process one sample through the filter

SIGNATURE

```
next(mut self, input: Float64, coef: Float64) -> Float64
```

ARGUMENTS

- **input:** Float64
- **coef:** Float64

RETURNS

**Type:** Float64

---### Integrator

Simple one-pole IIR filter that can be configured as lowpass or highpass

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

3.2.20 Functions

**fn next**

Process one sample through the filter

SIGNATURE

```
next(mut self, input: Float64, coef: Float64) -> Float64
```

ARGUMENTS

- **input:** Float64
- **coef:** Float64

RETURNS

**Type:** `Float64`

---### DCTrap

DC Trap from Digital Sound Generation by Beat Frei.

Arguments: input: The input signal to process.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. **N:** `Int`

### 3.2.21 Functions

**fn next**

Process one sample through the DC blocker filter

SIGNATURE

```
next(mut self, in_: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

• **in\_:** `SIMD`

RETURNS

**Type:** `SIMD`

---### VAOnePole

Simple one-pole IIR filter that can be configured as lowpass or highpass}

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. **N:** `Int`

### 3.2.22 Functions

**fn lpf**

Process one sample through the filter

SIGNATURE

```
lpf(mut self, input: SIMD[DType.float64, N], freq: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

• **input:** `SIMD`

• **freq:** `SIMD`

RETURNS

**Type:** `SIMD`

`fn hpf`

SIGNATURE

```
hpf(mut self, input: SIMD[DType.float64, N], freq: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- `input`: SIMD
- `freq`: SIMD

RETURNS

Type: SIMD

---### VAMoogLadder

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

3.2.23 Functions

`fn next`

SIGNATURE

```
next(mut self, sig: SIMD[DType.float64, N], freq: SIMD[DType.float64, N], q_val: SIMD[DType.float64, N], os_index: Int = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- `sig`: SIMD
- `freq`: SIMD
- `q_val`: SIMD
- `os_index`: Int = 0

RETURNS

Type: SIMD

---### FIR

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

3.2.24 Functions

`fn next`

SIGNATURE

```
next(mut self: FIR[N], input: SIMD[DType.float64, N], coeffs: List[SIMD[DType.float64, N]]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- `input`: SIMD
- `coeffs`: List



RETURNS

Type: SIMD

---### IIR

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

3.2.25 Functions

fn next

SIGNATURE

```
next(mut self: IIR[N], input: SIMD[DType.float64, N], coeffsbv: List[SIMD[DType.float64, N]], coeffsav: List[SIMD[DType.float64, N]]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- input: SIMD
- coeffsbv: List
- coeffsav: List

RETURNS

Type: SIMD

---### tf2

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

3.2.26 Functions

fn next

SIGNATURE

```
next(mut self: tf2[N], input: SIMD[DType.float64, N], coeffs: List[SIMD[DType.float64, N]]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- input: SIMD
- coeffs: List

RETURNS

Type: SIMD

---### Reson

Parent Traits: AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: `Int`

3.2.27 Functions

`fn` **lpf**

SIGNATURE

```
lpf(mut self: Reson[N], input: SIMD[DType.float64, N], freq: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** `SIMD`
- **freq:** `SIMD`
- **q:** `SIMD`
- **gain:** `SIMD`

RETURNS

**Type:** `SIMD`

`fn` **hpf**

SIGNATURE

```
hpf(mut self: Reson[N], input: SIMD[DType.float64, N], freq: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** `SIMD`
- **freq:** `SIMD`
- **q:** `SIMD`
- **gain:** `SIMD`

RETURNS

**Type:** `SIMD`

`fn` **bpf**

SIGNATURE

```
bpf(mut self: Reson[N], input: SIMD[DType.float64, N], freq: SIMD[DType.float64, N], q: SIMD[DType.float64, N], gain: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** `SIMD`
- **freq:** `SIMD`
- **q:** `SIMD`
- **gain:** `SIMD`

RETURNS

**Type:** `SIMD`

.....

.....

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## 3.2.28 Env

Envelope generator module.

This module provides an envelope generator class that can create complex envelopes with multiple segments, curves, and looping capabilities.

### Functions

#### MIN\_ENV

Create a minimum envelope with specified ramp and duration.

##### Signature

```
min_env(N: Int = 1)(ramp: SIMD[DType.float64, N] = 0.01, dur: SIMD[DType.float64, N] = 0.10000000000000001, rise: SIMD[DType.float64, N] = 0.001) -> SIMD[DType.float64, N]
```

##### Parameters

- **N:** Int

##### Arguments

- **ramp:** SIMD = 0.01 - **dur:** SIMD = 0.10000000000000001 - **rise:** SIMD = 0.001

##### Returns

**Type:** SIMD

#### ENV

Envelope generator.

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

## 3.2.29 Functions

#### fn reset\_vals

Reset internal values.

##### SIGNATURE

```
reset_vals(mut self, times: List[Float64])
```

##### ARGUMENTS

- **times:** List

#### fn next

Generate the next envelope sample.

##### SIGNATURE

```
next(mut self, values: List[Float64] = List[Float64, False](0, 1, 0, Tuple[]()), times: List[Float64] = List[Float64, False](1, 1, Tuple[]()), curves: List[Float64] = List[Float64, False](1, Tuple[]()), loop: Int64 = 0, trig: Float64 = 1, time_warp: Float64 = 1) -> Float64
```

##### ARGUMENTS

- **values:** List = List[Float64, False](0, 1, 0, Tuple[]())
- **times:** List = List[Float64, False](1, 1, Tuple[]())

- **curves:** `List = List[Float64, False](1, Tuple[]())`
- **loop:** `Int64 = 0`
- **trig:** `Float64 = 1`
- **time\_warp:** `Float64 = 1`

RETURNS

**Type:** `Float64`

.....

.....

.....

