

RTMHA Library  
2020a

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# Contents

<b>1</b>	<b>Hierarchical Index</b>	<b>1</b>
1.1	Class Hierarchy . . . . .	1
<b>2</b>	<b>Class Index</b>	<b>3</b>
2.1	Class List . . . . .	3
<b>3</b>	<b>File Index</b>	<b>5</b>
3.1	File List . . . . .	5
<b>4</b>	<b>Class Documentation</b>	<b>7</b>
4.1	adaptive_filter Class Reference . . . . .	7
4.1.1	Detailed Description . . . . .	8
4.1.2	Constructor & Destructor Documentation . . . . .	9
4.1.2.1	adaptive_filter() . . . . .	9
4.1.3	Member Function Documentation . . . . .	9
4.1.3.1	get_adaptation_type() . . . . .	10
4.1.3.2	get_max_frame_size() . . . . .	10
4.1.3.3	get_params() . . . . .	10
4.1.3.4	get_step_size_weights_IPNLMS() . . . . .	11
4.1.3.5	get_step_size_weights_SLMS() . . . . .	11
4.1.3.6	set_params() . . . . .	12
4.1.3.7	update_taps() . . . . .	12
4.2	afc Class Reference . . . . .	13

4.2.1	Detailed Description	14
4.2.2	Constructor & Destructor Documentation	15
4.2.2.1	afc()	15
4.2.3	Member Function Documentation	16
4.2.3.1	get_afc_on_off()	16
4.2.3.2	get_delay()	16
4.2.3.3	get_y_hat()	16
4.2.3.4	reset()	17
4.2.3.5	set_afc_on_off()	17
4.2.3.6	set_delay()	18
4.3	array_file Class Reference	18
4.3.1	Detailed Description	19
4.3.2	Constructor & Destructor Documentation	19
4.3.2.1	array_file()	19
4.3.3	Member Function Documentation	19
4.3.3.1	get_len()	19
4.3.3.2	get_ptr()	20
4.4	AudioFile< T > Class Template Reference	20
4.4.1	Detailed Description	21
4.4.2	Constructor & Destructor Documentation	21
4.4.2.1	AudioFile()	21
4.4.3	Member Function Documentation	21
4.4.3.1	getBitDepth()	21
4.4.3.2	getLengthInSeconds()	21
4.4.3.3	getNumChannels()	22
4.4.3.4	getNumSamplesPerChannel()	22
4.4.3.5	getSampleRate()	22
4.4.3.6	isMono()	22

4.4.3.7	<a href="#">isStereo()</a>	23
4.4.3.8	<a href="#">load()</a>	23
4.4.3.9	<a href="#">printSummary()</a>	23
4.4.3.10	<a href="#">save()</a>	23
4.4.3.11	<a href="#">setAudioBuffer()</a>	24
4.4.3.12	<a href="#">setAudioBufferSize()</a>	24
4.4.3.13	<a href="#">setBitDepth()</a>	24
4.4.3.14	<a href="#">setNumChannels()</a>	24
4.4.3.15	<a href="#">setNumSamplesPerChannel()</a>	25
4.4.3.16	<a href="#">setSampleRate()</a>	25
4.4.4	<a href="#">Member Data Documentation</a>	25
4.4.4.1	<a href="#">samples</a>	25
4.5	<a href="#">beamformer Class Reference</a>	26
4.5.1	<a href="#">Detailed Description</a>	27
4.5.2	<a href="#">Constructor &amp; Destructor Documentation</a>	27
4.5.2.1	<a href="#">beamformer()</a>	27
4.5.3	<a href="#">Member Function Documentation</a>	28
4.5.3.1	<a href="#">get_bf_params()</a>	29
4.5.3.2	<a href="#">get_e()</a>	29
4.5.3.3	<a href="#">set_bf_params()</a>	30
4.5.3.4	<a href="#">update_bf_taps()</a>	30
4.6	<a href="#">circular_buffer Class Reference</a>	31
4.6.1	<a href="#">Detailed Description</a>	31
4.6.2	<a href="#">Constructor &amp; Destructor Documentation</a>	31
4.6.2.1	<a href="#">circular_buffer()</a>	31
4.6.3	<a href="#">Member Function Documentation</a>	32
4.6.3.1	<a href="#">delay_block()</a>	32
4.6.3.2	<a href="#">get()</a>	32

4.6.3.3	set()	33
4.6.3.4	size()	33
4.7	file_play Class Reference	34
4.7.1	Detailed Description	34
4.8	file_record Class Reference	35
4.8.1	Detailed Description	36
4.9	filter Class Reference	36
4.9.1	Detailed Description	37
4.9.2	Constructor & Destructor Documentation	38
4.9.2.1	filter()	38
4.9.3	Member Function Documentation	38
4.9.3.1	cirfir() [1/2]	38
4.9.3.2	cirfir() [2/2]	39
4.9.3.3	get_size()	39
4.9.3.4	get_taps()	39
4.9.3.5	set_taps()	40
4.10	fir_formii Class Reference	40
4.10.1	Detailed Description	41
4.10.2	Constructor & Destructor Documentation	41
4.10.2.1	fir_formii()	41
4.10.3	Member Function Documentation	42
4.10.3.1	get_size()	42
4.10.3.2	get_taps()	42
4.10.3.3	process()	42
4.10.3.4	set_taps()	43
4.11	freping Class Reference	43
4.11.1	Detailed Description	44
4.11.2	Constructor & Destructor Documentation	44

4.11.2.1	<a href="#">freping()</a>	44
4.11.3	<a href="#">Member Function Documentation</a>	45
4.11.3.1	<a href="#">allpass_chain()</a>	45
4.11.3.2	<a href="#">freping_proc()</a>	46
4.11.3.3	<a href="#">get_params()</a>	46
4.11.3.4	<a href="#">overlap_add()</a>	46
4.11.3.5	<a href="#">set_params()</a>	47
4.11.3.6	<a href="#">windowing()</a>	47
4.12	<a href="#">GarbageCollector Class Reference</a>	48
4.12.1	<a href="#">Detailed Description</a>	48
4.13	<a href="#">libModule Class Reference</a>	48
4.13.1	<a href="#">Detailed Description</a>	49
4.14	<a href="#">noise_management Class Reference</a>	49
4.14.1	<a href="#">Detailed Description</a>	49
4.14.2	<a href="#">Constructor &amp; Destructor Documentation</a>	49
4.14.2.1	<a href="#">noise_management()</a>	49
4.14.3	<a href="#">Member Function Documentation</a>	50
4.14.3.1	<a href="#">get_param()</a>	50
4.14.3.2	<a href="#">set_param()</a>	50
4.14.3.3	<a href="#">speech_enhancement()</a>	51
4.15	<a href="#">peak_detect Class Reference</a>	51
4.15.1	<a href="#">Detailed Description</a>	52
4.15.2	<a href="#">Constructor &amp; Destructor Documentation</a>	52
4.15.2.1	<a href="#">peak_detect()</a>	52
4.15.3	<a href="#">Member Function Documentation</a>	52
4.15.3.1	<a href="#">get_param()</a>	52
4.15.3.2	<a href="#">get_spl()</a>	53
4.15.3.3	<a href="#">set_param()</a>	53

4.16	<a href="#">file_play::playfile_param_t Struct Reference</a>	54
4.16.1	<a href="#">Detailed Description</a>	54
4.17	<a href="#">polyphase_hb_downsampler Class Reference</a>	54
4.17.1	<a href="#">Detailed Description</a>	55
4.18	<a href="#">polyphase_hb_upsampler Class Reference</a>	55
4.18.1	<a href="#">Detailed Description</a>	55
4.19	<a href="#">resample Class Reference</a>	55
4.19.1	<a href="#">Detailed Description</a>	56
4.19.2	<a href="#">Constructor &amp; Destructor Documentation</a>	56
4.19.2.1	<a href="#">resample()</a>	56
4.19.3	<a href="#">Member Function Documentation</a>	56
4.19.3.1	<a href="#">resamp()</a>	56
4.20	<a href="#">rk_sema Struct Reference</a>	57
4.20.1	<a href="#">Detailed Description</a>	57
4.21	<a href="#">tenband_filterbank Class Reference</a>	57
4.21.1	<a href="#">Detailed Description</a>	58
4.21.2	<a href="#">Constructor &amp; Destructor Documentation</a>	58
4.21.2.1	<a href="#">tenband_filterbank()</a>	58
4.21.3	<a href="#">Member Function Documentation</a>	58
4.21.3.1	<a href="#">get()</a>	58
4.21.3.2	<a href="#">process()</a>	59
4.21.3.3	<a href="#">set()</a>	59
4.22	<a href="#">wdrc Class Reference</a>	59
4.22.1	<a href="#">Detailed Description</a>	60
4.22.2	<a href="#">Constructor &amp; Destructor Documentation</a>	60
4.22.2.1	<a href="#">wdrc()</a>	60
4.22.3	<a href="#">Member Function Documentation</a>	61
4.22.3.1	<a href="#">get_param()</a>	61
4.22.3.2	<a href="#">process()</a>	61
4.22.3.3	<a href="#">set_param()</a>	62



<b>5</b>	<b>File Documentation</b>	<b>63</b>
5.1	<a href="#">adaptive_filter.cpp File Reference</a>	63
5.1.1	<a href="#">Detailed Description</a>	63
5.2	<a href="#">adaptive_filter.hpp File Reference</a>	64
5.2.1	<a href="#">Detailed Description</a>	65
5.3	<a href="#">afc.cpp File Reference</a>	66
5.3.1	<a href="#">Detailed Description</a>	66
5.4	<a href="#">afc.hpp File Reference</a>	67
5.4.1	<a href="#">Detailed Description</a>	68
5.5	<a href="#">afc_init_filter.h File Reference</a>	68
5.5.1	<a href="#">Detailed Description</a>	68
5.6	<a href="#">array_file.cpp File Reference</a>	69
5.6.1	<a href="#">Detailed Description</a>	69
5.7	<a href="#">array_file.hpp File Reference</a>	70
5.7.1	<a href="#">Detailed Description</a>	70
5.8	<a href="#">array_utilities.cpp File Reference</a>	71
5.8.1	<a href="#">Detailed Description</a>	72
5.8.2	<a href="#">Function Documentation</a>	73
5.8.2.1	<a href="#">array_add_array()</a>	73
5.8.2.2	<a href="#">array_add_const()</a>	73
5.8.2.3	<a href="#">array_dot_product()</a>	74
5.8.2.4	<a href="#">array_element_divide_array()</a>	74
5.8.2.5	<a href="#">array_element_multiply_array()</a>	75
5.8.2.6	<a href="#">array_flip()</a>	75
5.8.2.7	<a href="#">array_mean()</a>	76
5.8.2.8	<a href="#">array_mean_square()</a>	76
5.8.2.9	<a href="#">array_min()</a>	77
5.8.2.10	<a href="#">array_multiply_const()</a>	77

5.8.2.11	<a href="#">array_print()</a>	78
5.8.2.12	<a href="#">array_right_shift()</a>	78
5.8.2.13	<a href="#">array_square()</a>	78
5.8.2.14	<a href="#">array_subtract_array()</a>	79
5.8.2.15	<a href="#">array_sum()</a>	79
5.9	<a href="#">array_utilities.hpp File Reference</a>	80
5.9.1	<a href="#">Detailed Description</a>	81
5.9.2	<a href="#">Function Documentation</a>	82
5.9.2.1	<a href="#">array_add_array()</a>	82
5.9.2.2	<a href="#">array_add_const()</a>	82
5.9.2.3	<a href="#">array_dot_product()</a>	83
5.9.2.4	<a href="#">array_element_divide_array()</a>	83
5.9.2.5	<a href="#">array_element_multiply_array()</a>	84
5.9.2.6	<a href="#">array_flip()</a>	84
5.9.2.7	<a href="#">array_mean()</a>	85
5.9.2.8	<a href="#">array_mean_square()</a>	85
5.9.2.9	<a href="#">array_min()</a>	85
5.9.2.10	<a href="#">array_multiply_const()</a>	86
5.9.2.11	<a href="#">array_print()</a>	86
5.9.2.12	<a href="#">array_right_shift()</a>	87
5.9.2.13	<a href="#">array_square()</a>	87
5.9.2.14	<a href="#">array_subtract_array()</a>	88
5.9.2.15	<a href="#">array_sum()</a>	88
5.10	<a href="#">AudioFile.cpp File Reference</a>	88
5.10.1	<a href="#">Detailed Description</a>	89
5.10.2	<a href="#">Variable Documentation</a>	89
5.10.2.1	<a href="#">aiffSampleRateTable</a>	90
5.11	<a href="#">AudioFile.h File Reference</a>	90

5.11.1 Detailed Description . . . . .	91
5.11.2 Enumeration Type Documentation . . . . .	92
5.11.2.1 AudioFileFormat . . . . .	92
5.12 bandlimited_filter.h File Reference . . . . .	92
5.12.1 Detailed Description . . . . .	92
5.12.2 Variable Documentation . . . . .	93
5.12.2.1 bandlimited_filter . . . . .	93
5.13 beamformer.cpp File Reference . . . . .	93
5.13.1 Detailed Description . . . . .	94
5.14 beamformer.hpp File Reference . . . . .	94
5.14.1 Detailed Description . . . . .	95
5.15 circular_buffer.cpp File Reference . . . . .	96
5.15.1 Detailed Description . . . . .	96
5.16 circular_buffer.hpp File Reference . . . . .	97
5.16.1 Detailed Description . . . . .	97
5.17 filter.cpp File Reference . . . . .	98
5.17.1 Detailed Description . . . . .	99
5.18 filter.hpp File Reference . . . . .	99
5.18.1 Detailed Description . . . . .	100
5.19 fir_formii.cpp File Reference . . . . .	101
5.19.1 Detailed Description . . . . .	101
5.20 fir_formii.h File Reference . . . . .	102
5.20.1 Detailed Description . . . . .	103
5.21 freping.cpp File Reference . . . . .	103
5.21.1 Detailed Description . . . . .	104
5.22 freping.hpp File Reference . . . . .	104
5.22.1 Detailed Description . . . . .	105
5.23 hamming_window128.h File Reference . . . . .	106
5.23.1 Detailed Description . . . . .	106
5.24 hamming_window64.h File Reference . . . . .	106
5.24.1 Detailed Description . . . . .	107
5.25 prefilter.h File Reference . . . . .	107
5.25.1 Detailed Description . . . . .	107
5.25.2 Variable Documentation . . . . .	108
5.25.2.1 prefilter . . . . .	108
5.26 sokolovaharris_filtercoef.h File Reference . . . . .	108
5.26.1 Detailed Description . . . . .	109



# Chapter 1

## Hierarchical Index

### 1.1 Class Hierarchy

This inheritance list is sorted roughly, but not completely, alphabetically:

array_file . . . . .	18
AudioFile< T > . . . . .	20
AudioFile< float > . . . . .	20
circular_buffer . . . . .	31
file_play . . . . .	34
file_record . . . . .	35
filter . . . . .	36
adaptive_filter . . . . .	7
afc . . . . .	13
beamformer . . . . .	26
fir_formii . . . . .	40
freping . . . . .	43
GarbageCollector . . . . .	48
libModule . . . . .	48
noise_management . . . . .	49
peak_detect . . . . .	51
file_play::playfile_param_t . . . . .	54
polyphase_hb_downsampler . . . . .	54
polyphase_hb_upsampler . . . . .	55
resample . . . . .	55
rk_sema . . . . .	57
tenband_filterbank . . . . .	57
wdrac . . . . .	59



## Chapter 2

# Class Index

### 2.1 Class List

Here are the classes, structs, unions and interfaces with brief descriptions:

<a href="#">adaptive_filter</a>	Adaptive Filter Class . . . . .	7
<a href="#">afc</a>	Adaptive Feedback Cancellation (AFC) Class . . . . .	13
<a href="#">array_file</a>	Array File Class . . . . .	18
<a href="#">AudioFile&lt; T &gt;</a>	. . . . .	20
<a href="#">beamformer</a>	Beamformer Class . . . . .	26
<a href="#">circular_buffer</a>	Circular Buffer Class . . . . .	31
<a href="#">file_play</a>	. . . . .	34
<a href="#">file_record</a>	. . . . .	35
<a href="#">filter</a>	Filter Class . . . . .	36
<a href="#">fir_formii</a>	Filter Class . . . . .	40
<a href="#">freping</a>	Freping Class . . . . .	43
<a href="#">GarbageCollector</a>	. . . . .	48
<a href="#">libModule</a>	A template for library modules . . . . .	48
<a href="#">noise_management</a>	Noise Management Class . . . . .	49
<a href="#">peak_detect</a>	Peak Detector Class . . . . .	51
<a href="#">file_play::playfile_param_t</a>	. . . . .	54
<a href="#">polyphase_hb_downsampler</a>	. . . . .	54
<a href="#">polyphase_hb_upsampler</a>	. . . . .	55
<a href="#">resample</a>	Resample Class . . . . .	55

