

# NatureDSP Signal for HiFi4

**Digital Signal Processing** 

**Library Reference** 

Library Release 4.1.0 API Revision 4.30 Updated by IntegrIT August, 2020

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# **Preface**

#### About This Manual

Welcome to the **NatureDSP Signal Processing Library**, or **NatureDSP Signal** or library for short. The library is a collection of number highly optimized DSP functions for the DSP targets.

This source code library includes C-callable functions (ANSI-C language compatible) for general signal processing (filtering, correlation, convolution), math and vector functions. Library supports both fixed-point and single precision floating data types.

# Supported Targets

Library supports Cadence HiFi4 with VFPU/SFPU little endian targets. Call **IntegrIT** to support more targets or other CPUs.

In general, library API is the same for all supported cores, but some functionality might be missing depending on core abilities. Common rules are:

• functions with floating point inputs/outputs require VFPU/SFPU option of the core Presence of specific functions might be detected in runtime using library identification routines, see para 2.13.3.

# About this Release

This is version 4.1.0 of the library which is tested on the Xtensa Xplorer (8.0.13) and tools version RI-2020.4. This release additionally contains implementation of functions related to imaging kernels. Please refer to release notes for details about new functions. Support for LLVM compilation is also added. This library has functions fine-tuned for better performance on HiFi4 DSP on RI-2020.4, and floating point variants require HiFi4 DSP with VFPU option enabled.

Owing to many optional features & add-ons available with HiFi4 core, many different variants of HiFi4 configuration can be created by the user. The present HiFi4 NatureDSP Library is tested on the HiFi4 configurations that use (i) Xtensa C library, (ii) 2 DRAM banks and (iii) big on-chip memory with RI-2020.4 tools. For toolsets other than RI-2020.4, performance numbers may not match or for some regression cases, failures might be observed. Careful fine tuning of restrict qualifier in some of the functions such as fft will help in overcoming the failures / regaining the performance.

Performance measurement is covered in a separate document.

# **Notations**

This document uses the following conventions:

- program listings, program examples, interactive displays, filenames, variables and another software elements are shown in a special typeface (Courier);
- · tables use smaller fonts.

# **Abbreviations**

API Application program interface DCT Discrete Cosine Transform

DSP Digital signal processing FFT Fast Fourier transform FIR Finite impulse response

IDE Integrated development environment IFFT Inverse Fast Fourier transform IIR Infinite impulse response IR Impulse response LMS Least mean squares

MFCC Mel frequency cepstrum coefficients

# General Library Organization

# 1.1 Headers

The library is delivered in several packages. The API for each package is defined in the appropriate header file which describes particular functions in the package. When the appropriate #include preprocessor directive is included in your source, the compiler uses the prototypes to check that each function is called with the correct arguments.

./library/include/NatureDSP_types.h	Declarations of basic data types and compiler auto detection	1.3
./library/include/NatureDSP_Signal.h	Declarations of all library functions	
./library/include/NatureDSP_Signal_fir.h	FIR Filters and Related Functions	2.1
./library/include/NatureDSP_Signal_iir.h	IIR Filters	2.1.13
./library/include/NatureDSP_Signal_math.h	Math Functions	2.3
./library/include/NatureDSP_Signal_complex.h	Complex Math Functions	2.4
./library/include/NatureDSP_Signal_vector.h	Vector Operations	2.5, 2.6
./library/include/NatureDSP_Signal_matop.h	Matrix Operations	2.7
./library/include/NatureDSP_Signal_matinv.h	Matrix Decomposition and Inversion Functions	2.8
./library/include/NatureDSP_Signal_fit.h	Fitting/interpolation	2.9
./library/include/NatureDSP_Signal_fft.h	FFT/DCT Routines	2.10
./library/include/NatureDSP_ Signal_audio.h	Mel frequency cepstrum coefficients functions	2.11
./library/include/NatureDSP_ Signal_img.h	Image processing functions	2.12
./library/include/NatureDSP_ Signal_id.h	Identification functions	2.13

# 1.2 Static Variables and Usage of C Standard Libraries

All library functions are re-entrant. Library functions do not call functions from standard C-library.

# 1.3 Types

Library uses the following C types with defined length

Name Description		Alignment, bytes
int8_t	8-bit signed value	1
uint8_t	8-bit unsigned value	1
int16_t	16-bit signed value	2
f24	24-bit fractional type	4
int32_t	32-bit signed value	4
uint32_t 32-bit unsigned value		4
int64_t 64-bit signed value		8
float32_t 32-bit single precision floating point value		4
complex_float complex single precision floating point (pair of two 32-bit values)		8
complex_fract16 complex 16-bit factional value (pair of two 16-bit values)		4
complex_fract32	complex 32-bit factional value (pair of two 32-bit values)	8

It is assumed throughout this Reference that constant pointers passed through function arguments point at read-only data

Data of given type should be aligned on its sizeof(), see table above.

# 1.4 Fractional Formats

Natively, HiFi CPU manipulates with standard 16 and 32-bit data. However, for backward compatibility with HiFi2/HiFi3 it supports special fractional type £24 which is stored in a memory as 32 bit word keeping significant bits in bits 8 through 31. So, from that perspective it may be treated as Q31 number. But users should take into account that 8 LSB are ignored. Unless specifically noted, library functions use that Q31 format, or, in another words, Q0.31.

In a Qm.n format, there are m bits used to represent the two's complement integer portion of the number, and n bits used to represent the two's complement fractional portion. m+n+1 bits are needed to store a general Qm.n number. The extra bit is needed to store the sign of the number in the most-significant bit position. The representable integer range is specified by  $[-2^m, 2^m-1]$  and the finest fractional resolution is  $2^{-n}$ . Normally, m from Q notation is omitted (because total length is defined of data type used for operand) and it is simply written as Qn.

Example data type and their formats are collected in the table below:

Data type Format Range		Resolution	Minimum value	Maximum value	
int16 t	Q0.15	-1 0 <b>,</b> 999969	3e-5	-32768	32767
int16 t	Q6.9	-64 63 <b>,</b> 998	2e-3	-32768	32767
int32 t	Q1.30	-2 1 <b>,</b> 999999991	9e-10	-2147483648	2147483647
int32 t	Q0.31	-1 0 <b>,</b> 999999999	5e-10	-2147483648	2147483647
int32 t	Q6.25	-64 63 <b>,</b> 999999970	3e-8	-2147483648	2147483647
int32_t	Q16.15	-65536 65535 <b>,</b> 99997	3e-5	-2147483648	2147483647
int32 t	Q23.8	-8388608 8388607.996	4e-3	-2147483648	2147483647
f24	Q1.30	-2 1 <b>,</b> 9999997625	2e-7	-2147483648	2147483392
f24	Q0.31	-1 0 <b>,</b> 9999998784	1e-7	-2147483648	2147483392
f24 Q6.25 -64 63,99999240		-64 63 <b>,</b> 99999240	8e-6	-2147483648	2147483392
f24 Q16.15 -6553665535,9921875		8e-3	-2147483648	2147483392	

The most-significant binary digit is interpreted as the sign bit in any  $\mathbb Q$  format number. Thus, in  $\mathbb Q15$  format, the decimal point is placed immediately to the right of the sign bit. The fractional portion to the right of the sign bit is stored in regular two's complement format.

# 1.5 Compiler Requirements

When building the library source files or library-dependent modules it is assumed that the target is a Cadence processor implementing the Xtensa HiFi4 Audio Engine Instruction Set Architecture with VFPU/SFPU option.

# 1.6 Call Conventions

Library uses ANSI-C call conventions.

# 1.7 Overflow Control and Intermediate Data Format

If not especially noted, library does not check real dynamic range of input data so it is user's responsibility to select parameters and the scale of input data according to specific case. However, if possible library use saturated arithmetic to prevent overflows.

In the most fixed-point routines operating with summing of multiple elements (i.e. FIR, matrix multiplies, etc.), library stores intermediate values in 64-bit accumulators using Q16.47 fixed-point

representation thus protecting from the overflows in the intermediate stages. Floating point routines use single precision floating point format for storing intermediate data.

The user is expected to conform to the range requirements if specified and take care to restrict the input range in such a way that the outputs do not overflow.

# 1.8 Exceptions and Processor Control Registers

Except for some mathematical routines, compatible with IEEE-754 and C99 standards (see para 2.3), all library functions do not touch global errno variable and do not modify the FPU enabled bits. FPU flags may be set during the execution of the routines. It is up to the caller to decide how to proceed given the flags.

Example of use cases are:

- The caller could enable floating point control bits before calling functions. This would result
  in an external signal that indicates an exceptional condition has occurred. We expect the
  customer to use that signal to control an external interrupt thus enabling an imprecise
  interrupt.
- The caller could zero the status flags before a function and check them when the function returns to see if any exceptional conditions occurred.

# 1.9 Special Numbers

The IEEE754 standard specifies some special values, and their representation: positive infinity ( $+\infty$  or +Inf), negative infinity ( $-\infty$  or -Inf), a negative zero (-0) distinct from ordinary ("positive") zero (+0), and "not a number" values (NaNs). In general, the following rules are applied:

- negative zero is treated as usual negative number
- the result of operations under NaN is NaN
- operations with infinity return  ${\tt NaN}$  except for few routines which require to interpret only the sign of infinity
- If a result depends on several values (E.g. in filters and correlations), and one or more of them is NaN or Inf, the propagation of those special values is complicated. The library routines will propagate the value in a way that minimizes cycles and code size. A special value will still appear in the output.
- outputs for mathematical functions for special numbers on their inputs follows ISO/IEC 9899 if not explicitly mentioned

# 1.10 Endianess

Library supports little-endian mode.

# 1.11 Performance Issues

Real-time performance of all functions depends on fulfillment special restrictions applied to input/output arguments. Typically, for maximum performance, user have to use **aligned data arrays** (on 8 byte boundary) for storing input and output arguments, number of data should be **multiple of 2 or 4** and should be **greater than 4**. Specific requirements are given for each function in its API description.

Data alignment may be achieved by several methods:

- placing the data into special data section and make alignment at the link-time
- use attribute ((aligned(x))) modifiers in the data declarations
- dynamically allocate arrays of slighter bigger size and align pointers<sup>1</sup>

Test examples use two last methods.

# 1.12 Object Model

Effective use of all HiFi4 core benefits requires specific processing and special data moves minimizing the overhead. That is why many functions are supplied with object-like interface simplifying real-time processing chain but requiring special initialization before processing. Besides, function wrapped by object-like interface use best possible alignment for data storage and may utilize HiFi4 core better in some cases.

Initialization normally done once at the initialization time and do not affect to the real-time performance. Sequence consists of three stages

- call <obj>\_alloc() function with parameters that define the block size, filter length, etc. This function/macro returns the size of memory has to be allocated for object for that specific parameters
- allocate the memory somehow. It may be done dynamically if <obj>\_alloc() function is used
- pass the pointer to allocated memory to the function <obj>\_init. It cleans up that memory block, reorder filter coefficients appropriately, etc. and returns the handle to the object. This handle will be used later for data processing by this given object, .i.e. block filtering.

Here we denote the symbolic name of object as <obj>. For example, corresponding functions for block FIR filtering will be named as:

```
bkfir_alloc()request the memory size for objectbkfir_init()initialize the objectbkfir_process()make filtering of block
```

# 1.13 Scratch Memory

Some functions require scratch memory area for saving temporary data during the execution. It can be shared between objects if they are called in the same processing thread. Scratch memory area must always be aligned on a 8-bytes boundary. For each algorithm that requires a scratch area, its minimum size (either in bytes or in 16-bit words) is specified in such a way that both static and dynamic allocation are possible. Most often the required scratch size is specified through a macro or function that takes a number of algorithm parameters. For example, one might use the following wrapping function to perform a autocorrelation (see 2.1.10) with the scratch area being allocated on the stack:

```
#define N 80
int run_autocorrelation( int16_t * y, const int16_t *x )
{
    // Allocate the scratch memory. Macro returns required size in bytes
    int8_t scratch[FIR_ACORRA16X16_SCRATCH_SIZE(N) attribute__ ((aligned(8)));
    // Invoke the autocorrelation routine
    fir_acorra16x16 (scratch, y, x, N);
}
```

<sup>&</sup>lt;sup>1</sup> xcc malloc() always returns pointer aligned on 64-bit boundary special additional alignment procedure is not required

# 1.14 Brief Function List

Vectorized version	Scalar version	Purpose	Reference
FIR filters and related f	unctions		
bkfir		Block real FIR filter	2.1.1, 2.1.2
cxfir		Complex block FIR filter	2.1.3
firdec		Decimating block real FIR filter	2.1.4
firinterp		Interpolating block real FIR filter	2.1.5
fir convol,		Circular/linear convolution	2.1.6, 2.1.7
cxfir_convol		Official/fillical convolution	2.1.0, 2.1.7
fir_xcorr		Circular/linear correlation	2.1.8, 2.1.9
fir_acorr		Circular/linear autocorrelation	2.1.10, 2.1.11
fir_blms		Blockwise Adaptive LMS algorithm	2.1.12
IIR filters			
bqriir, bqciir		Biquad Real block IIR	2.2.1
latr		Lattice block Real IIR	2.2.2
Mathamatica		Edition proof (Con III C	L.L.L
Mathematics	Tani marin	The state of the s	
vec_recip	scl_recip	Reciprocal on a vector of Q31 numbers	2.3.1
vec_divide	scl_divide	Division	2.3.2
vec_logn	scl_logn	Different kinds of logarithm	2.3.3
vec_log2 vec_logn	scl_log2 scl logn		
vec_recip	scl recip	Reciprocal on a vector of Q31 numbers	2.3.1
vec divide	scl divide	Division	2.3.2
vec logn	scl logn	Different kinds of logarithm	2.3.3
vec_log2	scl_log2	Different kinds of logarithm	2.0.0
vec_logn	scl_logn		
vec_antilog2 vec_antilog10	scl_antilog2 scl_antilog10	Different kinds of antilogarithm	2.3.4
vec_antilogio	scl_antilogio	-	
vec pow		Power function	2.3.5
vec sqrt	scl sqrt	Square root	2.3.6
vec rsqrt	scl rsqrt	Reciprocal square root	2.3.7
vec sine	scl sine	Sine	2.3.8
vec cosine	scl cosine	Cosine	2.0.0
vec_tan	scl tan	Tangent	2.3.9
vec_atan,	scl_atan, scl_atan2	Arctangent	2.3.10, 2.3.11
vec_atan2 vec_tanh	vec tanh	Lly marked to ten mark	0.2.40
	_	Hyperbolic tangent	2.3.12
vec_sigmoid	scl_sigmoid	Sigmoid	2.3.13
vec_relu vec softmax	scl_relu	Rectifier function (ReLU)	2.3.14
_		Softmax	2.3.15
vec_int2float	scl_int2float	Integer to float conversion	2.3.16
vec_float2int	scl_float2int	Float to integer conversion	2.3.17
<b>Complex Mathematics</b>			
vec_complex2mag	scl_complex2mag	Complex magnitude	2.4.1
vec_complex2invmag	scl_complex2invmag	Reciprocal of complex magnitude	2.4.1
Vector operations			
vec_dot		Vector dot product	2.5.1, 2.5.2, 2.6.4
vec_add		Vector sum	2.5.3, 2.6.1
vec mul		Vector multiply, multiply-accumulate	2.6.2, 2.6.3
vec_power		Power of a vector	2.5.4
vec shift		Vector scaling with saturation	2.5.5
vec_scale		Toolor ooding with odtalation	2.0.0