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### 3.2.30 Delays

DELAY

A delay line with Lagrange interpolation.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. `N`: `Int` - size of the SIMD vector - defaults to 1

### 3.2.31 Functions

**fn next**

Process one sample through the delay line. This function computes the average of two values.`next(input, delay_time)`

SIGNATURE

```
next(mut self, input: SIMD[DType.float64, N], delay_time: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- `input`: `SIMD` - The input sample to process.
- `delay_time`: `SIMD` - The amount of delay to apply (in seconds).

RETURNS

**Type:** `SIMD` The processed output sample.

**fn lagrange4**

Perform Lagrange interpolation for 4th order case (from JOS Faust Model).

SIGNATURE

```
lagrange4(mut self, input: SIMD[DType.float64, N], delay_time: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- `input`: `SIMD`
- `delay_time`: `SIMD`

RETURNS

**Type:** `SIMD`

---### Comb

A simple comb filter using a delay line with feedback.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. `N`: `Int` - size of the SIMD vector - defaults to 1

### 3.2.32 Functions

**fn next**

Process one sample through the comb filter.`next(input, delay_time=0.0, feedback=0.0, interp=0)`

SIGNATURE

```
next(mut self, input: SIMD[DType.float64, N], delay_time: SIMD[DType.float64, N] = 0, feedback: SIMD[DType.float64, N] = 0, interp: Int64 = 0) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD - The input sample to process.
- **delay\_time:** SIMD = 0 - The amount of delay to apply (in seconds).
- **feedback:** SIMD = 0 - The amount of feedback to apply (0.0 to 1.0).
- **interp:** Int64 = 0 - The interpolation method to use (0 = linear, 1 = cubic, 2 = Lagrange).

RETURNS

**Type:** SIMD The processed output sample.

---### FBDelay

Like a Comb filter but with any amount of feedback and a tanh function.

```
Parameters:  
N: size of the SIMD vector - defaults to 1
```

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

Parameters

1. N: Int

### 3.2.33 Functions

**fn next**

Process one sample or SIMD vector through the feedback delay.`next(input, delay_time=0.0, feedback=0.0, interp=0)`

SIGNATURE

```
next(mut self, input: SIMD[DType.float64, N], delay_time: SIMD[DType.float64, N], feedback: SIMD[DType.float64, N]) -> SIMD[DType.float64, N]
```

ARGUMENTS

- **input:** SIMD - The input sample to process.
- **delay\_time:** SIMD - The amount of delay to apply (in seconds).
- **feedback:** SIMD - The amount of feedback to apply (0.0 to 1.0).

RETURNS

**Type:** SIMD The processed output sample or SIMD vector.

### 3.2.34 Distortion

Functions

VTANH

Signature

```
vtanh(in_samp: Float64, gain: Float64, offset: Float64) -> Float64
```

Arguments

- **in\_samp:** Float64 - **gain:** Float64 - **offset:** Float64

Returns

**Type:** Float64

BITCRUSHER

Signature

```
bitcrusher(in_samp: Float64, bits: Int64) -> Float64
```

Arguments

- **in\_samp:** Float64 - **bits:** Int64

Returns

**Type:** Float64

LATCH

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.35 Functions

fn next

SIGNATURE

```
next(mut self, in_samp: Float64, trig: Float64) -> Float64
```

ARGUMENTS

- **in\_samp:** Float64
- **trig:** Float64

RETURNS

**Type:** Float64



### 3.2.36 Noise

#### WHITENOISE

Generate white noise samples.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `UnknownDestructibility`

### 3.2.37 Functions

#### `fn next`

Generate the next white noise sample.

#### SIGNATURE

```
next(self, gain: Float64 = 1) -> Float64
```

#### ARGUMENTS

- **gain:** `Float64` = 1

#### RETURNS

**Type:** `Float64` A random value between -gain and gain.

#### ---### PinkNoise

Generate pink noise samples.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `UnknownDestructibility`

### 3.2.38 Functions

#### `fn next`

Generate the next pink noise sample.

#### SIGNATURE

```
next(mut self, gain: Float64 = 1) -> Float64
```

#### ARGUMENTS

- **gain:** `Float64` = 1 - Amplitude scaling factor.

#### RETURNS

**Type:** `Float64`

#### ---### BrownNoise

Generate brown noise samples.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `UnknownDestructibility`

### 3.2.39 Functions

#### `fn next`

Generate the next brown noise sample.

SIGNATURE

```
next(mut self, gain: Float64 = 1) -> Float64
```

ARGUMENTS

- **gain:** `Float64 = 1` - Amplitude scaling factor.

RETURNS

**Type:** `Float64` A brown noise sample scaled by gain.

.....

.....

.....

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### 3.2.40 Pan

PAN2

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.41 Functions

**fn next**

SIGNATURE

```
next(mut self, sample: Float64, mut pan: Float64) -> SIMD[DType.float64, 2]
```

ARGUMENTS

- **sample:** Float64
- **pan:** Float64

RETURNS

**Type:** SIMD

---### PanAz

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.42 Functions

**fn next**

SIGNATURE

```
next[N: Int](mut self, sample: Float64, pan: Float64, num_speakers: Int64, width: Float64 = 2, orientation: Float64 = 0.5) -> SIMD[DType.float64, N]
```

PARAMETERS

- **N:** Int

ARGUMENTS

- **sample:** Float64
- **pan:** Float64
- **num\_speakers:** Int64
- **width:** Float64 = 2
- **orientation:** Float64 = 0.5

RETURNS

**Type:** SIMD

### 3.2.43 PlayBuf

PLAYBUF

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.44 Functions

**fn next**

get the next sample from an audio buffer - can take both Buffer or InterleavedBuffer. Arguments: buffer: The audio buffer to read from (can be Buffer or InterleavedBuffer). rate: The playback rate. 1 is the normal speed of the buffer. loop: Whether to loop the buffer (default: True). trig: Trigger starts the synth at start\_frame (default: 1.0). start\_frame: The start frame for playback (default: 0) upon receiving a trigger. end\_frame: The end frame for playback (default: -1).

SIGNATURE

```
next[T: Buffable, N: Int = 1](mut self, mut buffer: T, start_chan: Int, rate: Float64, loop: Bool = True, trig: Float64 = 1, start_frame: Float64 = 0, end_frame: Float64 = -1) -> SIMD[DType.float64, N]
```

PARAMETERS

- **T:** Buffable
- **N:** Int

ARGUMENTS

- **buffer:** T
- **start\_chan:** Int
- **rate:** Float64
- **loop:** Bool = True
- **trig:** Float64 = 1
- **start\_frame:** Float64 = 0
- **end\_frame:** Float64 = -1

RETURNS

**Type:** SIMD

**fn get\_phase**

SIGNATURE

```
get_phase(mut self) -> Float64
```

RETURNS

**Type:** Float64

**fn get\_win\_phase**

SIGNATURE

```
get_win_phase(mut self) -> Float64
```

RETURNS

**Type:** Float64

---### Grain

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.45 Functions

**fn next**

SIGNATURE