<a href="#">rk_sema</a> . . . . .	57
<a href="#">tenband_filterbank</a>	
The Ten Band Filter bank is a 10 Band Multirate Filter Bank with its center frequencies located at 250Hz, 500Hz, 750Hz, 1kHz, 1.5kHz, 2kHz, 3kHz, 4kHz, 6kHz, 8kHz . . . . .	57
<a href="#">wdrc</a>	
Wide Dynamic Range Compression (WDRC) Class . . . . .	59



## Chapter 3

# File Index

### 3.1 File List

Here is a list of all documented files with brief descriptions:

<b>32_48_filter.h</b>	??
<b>48_32_filter.h</b>	??
<a href="#">adaptive_filter.cpp</a>	63
<a href="#">adaptive_filter.hpp</a>	64
<a href="#">afc.cpp</a>	66
<a href="#">afc.hpp</a>	67
<a href="#">afc_init_filter.h</a>	68
<a href="#">array_file.cpp</a>	69
<a href="#">array_file.hpp</a>	70
<a href="#">array_utilities.cpp</a>	71
<a href="#">array_utilities.hpp</a>	80
<a href="#">AudioFile.cpp</a>	88
<a href="#">AudioFile.h</a>	90
<a href="#">bandlimited_filter.h</a>	92
<a href="#">beamformer.cpp</a>	93
<a href="#">beamformer.hpp</a>	94
<a href="#">circular_buffer.cpp</a>	96
<a href="#">circular_buffer.hpp</a>	97
<b>example_template.cpp</b>	??
<b>file_record.cpp</b>	??
<b>file_record.h</b>	??
<a href="#">filter.cpp</a>	98
<a href="#">filter.hpp</a>	99
<a href="#">fir_formii.cpp</a>	101
<a href="#">fir_formii.h</a>	102
<a href="#">freping.cpp</a>	103
<a href="#">freping.hpp</a>	104
<b>GarbageCollector.hpp</b>	??
<a href="#">hamming_window128.h</a>	106
<a href="#">hamming_window64.h</a>	106
<b>noise_management.cpp</b>	??

noise_management.hpp	??
peak_detect.cpp	??
peak_detect.hpp	??
playfile.cpp	??
playfile.h	??
polyphase_hb_downsampler.cpp	??
polyphase_hb_downsampler.h	??
polyphase_hb_upsampler.cpp	??
polyphase_hb_upsampler.h	??
prefilter.h	107
resample.cpp	??
resample.hpp	??
sema.hpp	??
sokolovaharris_filtercoef.h	108
tenband_filterbank.cpp	??
tenband_filterbank.h	??
wdrc.cpp	??
wdrc.hpp	??

## Chapter 4

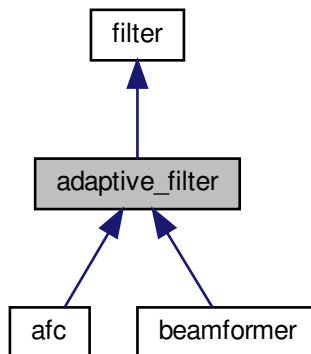
# Class Documentation

### 4.1 adaptive\_filter Class Reference

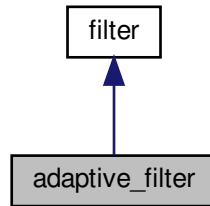
Adaptive Filter Class.

```
#include <adaptive_filter.hpp>
```

Inheritance diagram for adaptive\_filter:



Collaboration diagram for `adaptive_filter`:



## Public Member Functions

- `adaptive_filter` (float \*adaptive\_filter\_taps, size\_t adaptive\_filter\_tap\_len, size\_t max\_frame\_size, int adaptation\_type, float mu, float delta, float rho, float alpha, float beta, float p, float c, float power\_estimate)  
*Adaptive filter constructor.*
- `~adaptive_filter` ()  
*Adaptive filter destructor.*
- int `update_taps` (float \*u\_ref, float \*e\_ref, size\_t ref\_size)  
*To update the taps of this adaptive filter based on the reference signals u\_ref and e\_ref.*
- size\_t `get_max_frame_size` ()  
*Getting the maximum frame size.*
- void `get_params` (float &mu, float &rho, float &delta, float &alpha, float &beta, float &p, float &c, int &adaptation\_type)  
*Getting all parameters from this adaptive filter.*
- void `set_params` (float mu, float rho, float delta, float alpha, float beta, float p, float c, int adaptation\_type)  
*Setting all parameters from this adaptive filter.*

## Protected Member Functions

- int `get_adaptation_type` ()  
*A function to get the adaptation type.*
- void `get_step_size_weights_IPNLMS` (float \*taps, float \*step\_size\_weights, float alpha, float beta, float delta, size\_t tap\_len)  
*A function computing the step size control matrix for IPNLMS-I\_0.*
- void `get_step_size_weights_SLMS` (float \*taps, float \*step\_size\_weights, float p, float c, size\_t tap\_len)  
*A function computing the step size control matrix for SLMS.*

### 4.1.1 Detailed Description

Adaptive Filter Class.

This adaptive filter class implements several popular LMS-based algorithms including Modified LMS [Greenberg, 1998], IPNLMS-I\_0 [Paleologu et al., 2010] and SLMS [Lee et al., 2017].

Definition at line 39 of file `adaptive_filter.hpp`.

## 4.1.2 Constructor & Destructor Documentation

### 4.1.2.1 adaptive\_filter()

```
adaptive_filter::adaptive_filter (
    float * adaptive_filter_taps,
    size_t adaptive_filter_tap_len,
    size_t max_frame_size,
    int adaptation_type,
    float mu,
    float delta,
    float rho,
    float alpha,
    float beta,
    float p,
    float c,
    float power_estimate ) [explicit]
```

Adaptive filter constructor.

#### Parameters

in	<i>adaptive_filter_taps</i>	The initial filter taps for adaptive filter
in	<i>adaptive_filter_tap_len</i>	The number of filter taps of adaptive filter
in	<i>max_frame_size</i>	The maximum processing frame size in adaptive filter
in	<i>adaptation_type</i>	-1: 0 $\hat{y}$ , 0: stop adaptation, 1: Modified LMS, 2: IPNLMS-I_0, 3: SLMS
in	<i>mu</i>	The gradient descent step size (learning rate) for LMS-based algorithms
in	<i>delta</i>	A small positive number to prevent dividing zero
in	<i>rho</i>	The forgetting factor for power estimate
in	<i>alpha</i>	A number between -1 to 1 for different degrees of sparsity in IPNLMS-I_0
in	<i>beta</i>	A number between 0 to 500 for different degrees of sparsity in IPNLMS-I_0
in	<i>p</i>	A number between 0 to 2 for fitting different degrees of sparsity in SLMS
in	<i>c</i>	A small positive number for preventing stagnation in SLMS
in	<i>power_estimate</i>	An initial power estimate for adaptation

Definition at line 33 of file adaptive\_filter.cpp.

## 4.1.3 Member Function Documentation

#### 4.1.3.1 `get_adaptation_type()`

```
int adaptive_filter::get_adaptation_type ( ) [protected]
```

A function to get the adaptation type.

##### Returns

Adaptation type

Definition at line 164 of file `adaptive_filter.cpp`.

#### 4.1.3.2 `get_max_frame_size()`

```
size_t adaptive_filter::get_max_frame_size ( )
```

Getting the maximum frame size.

##### Returns

Maximum frame size

Definition at line 128 of file `adaptive_filter.cpp`.

#### 4.1.3.3 `get_params()`

```
void adaptive_filter::get_params (
    float & mu,
    float & rho,
    float & delta,
    float & alpha,
    float & beta,
    float & p,
    float & c,
    int & adaptation_type )
```

Getting all parameters from this adaptive filter.

##### Parameters

out	<i>mu</i>	The gradient descent step size (learning rate) for LMS-based algorithms
out	<i>rho</i>	The forgetting factor for power estimate
out	<i>delta</i>	A small positive number to prevent dividing zero
out	<i>alpha</i>	A number between -1 to 1 for different degrees of sparsity in IPNLMS-I_0
out	<i>beta</i>	A number between 0 to 500 for different degrees of sparsity in IPNLMS-I_0
out	<i>p</i>	A number between 0 to 2 for fitting different degrees of sparsity in SLMS
out	<i>c</i>	A small positive number for preventing stagnation in SLMS
out	<i>adaptation_type</i>	-1: 0 y_hat, 0: stop adaptation, 1: Modified LMS, 2: IPNLMS-I_0, 3: SLMS

Definition at line 133 of file adaptive\_filter.cpp.

#### 4.1.3.4 get\_step\_size\_weights\_IPNLMS()

```
void adaptive_filter::get_step_size_weights_IPNLMS (
    float * taps,
    float * step_size_weights,
    float alpha,
    float beta,
    float delta,
    size_t tap_len ) [protected]
```

A function computing the step size control matrix for IPNLMS-I\_0.

##### Parameters

in	<i>taps</i>	The current filter taps of the adaptive filter
out	<i>step_size_weights</i>	The step size control matrix (it is an 1-D array due to the diagonal matrix)
in	<i>alpha</i>	A number between -1 to 1 for different degrees of sparsity in IPNLMS-I_0
in	<i>beta</i>	A number between 0 to 500 for different degrees of sparsity in IPNLMS-I_0
in	<i>delta</i>	A small positive number to prevent dividing zero
in	<i>tap_len</i>	The number of taps of the adaptive filter

Definition at line 170 of file adaptive\_filter.cpp.

#### 4.1.3.5 get\_step\_size\_weights\_SLMS()

```
void adaptive_filter::get_step_size_weights_SLMS (
    float * taps,
    float * step_size_weights,
    float p,
    float c,
    size_t tap_len ) [protected]
```

A function computing the step size control matrix for SLMS.

##### Parameters

in	<i>taps</i>	The current filter taps of the adaptive filter
out	<i>step_size_weights</i>	The step size control matrix (it is an 1-D array due to the diagonal matrix)
in	<i>p</i>	A number between 0 to 2 for fitting different degrees of sparsity in SLMS
in	<i>c</i>	A small positive number for preventing stagnation in SLMS
in	<i>tap_len</i>	The number of taps of the adaptive filter

Definition at line 184 of file adaptive\_filter.cpp.

#### 4.1.3.6 set\_params()

```
void adaptive_filter::set_params (
    float mu,
    float rho,
    float delta,
    float alpha,
    float beta,
    float p,
    float c,
    int adaptation_type )
```

Setting all parameters from this adaptive filter.

##### Parameters

in	<i>mu</i>	The gradient descent step size (learning rate) for LMS-based algorithms
in	<i>rho</i>	The forgetting factor for power estimate
in	<i>delta</i>	A small positive number to prevent dividing zero
in	<i>alpha</i>	A number between -1 to 1 for different degrees of sparsity in IPNLMS-I_0
in	<i>beta</i>	A number between 0 to 500 for different degrees of sparsity in IPNLMS-I_0
in	<i>p</i>	A number between 0 to 2 for fitting different degrees of sparsity in SLMS
in	<i>c</i>	A small positive number for preventing stagnation in SLMS
in	<i>adaptation_type</i>	-1: 0 y_hat, 0: stop adaptation, 1: Modified LMS, 2: IPNLMS-I_0, 3: SLMS

Definition at line 147 of file adaptive\_filter.cpp.

#### 4.1.3.7 update\_taps()

```
int adaptive_filter::update_taps (
    float * u_ref,
    float * e_ref,
    size_t ref_size )
```

To update the taps of this adaptive filter based on the reference signals *u\_ref* and *e\_ref*.

##### Parameters

in	<i>u_ref</i>	A reference input signal for adaptation
in	<i>e_ref</i>	A reference error signal for adaptation
in	<i>ref_size</i>	The size of each reference signal ( <i>u_ref</i> and <i>e_ref</i> have the same size)



### Returns

A flag indicating the success of adaptation

Definition at line 70 of file adaptive\_filter.cpp.

The documentation for this class was generated from the following files:

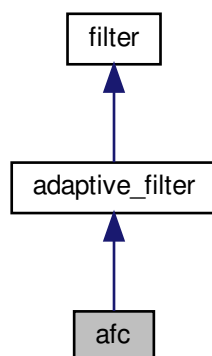
- [adaptive\\_filter.hpp](#)
- [adaptive\\_filter.cpp](#)

## 4.2 afc Class Reference

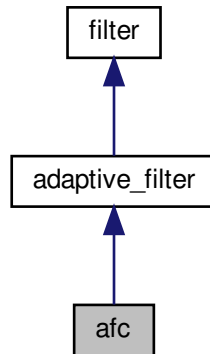
Adaptive Feedback Cancellation (AFC) Class.

```
#include <afc.hpp>
```

Inheritance diagram for afc:



Collaboration diagram for `afc`:



## Public Member Functions

- `afc` (float \*bandlimited\_filter\_taps, size\_t bandlimited\_filter\_tap\_len, float \*prefilter\_taps, size\_t prefilter\_tap\_len, float \*adaptive\_filter\_taps, size\_t adaptive\_filter\_tap\_len, size\_t max\_frame\_size, int adaptation\_type, float mu, float delta, float rho, float alpha, float beta, float p, float c, float power\_estimate, size\_t delay\_len, int afc\_on\_off)  
*AFC constructor.*
- `~afc` ()  
*AFC destructor.*
- int `get_y_hat` (float \*y\_hat, float \*e, float \*s, size\_t ref\_size)  
*Getting y\_hat signal (an estimated feedback signal)*
- void `get_delay` (size\_t &delay\_len)  
*Getting the length of delay line in samples.*
- int `set_delay` (size\_t delay\_len)  
*Setting the length of delay line in samples.*
- void `set_afc_on_off` (int afc\_on\_off)  
*Setting the ON/OFF for the AFC.*
- void `get_afc_on_off` (int &afc\_on\_off)  
*Getting the ON/OFF for the AFC.*
- void `reset` (float \*default\_taps, size\_t len)  
*Reset the AFC filter to all zeros.*

## Additional Inherited Members

### 4.2.1 Detailed Description

Adaptive Feedback Cancellation (AFC) Class.

Under the FXLMS framework, this AFC class utilizes an adaptive filter to estimate the feedback signal, namely, `y_hat`.

Definition at line 40 of file `afc.hpp`.

## 4.2.2 Constructor & Destructor Documentation

### 4.2.2.1 afc()

```
afc::afc (
    float * bandlimited_filter_taps,
    size_t bandlimited_filter_tap_len,
    float * prefilter_taps,
    size_t prefilter_tap_len,
    float * adaptive_filter_taps,
    size_t adaptive_filter_tap_len,
    size_t max_frame_size,
    int adaptation_type,
    float mu,
    float delta,
    float rho,
    float alpha,
    float beta,
    float p,
    float c,
    float power_estimate,
    size_t delay_len,
    int afc_on_off ) [explicit]
```

AFC constructor.

#### Parameters

in	<i>bandlimited_filter_taps</i>	The filter taps for bandlimited filter in AFC
in	<i>bandlimited_filter_tap_len</i>	The number of taps of bandlimited filter in AFC
in	<i>prefilter_taps</i>	The filter taps for whitening filter in AFC
in	<i>prefilter_tap_len</i>	The number of taps of prefilter in AFC
in	<i>adaptive_filter_taps</i>	The initial filter taps for adaptive filter in AFC
in	<i>adaptive_filter_tap_len</i>	The number of filter taps of adaptive filter in AFC
in	<i>max_frame_size</i>	The maximum processing frame size in adaptive filter
in	<i>adaptation_type</i>	The adaptation type for adaptive filter
in	<i>mu</i>	A parameter for adaptive filter
in	<i>delta</i>	A parameter for adaptive filter
in	<i>rho</i>	A parameter for adaptive filter
in	<i>alpha</i>	A parameter for adaptive filter
in	<i>beta</i>	A parameter for adaptive filter
in	<i>p</i>	A parameter for adaptive filter
in	<i>c</i>	A parameter for adaptive filter
in	<i>power_estimate</i>	A parameter for adaptive filter
in	<i>delay_len</i>	The number of delay in samples
in	<i>afc_on_off</i>	A flag to turn the AFC on (0: OFF, 1: ON)

See also

[adaptive\\_filter](#)

Definition at line 29 of file `afc.cpp`.