Vectorized version	Scalar version	Purpose	Reference
vec_bexp	scl_bexp	Common exponent	2.5.6
vec_min, vec_max		Find a maximum/minimum in a vector	2.5.7
Matrix operations			- 1
mtx_mpy		Matrix multiply	2.7.1
mtx_vecmpy		Matrix by vector multiple	2.7.2
	and Inversion Functions	Matrix by voctor manapie	2.1.2
•			1004000
mtx_inv		Gauss-Jordan matrix inversion	2.8.1, 2.8.2
matchol		Cholesky decomposition, pseudoinversion	2.8.3, 2.8.4,
			2.8.5, 2.8.6,
			2.8.7, 2.8.8
Fitting/Interpolation			
vec_poly		Polynomial approximation	2.9.1
FFT/DCT			1
fft cplx		FFT on complex data	2.10.1
fft real		FFT on real data	2.10.2
ifft cplx		Inverse FFT on complex data	2.10.3
ifft real		Inverse FFT forming real data	2.10.4
fft cplx ie		FFT on complex data with optimized memory usage	2.10.5
fft real ie		FFT on real data with optimized memory usage	2.10.6
ifft cplx ie		Inverse FFT on complex data with optimized memory	2.10.7
		usage	2.10.7
ifft_real_ie		Inverse FFT forming real data with optimized memory	2.10.8
		usage	
fft_powerspectrum			2.10.9
dct, mdct		Discrete cosine transform	2.10.10, 2.10.11
dct2d, idct2d		2D Discrete cosine transforms	2.10.12, 2.10.13
Mel frequency cepstru	m coefficients functions		•
LogMel		Mel frequency extraction	2.11.1
MFCC		MFCC	2.11.2
Image processing			•
imgpad		padding/cropping	2.12.1
imghist		histogram	2.12.2
imgnorm		normalization	2.12.3
imgconvert		RGB/YUV conversion	2.12.4
imginterleave,		interleaving/definterleaving	2.12.5
imgdeinterleave imgresize		racining	2.10.6
imgresize		resizing	2.12.6
imgfft, imgifft	1	rotation	2.12.7
Identification		2D FFT for images	2.12.8
NatureDSP Signal ge	et library version	Library Varaian Request	2 12 1
	et_library_version et_library_api_versio	Library Version Request	2.13.1
NatureDSP_Signal_ge		=======================================	2.13.2 2.13.3
Macarenor stallar 1:	DI T.C9CII.C	Library API Capability Request	Z. 13.3

# 2 Reference

# 2.1 FIR Filters and Related Functions

FIR filtering APIs excepting correlation/convolution, autocorrelation and blockwise LMS algorithm require instantiation. In particular, filter objects encapsulate the delay line buffer, which is organized in such a way that advanced processor capabilities (e.g. circular data addressing) are efficiently utilized. When allocating and initializing a filter instance through  ${\tt xfir\_alloc}()$  and  ${\tt xfir\_init}()$  function calls, the user has to specify the length of filters and its coefficients. On the data processing stage the user application sequentially calls an  ${\tt xfir\_process}()$  function, providing it with a block of N input samples on each call.  ${\tt xfir\_process}()$  function updates the internal delay line with input samples, and computes N filter output samples, which are returned to the calling application via the output data buffer argument.

#### 2.1.1 Block Real FIR Filter

#### Description

Computes a real FIR filter (direct-form) using IR stored in vector  $\mathbf{h}$ . The real data input is stored in vector  $\mathbf{x}$ . The filter output result is stored in vector  $\mathbf{y}$ . The filter calculates  $\mathbf{n}$  output samples using  $\mathbf{m}$  coefficients and requires last  $\mathbf{m}-1$  samples in the delay line which is updated in circular manner for each new sample. User has an option to set IR externally or copy from original location (i.e. from the slower constant memory). In the first case, user is responsible for right alignment, ordering and zero padding of filter coefficients – usually array is composed from zeroes (left padding), reverted IR and right zero padding. 6 variants available:

Precision

Туре	Description	
16x16	16-bit data, 16-bit coefficients, 16-bit outputs. Ordinary (single channel) variant and stereo	
24x24p	use 24-bit data packing for internal delay line buffer and internal coefficients storage	
32x16	32-bit data, 16-bit coefficients, 32-bit outputs	
32x32	32-bit data, 32-bit coefficients, 32-bit outputs. Ordinary (single channel) variant and	
	stereo	
32x32ep	32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate	
computations. Available for HiFi4 only		
f floating point. Requires VFPU/SFPU core option. Ordinary (single channel) va		
	and stereo	

#### **Algorithm**

$$y_n = \sum_{m=0}^{M-1} h_{M-1-m} x_{n+m}, n = \overline{0...N-1}$$

NOTE

This is formal description of algorithm, in reality processing in done using circular buffers, so user application is not responsible for management of delay lines

#### Object allocation

```
size_t bkfir24x24p_alloc(int M, int extIR)
size_t bkfir16x16_alloc (int M, int extIR)
size_t bkfir32x16_alloc (int M, int extIR)
size_t bkfir32x32_alloc (int M, int extIR)
size_t bkfir32x32_alloc (int M, int extIR)
size_t bkfir32x32ep_alloc(int M, int extIR)
size_t bkfirf_alloc (int M, int extIR)
size_t stereo_bkfir16x16_alloc (int M, int extIR)
size_t stereo_bkfir32x32_alloc (int M, int extIR)
size_t stereo_bkfirf_alloc (int M, int extIR)
```

Туре	Name	Size	Description
Input			

int	М	length of filter, should be a multiple of 4
int	extIR	if zero, IR is copied from original location, otherwise not but user should keep alignment, order of coefficients and zero padding requirements shown below

Returns: size of memory in bytes to be allocated

#### NOTE:

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes			
	extIR=0	extIR!=0		
bkfir24x24p alloc	80+M*6	56+M*3		
bkfir16x16 alloc	72+M*4	56+M*2		
bkfir32x16 alloc	72+M*6	64+M*4		
bkfir32x32 alloc	72+M*8	48+M*4		
bkfir32x32ep alloc	72+M*8	48+M*4		
bkfirf alloc	72+M*8	48+M*4		
stereo bkfir16x16 alloc	72+M*8	56+M*4		
stereo bkfir32x32 alloc	72+M*16	48+M*8		
stereo bkfirf alloc	72+M*16	48+M*8		

#### Object initialization

```
bkfir24x24p_handle_t bkfir24x24p_init
          \overline{\text{(void * objmem, int M, int extIR, const f24 * h)}}
bkfir16x16 handle t bkfir16x16 init
          (void * objmem, int M, int extIR, const int16_t * h)
bkfir32x32ep handle_t bkfir32x32ep_init
          (void * objmem, int M, int extIR, const int32 t* h)
bkfirf handle t bkfirf init
          (void * objmem, int M, int extIR, const float32_t* h)
stereo bkfir16x16 handle t stereo bkfir16x16 init
          (void \overline{\phantom{m}} objmem, int M, int extIR,
const int16_t * hl, const int16_t * hr)
stereo_bkfir32x32_handle_t stereo_bkfir32x32_init
          (void * objmem, int M, int extIR,
           const int32_t * hl, const int32_t * hr)
stereo bkfirf handle t stereo bkfirf init
          (void * objmem, int M, int extIR,
           const float32 t * hl, const float32 t * hr)
```

Туре	Name	Size	Description		
Input	Input				
void*	objmem		allocated memory block		
f24, int16_t, int32_t, float32_t	h	М	filter coefficients; h[0] is to be multiplied with the newest sample		
<pre>int16_t, int32_t, float32_t</pre>	hl	М	for stereo filters: filter coefficients for left channel		
int16_t, int32_t, float32 t	hr	М	for stereo filters: filter coefficients for right channel		
int	М		length of filter		
int	extIR		if zero, IR is copied from original location, otherwise not but user should keep alignment, order of coefficients and zero padding requirements shown below		

Returns: handle to the object

Alignment, ordering and zero padding for external IR (extin!=0)

Function	Alignment, bytes	Left zero padding, bytes	Coefficient order	Right zero padding, bytes
bkfir24x24p_init	8	((-M&4)+5)*3	inverted	7
bkfir16x16_init	8	2	inverted	6
bkfir32x16_init (M>32)	8	10	inverted	6
bkfir32x16_init (M<=32)	8	2	inverted	6
bkfir32x32_init	8	4	inverted	12
bkfir32x32ep_init	8	4	inverted	12
bkfirf_init	8	0	direct	0
stereo_bkfir16x16_init	8	2	inverted	6
stereo_bkfir32x32_init	8	4	inverted	12
stereo_bkfirf_init	8	0	direct	0

Update the delay line and compute filter output

```
void bkfir24x24p process( bkfir24x24p handle t handle,
                        f24* y, const f24 * x, int N )
void bkfir16x16 process (
              bkfir16x16 handle t handle,
              int16 t * y, const int16 t * x, int N )
void bkfir32x16 process (
              bkfir32x16_handle_t handle,
int32_t * y, const int32_t * y, int N) void bkfir32x32 process (
              bkfir32x32_handle_t handle,
int32_t * y, const int32_t * y, int N)
void bkfir32x32ep_process ( bkfir32x32ep_handle_t handle,
                       int32 t * y, const int32 t * y, int N)
void bkfirf process (
              bkfirf handle t handle,
               float32_t * y, const float32_t * x, int N);
void stereo_bkfir16x16_process (
              stereo bkfir16x16 handle t handle,
              int16 \overline{t} * y, const int1\overline{6} t * x, int N )
void stereo bkfir32x32 process (
              stereo bkfir32x32_handle_t handle,
               int32 t * y, const int32 t * x, int N)
void stereo bkfirf process (
               stereo bkfirf_handle_t handle,
               float32 t * y, const float32 t * x, int N);
```

Туре	Name	Size	Description
Input			
f24, int16_t, int32_t, float32_t	Х	N*S	input samples
int	N		length of sample block
	S		1 for ordinary (single channel) filters, 2 - for stereo variant
Output			
f24, int16_t, int32_t, float32_t	У	N*S	output samples

Returns: none

Restrictions

x,y-should not overlap

x,h - aligned on a 8-bytes boundary

N,M - multiples of 4

# 2.1.2 Block Real FIR Filter with Arbitrary Parameters

Description

These functions implement FIR filter described in previous chapter with no limitation on size of data block,

alignment and length of impulse response for the cost of performance.

#### Precision

5 variants available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
32x32ep	32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate computations
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

$$y_n = \sum_{m=0}^{M-1} h_{M-1-m} x_{n+m}, n = \overline{0...N-1}$$

#### NOTE:

This is formal description of algorithm, in reality processing in done using circular buffers, so user application is not responsible for management of delay lines

#### **Object allocation**

```
size_t bkfiral6x16_alloc(int M)
size_t bkfira32x16_alloc(int M)
size_t bkfira32x32_alloc(int M)
size_t bkfira32x32_ep_alloc(int M)
size_t bkfiraf alloc(int M)
```

Туре	Name	Size	Description
Input			
int	М		length of filter

Returns: size of memory in bytes to be allocated

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes
bkfira16x16_alloc	72+ M*4
bkfira32x16_alloc	72+ M*6
bkfira32x32_alloc	80+ M*8
bkfira32x32ep_alloc	80+ M*8
bkfiraf_alloc	80+ M*8

#### **Object initialization**

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int16_t, int32_t, float32_t	h	М	filter coefficients; h [0] is to be multiplied with the newest sample
int	М		length of filter

Returns: handle to the object

Update the delay line and compute filter output

Туре	Name	Size	Description		
Input					
int16_t,	X	N	input samples		
int32_t,					
float32_t					
int	N		length of sample block		
Output					
int16_t,	У	N	output samples		
int32_t,					
float32_t					

Returns: none

Restrictions

x,y-should not overlap

#### 2.1.3 Complex Block FIR Filter

#### Description

Computes a complex FIR filter (direct-form) using complex IR stored in vector  $\mathbf{h}$ . The complex data input is stored in vector  $\mathbf{x}$ . The filter output result is stored in vector  $\mathbf{y}$ . The filter calculates  $\mathbf{n}$  output samples using  $\mathbf{m}$  coefficients, requires last  $\mathbf{m}-\mathbf{1}$  samples in the delay line which is updated in circular manner for each new sample. Real and imaginary parts are interleaved and real parts go first (at even indexes). User has an option to set IR externally or copy from original location (i.e. from the slower constant memory). In the first case, user is responsible for right alignment, ordering and zero padding of filter coefficients – usually array is composed from zeroes (left padding), reverted IR and right zero padding. 5 variants available:

Precision

# Type Description 16x16 16-bit data, 16-bit coefficients, 16-bit outputs 32x16 32-bit data, 16-bit coefficients, 32-bit outputs 32x32 32-bit data, 32-bit coefficients, 32-bit outputs 32x32ep 32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate computations f floating point. Requires VFPU/SFPU core option

**Algorithm** 

$$y_n = \sum_{m=0}^{M-1} h_{M-1-m} x_{n+m}, n = \overline{0...N-1}$$

NOTE

This is formal description of algorithm, in reality processing in done using circular buffers, so user application is not responsible for management of delay lines

**Object allocation** 

```
size_t cxfir16x16_alloc(int M, int extIR)
size_t cxfir32x16_alloc(int M, int extIR)
size_t cxfir32x32_alloc(int M, int extIR)
size_t cxfir32x32ep_alloc(int M, int extIR)
size_t cxfirf alloc(int M, int extIR)
```

Туре	Name	Size	Description
Input			

int   M   length of filter
----------------------------

Returns: size of memory in bytes to be allocated

NOTE:

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes			
	extIR=0	extIR!=0		
cxfir16x16 alloc	80+12*M	40+4*M		
cxfir32x16 alloc	80+12*M	64+8*M		
cxfir32x32 alloc	80+16*M	72+8*M		
cxfir32x32ep_alloc	80+16*M	72+8*M		
cxfirf_alloc	64+16*M	72+8*M		

#### Object initialization

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
<pre>complex_fract32, complex_fract16, complex_float</pre>	h	М	complex filter coefficients; h[0] is to be multiplied with the newest sample, Q31, Q15 or floating point
int	М		length of filter
int	extIR		if zero, IR is copied from original location, otherwise not but user should keep alignment, order of coefficients and zero padding requirements shown below

Returns: handle to the object

Alignment, ordering and zero padding for external IR (extIR!=0)

Function	Alignment, bytes	· · · · · · · · · · · · · · · · · · ·		Right zero padding, bytes
cxfir16x16_alloc	8	2 before each copy	inverted: conjugated copy and (imaginary; real) copy at 4*(M+4) bytes offset	6 after each copy
cxfir32x16_init	8	4	inverted	4
cxfir32x32_init	8	0	inverted and conjugated	0
cxfir32x32ep_init	8	0	inverted and conjugated	0
cxfirf_init	8	0	direct	0

Update the delay line and compute filter output

Туре	Name	Size	Description	
Input				
complex_fract16, complex_fract32, complex_float	Х	N	input samples , Q15, Q31 or floating point	
int	N		length of sample block	
Output				
<pre>complex_fract16, complex_fract32, complex_float</pre>	У	N	output samples , Q15, Q31 or floating point	

Returns: none

Restrictions

x, y - should not overlap

x, h - aligned on a 8-bytes boundary

N, M - multiples of 4

# 2.1.4 Decimating Block Real FIR Filter

#### Description

Computes a real FIR filter (direct-form) with decimation using IR stored in vector  $\mathbf{n}$ . The real data input is stored in vector  $\mathbf{x}$ . The filter output result is stored in vector  $\mathbf{y}$ . The filter calculates  $\mathbf{n}$  output samples from  $\mathbf{n}^*\mathbf{D}$  input samples using  $\mathbf{m}$  coefficients, requires last  $\mathbf{m}-1$  samples on the delay line and updated in circular manner for each new  $\mathbf{n}$  samples.