```
next[T: Buffable, N: Int = 1](mut self, mut buffer: T, start_chan: Int, trig: Float64 = 0, rate: Float64 = 1, start_frame: Float64 = 0, duration: Float64 = 0, pan: Float64 = 0, gain: Float64 = 1) -> SIMD[DType.float64, 2]
```

PARAMETERS

- **T:** Buffable
- **N:** Int

ARGUMENTS

- **buffer:** T
- **start\_chan:** Int
- **trig:** Float64 = 0
- **rate:** Float64 = 1
- **start\_frame:** Float64 = 0
- **duration:** Float64 = 0
- **pan:** Float64 = 0
- **gain:** Float64 = 1

RETURNS

**Type:** SIMD

---### TGrains

Triggered granular synthesis. Each trigger starts a new grain.

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.46 Functions

**fn next**

Generate the next set of grains. Arguments: buffer: Audio buffer containing the source sound. trig: Trigger signal (>0 to start a new grain). rate: Playback rate of the grains (1.0 = normal speed). start\_frame: Starting frame position in the buffer. duration: Duration of each grain in seconds. pan: Panning position from -1.0 (left) to 1.0 (right). gain: Amplitude scaling factor for the grains.

SIGNATURE

```
next[T: Buffable, N: Int = 1](mut self, mut buffer: T, buf_chan: Int, trig: Float64 = 0, rate: Float64 = 1, start_frame: Float64 = 0, duration: Float64 = 0.10 000000000000001, pan: Float64 = 0, gain: Float64 = 1) -> SIMD[DType.float64, 2]
```

PARAMETERS

- **T:** Buffable
- **N:** Int

ARGUMENTS

- **buffer:** T

- **buf\_chan:** `Int`
- **trig:** `Float64 = 0`
- **rate:** `Float64 = 1`
- **start\_frame:** `Float64 = 0`
- **duration:** `Float64 = 0.100000000000000001`
- **pan:** `Float64 = 0`
- **gain:** `Float64 = 1`

RETURNS

**Type:** `SIMD` List of output samples for all channels.

.....

.....

.....

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### 3.2.47 RecordBuf

RECORDBUF

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.2.48 Functions

**fn write**

SIGNATURE

```
write[N: Int = 1](mut self, value: SIMD[DType.float64, N], mut buffer: Buffer)
```

PARAMETERS

- **N:** Int

ARGUMENTS

- **value:** SIMD
- **buffer:** Buffer

.....  
.....  
.....  
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### 3.2.49 Oversampling

OVERSAMPLING

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

Parameters

1. **N:** `Int`

### 3.2.50 Functions

**fn** `set_os_index`

SIGNATURE

```
set_os_index(mut self, index: Int)
```

ARGUMENTS

- **index:** `Int`

**fn** `add_sample`

Add a sample to the oversampling buffer.

SIGNATURE

```
add_sample(mut self, sample: SIMD[DType.float64, N])
```

ARGUMENTS

- **sample:** `SIMD`

**fn** `get_sample`

get the next sample from a filled oversampling buffer.

SIGNATURE

```
get_sample(mut self) -> SIMD[DType.float64, N]
```

RETURNS

**Type:** `SIMD`



### 3.2.51 MLP

MLP

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

### 3.2.52 Functions

`fn next`

SIGNATURE

```
next[N: Int = 16](mut self, input: List) -> SIMD[DType.float64, N]
```

PARAMETERS

- `N: Int`

ARGUMENTS

- `input: List`

RETURNS

**Type:** `SIMD`

RAISES

.....

.....

.....

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## 3.3 Core Framework

---

### 3.3.1 MMMGraphs

---

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### 3.3.2 MMMTraits

**Traits**

GRAPHABLE

Signature

Required Methods

next Signature

```
next(mut self: _Self) -> List[Float64]
```

Arguments

- self: \_Self

Returns

**Type:** List

BUFFABLE

Signature

Required Methods

read Signature

```
read(N: Int = 1)(mut self: _Self, start_chan: Int64, phase: Float64, interp: Int64 = 0) -> SIMD[DType.float64, N]
```

Parameters

- N: Int

Arguments

- self: \_Self - start\_chan: Int64 - phase: Float64 - interp: Int64 = 0

Returns

**Type:** SIMD

get\_num\_frames Signature

```
get_num_frames(self: _Self) -> Float64
```

Arguments

- self: \_Self

Returns

**Type:** Float64

get\_duration Signature

```
get_duration(self: _Self) -> Float64
```

Arguments

- **self:** `_Self`

Returns

**Type:** `Float64`

get\_buf\_sample\_rate Signature

```
get_buf_sample_rate(self: _Self) -> Float64
```

Arguments

- **self:** `_Self`

Returns

**Type:** `Float64`

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### 3.3.3 MMMWorld

MMMWORLD

**Parent Traits:** AnyType, Copyable, Movable, Representable, UnknownDestructibility

### 3.3.4 Functions

**fn set\_channel\_count**

SIGNATURE

```
set_channel_count(mut self, num_in_chans: Int64, num_out_chans: Int64)
```

ARGUMENTS

- num\_in\_chans: Int64
- num\_out\_chans: Int64

**fn send\_msg**

SIGNATURE

```
send_msg(mut self, key: String, mut list: List[Float64])
```

ARGUMENTS

- key: String
- list: List

**fn get\_msg**

SIGNATURE

```
get_msg(mut self, key: String) -> Optional[List[Float64]]
```

ARGUMENTS

- key: String

RETURNS

**Type:** Optional

**fn send\_text\_msg**

SIGNATURE

```
send_text_msg(mut self, key: String, mut list: List[String])
```

ARGUMENTS

- key: String
- list: List

**fn get\_text\_msg**

## SIGNATURE

```
get_text_msg(mut self, key: String) -> Optional[List[String]]
```

## ARGUMENTS

- **key:** String

## RETURNS

**Type:** Optional

---

**fn get\_midi**

## SIGNATURE

```
get_midi(mut self, key: String, chan: Int64 = -1, param: Int64 = -1) -> Optional[List[List[Int64]]]
```

## ARGUMENTS

- **key:** String
- **chan:** Int64 = -1
- **param:** Int64 = -1

## RETURNS

**Type:** Optional

---

**fn clear\_midi**

## SIGNATURE