### 4.2.3 Member Function Documentation

#### 4.2.3.1 `get_afc_on_off()`

```
void afc::get_afc_on_off (
    int & afc_on_off )
```

Getting the ON/OFF for the AFC.

Parameters

in	<i>afc_on_off</i>	A flag indicating the ON/OFF of AFC (False: OFF, True: ON)
----	-------------------	--

Definition at line 120 of file `afc.cpp`.

#### 4.2.3.2 `get_delay()`

```
void afc::get_delay (
    size_t & delay_len )
```

Getting the length of delay line in samples.

Parameters

out	<i>delay_len</i>	The number of delay in samples
-----	------------------	--------------------------------

Definition at line 91 of file `afc.cpp`.

#### 4.2.3.3 `get_y_hat()`

```
int afc::get_y_hat (
    float * y_hat,
```

```
float * e,
float * s,
size_t ref_size )
```

Getting  $y_{\hat{}}$  signal (an estimated feedback signal)

#### Parameters

out	<i>y_hat</i>	An estimated feedback signal
in	<i>e</i>	An error signal for AFC (the output of hearing aid processing)
in	<i>s</i>	An input signal for AFC (the input of hearing aid processing)
in	<i>ref_size</i>	The size of each signal (e and s have the same size)

#### Returns

A flag indicating the success of getting correct  $y_{\hat{}}$  according to the adaptation type

Definition at line 63 of file afc.cpp.

#### 4.2.3.4 reset()

```
void afc::reset (
    float * default_taps,
    size_t len )
```

Reset the AFC filter to all zeros.

#### Parameters

in	<i>default_taps</i>	The default AFC filter
in	<i>len</i>	Length of the default AFC filter

Definition at line 126 of file afc.cpp.

#### 4.2.3.5 set\_afc\_on\_off()

```
void afc::set_afc_on_off (
    int afc_on_off )
```

Setting the ON/OFF for the AFC.

**Parameters**

in	<i>afc_on_off</i>	A flag to turn the AFC on (False: OFF, True: ON)
----	-------------------	--

Definition at line 111 of file `afc.cpp`.

**4.2.3.6 set\_delay()**

```
int afc::set_delay (
    size_t delay_len )
```

Setting the length of delay line in samples.

**Parameters**

in	<i>delay_len</i>	The number of delay in samples
----	------------------	--------------------------------

**Returns**

A flag indicating the success of setting `delay_len`

Definition at line 97 of file `afc.cpp`.

The documentation for this class was generated from the following files:

- [afc.hpp](#)
- [afc.cpp](#)

**4.3 array\_file Class Reference**

Array File Class.

```
#include <array_file.hpp>
```

**Public Member Functions**

- [array\\_file](#) (std::string path)  
*Array file constructor.*
- [~array\\_file](#) ()  
*Array file destructor.*
- size\_t [get\\_len](#) ()  
*Getting the length of the array.*
- float \* [get\\_ptr](#) ()  
*Getting the pointer which points to the array.*

### 4.3.1 Detailed Description

Array File Class.

Reading a binary file into an array in single-precision floating-point format

Definition at line 33 of file array\_file.hpp.

### 4.3.2 Constructor & Destructor Documentation

#### 4.3.2.1 array\_file()

```
array_file::array_file (
    std::string path )
```

Array file constructor.

##### Parameters

<i>path</i>	The path of the binary file
-------------	-----------------------------

Definition at line 32 of file array\_file.cpp.

### 4.3.3 Member Function Documentation

#### 4.3.3.1 get\_len()

```
size_t array_file::get_len ( )
```

Getting the length of the array.

##### Returns

The length of the array

Definition at line 58 of file array\_file.cpp.

#### 4.3.3.2 `get_ptr()`

```
float * array_file::get_ptr ( )
```

Getting the pointer which points to the array.

##### Returns

The pointer which points to the array

Definition at line 63 of file `array_file.cpp`.

The documentation for this class was generated from the following files:

- [array\\_file.hpp](#)
- [array\\_file.cpp](#)

## 4.4 `AudioFile< T >` Class Template Reference

### Public Types

- typedef std::vector< std::vector< T > > **AudioBuffer**

### Public Member Functions

- [AudioFile](#) ()
- bool [load](#) (std::string filePath)
- bool [save](#) (std::string filePath, [AudioFileFormat](#) format=AudioFileFormat::Wave)
- uint32\_t [getSampleRate](#) () const
- int [getNumChannels](#) () const
- bool [isMono](#) () const
- bool [isStereo](#) () const
- int [getBitDepth](#) () const
- int [getNumSamplesPerChannel](#) () const
- double [getLengthInSeconds](#) () const
- void [printSummary](#) () const
- bool [setAudioBuffer](#) (AudioBuffer &newBuffer)
- void [setAudioBufferSize](#) (int numChannels, int numSamples)
- void [setNumSamplesPerChannel](#) (int numSamples)
- void [setNumChannels](#) (int numChannels)
- void [setBitDepth](#) (int numBitsPerSample)
- void [setSampleRate](#) (uint32\_t newSampleRate)

### Public Attributes

- AudioBuffer [samples](#)



### 4.4.1 Detailed Description

```
template<class T>  
class AudioFile< T >
```

Definition at line 47 of file AudioFile.h.

### 4.4.2 Constructor & Destructor Documentation

#### 4.4.2.1 AudioFile()

```
template<class T >  
AudioFile< T >::AudioFile ( )
```

Constructor

Definition at line 54 of file AudioFile.cpp.

### 4.4.3 Member Function Documentation

#### 4.4.3.1 getBitDepth()

```
template<class T >  
int AudioFile< T >::getBitDepth ( ) const
```

the bit depth of each sample

Definition at line 93 of file AudioFile.cpp.

#### 4.4.3.2 getLengthInSeconds()

```
template<class T >  
double AudioFile< T >::getLengthInSeconds ( ) const
```

the length in seconds of the audio file based on the number of samples and sample rate

Definition at line 110 of file AudioFile.cpp.

#### 4.4.3.3 getNumChannels()

```
template<class T >
int AudioFile< T >::getNumChannels ( ) const
```

the number of audio channels in the buffer

Definition at line 72 of file AudioFile.cpp.

#### 4.4.3.4 getNumSamplesPerChannel()

```
template<class T >
int AudioFile< T >::getNumSamplesPerChannel ( ) const
```

the number of samples per channel

Definition at line 100 of file AudioFile.cpp.

#### 4.4.3.5 getSampleRate()

```
template<class T >
uint32_t AudioFile< T >::getSampleRate ( ) const
```

the sample rate

Definition at line 65 of file AudioFile.cpp.

#### 4.4.3.6 isMono()

```
template<class T >
bool AudioFile< T >::isMono ( ) const
```

true if the audio file is mono

Definition at line 79 of file AudioFile.cpp.

#### 4.4.3.7 isStereo()

```
template<class T >
bool AudioFile< T >::isStereo ( ) const
```

true if the audio file is stereo

Definition at line 86 of file AudioFile.cpp.

#### 4.4.3.8 load()

```
template<class T >
bool AudioFile< T >::load (
    std::string filePath )
```

Loads an audio file from a given file path. true if the file was successfully loaded

Definition at line 221 of file AudioFile.cpp.

#### 4.4.3.9 printSummary()

```
template<class T >
void AudioFile< T >::printSummary ( ) const
```

Prints a summary of the audio file to the console

Definition at line 117 of file AudioFile.cpp.

#### 4.4.3.10 save()

```
template<class T >
bool AudioFile< T >::save (
    std::string filePath,
    AudioFileFormat format = AudioFileFormat::Wave )
```

Saves an audio file to a given file path. true if the file was successfully saved

Definition at line 525 of file AudioFile.cpp.

#### 4.4.3.11 `setAudioBuffer()`

```
template<class T >
bool AudioFile< T >::setAudioBuffer (
    AudioBuffer & newBuffer )
```

Set the audio buffer for this [AudioFile](#) by copying samples from another buffer. true if the buffer was copied successfully.

Definition at line 130 of file `AudioFile.cpp`.

#### 4.4.3.12 `setAudioBufferSize()`

```
template<class T >
void AudioFile< T >::setAudioBufferSize (
    int numChannels,
    int numSamples )
```

Sets the audio buffer to a given number of channels and number of samples per channel. This will try to preserve the existing audio, adding zeros to any new channels or new samples in a given channel.

Definition at line 162 of file `AudioFile.cpp`.

#### 4.4.3.13 `setBitDepth()`

```
template<class T >
void AudioFile< T >::setBitDepth (
    int numBitsPerSample )
```

Sets the bit depth for the audio file. If you use the [save\(\)](#) function, this bit depth rate will be used

Definition at line 207 of file `AudioFile.cpp`.

#### 4.4.3.14 `setNumChannels()`

```
template<class T >
void AudioFile< T >::setNumChannels (
    int numChannels )
```

Sets the number of channels. New channels will have the correct number of samples and be initialised to zero

Definition at line 186 of file `AudioFile.cpp`.

#### 4.4.3.15 `setNumSamplesPerChannel()`

```
template<class T >
void AudioFile< T >::setNumSamplesPerChannel (
    int numSamples )
```

Sets the number of samples per channel in the audio buffer. This will try to preserve the existing audio, adding zeros to new samples in a given channel if the number of samples is increased.

Definition at line 170 of file `AudioFile.cpp`.

#### 4.4.3.16 `setSampleRate()`

```
template<class T >
void AudioFile< T >::setSampleRate (
    uint32_t newSampleRate )
```

Sets the sample rate for the audio file. If you use the `save()` function, this sample rate will be used

Definition at line 214 of file `AudioFile.cpp`.

### 4.4.4 Member Data Documentation

#### 4.4.4.1 `samples`

```
template<class T>
AudioBuffer AudioFile< T >::samples
```

A vector of vectors holding the audio samples for the `AudioFile`. You can access the samples by channel and then by sample index, i.e:

```
samples[channel][sampleIndex]
```

Definition at line 126 of file `AudioFile.h`.

The documentation for this class was generated from the following files:

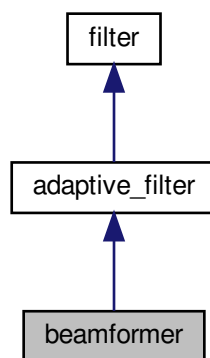
- [AudioFile.h](#)
- [AudioFile.cpp](#)

## 4.5 beamformer Class Reference

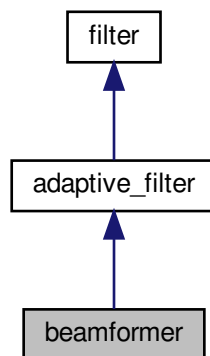
Beamformer Class.

```
#include <beamformer.hpp>
```

Inheritance diagram for beamformer:



Collaboration diagram for beamformer:



## Public Member Functions

- [beamformer](#) (size\_t delay\_len, float \*adaptive\_filter\_taps, size\_t adaptive\_filter\_tap\_len, size\_t max\_frame\_size, int adaptation\_type, float mu, float delta, float rho, float alpha, float beta, float p, float c, float power\_estimate, int bf\_on\_off, int bf\_nc\_on\_off, int bf\_amc\_on\_off, float nc\_thr, float amc\_thr, float amc\_forgetting\_factor)

*Beamformer constructor.*

- [~beamformer](#) ()

*Beamformer destructor.*

- void [get\\_e](#) (float \*e\_l, float \*e\_r, const float \*x\_l, const float \*x\_r, size\_t ref\_size)

*Getting e signal (the output signal of this beamformer)*

- int [update\\_bf\\_taps](#) (size\_t ref\_size)

*Update the taps of the beamformer.*

- void [get\\_bf\\_params](#) (int &bf\_on\_off, int &bf\_nc\_on\_off, int &bf\_amc\_on\_off, float &nc\_thr, float &amc\_thr, float &amc\_forgetting\_factor)

*Getting all parameters from this beamformer.*

- void [set\\_bf\\_params](#) (int bf\_on\_off, int bf\_nc\_on\_off, int bf\_amc\_on\_off, float nc\_thr, float amc\_thr, float amc\_forgetting\_factor)

*Setting all parameters from this beamformer.*

## Additional Inherited Members

### 4.5.1 Detailed Description

Beamformer Class.

This beamformer class implements the generalized sidelobe canceller (GSC) using SLMS [Lee et al., IHCON 2018].

Definition at line 39 of file beamformer.hpp.

### 4.5.2 Constructor & Destructor Documentation

#### 4.5.2.1 beamformer()

```
beamformer::beamformer (
    size_t delay_len,
    float * adaptive_filter_taps,
    size_t adaptive_filter_tap_len,
    size_t max_frame_size,
    int adaptation_type,
    float mu,
    float delta,
    float rho,
    float alpha,
    float beta,
```

```

float p,
float c,
float power_estimate,
int bf_on_off,
int bf_nc_on_off,
int bf_amc_on_off,
float nc_thr,
float amc_thr,
float amc_forgetting_factor ) [explicit]

```

Beamformer constructor.

#### Parameters

in	<i>delay_len</i>	The length of delay line in samples for beamformer
in	<i>adaptive_filter_taps</i>	The initial filter taps for adaptive filter in beamformer
in	<i>adaptive_filter_tap_len</i>	The number of filter taps of adaptive filter in beamformer
in	<i>max_frame_size</i>	The maximum processing frame size in adaptive filter
in	<i>adaptation_type</i>	The adaptation type for adaptive filter
in	<i>mu</i>	A parameter for adaptive filter
in	<i>delta</i>	A parameter for adaptive filter
in	<i>rho</i>	A parameter for adaptive filter
in	<i>alpha</i>	A parameter for adaptive filter
in	<i>beta</i>	A parameter for adaptive filter
in	<i>p</i>	A parameter for adaptive filter
in	<i>c</i>	A parameter for adaptive filter
in	<i>power_estimate</i>	A parameter for adaptive filter
in	<i>bf_on_off</i>	A flag for enabling beamformer
in	<i>bf_nc_on_off</i>	A flag for enabling norm-constrained adaptation
in	<i>bf_amc_on_off</i>	A flag for enabling adaptation mode controller
in	<i>nc_thr</i>	A threshold for the norm-constrained adaptation
in	<i>amc_thr</i>	A threshold for the adaptation mode controller

See also

[adaptive\\_filter](#)

Definition at line 32 of file beamformer.cpp.

### 4.5.3 Member Function Documentation



4.5.3.1 `get_bf_params()`

```
void beamformer::get_bf_params (
    int & bf_on_off,
    int & bf_nc_on_off,
    int & bf_amc_on_off,
    float & nc_thr,
    float & amc_thr,
    float & amc_forgetting_factor )
```

Getting all parameters from this beamformer.

## Parameters

out	<i>bf_on_off</i>	A flag to turn the beamformer on (0: OFF, 1: ON)
out	<i>bf_nc_on_off</i>	A flag to turn the norm-constrained adaptation on (0: OFF, 1: ON)
out	<i>bf_amc_on_off</i>	A flag to turn the adaptation mode controller on (0: OFF, 1: ON)
out	<i>nc_thr</i>	A threshold for the norm-constrained adaptation
out	<i>amc_thr</i>	A threshold for the adaptation mode controller
out	<i>amc_forgetting_factor</i>	A forgetting factor to compute the signal power

Definition at line 142 of file beamformer.cpp.

4.5.3.2 `get_e()`

```
void beamformer::get_e (
    float * e_l,
    float * e_r,
    const float * x_l,
    const float * x_r,
    size_t ref_size )
```

Getting e signal (the output signal of this beamformer)

## Parameters

out	<i>e_l</i>	The output signal of this beamformer on the left
out	<i>e_r</i>	The output signal of this beamformer on the right
in	<i>x_l</i>	The input signal from the left channel
in	<i>x_r</i>	the input signal from the right channel
in	<i>ref_size</i>	The size of each input signal ( <i>x_l</i> and <i>x_r</i> have the same size)

Definition at line 70 of file beamformer.cpp.

#### 4.5.3.3 set\_bf\_params()

```
void beamformer::set_bf_params (
    int bf_on_off,
    int bf_nc_on_off,
    int bf_amc_on_off,
    float nc_thr,
    float amc_thr,
    float amc_forgetting_factor )
```

Setting all parameters from this beamformer.