NOTE:

To avoid aliasing IR should be synthesized in such a way to be narrower than input sample rate divided to 2D.

#### **Precision**

#### 5 variants available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
32x32ep	32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate computations
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

$$r_n = \sum_{m=0}^{M-1} h_{M-1-m} x_{D \cdot n+m}, n = \overline{0...N-1}$$

NOTE:

This is formal description of algorithm, in reality processing in done using circular buffers, so user application is not responsible for management of delay lines

#### **Object allocation**

```
size_t firdec16x16_alloc(int D, int M)
size_t firdec32x16_alloc(int D, int M)
size_t firdec32x32_alloc(int D, int M)
size_t firdec32x32ep_alloc(int D, int M)size_t firdecf_alloc (int D, int M)
```

Туре	Name	Size	Description
Input			
int	D		decimation factor
int	М		length of filter

Returns: size of memory in bytes to be allocated

#### NOTE:

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes
firdec32x16_alloc	40+ (M+8*D) *4+ (M+4) *2
firdec16x16_alloc	40+ (M+8*D) *2+ (M+4) *2
firdec32x32_alloc	40+ (M+8*D) *4+ (M+4) *4
firdec32x32ep_alloc	40+ (M+8*D) *4+ (M+4) *4
firdecf alloc	40+ (M+8*D) *4+ (M+4) *4

#### Object initialization

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int32_t, int16_t, float32_t	h	М	filter coefficients; h[0] is to be multiplied with the newest sample, Q31, Q15 or floating point
int	D		decimation factor
int	М		length of filter

Returns: handle to the object

#### Update the delay line and compute decimator output

Туре	Name	Size	Description		
Input					
int16_t, int32_t, float32 t	Х	D*N	input samples , Q15, Q31 or floating point		
int	N		length of output sample block, should be a multiple of 8		
Output					
<pre>int16_t, int32_t, float32_t</pre>	У	N	output samples, Q15, Q31 or floating point		

Returns: none

Restrictions

x,h,r should not overlap

x, h - aligned on a 8-bytes boundary

N - multiple of 8

D >1

Conditions for optimum performance

D - 2, 3 or 4

# 2.1.5 Interpolating Block Real FIR Filter

#### Description

Computes a real FIR filter (direct-form) with interpolation using IR stored in vector  ${\tt h}$ . The real data input is stored in vector  ${\tt x}$ . The filter output result is stored in vector  ${\tt y}$ . The filter calculates  ${\tt N}^{\star}{\tt D}$  output samples using  ${\tt M}^{\star}{\tt D}$  coefficients from  ${\tt N}$  inputs. Delay line holds  ${\tt M}^{\star}{\tt D}-1$  last samples and updated in circular manner for each new sample.

#### Precision

#### 5 variants available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
32x32ep	32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate computations
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

$$y_{n \cdot D + d} = D \cdot \sum_{m=0}^{M-1} h_{D(M-1-m)+d} x_{n+m}, n = \overline{0...N-1}, d = \overline{0...D-1},$$

#### NOTE

This is formal description of algorithm, in reality processing in done using circular buffers, so user application is not responsible for management of delay lines

#### **Object allocation**

```
size_t firinterp16x16_alloc(int D, int M)
size_t firinterp32x16_alloc(int D, int M)
size_t firinterp32x32_alloc(int D, int M)
size_t firinterp32x32ep_alloc(int D, int M)
size t firinterpf alloc (int D, int M)
```

Туре	Name	Size	Description
Input			
int	D		interpolation ratio
int	М		length of subfilter. Total length of filter is м*□

Returns: size of memory in bytes to be allocated

NOTE:

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes
firinterp16x16_alloc	40+ (M+8) *2+ (M+4) *D*2
firinterp32x16_alloc	40+ (M+8) *4+ (M+4) *D*2
firinterp32x32_alloc	40+ (M+8) *4+ (M+4) *D*4
firinterp32x32ep_alloc	40+ (M+8) *4+ (M+4) *D*4
firinterpf alloc	40+ (M+8) *4+ (M+4) *D*4

#### **Object initialization**

```
firinterp16x16_handle_t firinterp16x16_init(void * objmem, int D, int M, const int16_t* h)
firinterp32x16_handle_t firinterp32x16_init(void * objmem, int D, int M, const int16_t * h)
firinterp32x32_handle_t firinterp32x32_init(void * objmem, int D, int M, const int32_t * h)
firinterp32x32ep_handle_t firinterp32x32ep_init(void * objmem, int D, int M, const int32_t * h)
firinterpf_handle_t firinterpf_init(void * objmem, int D, int M, const float32_t * h)
firinterpf_handle_t firinterpf_init(void * objmem, int D, int M, const float32_t * h)
```

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int32_t, int16_t, float32_t	h	M*D	filter coefficients; h[0] is to be multiplied with the newest sample,Q31, Q15 or floating point
int	D		interpolation ratio
int	М		length of subfilter. Total length of filter is м* D

Returns: handle to the object

# Update the delay line and compute interpolator output

Туре	Name	Size	Description
Input			
int16_t,	Х	N	input samples, Q15, Q31 or floating point
int32_t,			
float32_t			
int	N		length of input sample block
Output			
int16_t,	У	N*D	output samples, Q15, Q31 or floating point
int32_t,			
float32_t			

Returns: none

Restrictions x,h,y should not overlap

x, h - aligned on a 8-bytes boundary

 $_{\rm M}$  - multiples of 4  $_{\rm N}$  - multiples of 8  $_{\rm D}$  should be >1  $_{\rm D}$  - 2, 3 or 4

Conditions for optimum performance

#### 2.1.6 Circular convolution

#### Description

Performs circular convolution between vectors  $\mathbf{x}$  (of length  $\mathbf{n}$ ) and  $\mathbf{y}$  (of length  $\mathbf{m}$ ) resulting in vector  $\mathbf{r}$  of length  $\mathbf{n}$ .

Two versions of these functions available: faster version (fir\_convol16x16, fir\_convol32x16, fir\_convol32x32, fir\_convol32x32ep, cxfir\_convol32x16, fir\_convolf) with some restrictions on input arguments and slower version (fir\_convola16x16, fir\_convola32x16, fir\_convola32x32ep, cxfir\_convola32x16, fir\_convolaf) for arbitrary arguments. In addition, these slower version implementations require scratch memory area. 5 variants available:

#### Precision

Type Description			
16x16	16x16-bit data, 16-bit outputs		
32x16 32x16-bit data, 32-bit outputs (both real and complex)			
32x32 32x32-bit data, 32-bit outputs			
32x32ep 32-bit data, 32-bit outputs, use 72-bit accumulators for intermediate computations			
f floating point. Requires VFPU/SFPU core option			

#### Algorithm

$$r_k = \sum_{m=0}^{M-1} x_{\text{mod}(k-m,N)} y_m, k = \overline{0...(N-1)}$$

```
void fir convol16x16 (int16 t * r, const int16 t * x,const int16 t * y,
                     int N, int M);
void fir convol32x16 (int32 t * r, const int32 t * x, const int16 t * y,
                     int N, int M);
void fir convol32x32 (int32_t * r, const int32_t * x, const int32_t * y,
                     int N, int M);
void fir_convol32x32ep( int32_t * r, const int32_t * x, const int32_t * y,
                     int N, int M);
void cxfir_convol32x16( complex_fract32 * r,
                     const complex fract32 * x, const complex fract16 * y,
                     int N, int M);
void fir convolf
                    ( float32 t * r,
                     const float32 t * x, const float32 t * y,
                     int N, int M);
int N, int M);
void fir convola32x16 ( void
                     int32_t * r, const int32_t * x, const int16_t * y,
                     int N, int M);
void fir convola32x32 ( void
                              s,
                      int32_t * r, const int32_t * x, const int32 t * y,
                     int N, int M);
void fir convola32x32ep(void
                      int32 t * r, const int32 t * x, const int32 t * y,
                     int N, int M);
void cxfir convola32x16(void
                     complex fract32 * r,
                     const complex fract32 * x, const complex fract16 * y,
                     int N, int M);
void fir convolaf
                     (void
                      float32 t * r,
                     const float32_t * x, const float32 t * y,
                     int N, int M);
```

Туре	Name	Size	Description
Input	•	•	
<pre>int16_t, int32_t, complex_fract32, float32_t</pre>	Х	N	input data (Q15, Q31 or floating point)
<pre>int16_t, complex_fract16, or float32_t</pre>	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	М		length of y
Output			
int16_t, int32_t, complex_fract32, float32_t	r	N	output data, Q15, Q31 or floating point
Temporary			
void	s		Scratch memory,  FIR_CONVOLA16X16_SCRATCH_SIZE( N, M )  FIR_CONVOLA32X16_SCRATCH_SIZE( N, M )  FIR_CONVOLA32X32_SCRATCH_SIZE( N, M )  FIR_CONVOLA32X32EP_SCRATCH_SIZE( N, M )  CXFIR_CONVOLA32X16_SCRATCH_SIZE( N, M )  FIR_CONVOLAF_SCRATCH_SIZE( N, M )  bytes

#### Returned value

none

Restrictions

For slow versions (fir\_convola16x16, fir\_convola32x16, fir\_convola32x32, fir\_convola32x32ep, cxfir\_convola32x16, fir\_convolaf): x,y,r,s should not overlap s should be aligned on 8-byte boundary N>=M-1

For fast versions (fir\_convol16x16, fir\_convol32x16, fir\_convol32x32, fir\_convol32x32ep, cxfir\_convol32x16, fir\_convolf): x,y,r should not overlap x,y,r should be aligned on 8-byte boundary N,M - multiples of 4

#### 2.1.7 Linear Convolution

#### Description

Functions perform linear convolution between vectors x (of length n) and y (of length m) resulting in vector x of length n+m-1.

#### Precision

2 variants available:

Туре	Type Description	
16x16	16x16-bit data, 16-bit outputs	
32x32 32x32-bit data, 32-bit outputs		

#### Algorithm

$$r_k = \sum_{j=\max(k-M+1,0)}^{\min(N-1,k)} x_j y_{k-j}, k = \overline{0...(M+N-2)}$$

Туре	Name	Size	Description
Input			
int16_t, int32_t	X	N	input data (Q15, Q31)
int16_t, int32_t	У	М	input data (Q31, Q15)
int	N		length of x
int	М		length of y
Output			
int16_t, int32_t	r	M+N-1	output data, Q15, Q31
Temporary			
void	Ø		Scratch memory,  FIR_LCONVOLA16X16_SCRATCH_SIZE( N, M)  FIR_LCONVOLA32X32_SCRATCH_SIZE( N, M)  bytes

#### Returned value

none

Restrictions

x, y, r, s should not overlap s should be aligned on 8-byte boundary

N>0, M>0N>=M-1

#### 2.1.8 Circular Correlation

#### Description

Estimates the circular cross-correlation between vectors  $\mathbf{x}$  (of length  $\mathbf{n}$ ) and  $\mathbf{y}$  (of length  $\mathbf{m}$ ) resulting in vector  $\mathbf{x}$  of length  $\mathbf{n}$ . It is a similar to correlation but  $\mathbf{y}$  is read in opposite direction.

Two versions of these functions available: faster version ( $fir_xcorr16x16$ ,  $fir_xcorr32x16$ ,  $fir_xcorr32x32$ ,  $fir_xcorr32x32ep$ ,  $fir_xcorrf$ ,  $cxfir_xcorrf$ ) with some restrictions on input arguments and slower version ( $fir_xcorra16x16$ ,  $fir_xcorra32x16$ ,  $fir_xcorra32x32ep$ ,  $fir_xcorra16x16$ ,  $fir_xcorra32x32ep$ ,  $fir_xcorra16x16$ ,  $fir_xcorra$ 

Precision

5 variants available:

Туре	Description	
16x16	16x16-bit data, 16-bit outputs	
32x16	32x16-bit data, 32-bit outputs	
32x32	32x32-bit data, 32-bit outputs (both real and complex data)	
32x32ep	32-bit data, 32-bit outputs, use 72-bit accumulators for intermediate computations.	
f	floating point (both real and complex data). Requires VFPU/SFPU core option	

Algorithm

$$r_k = \sum_{m=0}^{M-1} x_{\text{mod}(k+m,N)} y_m, k = \overline{0...(N-1)}$$

```
void fir_xcorr16x16 ( int16_t * r, const int16_t * x, const int16_t * y,
                  int N, int M);
void fir_xcorr32x16 ( int32_t * r, const int32_t * x, const int16 t * y,
                   int N, nt M);
void fir xcorr32x32 ( int32 t * r, const int32 t * x, const int32 t * y,
                  int N, int M);
void cxfir xcorr32x32 ( complex fract32 *
                  const complex fract32 * x, const complex fract32 * y,
                  int N, int M);
                (float32 t * r,
void fir_xcorrf
                  const float32 t * x, const float32 t * y,
                  int N, int M);
void cxfir xcorrf
                  (complex float * r,
                  const complex float * x, const complex float * y,
                  int N, int M);
```

```
void fir xcorra32x16 (void
                       int32 t * r, const int32 t * x, const int16 t * y,
                       int N, int M);
void fir xcorra32x32 (void * s,
                       int32 t * r, const int32 t * x, const int32 t * y,
int N, int M);
void cxfir_xcorra32x32 (void * s, complex_fract32 * r,
                         const complex fract32 * x, const complex fract32 * y,
                         int N, int M);
void fir xcorra32x32ep(void *
                      p(void * s,
int32_t * r, const int32_t * x, const int32_t * y,
                       int N, int M);
void fir xcorraf
                      (void
                      (void     * s,
float32_t * r, const float32_t * x, const float32_t * y,
                      int N, \overline{i}nt M);
                               * s,
void cxfir xcorraf
                      (void
                      complex float *
                                        r,
                      const complex_float * x, const complex_float * y,
                      int N, int M);
```