```
clear_midi(mut self)
```

---

**fn send\_midi**

## SIGNATURE

```
send_midi(mut self, msg: PyObject)
```

## ARGUMENTS

- **msg:** PyObject

## RAISES

**fn clear\_msgs**

## SIGNATURE

```
clear_msgs(mut self)
```

---

**fn print**

## SIGNATURE

```
print[T: Writable](mut self, value: T, label: String = "", freq: Float64 = 10)
```

PARAMETERS

- **T**: Writable

ARGUMENTS

- **value**: T
- **label**: String = ""
- **freq**: Float64 = 10

.....

.....

.....

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## 3.4 Utilities

---

### 3.4.1 MMM\_FFT

---

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## 3.4.2 Windows

### Functions

#### BESSEL\_I0

Calculate the modified Bessel function of the first kind, order 0 (I<sub>0</sub>). Uses polynomial approximation for accurate results.

##### Signature

```
bessel_i0(x: Float64) -> Float64
```

##### Arguments

- **x:** Float64 - Input value

##### Returns

**Type:** Float64

#### KAISER\_WINDOW

Create a Kaiser window of length n with shape parameter beta.

##### Signature

```
kaiser_window(size: Int64, beta: Float64) -> List[Float64]
```

##### Arguments

- **size:** Int64 - **beta:** Float64 - Shape parameter that controls the trade-off between main lobe width and side lobe level
- beta = 0: rectangular window
- beta = 5: similar to Hamming window
- beta = 6: similar to Hanning window
- beta = 8.6: similar to Blackman window

##### Returns

**Type:** List DynamicVector[Float64] containing the Kaiser window coefficients

#### BUILD\_SINC\_TABLE

Build a sinc function lookup table.

##### Signature

```
build_sinc_table(size: Int64, ripples: Int64 = 4) -> List[Float64]
```

##### Arguments

- **size:** Int64 - Number of points in the table- **ripples:** Int64 = 4 - Number of ripples/lobes on each side of the main lobe

##### Returns

**Type:** List List containing the sinc function values

#### HANN\_WINDOW

Generate a Hann window of length n.

##### Signature

```
hann_window(n: Int64) -> List[Float64]
```

Arguments

- **n:** Int64 - Length of the window

Returns

**Type:** List

---

HAMMING\_WINDOW

Generate a Hamming window of length n.

Signature

```
hamming_window(n: Int64) -> List[Float64]
```

Arguments

- **n:** Int64

Returns

**Type:** List

---

BLACKMAN\_WINDOW

Generate a Blackman window of length n. Args: n: Length of the window Returns: List containing the Blackman window values

Signature

```
blackman_window(n: Int64) -> List[Float64]
```

Arguments

- **n:** Int64

Returns

**Type:** List

---

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### 3.4.3 Print

PRINT

A struct for printing values in the MMMWorld environment.

**Parent Traits:** `AnyType`, `Copyable`, `Movable`, `Representable`, `UnknownDestructibility`

### 3.4.4 Functions

`fn next`

Print the value at a given frequency. Arguments: value: The value to print. label: An optional label to prepend to the printed value. freq: The frequency (in Hz) at which to print the value.

SIGNATURE

```
next[T: Writable](mut self, value: T, label: String = "", freq: Float64 = 10)
```

PARAMETERS

- **T:** `Writable`

ARGUMENTS

- **value:** `T`
- **label:** `String` = `""`
- **freq:** `Float64` = `10`

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## 4. Examples

---

### 4.1 Examples

---

This section contains practical examples demonstrating how to use MMMAudio for various audio processing tasks.

#### 4.1.1 Basic Examples

---

- **Default Graph**: Basic audio graph setup
- **In2Out**: Simple input to output routing

#### 4.1.2 Synthesis Examples

---

- **Many Oscillators**: Multiple oscillator management
- **Grains**: Granular synthesis techniques

#### 4.1.3 Effects Examples

---

- **Feedback Delays**: Delay-based effects
- **Pan Az**: Spatial audio panning

#### 4.1.4 Advanced Examples

---

- **MIDI Sequencer**: MIDI-controlled sequencing
- **Torch MLP**: Neural network audio processing
- **Record**: Audio recording and playback

#### 4.1.5 Running Examples

---

Most examples can be run directly with Python:

```
python examples/default.py
```

Or with Mojo for the .mojo examples:

```
mojo examples/Default_Graph.mojo
```

## 4.2 Pan\_Az

For more information about the examples, such as how the Python and Mojo files interact with each other, see the [Examples Overview](#)

### 4.2.1 Python Code

```
from mmm_src.MMMAudio import MMMAudio

# instantiate and load the graph

# PanAz is not quite right as of yet
mmm_audio = MMMAudio(128, graph_name="Pan_Az", package_name="examples", num_output_channels=5)
mmm_audio.start_audio()

mmm_audio.stop_audio()

from random import random
mmm_audio.send_msg("osc_freq", random() * 500 + 100 ) # set the frequency to a random value
```

### 4.2.2 Mojo Code

```
from mmm_src.MMMWorld import MMMWorld
from mmm_utils.functions import *
from mmm_src.MMMTraits import *

from mmm_dsp.Osc import Phasor, Osc
from mmm_dsp.Pan import PanAz

struct PanAz_Synth(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var osc: Osc
    var freq: Float64

    var pan_osc: Phasor
    var pan_az: PanAz

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.osc = Osc(self.world_ptr)
        self.freq = 440.0

        self.pan_osc = Phasor(self.world_ptr)
        self.pan_az = PanAz(self.world_ptr)

    fn __repr__(self) -> String:
        return String("Default")

    fn next(mut self) -> SIMD[DType.float64, 8]:

        self.get_msgs()

        # PanAz needs to be given a SIMD size that is a power of 2, in this case [8], but the speaker size can be anything smaller than that
        panned = self.pan_az.next[8](self.osc.next(self.freq, osc_type=2), self.pan_osc.next(0.1), 2, 2) * 0.1

        return panned

    fn get_msgs(mut self: Self):
        # Get messages from the world
        msg = self.world_ptr[0].get_msg("osc_freq")
        if msg:
            self.freq = msg.value()[0]

# there can only be one graph in an MMMAudio instance
# a graph can have as many synths as you want
struct Pan_Az(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var synth: PanAz_Synth