##### Parameters

in	<i>bf_on_off</i>	A flag to turn the beamformer on (0: OFF, 1: ON)
in	<i>bf_nc_on_off</i>	A flag to turn the norm-constrained adaptation on (0: OFF, 1: ON)
in	<i>bf_amc_on_off</i>	A flag to turn the adaptation mode controller on (0: OFF, 1: ON)
in	<i>nc_thr</i>	A threshold for the norm-constrained adaptation
in	<i>amc_thr</i>	A threshold for the adaptation mode controller
in	<i>amc_forgetting_factor</i>	A forgetting factor to compute the signal power

Definition at line 153 of file beamformer.cpp.

#### 4.5.3.4 update\_bf\_taps()

```
int beamformer::update_bf_taps (
    size_t ref_size )
```

Update the taps of the beamformer.

##### Parameters

in	<i>ref_size</i>	The size of each input signal ( <i>x_l</i> and <i>x_r</i> have the same size)
----	-----------------	---

##### Returns

A flag indicating the success of adaptation in adaptive filter

Definition at line 102 of file beamformer.cpp.

The documentation for this class was generated from the following files:

- [beamformer.hpp](#)
- [beamformer.cpp](#)

## 4.6 circular\_buffer Class Reference

Circular Buffer Class.

```
#include <circular_buffer.hpp>
```

### Public Member Functions

- `circular_buffer` (size\_t `size`, float `reset`)  
*Circular buffer constructor.*
- `~circular_buffer` ()  
*Default destructor.*
- void `set` (const float \*item, size\_t buf\_size)  
*This is the set command for the circular buffer.*
- void `get` (float \*data, size\_t buf\_size)  
*This is the get function for the circular buffer.*
- void `delay_block` (float \*data, size\_t buf\_size, size\_t delay)  
*This is the get function for the circular buffer with a delay.*
- void `reset` ()  
*This is the reset command for circular buffer. It resets all of the values in the buffer to the default value the user entered in the constructor.*
- size\_t `size` () const  
*Function to get the size of the buffer.*

### Public Attributes

- float \* `buf_`
- std::atomic< size\_t > `head_`
- size\_t `size_`
- size\_t `mask_`
- float `reset_`

#### 4.6.1 Detailed Description

Circular Buffer Class.

Definition at line 36 of file circular\_buffer.hpp.

#### 4.6.2 Constructor & Destructor Documentation

##### 4.6.2.1 circular\_buffer()

```
circular_buffer::circular_buffer (
    size_t size,
    float reset ) [explicit]
```

Circular buffer constructor.

**Parameters**

in	<i>size</i>	The maximum size you would want your circular buffer to be
in	<i>reset</i>	The value you want to reset all of the values in the circular buffer to

Definition at line 34 of file circular\_buffer.cpp.

### 4.6.3 Member Function Documentation

#### 4.6.3.1 delay\_block()

```
void circular_buffer::delay_block (
    float * data,
    size_t buf_size,
    size_t delay )
```

This is the get function for the circular buffer with a delay.

**Parameters**

out	<i>data</i>	A buffer to put your data in.
in	<i>buf_size</i>	The amount of data you want from the circular buffer
in	<i>delay</i>	The delay you want form the delay block in terms of time step

Definition at line 87 of file circular\_buffer.cpp.

#### 4.6.3.2 get()

```
void circular_buffer::get (
    float * data,
    size_t buf_size )
```

This is the get function for the circular buffer.

**Parameters**

out	<i>data</i>	A buffer to put your data in.
in	<i>buf_size</i>	The amount of data you want from the circular buffer

Definition at line 71 of file circular\_buffer.cpp.

#### 4.6.3.3 set()

```
void circular_buffer::set (
    const float * item,
    size_t buf_size )
```

This is the set command for the circular buffer.

##### Parameters

in	<i>item</i>	The buffer of data you want to put in the circular buffer
in	<i>buf_size</i>	The size of the buffer.

Definition at line 52 of file circular\_buffer.cpp.

#### 4.6.3.4 size()

```
size_t circular_buffer::size ( ) const
```

Function to get the size of the buffer.

##### Returns

The size of the circular buffer which will be a power of 2

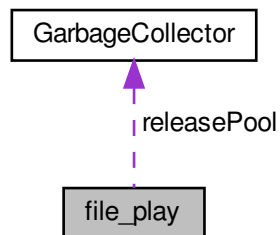
Definition at line 111 of file circular\_buffer.cpp.

The documentation for this class was generated from the following files:

- [circular\\_buffer.hpp](#)
- [circular\\_buffer.cpp](#)

## 4.7 file\_play Class Reference

Collaboration diagram for file\_play:



### Classes

- struct [playfile\\_param\\_t](#)

### Public Member Functions

- void **rtmha\_play** (int num\_sample, float \*out, int channel)
- void **set\_params** (const char \*, int, int, int)

### Public Attributes

- std::string **rootPath**
- std::shared\_ptr< [playfile\\_param\\_t](#) > **currentParam**
- [GarbageCollector](#) **releasePool**

### 4.7.1 Detailed Description

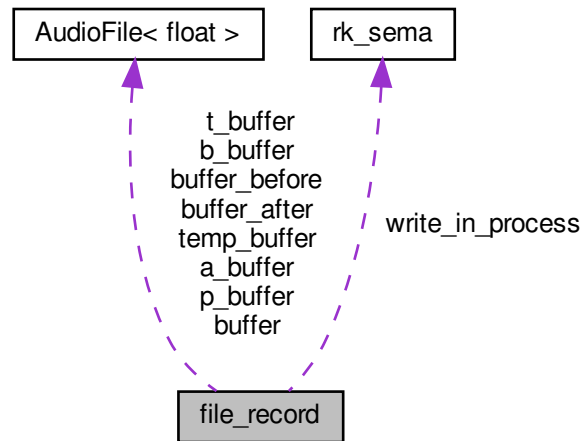
Definition at line 7 of file playfile.h.

The documentation for this class was generated from the following files:

- playfile.h
- playfile.cpp

## 4.8 file\_record Class Reference

Collaboration diagram for file\_record:



### Public Member Functions

- **file\_record** (int freq=48000, std::string file\_="sample.wav", float seconds=5)
- void **rtmha\_record** (int num\_sample, float \*in, int channel)
- void **record\_before** (int num\_sample, float \*in, int channel)
- void **record\_after** (int num\_sample, float \*in, int channel)
- void **set\_params** (int start\_, int stop\_, float seconds, const char \*file\_)
- void **get\_params** (float &seconds)
- void **write\_to\_file** ()

### Public Attributes

- `AudioFile< float >::AudioBuffer` **buffer**
- `AudioFile< float >::AudioBuffer` **temp\_buffer**
- `AudioFile< float >::AudioBuffer` **buffer\_before**
- `AudioFile< float >::AudioBuffer` **buffer\_after**
- std::string **rootPath**
- std::string **file**
- std::atomic< bool > **write**
- `AudioFile< float >::AudioBuffer *` **p\_buffer** =0
- `AudioFile< float >::AudioBuffer *` **t\_buffer** =0
- `AudioFile< float >::AudioBuffer *` **b\_buffer** =&buffer\_before
- `AudioFile< float >::AudioBuffer *` **a\_buffer** =&buffer\_after

- int **reset**
- int **start** =0
- int **stop** =0
- int **start\_before**
- int **start\_after** =0
- int **stop\_before** =0
- int **stop\_after**
- int **record**
- int **repeat**
- std::string **file\_path**
- int **current\_position\_l**
- int **current\_position\_r**
- int **current\_before\_l**
- int **current\_before\_r**
- int **current\_after\_l**
- int **current\_after\_r**
- int **sampling\_freq**
- int **numSamples**
- int **numSamples\_before**
- int **numSamples\_after**
- int **finish**
- int **mono**
- int **t**
- [rk\\_sema](#) \* **write\_in\_process**
- std::thread \* **record\_thread**

#### 4.8.1 Detailed Description

Definition at line 18 of file file\_record.h.

The documentation for this class was generated from the following files:

- file\_record.h
- file\_record.cpp

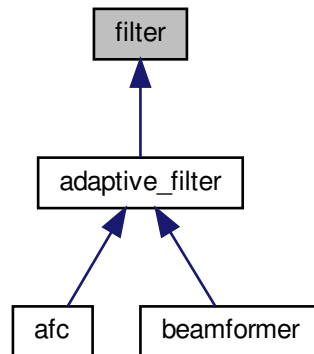
## 4.9 filter Class Reference

Filter Class.

```
#include <filter.hpp>
```



Inheritance diagram for filter:



## Public Member Functions

- [filter](#) (float \*taps, size\_t tap\_size, [circular\\_buffer](#) \*cir\_buf, size\_t max\_buf\_size)  
*Filter constructor.*
- [~filter](#) ()  
*Filter destructor.*
- int [set\\_taps](#) (const float \*taps, size\_t buf\_size)  
*Setting the filter taps.*
- int [get\\_taps](#) (float \*taps, size\_t buf\_size)  
*Getting the filter taps.*
- void [cirfir](#) (float \*data\_in, float \*data\_out, size\_t num\_samp)  
*Getting the output of this FIR filter by performing frame-based convolution.*
- size\_t [get\\_size](#) ()  
*Getting the number of taps of this FIR filter.*
- void [cirfir](#) (float \*data\_out, size\_t num\_samp)  
*Frame-based convolution for FIR filtering.*

### 4.9.1 Detailed Description

Filter Class.

This filter class implements the FIR filter

Definition at line 39 of file filter.hpp.

## 4.9.2 Constructor & Destructor Documentation

### 4.9.2.1 filter()

```
filter::filter (
    float * taps,
    size_t tap_size,
    circular_buffer * cir_buf,
    size_t max_buf_size ) [explicit]
```

Filter constructor.

#### Parameters

in	<i>taps</i>	The filter taps for this FIR filter
in	<i>tap_size</i>	The number of taps of this FIR filter
in	<i>cir_buf</i>	The circular buffer for this FIR filter to perform frame-based convolution
in	<i>max_buf_size</i>	The maximum size of circular buffer you need to specify if there is no circular buffer given in <i>cir_buf</i>

Definition at line 31 of file filter.cpp.

## 4.9.3 Member Function Documentation

### 4.9.3.1 cirfir() [1/2]

```
void filter::cirfir (
    float * data_in,
    float * data_out,
    size_t num_samp )
```

Getting the output of this FIR filter by performing frame-based convolution.

#### Parameters

in	<i>data_in</i>	The input signal
out	<i>data_out</i>	The output signal
	<i>num_samp</i>	The size of input and output signal ( <i>data_in</i> and <i>data_out</i> should have the same size)

Definition at line 96 of file filter.cpp.

#### 4.9.3.2 `cirfir()` [2/2]

```
void filter::cirfir (
    float * data_out,
    size_t num_samp )
```

Frame-based convolution for FIR filtering.

##### Parameters

out	<i>data_out</i>	The output signal
in	<i>num_samp</i>	The size of input and output signal

Definition at line 107 of file filter.cpp.

#### 4.9.3.3 `get_size()`

```
size_t filter::get_size ( )
```

Getting the number of taps of this FIR filter.

##### Returns

The number of taps of this FIR filter

Definition at line 102 of file filter.cpp.

#### 4.9.3.4 `get_taps()`

```
int filter::get_taps (
    float * taps,
    size_t buf_size )
```

Getting the filter taps.

##### Parameters

out	<i>taps</i>	The filter taps (1-D array)
in	<i>buf_size</i>	The size of the filter taps (this should be the same as <code>tap_size</code> passed in constructor)

**Returns**

A flag indicating the success of getting the filter taps

Definition at line 84 of file filter.cpp.

**4.9.3.5 set\_taps()**

```
int filter::set_taps (
    const float * taps,
    size_t buf_size )
```

Setting the filter taps.

**Parameters**

in	<i>taps</i>	The filter taps (an 1-D array)
in	<i>buf_size</i>	The size of the filter taps (this should be the same as tap_size passed in constructor)

**Returns**

A flag indicating the success of setting the filter taps

Definition at line 68 of file filter.cpp.

The documentation for this class was generated from the following files:

- [filter.hpp](#)
- [filter.cpp](#)

**4.10 fir\_formii Class Reference**

Filter Class.

```
#include <fir_formii.h>
```

## Public Member Functions

- [fir\\_formii](#) (float \*taps, size\_t tap\_size)  
*Filter constructor.*
- [~fir\\_formii](#) ()  
*Filter destructor.*
- void [set\\_taps](#) (const float \*taps, size\_t tap\_size)  
*Setting the filter taps.*
- void [get\\_taps](#) (float \*taps, size\_t &tap\_size)  
*Getting the filter taps.*
- void [process](#) (const float \*data\_in, float \*data\_out, size\_t num\_samp)  
*Getting the output of this FIR filter by performing frame-based convolution.*
- size\_t [get\\_size](#) ()  
*Getting the number of taps of this FIR filter.*

### 4.10.1 Detailed Description

Filter Class.

This filter class implements the FIR filter

Definition at line 41 of file fir\_formii.h.

### 4.10.2 Constructor & Destructor Documentation

#### 4.10.2.1 fir\_formii()

```
fir_formii::fir_formii (
    float * taps,
    size_t tap_size ) [explicit]
```

Filter constructor.

#### Parameters

in	<i>taps</i>	The filter taps for this FIR filter
in	<i>tap_size</i>	The number of taps of this FIR filter
in	<i>cir_buf</i>	The circular buffer for this FIR filter to perform frame-based convolution
in	<i>max_buf_size</i>	The maximum size of circular buffer you need to specify if there is no circular buffer given in <i>cir_buf</i>

Definition at line 29 of file fir\_formii.cpp.

### 4.10.3 Member Function Documentation

#### 4.10.3.1 `get_size()`

```
size_t fir_formii::get_size ( )
```

Getting the number of taps of this FIR filter.

##### Returns

The number of taps of this FIR filter

Definition at line 97 of file `fir_formii.cpp`.

#### 4.10.3.2 `get_taps()`

```
void fir_formii::get_taps (
    float * taps,
    size_t & tap_size )
```

Getting the filter taps.

##### Parameters

out	<i>taps</i>	The filter taps (1-D array)
in	<i>buf_size</i>	The size of the filter taps (this should be the same as <code>tap_size</code> passed in constructor)

##### Returns

A flag indicating the success of getting the filter taps

Definition at line 65 of file `fir_formii.cpp`.

#### 4.10.3.3 `process()`

```
void fir_formii::process (
    const float * data_in,
    float * data_out,
    size_t num_samp )
```

Getting the output of this FIR filter by performing frame-based convolution.

**Parameters**

in	<i>data_in</i>	The input signal
out	<i>data_out</i>	The output signal
	<i>num_samp</i>	The size of input and output signal ( <i>data_in</i> and <i>data_out</i> should have the same size)

Definition at line 73 of file `fir_formii.cpp`.

**4.10.3.4 set\_taps()**

```
void fir_formii::set_taps (
    const float * taps,
    size_t tap_size )
```

Setting the filter taps.

**Parameters**

in	<i>taps</i>	The filter taps (an 1-D array)
in	<i>buf_size</i>	The size of the filter taps (this should be the same as <i>tap_size</i> passed in constructor)

**Returns**

A flag indicating the success of setting the filter taps

Definition at line 49 of file `fir_formii.cpp`.

The documentation for this class was generated from the following files:

- [fir\\_formii.h](#)
- [fir\\_formii.cpp](#)

## 4.11 freping Class Reference

Freping Class.

```
#include <freping.hpp>
```

## Public Member Functions

- `freping` (int `allpass_chain_len`, int `frame_size`, float `alpha`, float `*window`, int `freping_on_off`)  
*Freping constructor.*
- `~freping` ()  
*Freping destructor.*
- void `get_params` (float &`alpha`, int &`freping_on_off`)  
*Get the parameter alpha.*
- void `set_params` (float `alpha`, int `freping_on_off`)  
*Set the parameter alpha.*
- void `freping_proc` (float `*in`, float `*out`)

## Protected Member Functions

- void `overlap_add` (const float `*in`, float `*out`)
- void `allpass_chain` (float `*in`, float `*out`, float `alpha_`, float `coeff_`)  
*The all-pass chain which has the length of `allpass_chain_len`.*
- void `windowing` (const float `*in`, float `*out`)

### 4.11.1 Detailed Description

Freping Class.