	Name	Size	Description
Input			
int16_t,	Х	N	input data (Q15, Q31 or floating point)
int32_t,			
float32_t,			
complex_fract32,			
complex_float			
int16_t,			
float32_t,	У	M	input data (Q31, Q15 or floating point)
complex_fract32, complex float			James Communication of the Com
int.	N		less the of
			length of x
int	M		length of y
Output			
int16_t,	r	N	output data, Q15, Q31 or floating point
int32_t,			<b>3 7 1 1 1 1 1 1 1 1 1 1</b>
float32_t,			
complex_fract32,			
complex_float			
Temporary			
void	S		Scratch memory,
			FIR XCORRA16X16 SCRATCH SIZE(N, M)
			FIR_XCORRA32X16_SCRATCH_SIZE(N,M)
			FIR_XCORRA32X32_SCRATCH_SIZE(N, M)
			FIR_XCORRA32X32EP_SCRATCH_SIZE(N, M)
			FIR_XCORRAF_SCRATCH_SIZE(N, M)
			CXFIR_XCORRA32X32_SCRATCH_SIZE(N, M)
			CXFIR_XCORRAF_SCRATCH_SIZE(N, M)
			bytes

#### Returned value

none

#### Restrictions

```
For slow versions (fir_xcorra16x16, fir_xcorra32x16, fir_xcorra32x32ep, fir_xcorra32x32, fir_cxcorra32x32, fir_xcorraf, cxfir_xcorraf): x,y,r,s should not overlap s should be aligned on 8-byte boundary N>=M-1

For fast versions (fir_xcorr16x16, fir_xcorr32x16, fir_xcorr32x32, fir_xcorr32x32ep, cxfir_xcorr32x32, fir_xcorrf, cxfir_xcorrf): x,y,r should not overlap x,y,r should be aligned on 8-byte boundary N,M - multiples of 4
```

#### 2.1.9 Linear Correlation

Description

Functions estimate the linear cross-correlation between vectors  $\mathbf{x}$  (of length  $\mathbf{N}$ ) and  $\mathbf{y}$  (of length  $\mathbf{M}$ ) resulting in vector  $\mathbf{r}$  of length  $\mathbf{N}+\mathbf{M}-1$ . It is a similar to convolution but  $\mathbf{y}$  is read in opposite direction.

Precision

2 variants available:

	Type Description	
16x16 16x16-bit data, 16-bit outputs		16x16-bit data, 16-bit outputs
32x32 32x32-bit data, 32-bit outputs		32x32-bit data, 32-bit outputs

Algorithm

$$r_k = \sum_{j=\max(k-M+1,0)}^{\min(N-1,k)} x_j y_{M-1-(k-j)}^*, k = 0...(M+N-2)$$

**Prototype** 

**Arguments** 

	111032_0		Const intoz_t wy, int N, int M),
Туре	Name	Size	Description
Input			
int16_t, int32_t	Х	N	input data (Q15, Q31)
int16_t, int32_t	У	М	input data (Q31, Q15)
int	N		length of x
int	М		length of y
Output			
int16_t, int32_t,	r	M+N-1	output data, Q15, Q31
Temporary			
void	S		Scratch memory,
			FIR LXCORRA16X16 SCRATCH SIZE(N, M)
			FIR_LXCORRA32X32_SCRATCH_SIZE(N, M)
			bytes

Returned value

none

Restrictions

x, y, r, s should not overlap s should be aligned on 8-byte boundary

N>0, M>0 N>=M-1

#### 2.1.10 Circular Autocorrelation

**Description** 

Estimates the auto-correlation of vector  ${\bf x}$ . Returns autocorrelation of length  ${\bf N}$ .

Two versions of these functions available: faster version ( $fir_acorr16x16$ ,  $fir_acorr32x32$ ,  $fir_acorr32x32ep$ ,  $fir_acorrf$ ) with some restrictions on input arguments and slower version ( $fir_acorra16x16$ ,  $fir_acorra32x32$ ,  $fir_acorra32x32ep$ ,  $fir_acorraf$ ) for arbitrary arguments. In addition, this slower version implementations require scratch memory area.

Precision

4 variants available:

Туре	Description		
16x16	16-bit data, 16-bit outputs		
32x32	32-bit data, 32-bit outputs		
32x32ep	32-bit data, 32-bit outputs, use 72-bit accumulators for intermediate computations.		
f	f floating point. Requires VFPU/SFPU core option		

Algorithm

$$r_k = \sum_{n=0}^{N-1} x_{\text{mod}(n+k,N)} x_n, k = \overline{0...(N-1)}$$

#### **Prototype**

```
void fir_acorr16x16 ( int16_t * r, const int16_t * x, int N);
void fir_acorr32x32 ( int32_t * r, const int32_t * x, int N);
void fir_acorr32x32ep( int32_t * r, const int32_t * x, int N);
void fir_acorrf ( float32_t* r, const float32_t* x, int N);

void fir_acorra16x16 (void* s, int16_t * r, const int16_t * x, int N);
void fir_acorra32x32 (void* s, int32_t * r, const int32_t * x, int N);
void fir_acorra32x32ep(void* s, int32_t * r, const int32_t * x, int N);
void fir_acorraf (void* s, float32_t* r, const float32_t* x, int N);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
int16_t, int32_t or float32_t	X	N	input data (Q15, Q31 or floating point)
int	N		length of x
Output			
int16_t, int32_t or float32_t	r	N	output data, Q15, Q31 or floating point
Temporary			
void	S		Scratch memory,  FIR ACORRA16X16 SCRATCH SIZE ( N )  FIR ACORRA32X32 SCRATCH SIZE ( N )  FIR ACORRAF SCRATCH SIZE ( N )  bytes

#### Returned value

none

#### Restrictions

For slow versions (fir\_acorr16x16, fir\_acorr32x32, fir\_acorr32x32ep, fir\_acorrf):

x, r, s should not overlap

 $_{\rm N}\;$  - must be non-zero

s - aligned on an 8-bytes boundary

For fast versions (fir\_acorra16x16, fir\_acorra32x32, fir\_acorra32x32ep, fir\_acorraf):

 $\mathtt{x}$ ,  $\mathtt{r}$  should not overlap

x, r should be aligned on 8-byte boundary

N - non-zero multiple of 4

#### 2.1.11 Linear Autocorrelation

Description

Functions estimate the linear auto-correlation of vector x. Returns autocorrelation of length N.

Precision

2 versions available:

Туре	Description
16x16	16-bit data, 16-bit outputs
32x32	32-bit data, 32-bit outputs

Algorithm

$$r_k = \sum_{n=0}^{N-k-1} x_{n+k} x_n, k = \overline{0...(N-1)}$$

**Prototype** 

Туре	Name	Size	Description	
Input				
int16_t, int32_t	X	N	input data (Q15, Q31)	
int	N		length of x	
Output				
int16_t, int32_t	r	N	output data, Q15, Q31	
Temporary				
void	S		Scratch memory,	
			FIR_LACORRA16X16_SCRATCH_SIZE( N )	
			FIR_LACORRA32X32_SCRATCH_SIZE( N )	
			bytes	

Returned value

none

Restrictions

x,r,s should not overlap

N>0

s - aligned on an 8-bytes boundary

#### 2.1.12 Blockwise Adaptive LMS Algorithm for Real Data

#### Description

Blockwise LMS algorithm performs filtering of reference samples x[N+M-1], computation of error e[N] over a block of input samples x[N] and makes blockwise update of IR to minimize the error output. Algorithm includes FIR filtering, calculation of correlation between the error output e[N] and reference signal x[N+M-1] and IR taps update based on that correlation. NOTES:

- 1. The algorithm must be provided with the normalization factor, which is the power of the reference signal times  $\tt N$  the number of samples in a data block. This can be calculated using the  $\tt vec\_power32x32()$  or  $\tt vec\_power16x16()$  function. In order to avoid the saturation of the normalization factor, it may be biased, i.e. shifted to the right. If it's the case, then the adaptation coefficient must be also shifted to the right by the same number of bit positions.
- 2. This algorithm consumes less CPU cycles per block than single sample algorithm at similar convergence rate.
- 3. Right selection of  $_{\rm N}$  depends on the change rate of impulse response: on static or slow varying channels convergence rate depends on selected  $_{\rm ML}$  and  $_{\rm ML}$ , but not on  $_{\rm NL}$ .
- 4. 16x16 routine may converge slower on small errors due to roundoff errors. In that cases, 16x32 routine will give better results although convergence rate on bigger errors is the same
- 5. Terms near-end and far-end come from echo cancellation theory where the LMS algorithm is used widely. For echo cancellation the term far-end means the output of speakerphone (far end designates that the origin of it is somewhere outside say came from the remote speaker). The near-end is a signal from the local microphone representing a sum of the echo, speech of local speaker and the noise. The LMS is used to estimate the equivalent impulse response of the echopath further compensation and removal of the echo from the near-end signal.

#### Precision

#### 5 variants available:

Туре	Description
16x16	16-bit coefficients, 16-bit data, 16-bit output
16x32	32-bit coefficients, 16-bit data, 16-bit output
32x32	32-bit coefficients, 32-bit data, 32-bit output, complex and real
32x32ep	32-bit data, 32-bit coefficients, 32-bit outputs, use 72-bit accumulators for intermediate computations
f	floating point, both complex and real data. Requires VFPU/SFPU core option

Algorithm

$$b = \frac{\mu}{norm}$$

$$e_n = r_n - \sum_{m=0}^{M-1} h_{M-1-m} x_{m+n}, n = \overline{0...N-1}$$

$$h_{M-1-m} = h_{M-1-m} + b \cdot \sum_{n=0}^{N-1} e_n^* x_{n+m}, m = \overline{0...M-1}$$

```
int N, int M);
void fir_blms16x32 ( int32_t * e, int32_t * h,
           const int16_t * r,
const int16_t * x,
           int32_t norm,int16_t mu,
int32_t norm, int32_t mu, int N, int M)
                             M);
void cxfir_blms32x32 (complex_fract32 * e, complex fract32 * h,
           const complex_fract32 * r, const complex_fract32 * x,
           int32_t norm, int32_t mu,
                  N, int
int32_t norm, int32_t mu, int N, int M)
           void fir_blmsf
           const float32_t * x,
           float32_t norm, float32_t mu,
const complex_float * r,
           const complex float * x,
           float32 t norm, float32 t mu,
                    N, int
```

Туре	Name	Size	Description
Input			
int16_t, int32 t,	h	М	impulse response, Q15, Q31 or floating point
complex fract32,			
float32 t,			
complex_float			
int16_t,	r	N	input (near end) data vector. First in time value is in
int32_t			r[0], Q31, Q15 or floating point
complex_fract32,			·[o], do i, d io oi nodang point
float32_t,			
complex_float		N+M-1	
int16_t, int32 t,	X	N+M-T	reference (far end) data vector. First in time value is in
complex fract32,			x[0], Q31, Q15 or floating point
float32 t,			
complex float			
int16 t,	norm		normalization factor: power of signal multiplied by N,
int32_t,			Q31, Q15 or floating point
float32_t			
int16_t	mu		adaptation coefficient in Q31, Q15 or floating point
int32_t,			(LMS step)
float32_t			
int	N		length of data block
int	M		length of h
Output			
int16_t,	е	N	estimated error, Q31, Q15 or floating point
int32_t,			g p
complex_fract32,			
float32_t,			
complex_float	<b>—</b>	.,	
int16_t,	h	M	updated impulse response, Q31, Q15 or floating point
int32_t, float32 t,			
complex float			
compies_iioat		l	

Returned value

none

Restrictions

h,x,r, e - should not overlap

 $\mathtt{x},\mathtt{e},\mathtt{h},\mathtt{r}\,$  - aligned on a 8-bytes boundary

N,M - multiples of 8

#### 2.1.13 2D Convolution

#### Description

The  $\mathtt{conv2d}()$  functions compute the two-dimensional convolution of input matrix  $\mathtt{x}[\mathtt{M}][\mathtt{N}]$  and  $\mathtt{y}[\mathtt{P}][\mathtt{Q}]$  and store the result in matrix  $\mathtt{z}[\mathtt{M}+\mathtt{P}-1][\mathtt{N}+\mathtt{Q}-1]$ 

Additional parameter rsh allows to control fixed point representation of output data.

#### Precision

#### 6 variants available:

Туре	Description
8x8	8-bit coefficients, 8-bit data, 8-bit output, Q7
8x16	8-bit coefficients Q7, 16-bit data, 16-bit output, Q15
16x16	16-bit coefficients, 16-bit data, 16-bit output, Q15
f	floating point

#### Algorithm

$$y_{i,j} = 2^{-rsh} \sum_{m = \max(i-P+1,0)}^{\min(i,M-1)} \sum_{n = \max(j-Q+1,0)}^{\min(j,N-1)} x_{m,n} \cdot y_{j-m,i-n} \text{ where } i = \overline{0...M+P-2}, j = \overline{0...N+Q-2}$$

		Scratch allocation function	Dimensio
Function	API		ns
			MxN
conv2d_3x3_8x8	conv2d_8x8	conv2d_3x3_8x8_getScratchSize	3x3
conv2d_5x5_8x8	conv2d_8x8	conv2d_5x5_8x8_getScratchSize	5x5
conv2d_11x7_8x8	conv2d_8x8	conv2d_11x7_8x8_getScratchSize	11x7
conv2d_3x3_8x16	conv2d_8x16	conv2d_3x3_8x16_getScratchSize	3x3
conv2d_5x5_8x16	conv2d_8x16	conv2d_5x5_8x16_getScratchSize	5x5
conv2d_11x7_8x16	conv2d_8x16	conv2d_11x7_8x16_getScratchSize	11x7
conv2d_3x3_16x16	conv2d_16x16	conv2d_3x3_16x16_getScratchSize	3x3
conv2d_5x5_16x16	conv2d_16x16	conv2d_5x5_16x16_getScratchSize	5x5
conv2d_11x7_16x16	conv2d_16x16	conv2d_11x7_16x16_getScratchSize	11x7
conv2d_3x3f	conv2df	conv2d_3x3f_getScratchSize	3x3
conv2d_5x5f	conv2df	conv2d_5x5f_getScratchSize	5x5
conv2d_11x7f	conv2df	conv2d_11x7f_getScratchSize	11x7

# Arguments

Туре	Nam	Size	Description
	е	Size	Description
Input			
<pre>int8_t, int16_t, float32 t</pre>	х	[M] [N]	input data, Q15, Q7 or floating point
int8_t, int16_t, float32 t	У	[P] [Q]	input data, Q15, Q7 or floating point
int	rsh		additional right shift (for fixed point API only)
	М		number of rows in the matrix x
	N		number of columns in the matrix x
int	P		number of rows in the matrix y
int	Q		number of columns in the matrix y
Temporary:			
pScr			scratch data. Should have size in bytes at least as requested by corresponding scratch allocation function
Output			
<pre>int8_t, int16_t, float32_t</pre>	Z	[M+P-1]*[N+Q-1]	output data, Q(7-rsh), Q(15-rsh), or floating point

# Returned value Restrictions

# None

M,N,P,Q n

must be positive

P, Q

must be bigger than zero and multiplies of 8

x,y,z

aligned on a 8-bytes boundary

# 2.2 IIR filters

#### 2.2.1 Bi-quad Block IIR

#### Description

Computes a IIR filter (cascaded IIR direct form I or II using 5 coefficients per bi-quad + gain term) . Input data are stored in vector  $\mathbf{x}$ . Filter output samples are stored in vector  $\mathbf{r}$ . The filter calculates  $\mathbf{N}$  output samples using SOS and G matrices.

Due to the performance reasons, IIR biquad filters may introduce additional algorithmic delay of several samples. So, all the filters are available in 2 implementations

- first fastest variant use some additional delay lines for better parallelization. Amount of above algorithmic delay depends on the filter structure, number of sections, type of the data and might be requested by xxx groupDelay() API
- another one (with suffix nd) selects slower (especially with the big number of biquads) delayless algorithms. For these functions xxx groupDelay() API functions return zero.

#### NOTES:

- Bi-quad coefficients may be derived from standard SOS and G matrices generated by MATLAB. However, typically biquad stages have big peaks in their step response which may cause undesirable overflows at the intermediate outputs. To avoid that the additional scale factors <code>coef\_g[M]</code> may be applied. These per-section scale factors may require some tuning to find a compromise between quantization noise and possible overflows. Output of the last section is directed to an additional multiplier, with the gain factor being a power of two, either negative or non-negative. It is specified through the total gain shift amount parameter <code>gain</code> of each filter initialization function
- 16x16 filters may suffer more from accumulation of the roundoff errors, so filters should be properly designed to match noise requirements

#### Precision

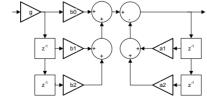
#### 4 variants available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit intermediate stage outputs (DF1, DF1 stereo, DF II form)
32x16	32-bit data, 16-bit coefficients, 32-bit intermediate stage outputs (DF1, DF1 stereo, DF II form)
32x32	32-bit data, 32-bit coefficients, 32-bit intermediate stage outputs (DF I, DF1 stereo, DF II form)
f	floating point (DF I, DF1 stereo, DF II and DF IIt). Requires VFPU/SFPU core option

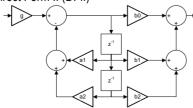
#### **Algorithm**

A block of N real input samples is sequentially passed through M bi-quad sections. There are two options for the implementation structure of a single section:

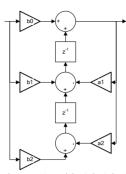
Direct Form I (DFI)



Direct Form II (DFII)



Direct Form II transposed (DF IIt)



#### **Object allocation**