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.synth = PanAz_Synth(self.world_ptr)

    fn __repr__(self) -> String:
        return String("PanAz")

    fn next(mut self) -> SIMD[DType.float64, 8]:

        sample = self.synth.next() # Get the next sample from the synth

        # the output will pan to the number of channels available
```

```
# if there are fewer than 5 channels, only those channels will be output  
return sample # Return the combined output samples
```

## 4.3 ManyOscillators

For more information about the examples, such as how the Python and Mojo files interact with each other, see the [Examples Overview](#)

Example showing how to use ManyOscillators.mojo with MMMAudio.

You can change the number of oscillators dynamically by sending a 'set\_num\_pairs' message.

### 4.3.1 Python Code

```
from mmm_src.MMMAudio import MMMAudio

mmm_audio = MMMAudio(128, graph_name="ManyOscillators", package_name="examples")

mmm_audio.start_audio() # start the audio thread - or restart it where it left off

mmm_audio.send_msg("set_num_pairs", 2) # set to 2 pairs of oscillators

mmm_audio.send_msg("set_num_pairs", 14) # change to 4 pairs of oscillators

mmm_audio.send_msg("set_num_pairs", 50) # change to 4 pairs of oscillators

mmm_audio.stop_audio() # stop/pause the audio thread
```

### 4.3.2 Mojo Code

```
from mmm_src.MMMWorld import MMMWorld
from mmm_utils.functions import *
from mmm_src.MMMTraits import *

from mmm_dsp.Osc import Osc
from random import random_float64
from mmm_dsp.Pan import Pan2
from mmm_dsp.OscBuffers import OscBuffers

# THE SYNTH

struct OscSynth(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var oscs: Osc[2] # An Osc instance with two internal Oscs
    var osc_freqs: SIMD[DType.float64, 2]
    var pan: Pan2
    var pan_osc: Osc
    var pan_freq: Float64
    var vol_osc: Osc
    var vol_osc_freq: Float64
    var temp: Float64

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld], center_freq: Float64):
        self.world_ptr = world_ptr
        self.oscs = Osc[2](world_ptr) # Initialize two Osc instances

        self.pan = Pan2(world_ptr)
        self.pan_osc = Osc(world_ptr)
        self.pan_freq = random_float64(0.03, 0.1)

        self.vol_osc = Osc(world_ptr)
        self.vol_osc_freq = random_float64(0.05, 0.2)
        self.osc_freqs = SIMD[DType.float64, 2](
            center_freq + random_float64(1.0, 5.0),
            center_freq - random_float64(1.0, 5.0)
        )
        self.temp = 0.0

    fn __repr__(self) -> String:
        return String("OscSynth")

    fn next(mut self) -> SIMD[DType.float64, 2]:

        temp = self.oscs.next(self.osc_freqs, interp = 0, os_index = 0)

        temp = temp * (self.vol_osc.next(self.vol_osc_freq) * 0.01 + 0.01)
        temp2 = temp[0] + temp[1]

        self.world_ptr[0].print(self.osc_freqs, "freqs", freq=1.0)

        pan_loc = self.pan_osc.next(self.pan_freq) # Get pan position

        return self.pan.next(temp2, pan_loc) # Pan the temp signal

# THE GRAPH
```

```

struct ManyOscillators(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]

    var osc_synths: List[OscSynth] # Instances of the Oscillator
    var num_pairs: Int

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr

        # initialize the list of oscillator pairs
        self.osc_synths = List[OscSynth]()
        # add 10 pairs to the list
        self.num_pairs = 10
        for _ in range(self.num_pairs):
            self.osc_synths.append(OscSynth(self.world_ptr, random_exp_float64(100.0, 1000.0)))

    fn __repr__(self) -> String:
        return String("ManyOscillators")

    fn next(mut self) -> SIMD[DType.float64, 2]:
        self.get_msgs()

        # sum all the stereo outs from the N oscillator pairs
        sum = SIMD[DType.float64, 2](0.0, 0.0)
        for i in range(self.num_pairs):
            sum += self.osc_synths[i].next()

        return sum

    fn get_msgs(mut self):
        # looking for a message that changes the number of osc pairs

        num = self.world_ptr[0].get_msg("set_num_pairs")
        if num:
            if num.value()[0] != self.num_pairs:
                print("Changing number of osc pairs to:", Int(num.value()[0]))
                # adjust the list of osc synths
                if Int(num.value()[0]) > self.num_pairs:
                    # add more
                    for _ in range(Int(num.value()[0]) - self.num_pairs):
                        self.osc_synths.append(OscSynth(self.world_ptr, random_exp_float64(100.0, 1000.0)))
                else:
                    # remove some
                    for _ in range(self.num_pairs - Int(num.value()[0])):
                        _ = self.osc_synths.pop()
            self.num_pairs = Int(num.value()[0])

```



## 4.4 FeedbackDelays

For more information about the examples, such as how the Python and Mojo files interact with each other, see the [Examples Overview](#)

use the mouse to control an overdriven feedback delay

### 4.4.1 Python Code

```
from mmm_src.MMMAudio import MMMAudio

mmm_audio = MMMAudio(128, graph_name="FeedbackDelays", package_name="examples")

mmm_audio.start_audio() # start the audio thread - or restart it where it left off
mmm_audio.stop_audio() # stop/pause the audio thread
```

### 4.4.2 Mojo Code

```
from mmm_src.MMMWorld import MMMWorld
from mmm_utils.functions import *
from mmm_src.MMMTraits import *

from mmm_dsp.Buffer import *
from mmm_dsp.PlayBuf import *
from mmm_dsp.Delays import *
from mmm_utils.functions import *

struct DelaySynth(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]

    var buffer: InterleavedBuffer # Interleaved buffer for audio samples
    var playBuf: PlayBuf
    var delays: FBDelay[2] # FBDelay for feedback delay effect
    var lag: Lag[2]
    var mouse_x: Float64
    var mouse_y: Float64

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.buffer = InterleavedBuffer(self.world_ptr, "resources/Shiverer.wav")
        self.playBuf = PlayBuf(self.world_ptr)
        # FBDelay is initialized as 2 channel
        self.delays = FBDelay[2](self.world_ptr)

        self.lag = Lag[2](self.world_ptr) # Initialize Lag with a default time constant

        self.mouse_x = 0.0
        self.mouse_y = 0.0

    fn next(mut self) -> SIMD[DType.float64, 2]:
        self.get_msgs() # Get messages from the world
        var sample = self.playBuf.next(N=2)(self.buffer, 0, 1.0, True) # Read samples from the buffer