This freping class provides frequency warping feature in real-time. Please refer to the following paper for details. Ching-Hua Lee, Kuan-Lin Chen, fred harris, Bhaskar D. Rao, and Harinath Garudadri, "On mitigating acoustic feedback in hearing aids with frequency warping by all-pass networks," in Annual Conference of the International Speech Communication Association (Interspeech), 2019.

Definition at line 44 of file `freping.hpp`.

### 4.11.2 Constructor & Destructor Documentation

#### 4.11.2.1 `freping()`

```
freping::freping (
    int allpass_chain_len,
    int frame_size,
    float alpha,
    float * window,
    int freping_on_off ) [explicit]
```

Freping constructor.



## Parameters

in	<i>allpass_chain_len</i>	The length of the all-pass chain
in	<i>frame_size</i>	The frame size of the caller (the real-time system)
in	<i>alpha</i>	The parameter to control the degree of frequency warping ( $1.0 \geq \alpha \geq -1.0$ )

As long as this is true we don't need circular buf function

Definition at line 29 of file freping.cpp.

## 4.11.3 Member Function Documentation

## 4.11.3.1 allpass\_chain()

```
void freping::allpass_chain (
    float * in,
    float * out,
    float alpha_,
    float coeff_ ) [protected]
```

The all-pass chain which has the length of allpass\_chain\_len.

## Parameters

in	<i>in</i>	The input frame which has the same size of the output frame (e.g., 128 samples)
out	<i>out</i>	The output frame (e.g., 128 samples)
in	<i>alpha</i> ↔ —	The parameter to control the degree of frequency warping ( $1.0 \geq \alpha \geq -1.0$ )
in	<i>coeff</i> ↔ —	a coefficient for the second IIR filter in the all-pass chain (a function of alpha_)

the first stage of the all-pass chain

the second stage of the all-pass chain

Pointer swaping cost less than copying memory

the remaining stages of the all-pass chain

Pointer swaping cost less than copying memory

Definition at line 78 of file freping.cpp.

#### 4.11.3.2 freping\_proc()

```
void freping::freping_proc (
    float * in,
    float * out )
```

Get the output signal of freping, the output signal is warped according to the alpha parameter

##### Parameters

in	<i>in</i>	The input frame
out	<i>out</i>	The output frame

in\_buf\_->set(in,frame\_size\_); // Do we really need if frame\_size is already a multiple of all\_pass\_len m\_ = ++run\_k\_; ///  
Why is this useful?

in\_buf\_->get(x\_buf\_,buf\_len\_); this->overlap\_framing(x\_buf\_,overlapped\_frame\_);

Pointer swaping cost less than copying memory

Definition at line 157 of file freping.cpp.

#### 4.11.3.3 get\_params()

```
void freping::get_params (
    float & alpha,
    int & freping_on_off )
```

Get the parameter alpha.

##### Parameters

out	<i>alpha</i>	The parameter to control the degree of frequency warping ( $1.0 \geq \alpha \geq -1.0$ )
-----	--------------	--

Definition at line 133 of file freping.cpp.

#### 4.11.3.4 overlap\_add()

```
void freping::overlap_add (
    const float * in,
    float * out ) [protected]
```

Overlap-add method

**Parameters**

in	<i>in</i>	The overlapped frame which has the twice size of the output frame (e.g., 128 samples)
out	<i>out</i>	The recovered frame (e.g., 64 samples)

This should do the same thing

Definition at line 119 of file freping.cpp.

**4.11.3.5 set\_params()**

```
void freping::set_params (
    float alpha,
    int freping_on_off )
```

Set the parameter alpha.

**Parameters**

in	<i>alpha</i>	The parameter to control the degree of frequency warping ( $1.0 \geq \alpha \geq -1.0$ )
----	--------------	--

Definition at line 139 of file freping.cpp.

**4.11.3.6 windowing()**

```
void freping::windowing (
    const float * in,
    float * out ) [protected]
```

Applying a window function such as Hamming window to the input frame

**Parameters**

in	<i>in</i>	The input frame which has the same size of the output frame (e.g., 128 samples)
out	<i>out</i>	The output frame (e.g., 128 samples)

Definition at line 72 of file freping.cpp.

The documentation for this class was generated from the following files:

- [freping.hpp](#)
- [freping.cpp](#)

## 4.12 GarbageCollector Class Reference

### Public Member Functions

- `template<typename T >`  
`void add (const std::shared_ptr< T > &object)`
- `void release ()`

### Public Attributes

- `std::vector< std::shared_ptr< void > > pool`
- `std::mutex m`

### 4.12.1 Detailed Description

Definition at line 36 of file `GarbageCollector.hpp`.

The documentation for this class was generated from the following file:

- `GarbageCollector.hpp`

## 4.13 libModule Class Reference

A template for library modules.

### Public Member Functions

- `libModule (...)`  
*`libModule` constructor*
- `~libModule ()`  
*`libModule` destructor*
- `void set\_param (...)`  
*Setting `libModule` parameters.*
- `void get\_param (...)`  
*Getting `libModule` parameters.*
- `void process (...)`  
*Real-time processing inside `libModule`.*

### 4.13.1 Detailed Description

A template for library modules.

Definition at line 16 of file example\_template.cpp.

The documentation for this class was generated from the following file:

- example\_template.cpp

## 4.14 noise\_management Class Reference

Noise Management Class.

```
#include <noise_management.hpp>
```

### Public Member Functions

- [noise\\_management](#) (int ntype, int stype, float sparam, float fsamp)  
*Noise management constructor.*
- [~noise\\_management](#) ()  
*Noise management destructor.*
- void [set\\_param](#) (int ntype, int stype, float sparam)  
*Setting all parameters in noise management.*
- void [get\\_param](#) (int &ntype, int &stype, float &sparam)  
*Getting all parameters in noise management.*
- void [speech\\_enhancement](#) (float \*data\_in, size\_t in\_len, float \*data\_out)  
*A function to perform speech enhancement.*

### 4.14.1 Detailed Description

Noise Management Class.

Speech enhancement using peak and valley detection, noise estimation and spectral subtraction

Definition at line 16 of file noise\_management.hpp.

### 4.14.2 Constructor & Destructor Documentation

#### 4.14.2.1 noise\_management()

```
noise_management::noise_management (
    int ntype,
    int stype,
    float sparam,
    float fsamp ) [explicit]
```

Noise management constructor.

**Parameters**

in	<i>ntype</i>	The type of noise estimation, 1: using limits on change (ref: Arslan et al.), 2: using the weighted averaging of Hirsch and Ehrlicher, 3: using MCRA of Cohen and Berdugo
in	<i>stype</i>	The type of spectral subtraction, 0: normal, 1: oversubtraction
in	<i>sparam</i>	A parameter for spectral subtraction
in	<i>fsamp</i>	The sampling rate

Definition at line 5 of file noise\_management.cpp.

**4.14.3 Member Function Documentation****4.14.3.1 get\_param()**

```
void noise_management::get_param (
    int & ntype,
    int & stype,
    float & sparam )
```

Getting all parameters in noise management.

**Parameters**

in	<i>ntype</i>	See constructor
in	<i>stype</i>	See constructor
in	<i>sparam</i>	See constructor

Definition at line 86 of file noise\_management.cpp.

**4.14.3.2 set\_param()**

```
void noise_management::set_param (
    int ntype,
    int stype,
    float sparam )
```

Setting all parameters in noise management.

**Parameters**

in	<i>ntype</i>	See constructor
in	<i>stype</i>	See constructor
in	<i>sparam</i>	See constructor

Definition at line 77 of file noise\_management.cpp.

#### 4.14.3.3 speech\_enhancement()

```
void noise_management::speech_enhancement (
    float * data_in,
    size_t in_len,
    float * data_out )
```

A function to perform speech enhancement.

##### Parameters

in	<i>data_in</i>	The input signal
in	<i>in_len</i>	Length of the input signal
out	<i>data_out</i>	The output signal, i.e., the enhanced speech signal

Definition at line 95 of file noise\_management.cpp.

The documentation for this class was generated from the following files:

- noise\_management.hpp
- noise\_management.cpp

## 4.15 peak\_detect Class Reference

Peak Detector Class.

```
#include <peak_detect.hpp>
```

### Public Member Functions

- [peak\\_detect](#) (float fsamp, float attack\_time, float release\_time)  
*Peak detector constructor.*
- [~peak\\_detect](#) ()  
*Peak detector destructor.*
- void [set\\_param](#) (float attack\_time, float release\_time)  
*Setting the parameters for peak detector (to have alpha and beta)*
- void [get\\_param](#) (float &attack\_time, float &release\_time)  
*Getting the parameters from peak detector (in terms of attach time and release time)*
- void [get\\_spl](#) (float \*data\_in, size\_t in\_len, float \*pdb\_out)  
*Getting the output from the peak detector in SPL.*

### 4.15.1 Detailed Description

Peak Detector Class.

This peak detector implements the algorithm according to Eq. (8.1) in [James M. Kates, Digital hearing aids, Plural publishing, 2008].

Definition at line 21 of file `peak_detect.hpp`.

### 4.15.2 Constructor & Destructor Documentation

#### 4.15.2.1 `peak_detect()`

```
peak_detect::peak_detect (
    float fsamp,
    float attack_time,
    float release_time ) [explicit]
```

Peak detector constructor.

#### Parameters

in	<i>fsamp</i>	The sampling rate of the system
in	<i>attack_time</i>	Attack time in milliseconds
in	<i>release_time</i>	Release time in milliseconds

Definition at line 4 of file `peak_detect.cpp`.

### 4.15.3 Member Function Documentation

#### 4.15.3.1 `get_param()`

```
void peak_detect::get_param (
    float & attack_time,
    float & release_time )
```

Getting the parameters from peak detector (in terms of attach time and release time)



**Parameters**

out	<i>attack_time</i>	attack_time Attack time in milliseconds
out	<i>release_time</i>	release_time Release time in milliseconds

Definition at line 37 of file peak\_detect.cpp.

**4.15.3.2 get\_spl()**

```
void peak_detect::get_spl (
    float * data_in,
    size_t in_len,
    float * pdb_out )
```

Getting the output from the peak detector in SPL.

**Parameters**

in	<i>data_in</i>	The input signal
in	<i>in_len</i>	The size of the input signal
out	<i>pdb_out</i>	The output of peak detector in SPL

Definition at line 45 of file peak\_detect.cpp.

**4.15.3.3 set\_param()**

```
void peak_detect::set_param (
    float attack_time,
    float release_time )
```

Setting the parameters for peak detector (to have alpha and beta)

**Parameters**

in	<i>attack_time</i>	Attack time in milliseconds
in	<i>release_time</i>	Release time in milliseconds

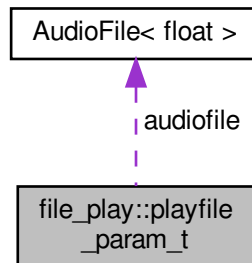
Definition at line 19 of file peak\_detect.cpp.

The documentation for this class was generated from the following files:

- peak\_detect.hpp
- peak\_detect.cpp

## 4.16 file\_play::playfile\_param\_t Struct Reference

Collaboration diagram for file\_play::playfile\_param\_t:



### Public Attributes

- int **reset**
- int **play**
- int **repeat**
- int **current\_position\_l**
- int **current\_position\_r**
- int **numSamples**
- bool **finish**
- int **mono**
- std::string **filename**
- [AudioFile< float >](#) \* **audiofile**

### 4.16.1 Detailed Description

Definition at line 14 of file `playfile.h`.

The documentation for this struct was generated from the following file:

- `playfile.h`

## 4.17 polyphase\_hb\_downsampler Class Reference

### Public Member Functions

- **polyphase\_hb\_downsampler** (float \*filter\_taps, size\_t num\_taps, size\_t max\_frame\_size)
- void **process** (float \*in, float \*out, size\_t frame\_size)

### 4.17.1 Detailed Description

Definition at line 13 of file polyphase\_hb\_downsampler.h.

The documentation for this class was generated from the following files:

- polyphase\_hb\_downsampler.h
- polyphase\_hb\_downsampler.cpp

## 4.18 polyphase\_hb\_upsampler Class Reference

### Public Member Functions

- **polyphase\_hb\_upsampler** (float \*filter\_taps, size\_t num\_taps, size\_t max\_frame\_size)
- void **process** (float \*in, float \*out, size\_t frame\_size)

### 4.18.1 Detailed Description

Definition at line 12 of file polyphase\_hb\_upsampler.h.

The documentation for this class was generated from the following files:

- polyphase\_hb\_upsampler.h
- polyphase\_hb\_upsampler.cpp

## 4.19 resample Class Reference

Resample Class.

```
#include <resample.hpp>
```

### Public Member Functions

- **resample** (float \*taps, size\_t tap\_size, size\_t max\_in\_buf\_size, int interp\_factor, int deci\_factor)  
*Resample constructor.*
- **~resample** ()  
*Resample destructor.*
- void **resamp** (float \*data\_in, size\_t in\_size, float \*data\_out, size\_t \*out\_size)  
*Getting the resampled signal.*

### 4.19.1 Detailed Description

Resample Class.

Resampling class implements L/M-fold resampling

Definition at line 15 of file resample.hpp.

### 4.19.2 Constructor & Destructor Documentation

#### 4.19.2.1 resample()

```
resample::resample (
    float * taps,
    size_t tap_size,
    size_t max_in_buf_size,
    int interp_factor,
    int deci_factor ) [explicit]
```

Resample constructor.

#### Parameters

in	<i>taps</i>	The filter taps of the lowpass filter (to reject images and prevent aliasing)
in	<i>tap_size</i>	The number of taps of the lowpass filter
in	<i>max_in_buf_size</i>	The maximum input buffer size
in	<i>interp_factor</i>	The interpolation factor L (to implement L-fold expander)
in	<i>deci_factor</i>	The decimation factor M (to implement M-fold decimator)

Definition at line 7 of file resample.cpp.

### 4.19.3 Member Function Documentation

#### 4.19.3.1 resamp()

```
void resample::resamp (
    float * data_in,
    size_t in_size,
    float * data_out,
    size_t * out_size )
```

Getting the resampled signal.

## Parameters

in	<i>data_in</i>	The signal in original sampling rate
in	<i>in_size</i>	The size of the original signal
out	<i>data_out</i>	The resampled signal
out	<i>out_size</i>	The size of the resampled signal

Definition at line 23 of file resample.cpp.

The documentation for this class was generated from the following files:

- resample.hpp
- resample.cpp

## 4.20 rk\_sema Struct Reference

### Public Attributes

- sem\_t **sem**

#### 4.20.1 Detailed Description

Definition at line 8 of file sema.hpp.

The documentation for this struct was generated from the following file:

- sema.hpp

## 4.21 tenband\_filterbank Class Reference

The Ten Band Filter bank is a 10 Band Multirate Filter Bank with its center frequencies located at 250Hz, 500Hz, 750Hz, 1kHz, 1.5kHz, 2kHz, 3kHz, 4kHz, 6kHz, 8kHz.

```
#include <tenband_filterbank.h>
```

### Public Member Functions

- [tenband\\_filterbank](#) (size\_t max\_frame\_size, bool aligned)  
*The constructor for the Ten Band Filter bank.*
- [~tenband\\_filterbank](#) ()  
*The destructor for the Ten Band Filter bank.*
- void [process](#) (float \*in, float \*\*out, size\_t frame\_size)  
*This function decomposes the incoming frame of data into the 10 subbands.*
- void [set](#) (bool aligned)  
*The compensation for the various group delay can be toggled on and off live.*
- void [get](#) (bool &aligned)  
*Used to read the current status of the aligned variable.*

### 4.21.1 Detailed Description

The Ten Band Filter bank is a 10 Band Multirate Filter Bank with its center frequencies located at 250Hz, 500Hz, 750Hz, 1kHz, 1.5kHz, 2kHz, 3kHz, 4kHz, 6kHz, 8kHz.

Definition at line 44 of file `tenband_filterbank.h`.

### 4.21.2 Constructor & Destructor Documentation

#### 4.21.2.1 `tenband_filterbank()`

```
tenband_filterbank::tenband_filterbank (
    size_t max_frame_size,
    bool aligned )
```

The constructor for the Ten Band Filter bank.

##### Parameters

<i>max_frame_size</i>	[in] - The maximum size of buffer that would need to be processed at any time step.
<i>aligned</i>	[in] - Enabling the delay blocks to compensate for the various group delays.

Definition at line 31 of file `tenband_filterbank.cpp`.