```
size_t bqriir16x16_df1_alloc(int M)
size_t bqriir32x16_df2_alloc(int M)
size_t bqriir32x16_df1_alloc(int M)
size_t bqriir32x16_df2_alloc(int M)
size_t bqriir32x32_df1_alloc(int M)
size_t bqriir32x32_df2_alloc(int M)
size_t bqriirf_df1_alloc(int M)
size_t bqriirf_df2_alloc(int M)
size_t bqriirf_df2_alloc(int M)
size_t bqriirf_df2_alloc(int M)
size_t bqriirf_df1_alloc(int M)
size_t stereo_bqriir16x16_df1_alloc(int M)
size_t stereo_bqriir32x16_df1_alloc(int M)
size_t stereo_bqriir32x32_df1_alloc(int M)
size_t stereo_bqriir32x32_df1_alloc(int M)
size_t stereo_bqriirf_df1_alloc (int M)
```

#### For delayless filters:

```
size_t bqriir16x16_df1_nd_alloc(int M);
size_t bqriir16x16_df2_nd_alloc(int M);
size_t bqriir32x16_df1_nd_alloc(int M);
size_t bqriir32x16_df2_nd_alloc(int M);
size_t bqriir32x32_df1_nd_alloc(int M);
size_t bqriir32x32_df2_nd_alloc(int M);
size_t bqriirf_df1_nd_alloc(int M);
size_t bqriirf_df2_nd_alloc(int M);
size_t bqriirf_df2_nd_alloc(int M);
size_t bqriirf_df2_nd_alloc(int M);
size_t bqciirf_df1_nd_alloc(int M);
size_t stereo_bqriir16x16_df1_nd_alloc(int M);
size_t stereo_bqriir32x32_df1_nd_alloc(int M);
size_t stereo_bqriir32x32_df1_nd_alloc(int M);
size_t stereo_bqriir32x32_df1_nd_alloc(int M);
size_t stereo_bqriir6df1_nd_alloc(int M);
```

Туре	Name	Size	Description
Input			
int	М		number of bi-quad sections

#### Returns: size of memory in bytes to be allocated

#### **Object initialization**

```
const int16_t * coef_sos, const int16_t * coef_g, int16_t gain ); bqriir16x16_df2_handle_t bqriir16x16_df2_init(void * objmem, int M,
            const int16 t * coef sos, const int16 t * coef g, int16 t gain);
bqriir32x16_df1_handle_t bqriir32x16_df1_init(void * objmem, int M,
            const int16 t * coef sos, const int16 t * coef g, int16 t gain);
bqriir32x16 df2 handle t bqriir32x16 df2 init(void * objmem, int M,
            const int16_t * coef_sos, const int16_t * coef_g, int16_t gain);
bqriir32x32_df1_handle_t bqriir32x32_df1_init(void * objmem, int M,
            const int32_t * coef_sos, const int16_t * coef_g, int16_t gain)
bqriir32x32 df2 handle t bqriir32x32 df2 init(void * objmem, int M,
const int32_t * coef_sos, const int16_t * coef_g, int16_t gain)
bqriirf_df1_handle_t bqriirf_df1_init(void * objmem, int M,
            const float32 t* coef sos, int16 t gain );
bqriirf df2 handle t bqriirf df2 init(void * objmem, int M,
const float32_t * coef_sos, int16_t gain); bqriirf_df2t_handle_t bqriirf_df2t_init(void * objmem, int M,
            const float32 t * coef sos, int16 t gain);
bqciirf_df1_handle_t bqciirf_df1_init(void * objmem, int M,
            const float32 t * coef sos, int16 t gain);
```

```
stereo bgriir16x16 dfl handle t stereo bgriir16x16 dfl init
          (void * objmem, int M,
          const int16_t * coef_sosl, const int16_t * coef_gl, int16_t gainl,
const int16 t * coef sosr, const int16 t * coef gr, int16 t gainr);
stereo bqriir32x16 df1 handle t stereo bqriir32x16 df1 init
          (void * objmem, int M,
           const int16 t * coef sosl, const int16 t * coef gl, int16 t gainl,
           const int16 t * coef sosr, const int16 t * coef gr, int16 t gainr);
stereo_bqriir32x32_df1_handle_t stereo_bqriir32x32_df1_init
          (void * objmem, int M,
          const int32 t * coef sosl, const int16 t * coef gl, int16 t gainl,
           const int32 t * coef sosr, const int16_t * coef_gr, int16_t gainr);
stereo bgriirf dfl handle t stereo bgriirf dfl init
          (void * objmem, int M,
           const float32 t* coef sosl, int16 t gainl,
           const float32 t* coef_sosr, int16_t gainr );
For delayless filters:
bqriir16x16 dfl nd handle t bqriir16x16 dfl nd init(void * objmem, int M,
   const int16_t * coef_sos,
const int16_t * coef_g,
    int16 t
                   gain);
bqriir16x16_df2_nd_handle_t bqriir16x16_df2_nd_init(void * objmem, int M,
    const int16 t * coef sos,
    const int16_t * coef_g,
    int16 t
                   gain);
bqriir32x16 df1 nd handle t bqriir32x16 df1 nd init(void * objmem, int M,
    const int16 t * coef sos,
    const int16_t * coef_g,
    int16 t
                   gain);
bqriir32x16 df2 nd handle t bqriir32x16 df2 nd init(void * objmem, int M,
    const int16 t * coef sos,
    const int16_t * coef_g,
    int16 t
                   gain);
bqriir32x32 df1 nd handle t bqriir32x32 df1 nd init(void * objmem, int M,
   const int32 t * coef sos,
    const int16_t * coef_g,
    int16 t
                   qain);
bqriir32x32 df2 nd handle t bqriir32x32 df2 nd init(void * objmem, int M,
    const int32 t * coef sos,
    const int16_t * coef_g,
    int16 t
                   gain);
bgriirf d\overline{f}1 nd handle t bgriirf df1 nd init(void * objmem, int M,
    const float32_t * coef_sos, int16_t gain);
bqriirf_df2t_nd_handle_t bqriirf_df2t_nd_init(void * objmem, int M,
    const float32 t * coef sos, int16 t gain);
bqciirf_df1_nd_handle_t bqciirf_df1_nd init(void * objmem, int M,
   const float32 t * coef sos, int16 t gain);
stereo bqriir16x16 df1 nd handle t stereo bqriir16x16 df1 nd init
                                           (void * objmem, int M,
    const int16 t * coef sosl, const int16_t * coef_gl, int16_t gainl,
    const int16 t * coef sosr, const int16 t * coef gr, int16 t gainr);
stereo bqriir32x16 df1 nd handle t stereo bqriir32x16 df1 nd init
                                           (void * objmem, int M,
    const int16 t * coef sosl, const int16 t * coef gl, int16 t gainl,
    const int16_t * coef_sosr, const int16_t * coef_gr, int16_t gainr);
stereo bqriir32x32 df1 nd handle t stereo bqriir32x32 df1 nd init
                                           (void * objmem, int M,
    const int32_t * coef_sosl, const int16_t * coef_gl, int16_t gainl,
    const int32 t * coef sosr, const int16 t * coef gr, int16 t gainr);
(void * objmem, int M,
    const float32 t* coef sosl, int16 t gainl,
    const float32 t* coef sosr, int16 t gainr);
```

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int	М		number of bi-quad sections

int32_t, int16_t, float32_t	coef_sos	M*5	filter coefficients stored in blocks of 5 numbers: b0 b1 b2 a1 a2. For fixed-point functions, fixed point format of filter coefficients is Q1.14 for 16x16 and 32x16, or Q1.30 for 32x32.
<pre>int32_t, int16_t, float32_t</pre>	coef_sosl	M*5	filter coefficients for the left channel (stereo filters only)
int32_t, int16_t, float32_t	coef_sosr	M*5	filter coefficients for the right channel (stereo filters only)
int16_t	coef_g	М	scale factor for each section, Q15 (for fixed-point functions only).
int16_t	coef_gl	М	scale factor for the left channel (stereo filters only)
int16_t	coef_gr	М	scale factor for the right channel (stereo filters only)
int16_t	gain		total gain shift amount, -4815
int16_t	gainl		total gain shift amount for the left channel (stereo filters only)
int16_t	gainr		total gain shift amount for the right channel (stereo filters only)

Returns: handle to the object

# Request algorithmic delay

```
size_t bqriir16x16_df1_groupDelay(bqriir16x16_df1_handle_t _bqriir);
size_t bqriir16x16_df2_groupDelay(bqriir16x16_df2_handle_t _bqriir);
size_t bqriir32x16_df1_groupDelay(bqriir32x16_df1_handle_t _bqriir);
size_t bqriir32x16_df2_groupDelay(bqriir32x16_df2_handle_t _bqriir);
size_t bqriir32x32_df1_groupDelay(bqriir32x32_df1_handle_t _bqriir);
size_t bqriir32x32_df2_groupDelay(bqriir32x32_df2_handle_t _bqriir);
size t bqriirf dfl groupDelay
                                                    (bqriirf_df1_handle_t _bqriir);
                                                     (bqriirf_df2_handle_t__bqriir);
(bqriirf_df2t_handle_t_bqriir);
(bqciirf_df1_handle_t_bqriir);
size_t bqriirf_df2_groupDelay
size_t bqriirf_df2t_groupDelay
size t bqciirf df1 groupDelay
size t stereo bqriir16x16 df1 groupDelay(stereo bqriir16x16 df1 handle t
 bariir);
size t stereo bqriir32x16 df1 groupDelay(stereo bqriir32x16 df1 handle t
bgriir);
__size_t stereo_bqriir32x32_df1_groupDelay(stereo bqriir32x32 df1 handle t
 bqriir);
size t stereo bqriirf df1 groupDelay
                                                                (stereo bqriirf dfl handle t
bqriir);
```

These functions return additional algorithmic delay introduced by the filter in samples, see more information in para 4.4

#### For delayless filters:

```
size_t bqriir16x16_df1_nd_groupDelay(bqriir16x16_df1_nd_handle_t _bqriir);
size_t bqriir16x16_df2_nd_groupDelay(bqriir16x16_df2_nd_handle_t _bqriir);
size_t bqriir32x16_df1_nd_groupDelay(bqriir32x16_df1_nd_handle_t _bqriir);
size_t bqriir32x16_df2_nd_groupDelay(bqriir32x16_df2_nd_handle_t _bqriir);
size_t bqriir32x32_df1_nd_groupDelay(bqriir32x32_df1_nd_handle_t _bqriir);
size_t bqriir32x32_df2_nd_groupDelay(bqriir32x32_df2_nd_handle_t _bqriir);
size_t bqriirf_df1_nd_groupDelay(bqriirf_df1_nd_handle_t _bqriir);
size_t bqriirf_df2_nd_groupDelay(bqriirf_df2_nd_handle_t _bqriir);
size_t bqriirf_df2_nd_groupDelay(bqriirf_df2t_nd_handle_t _bqriir);
size_t bqriirf_df1_nd_groupDelay(bqriirf_df2t_nd_handle_t _bqriir);
size_t stereo_bqriir16x16_df1_nd_groupDelay(stereo_bqriir16x16_df1_nd_handle_t _bqriir);
size_t stereo_bqriir32x16_df1_nd_groupDelay(stereo_bqriir32x16_df1_nd_handle_t _bqriir);
size_t stereo_bqriir32x32_df1_nd_groupDelay(stereo_bqriir32x32_df1_nd_handle_t _bqriir);
size_t stereo_bqriir32x32_df1_nd_groupDelay(stereo_bqriir32x32_df1_nd_handle_t _bqriir);
size_t stereo_bqriirf_df1_nd_groupDelay(stereo_bqriirf_df1_nd_handle_t _bqriir);
```

For delayless filters these functions return zero

Update the delay line and compute filter output