        # sending one value to the 2 channel lag gives both lags the same parameters
        # var del_time = self.lag.next(linlin(self.mouse_x, 0.0, 1.0, 0.0, self.buffer.get_duration()), 0.5)

        # this is a version with the 2 value SIMD vector as input each delay with have its own del_time
        var del_time = self.lag.next(SIMD[DType.float64, 2](
            linlin(self.mouse_x, 0.0, 1.0, 0.0, self.buffer.get_duration()),
            linlin(self.mouse_x, 0.0, 1.0, 0.0, self.buffer.get_duration()*0.9)
        ), SIMD[DType.float64, 2](0.5, 0.5))

        var feedback = SIMD[DType.float64, 2](self.mouse_y * 2.0, self.mouse_y * 2.1)

        sample = self.delays.next(sample, del_time, feedback)*0.8

        return sample

    fn __repr__(self) -> String:
        return String("DelaySynth")

    fn get_msgs(mut self):
        self.mouse_x = self.world_ptr[0].mouse_x
        self.mouse_y = self.world_ptr[0].mouse_y

struct FeedbackDelays(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var delay_synth: DelaySynth # Instance of the Oscillator

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.delay_synth = DelaySynth(world_ptr) # Initialize the DelaySynth with the world instance
```

```
fn __repr__(self) -> String:  
    return String("FeedbackDelays")  
  
fn next(mut self: FeedbackDelays) -> SIMD[DType.float64, 2]:  
    return self.delay_synth.next() # Return the combined output sample
```

## 4.5 Midi\_Sequencer

For more information about the examples, such as how the Python and Mojo files interact with each other, see the [Examples Overview](#)

### 4.5.1 `trig_synth(wait)` `async`

A counter coroutine

Source code in `examples/Midi_Sequencer.py` 

```

43  async def trig_synth(wait):
44      """A counter coroutine"""
45      count_to = np.random.choice([7, 11, 13, 17]).item()
46      mult_seq = Pseq(list(range(1, count_to + 1)))
47      fund_seq = Pxrnd([36, 37, 43, 42])
48      i = 0
49      fund = librosa.midi_to_hz(fund_seq.next())
50      while True:
51          pitch = mult_seq.next() * fund
52          mmm_audio.send_msg("t_trig", 1.0)
53          mmm_audio.send_msg("trig_seq_freq", pitch)
54          await asyncio.sleep(wait)
55          i = (i + 1) % count_to
56          if i == 0:
57              fund = librosa.midi_to_hz(fund_seq.next())
58              count_to = np.random.choice([7, 11, 13, 17]).item()
59              mult_seq = Pseq(list(range(1, count_to + 1)))

```

### 4.5.2 Python Code

```

from mmm_src.MMMAudio import MMMAudio

# instantiate and load the graph
mmm_audio = MMMAudio(128, graph_name="Midi_Sequencer", package_name="examples")
mmm_audio.start_audio()

# this next chunk of code is all about using a midi keyboard to control the synth-----

# the python host grabs the midi and sends the midi messages to the mojo audio engine

import mido
import time
import threading

# find your midi devices
mido.get_input_names()

# open your midi device - you may need to change the device name
in_port = mido.open_input('Oxygen Pro Mini USB MIDI')

def start_midi():
    while True:
        for msg in in_port.iter_pending():
            # print(msg)
            mmm_audio.send_midi(msg)
            time.sleep(0.01) # Small delay to prevent busy-waiting

midi_thread = threading.Thread(target=start_midi, daemon=True)
# once you start the midi_thread, it should register note_on, note_off, cc, etc from your device and send them to mmm
midi_thread.start()
midi_thread.stop()

# this chunk of code shows how to use the sequencer to trigger notes in the mmm_audio engine

# the scheduler can also sequence notes
from mmm_src.Patterns import * # some sc style patterns
import numpy as np
import asyncio
import librosa

scheduler = mmm_audio.scheduler

async def trig_synth(wait):
    """A counter coroutine"""
    count_to = np.random.choice([7, 11, 13, 17]).item()
    mult_seq = Pseq(list(range(1, count_to + 1)))
    fund_seq = Pxrnd([36, 37, 43, 42])
    i = 0
    fund = librosa.midi_to_hz(fund_seq.next())

```

```

while True:
    pitch = mult_seq.next() * fund
    mmm_audio.send_msg("t_trig", 1.0)
    mmm_audio.send_msg("trig_seq_freq", pitch)
    await asyncio.sleep(wait)
    i = (i + 1) % count_to
    if i == 0:
        fund = librosa.midi_to_hz(fund_seq.next())
        count_to = np.random.choice([7, 11, 13, 17]).item()
        mult_seq = Pseq(list(range(1, count_to + 1)))

scheduler.sched(trig_synth(0.1))

scheduler.stop_routes()

mmm_audio.stop_audio()
mmm_audio.start_audio()

```

### 4.5.3 Mojo Code

```

from mmm_src.MMMWorld import MMMWorld

from mmm_utils.functions import *
from mmm_dsp.Osc import *
from mmm_dsp.Filters import *
from mmm_dsp.Env import Env

from mmm_src.MMMTraits import *

struct TrigSynthVoice(Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld] # Pointer to the MMMWorld instance

    var env: Env

    var mod: Osc
    var car: Osc
    var lag: Lag

    var trig: Float64
    var freq: Float64

    var vol: Float64

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr

        self.mod = Osc(self.world_ptr)
        self.car = Osc(self.world_ptr)
        self.lag = Lag(self.world_ptr)

        self.env = Env(self.world_ptr)

        self.trig = 0.0
        self.freq = 100.0
        self.vol = 1.0

    fn next(mut self) -> Float64:
        if not self.env.is_active and self.trig <= 0.0:
            return 0.0 # Return 0 if the envelope is not active and no trigger
        else:
            var mod_value = self.mod.next(self.freq*1.5) # Get the next value from the modulator
            var env = self.env.next([0.0, 1.0, 0.75, 0.75, 0.0], [0.01, 0.1, 0.2, 0.5], [1.0], 0, self.trig)
            var mod_mult = linlin(env, 0.0, 1.0, 0.0, 0.25) # self.lag.next(linlin(self.mouse_x, 0.0, 1.0, 0.0, 8.0), 0.05)
            var car_value = self.car.next(self.freq, mod_value * mod_mult, osc_type=2, os_index=1) # Get the next value from the carrier
            car_value = car_value * 0.1 * env * self.vol

            return car_value

struct TrigSynth(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld] # Pointer to the MMMWorld instance

    var voices: List[TrigSynthVoice]
    var current_voice: Int64
    var trig: Float64
    var freq: Float64
    var num_voices: Int64
    var note_ons: List[List[Int64]]
    var ccs: List[List[Int64]]

    var svf: SVF
    var filt_lag: Lag
    var filt_freq: Float64

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld], num_voices: Int64 = 8):
        self.world_ptr = world_ptr
        self.trig = 0.0
        self.freq = 100.0
        self.num_voices = num_voices
        self.current_voice = 0

```