### 4.21.3 Member Function Documentation

#### 4.21.3.1 `get()`

```
void tenband_filterbank::get (
    bool & aligned )
```

Used to read the current status of the aligned variable.

##### Parameters

out	<i>aligned</i>	Grabbing the current status of the aligned variable.
-----	----------------	--

Definition at line 200 of file `tenband_filterbank.cpp`.

## 4.21.3.2 process()

```
void tenband_filterbank::process (
    float * in,
    float ** out,
    size_t frame_size )
```

This function decomposes the incoming frame of data into the 10 subbands.

## Parameters

in	<i>in</i>	An array of data the size of frame_size.
out	<i>out</i>	A 2D 10 by frame_size array of data for the 10 subbands
in	<i>frame_size</i>	The amount of data contained in the incoming array.

Definition at line 150 of file tenband\_filterbank.cpp.

## 4.21.3.3 set()

```
void tenband_filterbank::set (
    bool aligned )
```

The compensation for the various group delay can be toggled on and off live.

## Parameters

in	<i>aligned</i>	Enabling the delay blocks to compensate for the various group delays.
----	----------------	---

Definition at line 197 of file tenband\_filterbank.cpp.

The documentation for this class was generated from the following files:

- tenband\_filterbank.h
- tenband\_filterbank.cpp

## 4.22 wdrc Class Reference

Wide Dynamic Range Compression (WDRC) Class.

```
#include <wdrc.hpp>
```

## Public Member Functions

- [wdrc](#) (float gain50, float gain80, float knee\_low, float mpo\_limit)  
*wdrc constructor*
- [~wdrc](#) ()  
*wdrc destructor*
- void [set\\_param](#) (float gain50, float gain80, float knee\_low, float mpo\_limit)  
*Setting WDRC parameters.*
- void [get\\_param](#) (float &gain50, float &gain80, float &knee\_low, float &mpo\_limit)  
*Getting WDRC parameters.*
- void [process](#) (float \*input, float \*pdb, size\_t in\_len, float \*output)  
*Perform WDRC.*

### 4.22.1 Detailed Description

Wide Dynamic Range Compression (WDRC) Class.

Applying WDRC to a subband signal from an analysis filterbank

Definition at line 18 of file wdrc.hpp.

### 4.22.2 Constructor & Destructor Documentation

#### 4.22.2.1 wdrc()

```
wdrc::wdrc (
    float gain50,
    float gain80,
    float knee_low,
    float mpo_limit ) [explicit]
```

wdrc constructor

#### Parameters

in	<i>gain50</i>	Gain at 50 dB SPL of input level
in	<i>gain80</i>	Gain at 80 dB SPL of input level
in	<i>knee_low</i>	Lower knee-point
in	<i>mpo_limit</i>	Maximum power output (MPO)

Definition at line 3 of file wdrc.cpp.



### 4.22.3 Member Function Documentation

#### 4.22.3.1 get\_param()

```
void wdrc::get_param (
    float & gain50,
    float & gain80,
    float & knee_low,
    float & mpo_limit )
```

Getting WDRC parameters.

##### Parameters

out	<i>gain50</i>	Gain at 50 dB SPL of input level
out	<i>gain80</i>	Gain at 80 dB SPL of input level
out	<i>knee_low</i>	Lower knee-point
out	<i>mpo_limit</i>	MPO

Definition at line 43 of file wdrc.cpp.

#### 4.22.3.2 process()

```
void wdrc::process (
    float * input,
    float * pdb,
    size_t in_len,
    float * output )
```

Perform WDRC.

The peak detector output in dB SPL is needed as one of the inputs. The gain at 50 and 80 dB SPL is specified for the frequency sub-band, along with the lower and upper kneepoints in dB SPL. The compressor is linear below the lower kneepoint and applies compression limiting above the upper kneepoint

##### Parameters

in	<i>input</i>	The input signal (1-D array)
in	<i>pdb</i>	The output from the peak detector in SPL, i.e., the output from <a href="#">get_spl</a> member function in <a href="#">peak_detect</a> class
in	<i>in_len</i>	Length of the input signal
out	<i>output</i>	Pointer to a signal (1-D array) where the compressed output of the subband signal will be written, i.e., the output of WDRC

See also

[peak\\_detect](#)

Definition at line 52 of file wdrc.cpp.

#### 4.22.3.3 set\_param()

```
void wdrc::set_param (
    float gain50,
    float gain80,
    float knee_low,
    float mpo_limit )
```

Setting WDRC parameters.

##### Parameters

in	<i>gain50</i>	Gain at 50 dB SPL of input level
in	<i>gain80</i>	Gain at 80 dB SPL of input level
in	<i>knee_low</i>	Lower knee-point
in	<i>mpo_limit</i>	MPO

Definition at line 25 of file wdrc.cpp.

The documentation for this class was generated from the following files:

- wdrc.hpp
- wdrc.cpp

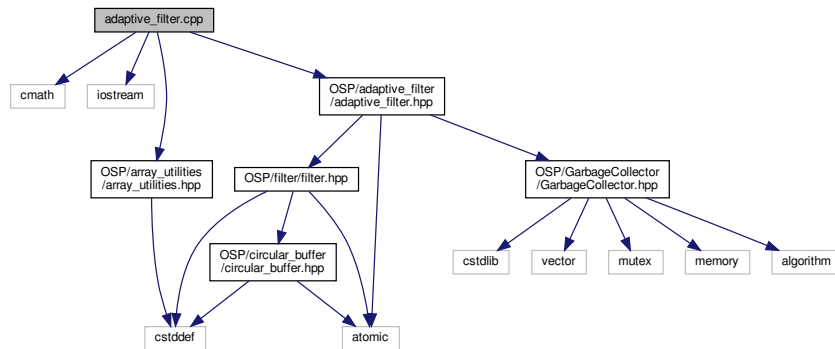
## Chapter 5

# File Documentation

### 5.1 adaptive\_filter.cpp File Reference

```
#include <cmath>
#include <iostream>
#include <OSP/array_utilities/array_utilities.hpp>
#include <OSP/adaptive_filter/adaptive_filter.hpp>
```

Include dependency graph for adaptive\_filter.cpp:



#### 5.1.1 Detailed Description

Author

Open Speech Platform (OSP) Team, UCSD

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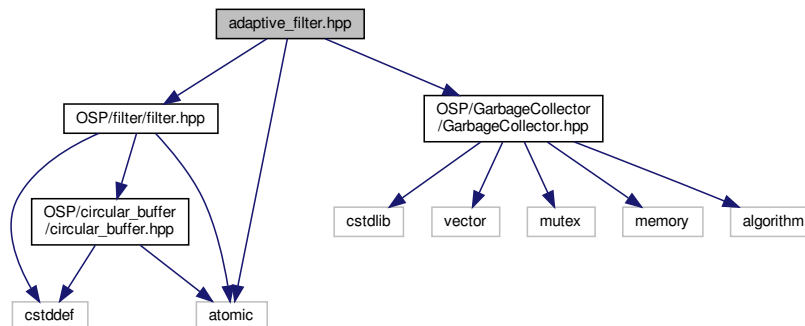
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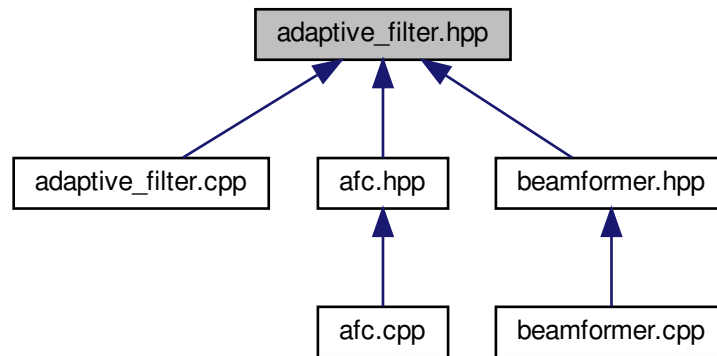
## 5.2 adaptive\_filter.hpp File Reference

```
#include <OSP/filter/filter.hpp>
#include <atomic>
#include <OSP/GarbageCollector/GarbageCollector.hpp>
```

Include dependency graph for adaptive\_filter.hpp:



This graph shows which files directly or indirectly include this file:



## Classes

- class [adaptive\\_filter](#)  
*Adaptive Filter Class.*

### 5.2.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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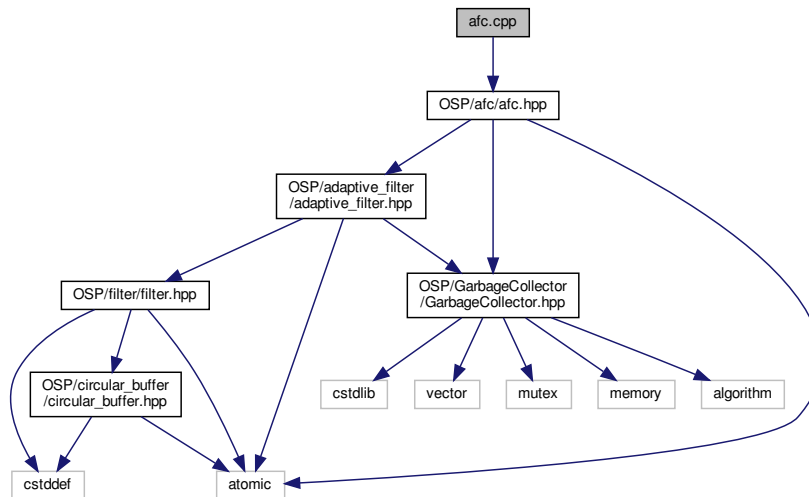
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### 5.3 afc.cpp File Reference

```
#include <OSP/afc/afc.hpp>
```

Include dependency graph for afc.cpp:



#### 5.3.1 Detailed Description

##### Author

Open Speech Platform (OSP) Team, UCSD

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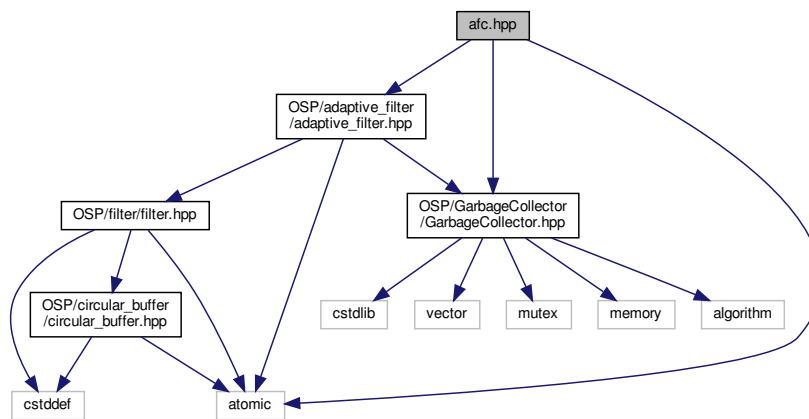
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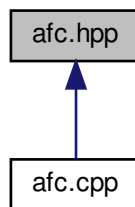
## 5.4 afc.hpp File Reference

```
#include <OSP/adaptive_filter/adaptive_filter.hpp>
#include <atomic>
#include <OSP/GarbageCollector/GarbageCollector.hpp>
```

Include dependency graph for afc.hpp:



This graph shows which files directly or indirectly include this file:



### Classes

- class [afc](#)  
*Adaptive Feedback Cancellation (AFC) Class.*

### Macros

- `#define MAX_DELAY_LEN 256`

### 5.4.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.5 `afc_init_filter.h` File Reference

#### Macros

- `#define AFC_INIT_FILTER_SIZE 160`

#### Variables

- float `afc_init_filter` [AFC\_INIT\_FILTER\_SIZE]

### 5.5.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD



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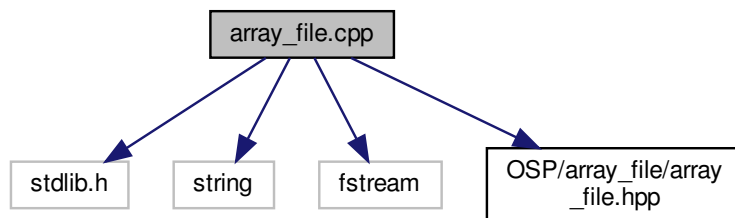
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## 5.6 array\_file.cpp File Reference

```
#include <stdlib.h>
#include <string>
#include <fstream>
#include <OSP/array_file/array_file.hpp>
```

Include dependency graph for array\_file.cpp:



### 5.6.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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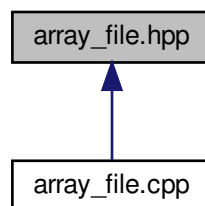
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## 5.7 array\_file.hpp File Reference

This graph shows which files directly or indirectly include this file:



## Classes

- class [array\\_file](#)  
*Array File Class.*

### 5.7.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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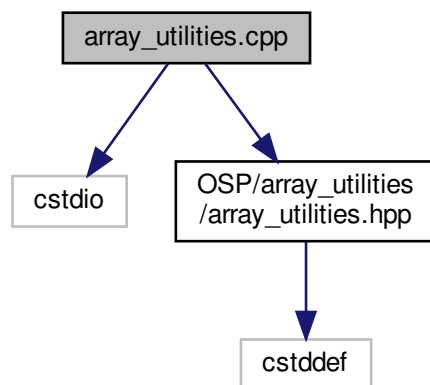
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## 5.8 array\_utilities.cpp File Reference

```
#include <stdio>
#include <OSP/array_utilities/array_utilities.hpp>
```

Include dependency graph for array\_utilities.cpp:



### Functions

- void `array_flip` (float \*arr, size\_t len)  
*Function to reverse an array.*

- float [array\\_sum](#) (const float \*arr, size\_t len)  
*Function to calculate the sum of an array.*
- float [array\\_dot\\_product](#) (const float \*in1, const float \*in2, size\_t len)  
*Function to calculate the dot-product of two 1-D vectors/arrays.*
- void [array\\_right\\_shift](#) (float \*arr, size\_t len)  
*Function to right shift an array by one place. Left most value will be replaced by zero.*
- void [array\\_multiply\\_const](#) (float \*arr, float constant, size\_t len)  
*Function to multiply each element of an array by a scalar constant.*
- void [array\\_add\\_const](#) (float \*arr, float constant, size\_t len)  
*Function to add a scalar constant to each element of an array.*
- void [array\\_add\\_array](#) (float \*in1, const float \*in2, size\_t len)  
*Function to do element wise addition of two arrays.*
- void [array\\_subtract\\_array](#) (float \*in1, const float \*in2, size\_t len)  
*Function to do element wise subtraction of two arrays.*
- void [array\\_element\\_multiply\\_array](#) (float \*in1, const float \*in2, size\_t len)  
*Function to do element wise multiplication of two arrays.*
- void [array\\_element\\_divide\\_array](#) (float \*in1, const float \*in2, size\_t len)  
*Function to do element wise division of two arrays.*
- float [array\\_min](#) (const float \*arr, size\_t len)  
*Function to return the minimum of the elements of an array.*
- float [array\\_mean](#) (float \*arr, size\_t len)  
*Function to calculate the mean of the elements of an array.*
- void [array\\_square](#) (const float \*in, float \*out, size\_t len)  
*Function to populate the output array with square of the elements of an input array.*
- float [array\\_mean\\_square](#) (const float \*arr, size\_t len)  
*Function to calculate the mean square of the elements of an array.*
- void [array\\_print](#) (const char \*str, float \*arr, size\_t len)  
*Function to print an array for debugging.*

### 5.8.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.8.2 Function Documentation

### 5.8.2.1 array\_add\_array()

```
void array_add_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise addition of two arrays.

#### Parameters

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

#### Warning

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 105 of file array\_utilities.cpp.

### 5.8.2.2 array\_add\_const()

```
void array_add_const (
    float * arr,
    float constant,
    size_t len )
```

Function to add a scalar constant to each element of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>constant</i>	The constant scalar adder
<i>len</i>	Length of the array

Definition at line 96 of file array\_utilities.cpp.

**5.8.2.3 array\_dot\_product()**

```
float array_dot_product (
    const float * in1,
    const float * in2,
    size_t len )
```

Function to calculate the dot-product of two 1-D vectors/arrays.

**Parameters**

<i>in1</i>	Pointer to the first vector
<i>in2</i>	Pointer to the second vector
<i>len</i>	Length of the vectors

**Returns**

Dot product (inner product) of the two vectors

**Warning**

Assumes both the vectors are of same length and takes only one length parameter

Definition at line 67 of file array\_utilities.cpp.

**5.8.2.4 array\_element\_divide\_array()**

```
void array_element_divide_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise division of two arrays.