```
void bgriir32x16 df1 (bgriir32x16 df1 handle t bgriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriir32x16 df2 (bqriir3\overline{2}x16 df2 handle t \overline{6}qriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriir32x32_df2(bqriir32x32_df2_handle_t _bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriirf_df1 (bqriirf_df1_handle_t,
                float32 t
                           * r, const float32 t * x, int N);
void bqriirf df2 (bqriirf_df2_handle_t,
                float32 t
                           * r, const float32 t * x, int N);
void bqciirf dfl (bqciirf dfl handle t,
                complex float* r, const complex float * x, int N);
void stereo bqriir16x16 dfl(stereo bqriir16x16 dfl handle t bqriir,
            void * s, int16 t * r, const int16 t *x, int N);
void stereo bqriir32x16 df1(stereo bqriir32x16 df1 handle t bqriir,
           void * s, int32 t * r, const int32 t *x, int N);
void stereo bqriir32x32 df1 (stereo bqriir32x32 df1 handle t bqriir,
           void * s, int32 t * r, const int32 t *x, int N);
void stereo bqriirf dfl (stereo bqriirf dfl handle t,
                float32 t
                            * r, const float32 \bar{t} * x, int N);
For delayless filters:
```

```
void bariir16x16 dfl nd(bariir16x16 dfl nd handle t bariir, void * s, int16 t *
r, const int16_t *x, int N);
void bqriir16x16 df2 nd(bqriir16x16 df2 nd handle t bqriir, void * s, int16 t *
r, const int16 t *x, int N);
void bgriir32x16 df1 nd(bgriir32x16 df1 nd handle t bgriir, void * s, int32 t *
r, const int32_t *x, int N);
\label{eq:cond_problem} \mbox{void bqriir} 32 \mbox{x} 16 \mbox{\_df2} \mbox{\_nd} \mbox{\_handle} \mbox{\_t bqriir}, \mbox{ void } \star \mbox{ s, int} 32 \mbox{ t } \star \mbox{\_t bqriir}, \mbox{\sc bqriir} \mbox{\sc bqriir}, \mbox{\sc bqriir} \mbox{\sc bqriir} \mbox{\sc bqriir}, \mbox{\sc bqriir} \mbox{\sc bqriir}, \mbox{\sc bqriir} \mbox{\sc bqriir}, \mbox{\sc bqriir} \mbox{\sc bqriir}, \mbox{\sc bqriir} \mb
r, const int32 t *x, int N);
void bqriir32x32_df1_nd(bqriir32x32_df1_nd_handle_t bqriir, void * s, int32 t *
r, const int32_t *x, int N);
void bqriir32x32_df2_nd(bqriir32x32_df2_nd_handle_t _bqriir, void * s, int32_t *
r, const int32 t *x, int N);
void bqriirf df1 nd(bqriirf df1 nd handle t bqriir, float32 t
                                                                                                                                                               * r, const
float32 t
                               * x, int N);
void bqriirf_df2_nd(bqriirf_df2_nd_handle_t _bqriir, float32_t
                                                                                                                                                               * r, const
                               * x, int N);
float32 t
void bqriirf_df2t_nd(bqriirf_df2t_nd_handle_t _bqriir, float32_t
                               * x, int N);
float32 t
void bqciirf dfl nd(bqciirf dfl nd handle t bqriir, complex float * r, const
complex float * x, int N);
void stereo bqriir16x16 dfl nd(stereo bqriir16x16 dfl nd handle t bqriir, void
 * s, int16 t * r, const int16 t *x, int N);
void stereo_bqriir32x16_df1_nd(stereo_bqriir32x16_df1_nd_handle_t _bqriir, void
* s, int32 t * r, const int32 t *x, int N);
void stereo bqriir32x32_df1_nd(stereo_bqriir32x32_df1_nd_handle_t _bqriir, void
* s, int32 t * r, const int32 t *x, int N);
void stereo_bqriirf_df1_nd(stereo_bqriirf_df1_nd_handle_t __bqriir, float32_t
  r, const float32 t
                                                          * x, int N);
```

Туре	Name	Size	Description
Input	•	•	
int16_t, int32_t, float32_t, complex_float	Х	N*S	input samples, Q31, Q15 or floating point. Stereo samples go in interleaved order (left, right)
int	N		length of input sample block
	S		1 for mono, 2 for stereo API
Output			
int16_t, int32_t, float32_t, complex_float	r	N*S	output data, Q31, Q15 or floating point . Stereo samples go in interleaved order (left, right)
Temporary			
void*	5		scratch memory area (for fixed-point functions only).  Minimum number of bytes depends on selected filter structure and precision (see spreadsheet below)  If a particular macro returns zero, then the corresponding IIR doesn't require a scratch area and parameter s may hold zero

#### Returns: none

Functon	Scratch memory, bytes
bqriir16x16 df1	BQRIIR16X16 DF1 SCRATCH SIZE(N,M)
bqriir16x16 df2	BQRIIR16X16 DF2 SCRATCH SIZE(N,M)
bqriir32x16_df1	BQRIIR32X16_DF1_SCRATCH_SIZE(N,M)
bqriir32x16_df2	BQRIIR32X16_DF2_SCRATCH_SIZE(N,M)
bqriir32x32_df1	BQRIIR32X32 DF1 SCRATCH SIZE(N,M)
bqriir32x32_df2	BQRIIR32X32_DF2_SCRATCH_SIZE(N,M)
stereo_bqriir16x16_df1	STEREO_BQRIIR16X16_DF1_SCRATCH_SIZE(N,M)
stereo bqriir32x16 df1	STEREO BQRIIR32X16 DF1 SCRATCH SIZE(N,M)
stereo_bqriir32x32_df1	STEREO BQRIIR32X32 DF1 SCRATCH SIZE (N, M)
stereo_bqriirf_df1	STEREO_BQRIIRF_DF1_SCRATCH_SIZE(N,M)
For delayless filters:	
bqriir16x16_df1_nd	BQRIIR16X16_DF1_ND_SCRATCH_SIZE(N,M)
bqriir16x16_df2_nd	BQRIIR16X16_DF2 ND_SCRATCH_SIZE(N,M)
bqriir32x16_df1_nd	BQRIIR32X16_DF1 ND_SCRATCH_SIZE(N,M)
bqriir32x16_df2_nd	BQRIIR32X16_DF2 ND_SCRATCH_SIZE(N,M)
bqriir32x32_df1_nd	BQRIIR32X32_DF1 ND_SCRATCH_SIZE(N,M)
bqriir32x32_df2_nd	BQRIIR32X32_DF2_ND_SCRATCH_SIZE(N,M)
stereo_bqriir16x16_df1_nd	STEREO_BQRIIR16X16_DF1_ND_SCRATCH_SIZE(N,M)
stereo_bqriir32x16_df1_nd	STEREO_BQRIIR32X16_DF1 ND_SCRATCH_SIZE(N,M)
stereo_bqriir32x32_df1_nd	STEREO_BQRIIR32X32_DF1_ND_SCRATCH_SIZE(N,M)
stereo_bqriirf_df1_nd	STEREO_BQRIIRF_DF1_ND_SCRATCH_SIZE(N,M)

# Returned value

none

Restrictions

 $x,r,s,coef\_g,coef\_sos$  must not overlap

- $_{
  m N}~$  must be a multiple of 2
- ${\tt s}\,\,$  whenever supplied must be aligned on an 8-bytes boundary

# 2.2.2 Lattice Block Real IIR

# Description

Computes a real cascaded lattice autoregressive IIR filter using reflection coefficients stored in vector  ${\tt k}$ . The real data input are stored in vector  ${\tt x}$ . The filter output result is stored in vector  ${\tt r}$ . Input scaling is done before the first cascade for normalization and overflow protection.

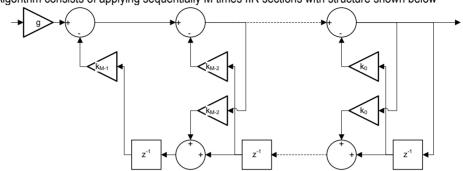
#### Precision

#### 4 variants available:

Туре	Description			
16x16	16-bit data, 16-bit coefficients			
32x16	32-bit data, 16-bit coefficients			
32x32	32-bit data, 32-bit coefficients			
f	floating point. Requires VFPU/SFPU core option			

# Algorithm

Algorithm consists of applying sequentially M times IIR sections with structure shown below



#### **Object allocation**

```
size_t latr16x16_alloc(int M);
size_t latr32x16_alloc(int M);
size_t latr32x32_alloc(int M);
size t latrf alloc (int M);
```

Туре	Name	Size	Description			
Input	Input					
int	М		number of sections			

Returns: size of memory in bytes to be allocated

#### Object initialization

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int	М		number of sections
int16_t, int32_t or float32 t	k	М	reflection coefficients, Q31, Q15 or floating point
int16_t, int32_t or float32 t	scale	М	input scale factor g, Q31, Q15 or floating point

Returns: handle to the object

# Update the delay line and compute filter output

Туре	Name	Size	Description
Input			
int16_t, int32_t or float32 t	Х	N	input samples, Q31, Q15 or floating point
int	N		length of input sample block
Output			· -
int16_t, int32_t or float32 t	r	N	output data, Q31, Q15 or floating point

Returns: none

#### Returned value

none

#### Restrictions

x,r,k should not overlap

Conditions for optimum performance

For optimum performance M should be in range 1...8

# 2.3 Mathematics

A number of DSP Library functions supersede standard floating-point mathematical functions similar to defined in <math.h>, as listed below:

ANSI function	Scalar function	reference
atanf	scl_atanf	2.3.10
atan2f	scl_atan2f	2.3.11
cosf	scl_cosinef	2.3.8
sinf	scl_sinef	2.3.8
tanf	scl_tanf	2.3.9
logf	scl_lognf	2.3.3
log2f	scl_log2f	2.3.3
log10f	scl_log10f	2.3.3
expf	scl_antilognf	2.3.4
exp2f	scl_antilog2f	2.3.4
alog10f	scl_antilog10f	2.3.4
tanhf	scl_tanhf	2.3.12
powf	scl_powf	2.3.5

All these functions conform to ISO/IEC 9899 standard (commonly referred to as C99) in respect to function semantics, parameters and return value specification. Moreover, floating-point mathematical functions handle error conditions in a way that differs from general DSP Library approach as stated in 1.8. Aforementioned functions follow the next ground rules:

- Each function executes as if it were a single operation, and may generate any of "invalid", "overflow" or "divide-by-zero" floating-point exceptions only to reflect the result of that operation.
- A domain error occurs if input argument(s) fall out of the function domain as defined in function specification. In such a case, the function assigns EDOM to the integer expression errno, raises the "invalid" floating-point exception, and returns a quiet NaN.
- NaN as an input argument is a special kind of domain error. Namely, the integer expression error acquires EDOM and returned value is a quiet NaN, but the function raises the "invalid" floating-point exception only if the input argument is a signaling NaN.
- A floating-point result overflows if the magnitude of the mathematical result is finite but so large that the target floating-point type cannot represent the mathematical result without extraordinary round-off error (for example, scl\_antilognf(100.0f)). If a function detects a floating-point result overflow, it assigns ERANGE to the integer expression error, raises the "overflow" floating-point exception and returns the properly signed infinity value.

The set of floating-point mathematical functions conforming to ISO/IEC 9899 includes vectorized variants of all the functions listed above. Due to the performance reasons, these vectorized functions do not handle  $\tt errno$  and may generate exceptions in bit different manner to minimize the overhead.

# 2.3.1 Reciprocal on Q63/Q31/Q15 Numbers

# Description

These routines return the fractional and exponential portion of the reciprocal of a vector  $\mathbf{x}$  of Q63, Q31 or Q15 numbers. Since the reciprocal is always greater than 1, it returns fractional portion  $\mathtt{frac}$  in Q(63- $\mathtt{exp}$ ), Q(31- $\mathtt{exp}$ ) or Q(15- $\mathtt{exp}$ ) format and exponent exp so true reciprocal value in the Q0.31/Q0.15 may be found by shifting fractional part left by exponent value.

NOTE:  $scl_recip64x64()$ ,  $scl_recip32x32()$  use packed output for mantissa/exponent. To take a full precision, just call vectorized counterparts.

Mantissa accuracy is 1 LSB, so relative accuracy is:

vec_recip16x16, scl_recip16x16	6.2e-5
scl_recip32x32	2.4e-7
vec_recip32x32	9.3e-10
vec_recip64x64	2.2e-19

#### Precision

#### 3 variants available:

Туре	Description	
64x64	64-bit input, 64-bit output.	
32x32	32-bit input, 32-bit output.	
16x16	16-bit input 16-bit output	

# **Algorithm**

$$frac_n \cdot 2^{exp_n} = 1/x_n, n = \overline{0...N-1}$$

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description			
Input						
int64_t, int32_t or int16_t	X	N	input data, Q63, Q31 or Q15			
int	N		length of vectors			
Output	Output					
int64_t, int32_t or int16_t	frac	N	fractional part of result, Q(63-exp), Q(31-exp) or Q(15-exp)			
int16_t	exp	N	exponent of result			

# Returned value

#### None

Restrictions

x, frac, exp should not overlap

Conditions for optimum performance

frac, x - aligned on 8-byte boundary

N - multiple of 4 and >4

#### Scalar versions

# Prototype

uint64\_t scl\_recip64x64 (int64\_t x)
uint32\_t scl\_recip32x32 (int32\_t x)
uint32\_t scl\_recip16x16 (int16\_t x)

#### **Arguments**

Туре	Name	Description
Input		
int64_t,	Х	input data, Q63, Q31 or Q15
int32 t or		
int16 t		

#### Returned value

```
packed value: scl_recip64x64(): bits 57...0 fractional part bits 63...58 exponent scl_recip32x32(): bits 23...0 fractional part bits 31...24 exponent scl_recip16x16(): bits 15...0 fractional part bits 31...16 exponent
```

# 2.3.2 Division of Q63/Q31/Q15 Numbers

#### Description

These routines perform pair-wise division of vectors written in Q63, Q31 or Q15 format. They return the fractional and exponential portion of the division result. Since the division may generate result greater than 1, it returns fractional portion frac in Q(63- $\exp$ ), Q(31- $\exp$ ) or or Q(15- $\exp$ ) format and exponent exp so true division result in the Q0.63, Q0.31 may be found by shifting fractional part left by exponent value. For division to 0, the result is not defined

Two versions of routines are available: regular versions (vec\_divide32x32, vec\_divide16x16) work with arbitrary arguments, faster versions (vec\_divide32x32\_fast, vec\_divide16x16\_fast) apply some restrictions.

NOTE:  $scl_divide32x32()$ ,  $scl_divide64x64()$  uses packed output for mantissa/exponent. To take a full precision, just call vectorized counterpart.

Mantissa accuracy is 2 LSB, so relative accuracy is:

vec_divide16x16, scl_divide16x16	1.2e-4
scl_divide32x32	4.8e-7
vec_divide32x32	1.8e-9
vec_divide64x64	4.3e-19

#### Precision

#### 4 variants available:

Туре	Description				
64x64	64-bit inputs, 64-bit output.				
64x32i	integer division, 64-bit nominator, 32-bit denominator, 32-bit output				
32x32	32-bit inputs, 32-bit output.				
16x16	16-bit inputs, 16-bit output.				

#### Algorithm

$$frac_n \cdot 2^{exp_n} = x_n / y_n, n = \overline{0...N-1}$$

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description
Input			

int64_t,	X	N	nominator, 64-bit integer, Q63, Q31 or Q15
int32_t or			
int16_t			
int64_t,	У	N	denominator, 32-bit integer, Q63, Q31 or Q15
int32 t or			
int16_t			
int	N		length of vectors
Output			
int64 t,	frac	N	fractional parts of result, Q(63-exp), Q(31-exp) or Q(15-
int32 t or			
int16_t			exp)
int16_t	exp	N	exponents of result

# Returned value

#### none

#### Restrictions