```

self.note_ons = List[List[Int64]]()
self.ccs = List[List[Int64]]()

self.voices = List[TrigSynthVoice]()
for _ in range(self.num_voices):
    self.voices.append(TrigSynthVoice(self.world_ptr))

self.svf = SVF(self.world_ptr)
self.filt_lag = Lag(self.world_ptr)
self.filt_freq = 1000.0

fn __repr__(self) -> String:
    return String("OscSynth")

fn next(mut self) -> SIMD[DType.float64, 2]:
    self.get_msgs()

    var out = 0.0

    for note_on in self.note_ons:
        print(note_on[0], note_on[1], note_on[2], end = "\n")
        self.current_voice = (self.current_voice + 1) % self.num_voices
        self.voices[self.current_voice].vol = Float64(note_on[2]) / 127.0
        self.voices[self.current_voice].trig = 1.0
        self.voices[self.current_voice].freq = midicps(note_on[1])
    self.note_ons.clear()

    # looking for midi cc on cc 34
    # this will control the frequency of the filter
    for cc in self.ccs:
        if cc[1] == 34: # Assuming CC 34 is for filter frequency
            self.filt_freq = linlin(Float64(cc[2]), 0.0, 127.0, 20.0, 1000.0) # Map CC value to frequency range

    if self.trig > 0.0:
        self.current_voice = (self.current_voice + 1) % self.num_voices
        self.voices[self.current_voice].trig = self.trig
        self.voices[self.current_voice].freq = self.freq

    # get the output of all the synths and reset the of the current voice (after getting audio)
    for i in range(len(self.voices)):
        out += self.voices[i].next()
        self.trig = 0.0
        self.voices[i].trig = 0.0 # Reset the trigger for the next iteration

    out = self.svf.lpf(out, self.filt_lag.next(self.filt_freq, 0.1), 2.0) * 0.6

    return out

fn get_msgs(mut self: Self):
    # calls to get_msg and get_midi return an Optional type
    # so you must get the value, then test the value to see if it exists, before using the value
    # get_msg returns a single list of values while get_midi returns a list of lists of values

    trig = self.world_ptr[0].get_msg("t_trig") # trig will be an Optional
    if trig: # if it trig is None, we do nothing
        self.trig = trig.value()[0]
    freq = self.world_ptr[0].get_msg("trig_seq_freq")
    if freq:
        self.freq = freq.value()[0]
    note_ons = self.world_ptr[0].get_midi("note_on", -1, -1) # Get all note on messages
    if note_ons:
        self.note_ons = note_ons.value().copy()

    ccs = self.world_ptr[0].get_midi("control_change", -1, -1)
    if ccs:
        self.ccs = ccs.value().copy()

struct Midi_Sequencer(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]

    var output: List[Float64] # Output buffer for audio samples

    var trig_synth: TrigSynth # Instance of the Oscillator

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.output = List[Float64](0.0, 0.0) # Initialize output list

        self.trig_synth = TrigSynth(world_ptr) # Initialize the TrigSynth with the world instance

    fn __repr__(self) -> String:
        return String("Midi_Sequencer")

    fn next(mut self: Midi_Sequencer) -> SIMD[DType.float64, 2]:
        return self.trig_synth.next() # Return the combined output sample

```

## 4.6 OleDusty

For more information about the examples, such as how the Python and Mojo files interact with each other, see the [Examples Overview](#)

### 4.6.1 Python Code

```
from mmm_src.MMMAudio import MMMAudio

# instantiate and load the graph
mmm_audio = MMMAudio(128, graph_name="OleDusty", package_name="examples")
mmm_audio.start_audio()
```

### 4.6.2 Mojo Code

```
from mmm_src.MMMWorld import MMMWorld
from mmm_dsp.Osc import Dust
from mmm_utils.functions import *
from mmm_dsp.Filters import *

# THE SYNTH

struct Dusty(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var dust: Dust[2]

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.dust = Dust[2](world_ptr)

    fn __repr__(self) -> String:
        return String("OleDusty")

    fn next(mut self, freq: Float64) -> SIMD[DType.float64, 2]:

        out = self.dust.next(freq)

        # uncomment below for use the phase of the Dust oscillator instead of the impulse
        # out = self.dust.get_phase()

        return out

# THE GRAPH

struct OleDusty(Representable, Movable, Copyable):
    var world_ptr: UnsafePointer[MMMWorld]
    var dusty: Dusty
    var reson: Reson[2]
    var freq: Float64

    fn __init__(out self, world_ptr: UnsafePointer[MMMWorld]):
        self.world_ptr = world_ptr
        self.dusty = Dusty(world_ptr)
        self.reson = Reson[2](world_ptr)
        self.freq = 20.0

    fn __repr__(self) -> String:
        return String("OleDusty")

    fn next(mut self) -> SIMD[DType.float64, 2]:

        freq = linterp(self.world_ptr[0].mouse_y, 0.0, 1.0, 100.0, 2000.0)

        out = self.dusty.next(linlin(self.world_ptr[0].mouse_x, 0.0, 1.0, 5.0, 200.0))

        # there is really no difference between ugens, synths, graphs
        # thus there is no reason you can't process the output of a synth directly in the graph
        # the reson filter uses SIMD to run 2 filters in parallel, each processing a channel of the dusty synth
        out = self.reson.hpf(out, freq, 10.0, 1.0) # apply a bandpass filter to the output of the Dusty synth

        return out
```

## 5. Contributing

---

### 5.1 Overview

---

test

## 5.2 Documentation Guide

---

This guide explains how to write and maintain documentation for MMAudio.

### 5.2.1 Documentation Structure

---

MMAudio uses MkDocs with Material theme for documentation generation. The documentation supports both Python and Mojo source files through different processing pipelines.

#### Python Documentation

Python files are documented using Google-style docstrings and processed by mkdocstrings:

```
def example_function(param1: int, param2: str = "default") -> bool:
    """Brief description of the function.