**Parameters**

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

**Warning**

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 132 of file array\_utilities.cpp.

**5.8.2.5 array\_element\_multiply\_array()**

```
void array_element_multiply_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise multiplication of two arrays.

**Parameters**

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

**Warning**

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 123 of file array\_utilities.cpp.

**5.8.2.6 array\_flip()**

```
void array_flip (
    float * arr,
    size_t len )
```

Function to reverse an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 31 of file array\_utilities.cpp.

**5.8.2.7 array\_mean()**

```
float array_mean (
    float * arr,
    size_t len )
```

Function to calculate the mean of the elements of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

**Returns**

Mean of the array elements

Definition at line 156 of file array\_utilities.cpp.

**5.8.2.8 array\_mean\_square()**

```
float array_mean_square (
    const float * arr,
    size_t len )
```

Function to calculate the mean square of the elements of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array



**Returns**

Mean square of the array elements

Definition at line 172 of file array\_utilities.cpp.

**5.8.2.9 array\_min()**

```
float array_min (
    const float * arr,
    size_t len )
```

Function to return the minimum of the elements of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

**Returns**

Minimum of the array elements

Definition at line 141 of file array\_utilities.cpp.

**5.8.2.10 array\_multiply\_const()**

```
void array_multiply_const (
    float * arr,
    float constant,
    size_t len )
```

Function to multiply each element of an array by a scalar constant.

**Parameters**

<i>arr</i>	Pointer to the array
<i>constant</i>	The constant scalar multiplier
<i>len</i>	Length of the array

Definition at line 87 of file array\_utilities.cpp.

### 5.8.2.11 array\_print()

```
void array_print (
    const char * str,
    float * arr,
    size_t len )
```

Function to print an array for debugging.

#### Parameters

<i>str</i>	String to use for debugging
<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 177 of file array\_utilities.cpp.

### 5.8.2.12 array\_right\_shift()

```
void array_right_shift (
    float * arr,
    size_t len )
```

Function to right shift an array by one place. Left most value will be replaced by zero.

#### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 78 of file array\_utilities.cpp.

### 5.8.2.13 array\_square()

```
void array_square (
    const float * in,
    float * out,
    size_t len )
```

Function to populate the output array with square of the elements of an input array.

**Parameters**

<i>in</i>	Pointer to the input array
<i>out</i>	Pointer to the output array
<i>len</i>	Length of the arrays

**Warning**

Assumes that output array already has memory allocated to it

Definition at line 163 of file array\_utilities.cpp.

**5.8.2.14 array\_subtract\_array()**

```
void array_subtract_array (  
    float * in1,  
    const float * in2,  
    size_t len )
```

Function to do element wise subtraction of two arrays.

**Parameters**

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

**Warning**

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 114 of file array\_utilities.cpp.

**5.8.2.15 array\_sum()**

```
float array_sum (  
    const float * arr,  
    size_t len )
```

Function to calculate the sum of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

**Returns**

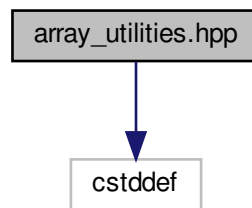
Sum of the array

Definition at line 47 of file array\_utilities.cpp.

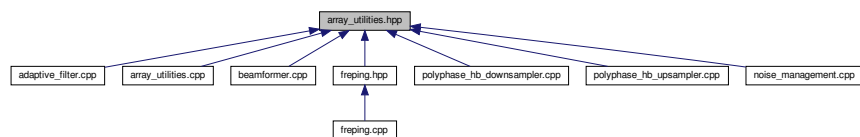
**5.9 array\_utilities.hpp File Reference**

```
#include <cstdint>
```

Include dependency graph for array\_utilities.hpp:



This graph shows which files directly or indirectly include this file:

**Functions**

- void [array\\_flip](#) (float \*arr, size\_t len)  
Function to reverse an array.
- float [array\\_sum](#) (const float \*arr, size\_t len)

*Function to calculate the sum of an array.*

- float [array\\_dot\\_product](#) (const float \*in1, const float \*in2, size\_t len)

*Function to calculate the dot-product of two 1-D vectors/arrays.*

- void [array\\_right\\_shift](#) (float \*arr, size\_t len)

*Function to right shift an array by one place. Left most value will be replaced by zero.*

- void [array\\_multiply\\_const](#) (float \*arr, float constant, size\_t len)

*Function to multiply each element of an array by a scalar constant.*

- void [array\\_add\\_const](#) (float \*arr, float constant, size\_t len)

*Function to add a scalar constant to each element of an array.*

- void [array\\_add\\_array](#) (float \*in1, const float \*in2, size\_t len)

*Function to do element wise addition of two arrays.*

- void [array\\_subtract\\_array](#) (float \*in1, const float \*in2, size\_t len)

*Function to do element wise subtraction of two arrays.*

- void [array\\_element\\_multiply\\_array](#) (float \*in1, const float \*in2, size\_t len)

*Function to do element wise multiplication of two arrays.*

- void [array\\_element\\_divide\\_array](#) (float \*in1, const float \*in2, size\_t len)

*Function to do element wise division of two arrays.*

- float [array\\_min](#) (const float \*arr, size\_t len)

*Function to return the minimum of the elements of an array.*

- float [array\\_mean](#) (float \*arr, size\_t len)

*Function to calculate the mean of the elements of an array.*

- void [array\\_square](#) (const float \*in, float \*out, size\_t len)

*Function to populate the output array with square of the elements of an input array.*

- float [array\\_mean\\_square](#) (const float \*arr, size\_t len)

*Function to calculate the mean square of the elements of an array.*

- void [array\\_print](#) (const char \*str, float \*arr, size\_t len)

*Function to print an array for debugging.*

### 5.9.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.9.2 Function Documentation

### 5.9.2.1 array\_add\_array()

```
void array_add_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise addition of two arrays.

#### Parameters

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

#### Warning

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 105 of file array\_utilities.cpp.

### 5.9.2.2 array\_add\_const()

```
void array_add_const (
    float * arr,
    float constant,
    size_t len )
```

Function to add a scalar constant to each element of an array.

#### Parameters

<i>arr</i>	Pointer to the array
<i>constant</i>	The constant scalar adder
<i>len</i>	Length of the array

Definition at line 96 of file array\_utilities.cpp.

### 5.9.2.3 array\_dot\_product()

```
float array_dot_product (
    const float * in1,
    const float * in2,
    size_t len )
```

Function to calculate the dot-product of two 1-D vectors/arrays.

#### Parameters

<i>in1</i>	Pointer to the first vector
<i>in2</i>	Pointer to the second vector
<i>len</i>	Length of the vectors

#### Returns

Dot product (inner product) of the two vectors

#### Warning

Assumes both the vectors are of same length and takes only one length parameter

Definition at line 67 of file array\_utilities.cpp.

### 5.9.2.4 array\_element\_divide\_array()

```
void array_element_divide_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise division of two arrays.

#### Parameters

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

#### Warning

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 132 of file array\_utilities.cpp.

#### 5.9.2.5 array\_element\_multiply\_array()

```
void array_element_multiply_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise multiplication of two arrays.

##### Parameters

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

##### Warning

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 123 of file array\_utilities.cpp.

#### 5.9.2.6 array\_flip()

```
void array_flip (
    float * arr,
    size_t len )
```

Function to reverse an array.

##### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 31 of file array\_utilities.cpp.



#### 5.9.2.7 array\_mean()

```
float array_mean (
    float * arr,
    size_t len )
```

Function to calculate the mean of the elements of an array.

##### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

##### Returns

Mean of the array elements

Definition at line 156 of file array\_utilities.cpp.

#### 5.9.2.8 array\_mean\_square()

```
float array_mean_square (
    const float * arr,
    size_t len )
```

Function to calculate the mean square of the elements of an array.

##### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

##### Returns

Mean square of the array elements

Definition at line 172 of file array\_utilities.cpp.

#### 5.9.2.9 array\_min()

```
float array_min (
    const float * arr,
    size_t len )
```

Function to return the minimum of the elements of an array.

**Parameters**

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

**Returns**

Minimum of the array elements

Definition at line 141 of file array\_utilities.cpp.

**5.9.2.10 array\_multiply\_const()**

```
void array_multiply_const (
    float * arr,
    float constant,
    size_t len )
```

Function to multiply each element of an array by a scalar constant.

**Parameters**

<i>arr</i>	Pointer to the array
<i>constant</i>	The constant scalar multiplier
<i>len</i>	Length of the array

Definition at line 87 of file array\_utilities.cpp.

**5.9.2.11 array\_print()**

```
void array_print (
    const char * str,
    float * arr,
    size_t len )
```

Function to print an array for debugging.

**Parameters**

<i>str</i>	String to use for debugging
<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 177 of file array\_utilities.cpp.

#### 5.9.2.12 array\_right\_shift()

```
void array_right_shift (
    float * arr,
    size_t len )
```

Function to right shift an array by one place. Left most value will be replaced by zero.

##### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

Definition at line 78 of file array\_utilities.cpp.

#### 5.9.2.13 array\_square()

```
void array_square (
    const float * in,
    float * out,
    size_t len )
```

Function to populate the output array with square of the elements of an input array.

##### Parameters

<i>in</i>	Pointer to the input array
<i>out</i>	Pointer to the output array
<i>len</i>	Length of the arrays

##### Warning

Assumes that output array already has memory allocated to it

Definition at line 163 of file array\_utilities.cpp.

#### 5.9.2.14 array\_subtract\_array()

```
void array_subtract_array (
    float * in1,
    const float * in2,
    size_t len )
```

Function to do element wise subtraction of two arrays.

##### Parameters

<i>in1</i>	Pointer to the first array
<i>in2</i>	Pointer to the second array
<i>len</i>	Length of the arrays

##### Warning

Assumes both the arrays are of same length and takes only one length parameter

Definition at line 114 of file array\_utilities.cpp.

#### 5.9.2.15 array\_sum()

```
float array_sum (
    const float * arr,
    size_t len )
```

Function to calculate the sum of an array.

##### Parameters

<i>arr</i>	Pointer to the array
<i>len</i>	Length of the array

##### Returns

Sum of the array

Definition at line 47 of file array\_utilities.cpp.

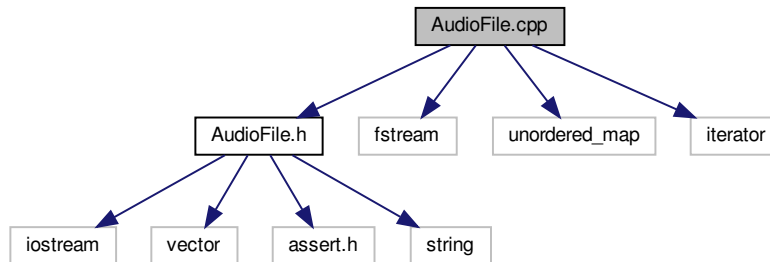
## 5.10 AudioFile.cpp File Reference

```
#include "AudioFile.h"
#include <fstream>
```

```
#include <unordered_map>
```

```
#include <iterator>
```

Include dependency graph for AudioFile.cpp:



## Variables

- `std::unordered_map< uint32_t, std::vector< uint8_t > > aiffSampleRateTable`

### 5.10.1 Detailed Description

#### Author

Adam Stark

#### Copyright

Copyright (C) 2017 Adam Stark

This file is part of the 'AudioFile' library

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### 5.10.2 Variable Documentation

### 5.10.2.1 aiffSampleRateTable

```
std::unordered_map<uint32_t, std::vector<uint8_t> > aiffSampleRateTable
```

**Initial value:**

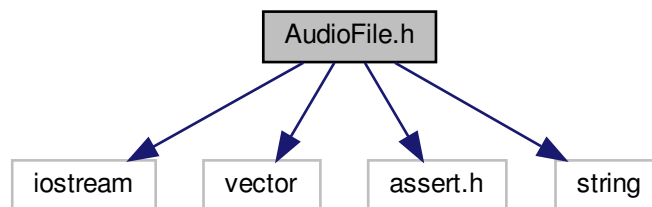
```
= {
    {8000, {64, 11, 250, 0, 0, 0, 0, 0, 0, 0}},
    {11025, {64, 12, 172, 68, 0, 0, 0, 0, 0, 0}},
    {16000, {64, 12, 250, 0, 0, 0, 0, 0, 0, 0}},
    {22050, {64, 13, 172, 68, 0, 0, 0, 0, 0, 0}},
    {32000, {64, 13, 250, 0, 0, 0, 0, 0, 0, 0}},
    {37800, {64, 14, 147, 168, 0, 0, 0, 0, 0, 0}},
    {44056, {64, 14, 172, 24, 0, 0, 0, 0, 0, 0}},
    {44100, {64, 14, 172, 68, 0, 0, 0, 0, 0, 0}},
    {47250, {64, 14, 184, 146, 0, 0, 0, 0, 0, 0}},
    {48000, {64, 14, 187, 128, 0, 0, 0, 0, 0, 0}},
    {50000, {64, 14, 195, 80, 0, 0, 0, 0, 0, 0}},
    {50400, {64, 14, 196, 224, 0, 0, 0, 0, 0, 0}},
    {88200, {64, 15, 172, 68, 0, 0, 0, 0, 0, 0}},
    {96000, {64, 15, 187, 128, 0, 0, 0, 0, 0, 0}},
    {176400, {64, 16, 172, 68, 0, 0, 0, 0, 0, 0}},
    {192000, {64, 16, 187, 128, 0, 0, 0, 0, 0, 0}},
    {352800, {64, 17, 172, 68, 0, 0, 0, 0, 0, 0}},
    {2822400, {64, 20, 172, 68, 0, 0, 0, 0, 0, 0}},
    {5644800, {64, 21, 172, 68, 0, 0, 0, 0, 0, 0}}
}
```

Definition at line 30 of file AudioFile.cpp.

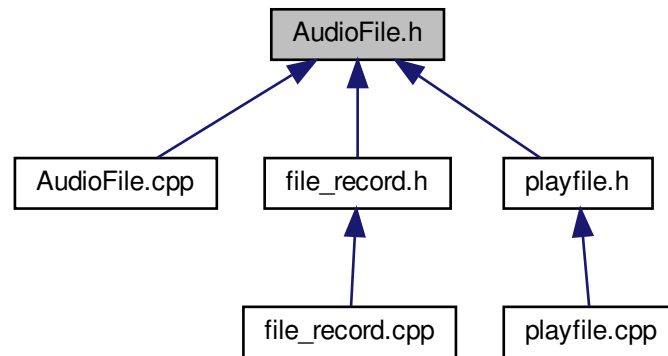
## 5.11 AudioFile.h File Reference

```
#include <iostream>
#include <vector>
#include <assert.h>
#include <string>
```

Include dependency graph for AudioFile.h:



This graph shows which files directly or indirectly include this file:



## Classes

- class [AudioFile< T >](#)

## Enumerations

- enum [AudioFileFormat](#) { **Error**, **NotLoaded**, **Wave**, **Aiff** }

### 5.11.1 Detailed Description

#### Author

Adam Stark

#### Copyright

Copyright (C) 2017 Adam Stark

This file is part of the '[AudioFile](#)' library

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## 5.11.2 Enumeration Type Documentation

### 5.11.2.1 AudioFileFormat

```
enum AudioFileFormat [strong]
```

The different types of audio file, plus some other types to indicate a failure to load a file, or that one hasn't been loaded yet

Definition at line 37 of file AudioFile.h.

## 5.12 bandlimited\_filter.h File Reference

### Macros

- `#define BANDLIMITED_FILTER_SIZE 3`

### Variables

- float **bandlimited\_filter** [BANDLIMITED\_FILTER\_SIZE]

### 5.12.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.12.2 Variable Documentation

### 5.12.2.1 bandlimited\_filter

```
float bandlimited_filter[BANDLIMITED_FILTER_SIZE]
```

**Initial value:**

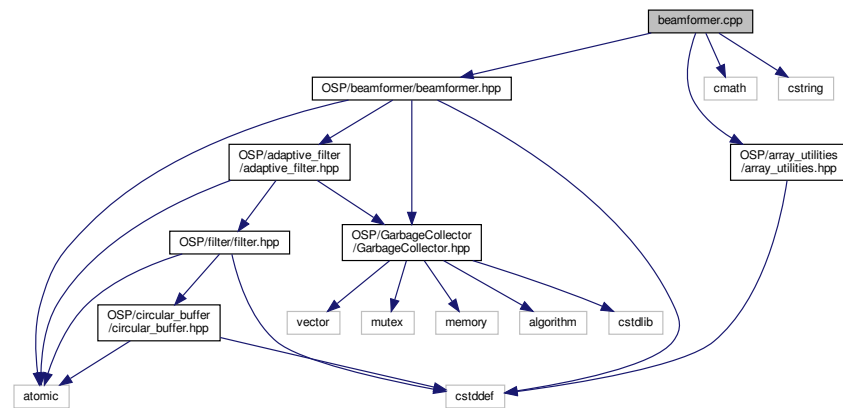
```
= {
    1.0f,
   -1.8f,
    0.81f
}
```

Definition at line 33 of file bandlimited\_filter.h.