```
For regular versions (vec_divide64x32i, vec_divide64x64, vec_divide32x32, vec_divide16x16):
x,y,frac,exp should not overlap
```

For faster versions (vec\_divide32x32\_fast, vec\_divide16x16\_fast):
x,y,frac,exp should not overlap
x, y, frac to be aligned by 8-byte boundary

 ${\scriptscriptstyle \rm N}$  - multiple of 4.

#### Scalar versions

#### **Prototype**

```
uint64_t scl_divide64x64(int64_t x,int64_t y);
int32_t scl_divide64x32(int64_t x,int32_t y);
uint32_t scl_divide32x32 (int32_t x, int32_t y)
uint32_t scl_divide16x16 (int16_t x, int16_t y)
```

#### **Arguments**

Туре	Name	Description
Input		
int64_t,	Х	nominator, 64-bit integer, Q63, Q31 or Q15
int32_t or int16 t		
int64 t,	7.7	demandantan 20 hitiatanan 002 024 an 045
int32 t or	У	denominator, 32-bit integer, Q63, Q31 or Q15
int16 t		

#### Returned value

```
scl_divide64x64() packed value: bits 57...0 fractional part,bits 63...58 exponent integer remainder scl_divide32x32() packed value: bits 23...0 fractional part,bits 31...24 exponent packed value: bits 15...0 fractional part, bits 31...16 exponent
```

# 2.3.3 Logarithm

#### Description

Different kinds of logarithm (base 2, natural, base 10). 32-bit fixed point functions interpret input as Q16.15, represent results in Q6.25 format or return 0x80000000 on negative of zero input Accuracy:

floating point	2 ULP
vec log10 32x32,scl log10 32x32	230 (6.9e-6)
vec logn 32x32,scl logn 32x32	510 (1.5e-5)
vec_log2_32x32,scl_log2_32x32	730 (2.2e-5)

#### NOTES:

- 1. Floating point functions are compatible with standard ANSI C routines and set errno and exception flags accordingly.
- 2. Floating point functions limit the range of allowable input values:
  - If x<0, the result is set to NaN. In addition, scalar floating point functions assign the value EDOM to errno and raise the "invalid" floating-point exception.
  - If x==0, the result is set to minus infinity. Scalar floating point functions assign the value ERANGE to errno and raise the "divide-by-zero" floating-point exception.

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit outputs
f	floating point. Requires VFPU/SFPU core option

**Algorithm** 

$$z_n = \log_K x_n, n = \overline{0...N-1}, K = 2, e, 10$$

**Prototypes** 

```
void vec_log2_32x32 ( int32_t * z, const int32_t * x, int N);
void vec_logn_32x32 ( int32_t * z, const int32_t * x, int N);
void vec_log10_32x32 ( int32_t * z, const int32_t * x, int N);
void vec_log2f (float32_t * z, const float32_t * x, int N);
void vec_lognf (float32_t * z, const float32_t * x, int N);
void vec_log10f (float32_t * z, const float32_t * x, int N);
```

**Arguments** 

Туре	Name	Size	Description		
Input	Input				
int32_t, float32_t	Х	N	input data, Q16.15 or floating point		
int	N		length of vectors		
Output					
int32_t, float32_t	Z	N	Q6.25 or floating point		

Returned value

none

Restrictions

x, z - should not overlap

#### Scalar versions

**Prototypes** 

```
int32_t scl_log2_32x32 (int32_t x);
int32_t scl_logn_32x32 (int32_t x);
int32_t scl_log10_32x32 (int32_t x);
float32_t scl_log2f (float32_t x);
float32_t scl_lognf (float32_t x);
float32_t scl_log10f (float32_t x);
```

**Arguments** 

Туре	Name	Description		
Input				
int32_t, float32 t	Х	input data, Q16.15 or floating point		

Returned value

result, Q6.25 or floating point

# 2.3.4 Antilogarithm

Description

These routines calculate antilogarithm (base2, natural and base10). 32-bit fixed-point functions accept inputs in Q6.25 and form outputs in Q16.15 format and return 0x7FFFFFFF in case of overflow and 0 in case of underflow.

## NOTES:

1. Floating point functions are compatible with standard ANSI C routines and set errno and exception flags accordingly.

**Precision** 

2 variants available:

- 14.14.110 4.14.114.11				
Туре	Description			
32x32	32-bit inputs, 32-bit outputs. Accuracy: 8e-6*y+1LSB			
f	floating point, Accuracy: 2 ULP, Requires VFPU/SFPU core option			

Algorithm

$$y_n = 2^{x_n}$$
$$y_n = e^{x_n}$$
$$y_n = 10^{x_n}$$

#### **Prototype**

```
void vec_antilog2_32x32(int32_t * y, const int32_t* x, int N);
void vec_antilogn_32x32(int32_t * y, const int32_t* x, int N);
void vec_antilog10_32x32(int32_t* y, const int32_t* x, int N);
void vec_antilog2f (float32_t * y, const float32_t* x, int N);
void vec_antilog1f (float32_t * y, const float32_t* x, int N);
void vec_antilog10f(float32_t * y, const float32_t* x, int N);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t, float32_t	Х	N	input data,Q6.25 or floating point
int	N		length of vectors
Output			
int32_t, float32_t	У	N	output data,Q16.15 or floating point

#### Returned value

#### none

Restrictions

x,y-should not overlap

Conditions for optimum performance

 $_{\rm N}$ ,  $_{\rm Y}$  - aligned on 8-byte boundary  $_{\rm N}$  - multiple of 2

# Scalar versions

#### **Prototypes**

```
int32_t scl_antilog2_32x32 (int32_t x);
int32_t scl_antilogn_32x32 (int32_t x);
int32_t scl_antilog10_32x32(int32_t x);
float32_t scl_antilog2f (float32_t x);
float32_t scl_antilognf (float32_t x);
float32_t scl_antilog10f (float32_t x);
```

#### **Arguments**

Туре	Name	Description			
Input					
int32_t, float32 t	х	input data, Q6.25 or floating point			

## Returned value

result, Q16.15 or floating point

# 2.3.5 Power Function

#### Description

These routines calculates power function for 32-bit fixed-point or floating point numbers. For the fixed point API, The base is represented in Q31, the exponent is represented in Q6.25. Results are represented as normalized fixed point number with separate mantissa in Q31 and exponent..

#### Notes:

- 1. Scalar floating point raise to a power function conform to ANSI C requirements on standard math library functions in respect to treatment of errno and floating-point exceptions. Vectorized function does not touch errno and may raise or not raise floating point exceptions.
- 2. For floating point API, If x<0 is finite, y is finite and not an integer value, then the respective result z is set to NaN, errno is assigned the value EDOM, and the "invalid" floating-point exception is raised.
- 3. For fixed point API, function returns zero for all non-positive x. Fixed point functions never touch errno

## Special cases:

x	у	Result	Extra conditions
floating poir	nt functions	3	
±0	У	±∞	odd y<0
±0	У	+∞	even y<0
±0	У	±0	odd y>0
±0	У	0	even y>0
±1	±∞	1	
1	У	1	any y including NaN

Х	±0	1	any x including NaN
Х	У	NaN	finite x<0 and finite non-integer y
			(see note 2 above)
Х	-∞	+∞	x <1
Х	-∞	0	x >1
Х	+∞	0	x <1
Х	+∞	+∞	x >1
-∞	У	-0	y an odd integer <0
-∞	У	0	y<0 and not an odd integer
-∞	У	-∞	y an odd integer >0
-∞	У	+∞	y>0 and not an odd integer
+∞	У	0	y<0
+∞	У	+∞	у>0
fixed poi	int functions		
Х	У	0	x<=0

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit outputs. Accuracy: 2 ULP
f	floating point input, floating point output, 2 ULP under condition that $ y  \le 100$

Algorithm

Fixed point API:

$$m_n \cdot 2^{e_n} = x_n^{y_n}$$

Floating point API:

$$z_n = x_n^{y_n}$$

**Prototype** 

# **Arguments**

Туре	Name	Size	Description
Input			
uint32_t, float32_t	Х	N	input data,Q1.31 (unsigned) or floating point
int32_t	У	N	input data,Q6.25 or floating point
int	N		length of vectors
Output			
int32_t	m	N	mantissa of output, Q31 - for fixed point API
int16_t	е	N	exponent of output - for fixed point API
float32_t	Z	N	result: for floating point API

Returned value

none

Restrictions

x, y, z, m – should not overlap

Prototype

float32\_t scl\_powf(float32\_t x, float32\_t y)

**Arguments** 

Туре	Name	Size	Description		
Input					
float32_t	Х		base		
float32_t	У		power		

Returned value

result

# 2.3.6 Square Root

Description

These routines calculate square root.

NOTE: functions return 0x80000000 on negative argument for 32-bit outputs or 0x8000 for 16-bit outputs

Two versions of functions available: regular version (vec\_sqrt16x16, vec\_sqrt32x32,

vec\_sqrt64x32, vec\_sqrt32x16) with arbitrary arguments and faster version
(vec\_sqrt32x32\_fast) that apply some restrictions.

# Precision

#### 4 variants available:

Туре	Description
16x16	16-bit inputs, 16-bit output. Accuracy: 2 LSB
32x32	32-bit inputs, 32-bit output. Accuracy: (2.6e-7*y+1LSB)
32x16	32-bit input, 16-bit output. Accuracy: 2 LSB
64x32	64-bit input, 32-bit output. Accuracy: 2 LSB

#### Algorithm

$$y_n = \sqrt{x_n}$$

# **Prototype**

#### **Arguments**

Туре	Name	Size	Description	
Input				
int64_t,	Х	N	input data, Q63, Q31, Q15	
int32_t,				
int16_t				
int	N		length of vectors	
Output				
int32_t,	У	N	output data, Q31, Q15	
int16 t				

# Returned value

#### none

## Restrictions

Regular versions (vec\_sqrt16x16, vec\_sqrt32x32, vec\_sqrt32x16, vec\_sqrt64x32):

x,y - should not overlap

Faster versions (vec sqrt32x32 fast):

x, y - should not overlap

x, y - aligned on 8-byte boundary

 $_{
m N}\,$  - multiple of 2

#### Scalar versions

# **Prototypes**

int16\_t scl\_sqrt16x16(int16\_t x);
int16\_t scl\_sqrt32x16(int32\_t x);
int32\_t scl\_sqrt32x32(int32\_t x);
int32\_t scl\_sqrt64x32(int64\_t x);

#### **Arguments**

Туре	Name	Description		
Input				
int64_t,	Х	input data, Q63, Q31, Q15		
int32_t,				
int.16 t.				

# Returned value

result, Q31, Q15

# 2.3.7 Reciprocal Square Root

#### Description

These routines return the fractional and exponential portion of the reciprocal square root of a vector  $\mathbf{x}$  of Q31 or Q15 numbers. Since the reciprocal square root is always greater than 1, they return fractional portion  $\mathtt{frac}$  in Q(31- $\mathtt{exp}$ ) or Q(15- $\mathtt{exp}$ ) format and exponent exp so true reciprocal value in the Q0.31/Q0.15 may be found by shifting fractional part left by exponent value.

NOTE:  $scl_rsqrt32x32()$  uses packed output for mantissa/exponent. To take a full precision, just call vectorized counterpart.

Mantissa accuracy is 1 LSB, so relative accuracy is:

vec_rsqrt16x16, scl_rsqrt16x16	6.2e-5
scl_rsqrt32x32	2.4e-7

# vec rsqrt32x32

#### Precision

#### 2 variants available:

Туре	Description
32x32	32-bit input, 32-bit output.
16x16	16-bit input, 16-bit output.

9.2e-10

#### **Algorithm**

$$frac_n \cdot 2^{exp_n} = 1/\sqrt{x_n}, n = \overline{0...N-1}$$

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t,int16_t	X	N	input data, Q31 or Q15
int	N		length of vectors
Output			
int32_t, int16_t	frac	N	fractional part of result, Q(31-exp) or Q(15-exp)
int16_t	exp	N	exponent of result

#### Returned value

#### None

#### Restrictions

x, frac, exp should not overlap

#### Scalar versions

#### **Prototype**

uint32\_t scl\_rsqrt32x32 (int32\_t x)
uint32\_t scl\_rsqrt16x16 (int16 t x)

#### **Arguments**

Туре	Name	Description	
Input			
int32_t, int16 t	Х	input data, Q31 or Q15	

#### Returned value

#### packed value:

scl\_rsqrt32x32(): bits 23...0 fractional part bits 31...24 exponent scl\_rsqrt16x16(): bits 15...0 fractional part bits 31...16 exponent

# 2.3.8 Sine/Cosine

#### Description

Fixed-point functions calculate  $\sin(pi^*x)$  or  $\cos(pi^*x)$  for numbers written in Q31 format. Return results in the same format. Floating point functions compute  $\sin(x)$  or  $\cos(x)$ .

Two versions of functions available: regular version (vec\_sine32x32, vec\_cosine32x32, vec\_sinef) with arbitrary arguments and faster version (vec\_sine32x32\_fast, vec\_cosine32x32\_fast) that apply some restrictions.

#### NOTE:

- 1. Scalar floating point functions are compatible with standard ANSI C routines and set errno and exception flags accordingly.
- 2. Floating point functions limit the range of allowable input values: [-102940.0, 102940.0] Whenever the input value does not belong to this range, the result is set to NaN.