    Longer description with more details about what the function does,
    its intended use cases, and any important behavior notes.

    Args:
        param1: Description of the first parameter.
        param2: Description of the second parameter with default value.

    Returns:
        Description of the return value.

    Raises:
        ValueError: When param1 is negative.

    Examples:
        Basic usage:

        ```python
        result = example_function(42, "test")
        ```

        With default parameter:

        ```python
        result = example_function(42)
        ```

    """
    return param1 > 0
```

#### Mojo Documentation

Mojo files are documented using triple-quoted docstrings and processed by our custom adapter:

```
fn example_function[N: Int = 1](
    param1: SIMD[DType.int32, N],
    param2: SIMD[DType.float64, N] = 1.0
) -> SIMD[DType.bool, N]:
    """Brief description of the function.

    Longer description with details about the function's behavior,
    SIMD optimization, and usage patterns.

    Parameters:
        N: SIMD vector width (defaults to 1).

    Args:
        param1: Description of the first parameter.
        param2: Description of the second parameter with default.

    Returns:
        Description of the return value.

    Examples:
        Single value processing:

        ```mojo
        result = example_function(42, 1.5)
        ```

        Vectorized processing:

        ```mojo
        values = SIMD[DType.int32, 4](1, 2, 3, 4)
        factors = SIMD[DType.float64, 4](1.0, 1.5, 2.0, 2.5)
```



```

        results = example_function[4](values, factors)
        ...
    """
    return param1 > 0

```

## 5.2.2 Building Documentation

### Prerequisites

Install documentation dependencies:

```
pip install -r requirements-docs.txt
```

### Generate Documentation

Run the documentation pipeline:

```
python documentation/generate_docs.py
```

This will:

1. Process all Mojo files and extract documentation
2. Create markdown stubs for Python files
3. Generate example documentation
4. Build the complete documentation site

### Serve Locally

To preview the documentation locally:

```
mkdocs serve
```

The documentation will be available at `http://localhost:8000`.

### Build for Production

To build the static documentation site:

```
mkdocs build
```

This creates a `site/` directory with the complete documentation.

### Build PDF

To generate a PDF version:

```

mkdocs build
# PDF is generated automatically by mkdocs-pdf plugin

```

## 5.2.3 Documentation Standards

### Writing Guidelines

1. **Be Clear and Concise:** Use simple, direct language
2. **Include Examples:** Every function should have practical examples
3. **Explain SIMD Usage:** For Mojo functions, explain vectorization benefits
4. **Cross-Reference:** Link to related functions and concepts
5. **Keep Updated:** Update docs when code changes

**Code Examples**

- Use real, runnable code examples
- Show both basic and advanced usage
- Include expected output when helpful
- Demonstrate error conditions when relevant

**Function Documentation**

Required sections: - Brief description (first line) - Detailed description - Parameters/Args with types and descriptions - Returns section - Examples section

Optional sections: - Raises (for error conditions) - Notes (for implementation details) - See Also (for related functions)

## 5.2.4 Maintenance

---

**Regular Updates**

- Review documentation when adding new features
- Update examples to use current best practices
- Check for broken links and outdated information
- Ensure all public APIs are documented

**Documentation Reviews**

Include documentation updates in code reviews: - Verify new functions are documented - Check that examples are correct and clear - Ensure docstring formatting is consistent - Validate that generated docs look correct

## 5.3 Contributing to MMMAudio

---

Thank you for your interest in contributing to MMMAudio! This guide will help you get started.

### 5.3.1 Development Setup

---

#### Prerequisites

- Python 3.9+
- Mojo compiler (latest version)
- Git

#### Installation

1. Fork and clone the repository:

```
git clone https://github.com/your-username/MMMAudio.git
cd MMMAudio
```

1. Install dependencies:

```
pip install -r requirements-docs.txt
```

1. Verify the setup:

```
python -c "import mmm_src.MMMAudio; print('Python setup OK!')"
mojo --version
```

### 5.3.2 Contributing Guidelines

---

#### Code Style

##### PYTHON CODE

- Follow PEP 8 style guidelines
- Use type hints for all function signatures
- Use Google-style docstrings
- Maximum line length: 88 characters (Black formatter)

##### MOJO CODE

- Follow Mojo style conventions
- Use SIMD types for performance-critical code
- Document all public functions with examples
- Use descriptive variable names

#### Documentation

- All public APIs must be documented
- Include practical examples for each function
- Update documentation when changing functionality
- Use the documentation templates in `documentation/`

## Testing

- Add tests for new functionality
- Ensure existing tests pass
- Test both Python and Mojo implementations
- Include performance benchmarks for critical paths

## Pull Request Process

1. Create a feature branch:

```
git checkout -b feature/your-feature-name
```

2. Make your changes with clear, atomic commits
3. Update documentation and tests
4. Generate and review documentation:

```
python documentation/generate_docs.py
mkdocs serve
```

5. Submit a pull request with:
6. Clear description of changes
7. Rationale for the changes
8. Any breaking changes noted
9. Screenshots of documentation updates

## Issue Reporting

When reporting issues: - Use the issue templates - Include minimal reproduction case - Specify OS and version information - Include relevant error messages and logs

## 5.3.3 Development Workflow

---

### Adding New DSP Functions

1. Implement in Mojo for performance (in `mmm_dsp/`)
2. Add comprehensive documentation with examples
3. Create Python wrapper if needed (in `mmm_src/`)
4. Add tests demonstrating functionality
5. Update relevant examples

### Adding New Examples

1. Create example in `examples/` directory
2. Include clear documentation in docstring
3. Ensure example runs without errors
4. Add to examples index in documentation

### Performance Optimization

- Profile before optimizing
- Use SIMD operations where possible
- Benchmark against reference implementations

- Document performance characteristics

## 5.3.4 Community

---

### Communication

- GitHub Issues: Bug reports and feature requests
- Discussions: General questions and community interaction
- Pull Requests: Code contributions and reviews

### Code of Conduct

Please be respectful and constructive in all interactions. We're building a welcoming community for audio developers of all skill levels.

## 5.3.5 Release Process

---

Releases follow semantic versioning: - Major: Breaking changes - Minor: New features (backward compatible) - Patch: Bug fixes

Documentation is automatically built and deployed with each release.