## 5.13 beamformer.cpp File Reference

```
#include <OSP/beamformer/beamformer.hpp>
#include <OSP/array_utilities/array_utilities.hpp>
#include <cmath>
#include <cstring>
```

Include dependency graph for beamformer.cpp:



### 5.13.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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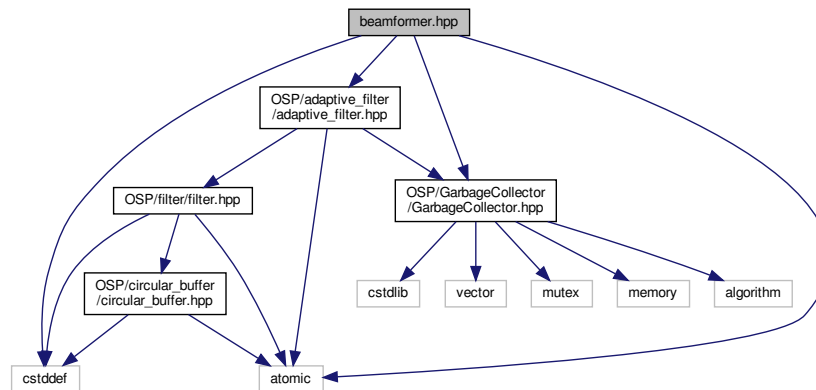
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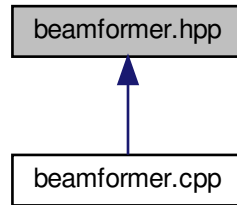
## 5.14 beamformer.hpp File Reference

```
#include <cstdint>
#include <OSP/adaptive_filter/adaptive_filter.hpp>
#include <atomic>
#include <OSP/GarbageCollector/GarbageCollector.hpp>
```

Include dependency graph for beamformer.hpp:



This graph shows which files directly or indirectly include this file:



## Classes

- class [beamformer](#)  
*Beamformer Class.*

### 5.14.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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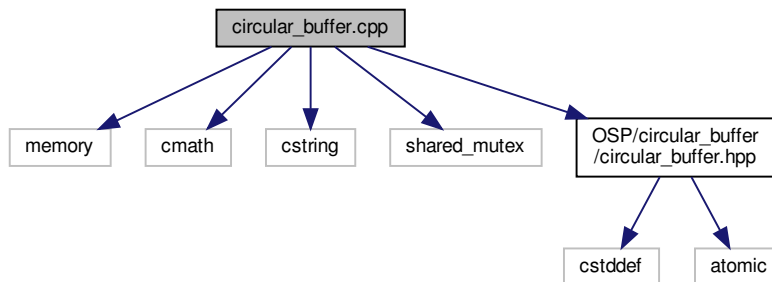
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## 5.15 circular\_buffer.cpp File Reference

```
#include <memory>
#include <cmath>
#include <cstring>
#include <shared_mutex>
#include <OSP/circular_buffer/circular_buffer.hpp>
```

Include dependency graph for circular\_buffer.cpp:



### 5.15.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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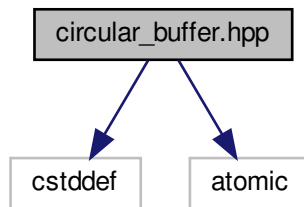
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## 5.16 circular\_buffer.hpp File Reference

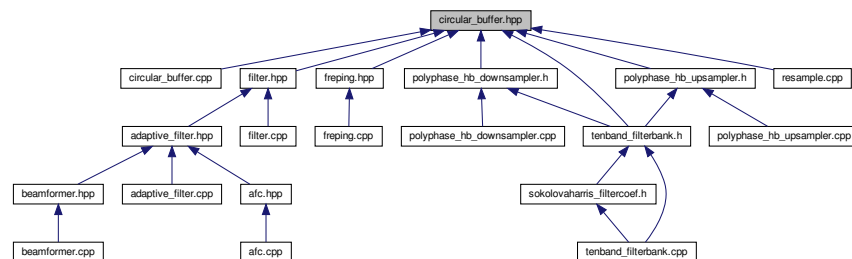
```
#include <cstdint>
```

```
#include <atomic>
```

Include dependency graph for circular\_buffer.hpp:



This graph shows which files directly or indirectly include this file:



### Classes

- class [circular\\_buffer](#)  
*Circular Buffer Class.*

### 5.16.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

## Copyright

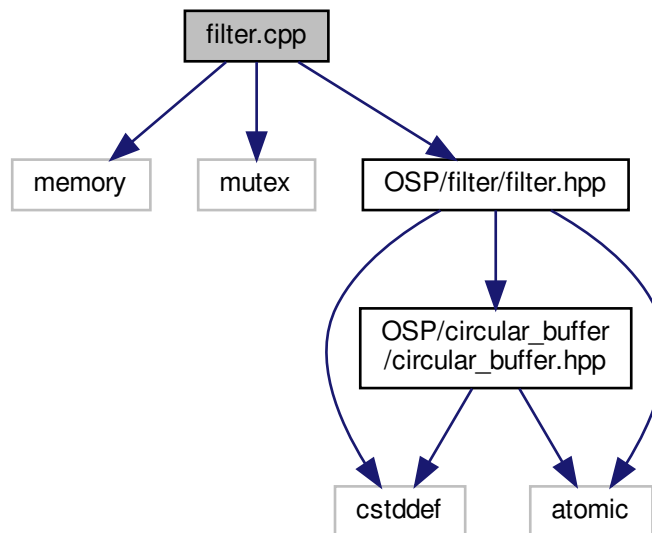
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## 5.17 filter.cpp File Reference

```
#include <memory>
#include <mutex>
#include <OSP/filter/filter.hpp>
Include dependency graph for filter.cpp:
```



### 5.17.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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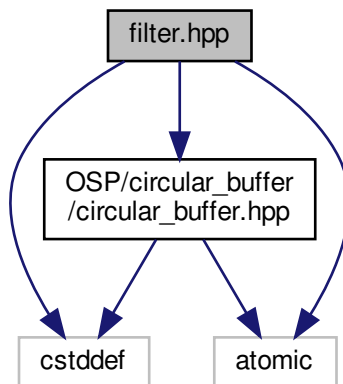
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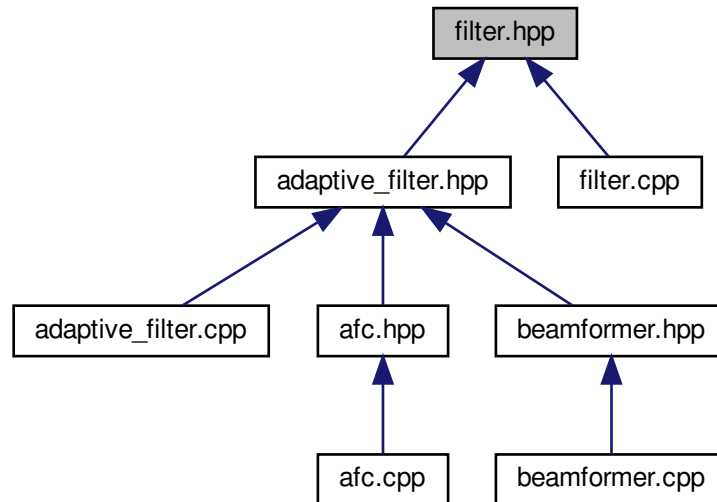
## 5.18 filter.hpp File Reference

```
#include <cstdint>
#include <atomic>
#include <OSP/circular_buffer/circular_buffer.hpp>
```

Include dependency graph for filter.hpp:



This graph shows which files directly or indirectly include this file:



## Classes

- class [filter](#)  
*Filter Class.*

### 5.18.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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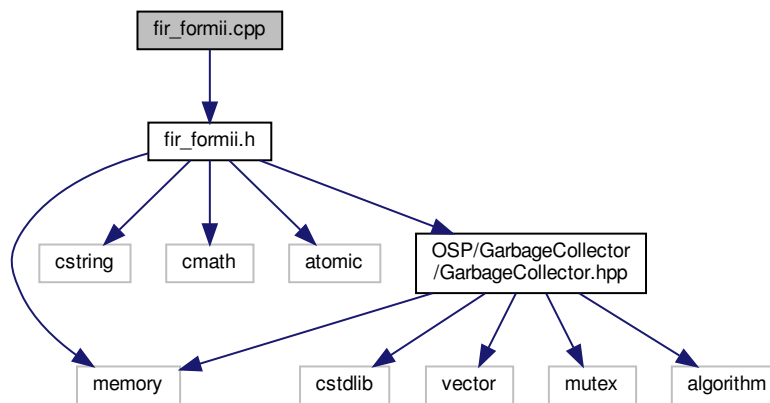


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## 5.19 fir\_formii.cpp File Reference

```
#include "fir_formii.h"
```

Include dependency graph for fir\_formii.cpp:



### 5.19.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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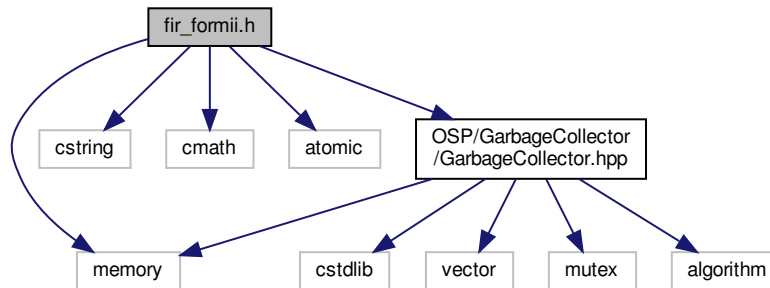
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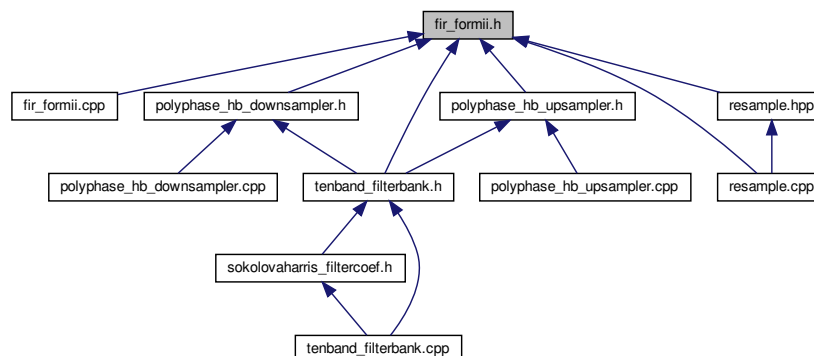
## 5.20 fir\_formii.h File Reference

```
#include <memory>
#include <cstring>
#include <cmath>
#include <atomic>
#include <OSP/GarbageCollector/GarbageCollector.hpp>
```

Include dependency graph for fir\_formii.h:



This graph shows which files directly or indirectly include this file:



## Classes

- class [fir\\_formii](#)  
*Filter Class.*

### 5.20.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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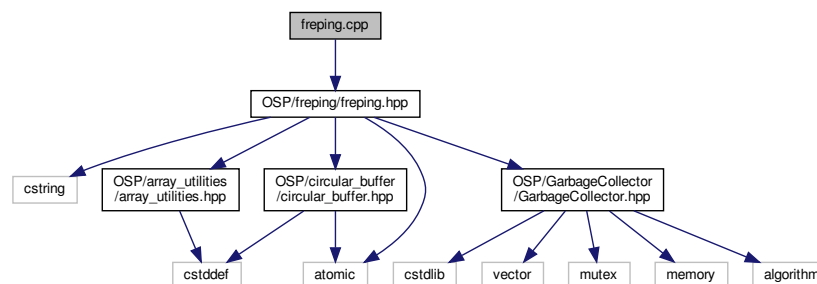
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## 5.21 freping.cpp File Reference

```
#include <OSP/freping/freping.hpp>
```

Include dependency graph for freping.cpp:



### 5.21.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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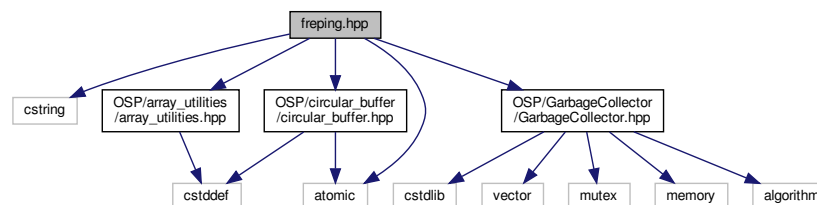
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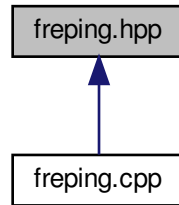
## 5.22 freping.hpp File Reference

```
#include <cstring>
#include <OSP/circular_buffer/circular_buffer.hpp>
#include <OSP/array_utilities/array_utilities.hpp>
#include <atomic>
#include <OSP/GarbageCollector/GarbageCollector.hpp>
```

Include dependency graph for freping.hpp:



This graph shows which files directly or indirectly include this file:



## Classes

- class `freping`  
*Freping Class.*

### 5.22.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.23 hamming\_window128.h File Reference

### Variables

- long **hamming\_window128\_length** = 128

### 5.23.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.24 hamming\_window64.h File Reference

### Variables

- long **hamming\_window64\_length** = 64

### 5.24.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.25 prefilter.h File Reference

#### Macros

- `#define PREFILTER_SIZE 3`

#### Variables

- `float prefilter [PREFILTER_SIZE]`

### 5.25.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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## 5.25.2 Variable Documentation

### 5.25.2.1 prefilter

```
float prefilter[PREFILTER_SIZE]
```

#### Initial value:

```
= {
    1.0f,
    -2.01f,
    1.0f
}
```

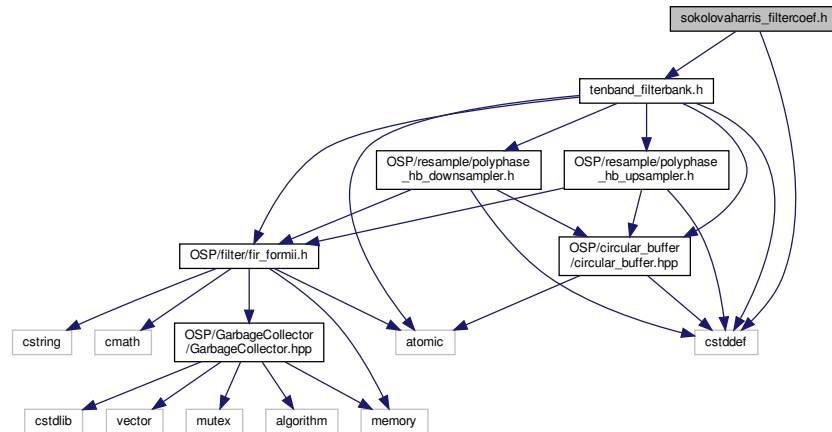
Definition at line 32 of file prefilter.h.

## 5.26 sokolovaharris\_filtercoef.h File Reference

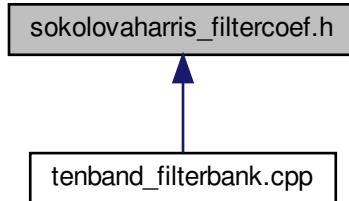
```
#include <cstdint>
#include "tenband_filterbank.h"
```



Include dependency graph for sokolovaharris\_filtercoef.h:



This graph shows which files directly or indirectly include this file:



## Variables

- `size_t half_band_filter1_length` = 35
- `size_t half_band_filter2_length` = 11
- `size_t half_band_filter3_length` = 7

### 5.26.1 Detailed Description

#### Author

Open Speech Platform (OSP) Team, UCSD

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