# Precision

#### 2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 1700 (7.9e-7)
f	floating point. Accuracy 2 ULP. Requires VFPU/SFPU core option

# **Algorithm**

For fixed point:

$$y_n = \sin(\pi x_n), n = \overline{0...N - 1} \text{ or}$$
$$y_n = \cos(\pi x_n), n = \overline{0...N - 1}$$

For floating point:

$$y_n = \sin(x_n), n = \overline{0...N - 1} \text{ or}$$
$$y_n = \cos(x_n), n = \overline{0...N - 1}$$

#### **Prototypes**

```
void vec_sine32x32 ( int32_t * y, const int32_t * x, int N);
void vec_cosine32x32(int32_t * y, const int32_t * x, int N);
void vec_sinef ( float32_t * y, const float32_t * x, int N);
void vec_cosinef (float32_t * y, const float32_t * x, int N);
void vec_sine32x32_fast (int32_t * y, const int32_t * x, int N);
void vec_cosine32x32_fast(int32_t * y, const int32_t * x, int N);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t, float32_t	Х	N	input data,Q31 or floating point
int	N		length of vectors
Output			
int32_t, float32_t	У	N	Result,Q31 or floating point

## Returned value

None

#### Restrictions

Regular versions (vec\_sine32x32, vec\_cosine32x32, vec\_sinef, vec\_cosinef): x,y - should not overlap

Faster versions (vec\_sine32x32\_fast, vec\_cosine32x32\_fast): x,y - should not overlap

x, y - aligned on 8-byte boundary

N - multiple of 2

#### Scalar versions

#### **Prototypes**

int32\_t scl\_sine32x32 (int32\_t x);
int32\_t scl\_cosine32x32 (int32\_t x);
float32\_t scl\_sinef (float32\_t x);
float32\_t scl\_cosinef (float32\_t x);

#### Arguments

Туре	Name	Description	
Input			
int32_t, float32_t	Х	input data, Q31 or floating point	

#### Returned value

result, Q31 or floating point

# 2.3.9 Tangent

# Description

Fixed point functions calculate tan(pi\*x) for number written in Q31. Floating point functions compute tan(x).

# NOTE:

- 1. Scalar floating point function is compatible with standard ANSI C routines and sets errno and exception flags accordingly.
- 2. Floating point functions limit the range of allowable input values: [-9099, 9099]. Whenever the input value does not belong to this range, the result is set to NaN.

#### Precision

2 variants available:

Type Description
------------------

32x32	32-bit inputs, 32-bit outputs. Accuracy: (1.3e- $4*_{y}$ +1 LSB) if abs (y) <=464873 (14.19
	in Q15) or abs(x) <pi*0.4776< th=""></pi*0.4776<>
f	floating point, Accuracy: 2 ULP. Requires VFPU/SFPU core option

Algorithm

for fixed point:

$$y_n = \tan(\pi x_n), n = \overline{0...N-1}$$

for floating point

$$y_n = \tan(x_n), n = \overline{0...N-1}$$

**Prototype** 

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t, float32_t	Х	N	input data, Q31 or floating point
int	N		length of vectors
Output			
int32_t, float32_t	У	N	result, Q16.15 or floating point

Returned value

none

Restrictions

x, y - should not overlap

Conditions for optimum

x, y - aligned on 8-byte boundary

N - multiple of 2

performance

# Scalar versions

**Prototype** 

int32\_t scl\_tan32x32 (int32\_t x);
float32\_t scl\_tanf (float32\_t x);

**Arguments** 

Туре	Name	Description	
Input			
int32_t,	Х	input data,Q31 or floating point	
float32_t			

Returned value

result, Q16.15 or floating point

# 2.3.10 Arctangent

# Description

Functions calculate arctangent of number. Fixed point functions scale down the output to pi

NOTE:

1. Scalar floating point function is compatible with standard ANSI C routines and sets  ${\tt errno}$  and exception flags accordingly

**Precision** 

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 42 (2.0e-8)
f	floating point. Accuracy: 2 ULP. Requires VFPU/SFPU core option

## **Algorithm**

for fixed point

$$z_n = \arctan(x_n)/\pi, n = \overline{0...N-1}$$

for floating point

$$z_n = \arctan(x_n), n = \overline{0...N-1}$$

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description	
Input				
int32_t, float32_t	Х	N	input data, Q31 or floating point	
int	N		length of vectors	
Output				
int32_t, float32_t	Z	N	result, Q31 or floating point	

#### Returned value

#### None

Restrictions x, z should not overlap

Conditions for optimum performance

 $_{\rm X}$ ,  $_{\rm Z}$  aligned on 8-byte boundary

N multiple of 2

#### Scalar versions

## **Prototype**

int32\_t scl\_atan32x32 (int32\_t x);
float32\_t scl\_atanf (float32\_t x);

#### **Arguments**

Туре	Name	Description	
Input			
int32_t,	Х	input data, Q31 or floating point	
float32_t			

# Returned value

result, Q31 or floating point

# 2.3.11 Full Quadrant Arctangent

# Description

The functions compute the full quadrant arc tangent of the ratio y/x. Floating point functions is in radians. Fixed point functions scale its output by pi.

#### NOTE:

- 1. Scalar floating point function is compatible with standard ANSI C routines and sets errno and exception flags accordingly
- 2. Scalar floating point function assigns EDOM to errno whenever y==0 and x==0.

#### Precision

#### 1 variant available:

Туре	Description
f	floating point. Accuracy: 2 ULP. Requires VFPU/SFPU core option

## Algorithm

$$z_n = \arctan(y_n / x_n), n = \overline{0...N - 1}$$

#### **Prototype**

void vec\_atan2f (float32\_t \* z, const float32\_t \* y, const float32\_t \* x,int N);

# **Arguments**

Туре	Name	Size	Description	
Input				
float32_t	X	N	input data, Q31 or floating point	
float32_t	У	N	input data, Q31 or floating point	
int	N		length of vectors	
Output				
float32_t	Z	N	result, Q31 or floating point	

#### Returned value

#### None

Restrictions

x, y, z should not overlap

#### Scalar versions

**Prototype** 

float32\_t scl\_atan2f (float32\_t y, float32\_t x);

**Arguments** 

Туре	Name	Description	
Input			
float32_t	У	input data, Q31 or floating point	
float32_t	Х	input data, Q31 or floating point	

Returned value

result, Q31 or floating point

# 2.3.12 Hyperbolic Tangent

**Description** The functions compute the hyperbolic tangent of input argument. 32-bit fixed-point functions accept inputs

in Q6.25 and form outputs in Q16.15 format.

Precision 2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 2 LSB
f	floating point input, floating point output, Accuracy: 2 ULP

Algorithm

$$y_n = \tanh(x_n), n = \overline{0...N - 1}$$

Prototype

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t, float32_t	Х	N	input data, Q6.25 or floating point
int	N		length of vectors
Output			
int32_t, float32 t	У	N	result, Q16.15 or floating point

Returned value

None

Restrictions

x, y should not overlap

#### Scalar versions

**Prototype** 

int32\_t scl\_tanh32x32 (int32\_t x);
float32\_t scl\_tanhf (float32\_t x);

**Arguments** 

Туре	Name	Description		
Input				
int32_t, float32 t	Х	input data, Q6.25 or floating point		

Returned value

result, Q16.15 or floating point

# 2.3.13 Sigmoid

Description

The functions compute the sigmoid of input argument. 32-bit fixed-point functions accept inputs in Q6.25 and form outputs in Q16.15 format.

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 2 LSB
f	floating point input, floating point output. Accuracy 2 ULP

Algorithm

$$y_n = \frac{1}{1 + \exp(-x_n)}, n = \overline{0...N - 1}$$

Prototype  $\begin{array}{c} \text{void vec\_sigmoid32x32} \quad \text{(int32\_t} \quad * \text{ y, const int32\_t} \quad * \text{ x,int N);} \\ \text{void vec sigmoidf} \quad \text{(float32 t * y, const float32 t * x,int N);} \\ \end{array}$ 

**Arguments** 

Туре	Name	Size	Description	
Input				
int32_t,	Х	N	input data, Q6.25 or floating point	
float32_t			<b>01</b>	
int	N		length of vectors	
Output				
int32_t,	У	N	result, Q16.15 or floating point	
float32_t			ζη	

Returned value

None

Restrictions

x, y should not overlap

Scalar versions

Prototype int32\_t scl\_sigmoid32x32 (int32\_t x);
float32 t scl sigmoidf (float32 t x);

**Arguments** 

Туре	Name	Description	
Input			
int32_t, float32_t	х	input data, Q6.25 or floating point	

Returned value

result, Q16.15 or floating point

#### 2.3.14 Rectifier function

Description

The functions compute the rectifier linear unit function of input argument. 32-bit fixed-point functions accept inputs in Q6.25 and form outputs in Q16.15 format. Parameter  $\kappa$  allows to set upper threshold for proper compression of output signal.

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 2 LSB
f	floating inputs, floating output. Accuracy: 2 ULP

Algorithm

 $y_n = \max(0, \min(x, K), n = 0...N - 1$ 

**Prototype** 

**Arguments** 

Туре	Name	Size	Description	
Input				
int32_t, float32_t	х	N	input data, Q6.25 or floating point	
int32_t, float32_t	K		threshold, Q16.15 or floating poin	
int	N		length of vectors	
Output				
int32_t, float32 t	У	N	result, Q16.15 or floating poin	

Returned value

None

Restrictions

x, y should not overlap

**Scalar versions** 

Prototype int32\_t scl\_relu32x32 (int32\_t x, int32\_t K); float32\_t scl\_reluf (float32\_t x, float32\_t K);

**Arguments** 

Type Name Description

Input		
int32_t	X	input data, Q6.25 or floating point
int32_t	K	threshold, Q16.15 or floating point

threshould

Returned value

result, Q16.15 or floating point

#### 2.3.15 Softmax

**Description** 

The function computes the softmax (normalized exponential function) of input data. 32-bit fixed-point functions accept inputs in Q6.25 and form outputs in Q16.15 format.

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output. Accuracy: 2 LSB (see Note below)
f	floating point input, floating point output

Note: Accuracy of function may depend on amount of data and their distribution. Given accuracy is achieved for N=2 for any pair of data from input domain.

**Algorithm** 

$$y_n = \frac{\exp(x_n)}{\sum_{k} \exp(x_k)}, n = \overline{0...N-1}$$

**Prototype** 

**Arguments** 

Туре	Name	Size	Description	
Input				
int32_t, float32 t	х	N	input data, Q6.25 or floating point	
int	N		length of vectors	
Output				
int32_t, float32_t	У	N	result, Q16.15 or floating point	

Returned value

None

Restrictions

x, y should not overlap

# 2.3.16 Integer to Float Conversion

Description

Routine converts integer to float and scales result up by 2^t.

**Precision** 

1 variant available:

T Variant a Vanas	<u>~</u>					
Туре	Description					
f	32-bit input, floating point output. Requires VFPU/SFPU core option					

Algorithm Prototype

$$y_n = x_n \cdot 2^t, n = 0...N - 1$$
void vec\_int2float
( float32\_t \* y,
const int32\_t \* x,
int t, int N);

**Arguments** 

Туре	Name	Size	Description	
Input				
int32_t	Х	N	input data, integer	
int	t		scale factor	
int	N		length of vectors	
Output				
float32_t	У	N	Conversion result, floating point	

Returned value None

Restrictions t should be in range -126...126

Scalar version

Arguments Type Name Description

Input
int32\_t x input data, integer

Returned value result, floating point

Restrictions t should be in range -126...126

# 2.3.17 Float to Integer Conversion

**Description** Routine scales floating point input down by 2^t and converts it to integer with saturation

Precision 1 variant available:

Type Description
f floating point input, 32-bit output. Requires VFPU core option

Algorithm  $y_n = x_n \cdot 2^{-t}, n = \overline{0...N-1}$ 

Prototype void vec\_float2int

( int32\_t \* y,
 const float32\_t \* x,
 int t, int N);

**Arguments** 

Туре	Name	Size	Description	
Input				
float32_t	Х	N	input data, floating point	
int	t		scale factor	
int	N		length of vectors	
Output				
int32_t	У	N	Conversion results, integers	

Returned value None

**Restrictions** t should be in range -126...126

Scalar version

**Arguments** 

Туре	Name	Description	
Input			
float32_t	X	input data, floating point	

Returned value result, integer

**Restrictions** t should be in range -126...126

# 2.4 Complex Mathematics

# 2.4.1 Complex Magnitude

**Description** Routines compute complex magnitude or its reciprocal

Precision 1 variant available:

Туре	Description
f	floating point input, 32-bit output. Requires VFPU/SFPU core option

Algorithm  $y_n = abs(x_n), n = \overline{0...N-1}$ 

 $y_n = 1/abs(x_n), n = \overline{0...N-1}$ 

Prototype void vec\_complex2mag (float32\_t \* y, const complex\_float \* x, int N); void vec\_complex2invmag (float32\_t \* y, const complex\_float \* x, int N);

Arguments Type Name Size Description

 Input
 x
 N
 input data

 int
 N
 length of vectors

 Output
 If loat 32\_t
 Y
 N
 magnitude or its reciprocal

Returned value None
Restrictions None

**Scalar version** 

Prototype float32\_t scl\_complex2mag (complex\_float x); float32\_t scl\_complex2invmag (complex\_float x);

Arguments Type Name Description

Returned value result, floating point

Restrictions None

# 2.5 Vector Operations

## 2.5.1 Vector Dot Product

#### Description

These routines take two vectors and calculates their dot product. Two versions of routines are available: regular versions (vec dot32x16, vec dot32x32, vec dot16x16, vec dotf) work with arbitrary arguments, faster versions (vec dot32x16 fast, vec dot32x32\_fast, vec\_dot16x16\_fast) apply some restrictions.

#### Precision

#### 7 variants available:

Туре	Description
64x32	64x32-bit data, 64-bit output (fractional multiply Q63xQ31->Q63)
64x64	64x64-bit data, 64-bit output (fractional multiply Q63xQ63->Q63)
64x64i	64x64-bit data, 64-bit output (low 64 bit of integer multiply)
32x16	32x16-bit data, 64-bit output
32x32	32x32-bit data, 64-bit output
16x16	16x16-bit data, 64-bit output for regular version and 32-bit for fast version
f	floating point. Requires VFPU/SFPU core option

# **Algorithm**

$$r = \sum_{n=0}^{N-1} x_n y_n$$

#### **Prototype**

```
int64_t vec_dot16x16 (const int16_t  *x, const int16_t  *y, int N);
int64_t vec_dot32x32 (const int32_t  *x, const int32_t  *y, int N);
                       (const float\overline{3}2 t *x, const float\overline{3}2 t* y, int N);
float32 t vec dotf
int64_t vec_dot64x32_fast (const int64_t * x,const int32_t * y,int N);
int64 t vec dot64x64 fast (const int64 t * x, const int64 t * y, int N);
int64 t vec dot64x64i fast(const int64 t * x, const int64 t * y, int N);
int64_t vec_dot32x16_fast (const int32_t * x, const int16_t * y, int N); int64_t vec_dot32x32_fast (const int32_t * x, const int32_t * y, int N); int32_t vec_dot16x16_fast (const int16_t * x, const int16_t * y, int N);
```

## **Arguments**

Туре	Name	Size	Description
Input			
int64_t, int32_t, int16_t, float32_t	х	N	input data, Q63, Q31, Q15 or floating point
int64_t, int16_t, float32_t	У	N	input data, Q63, Q31, Q15 or floating point
int	N		length of vectors

#### Returned value

dot product of all data pairs, Q63, Q31 or floating point

# Restrictions

Regular versions (vec dot64x32, vec dot64x64, vec dot64x64i, vec dot32x16, vec dot32x32, vec dot16x16, vec dotf):

None

Faster versions (vec dot64x32 fast, vec dot64x64 fast, vec dot64x64i fast, vec dot32x16 fast, vec dot32x32 fast, vec dot16x16 fast): x, y - aligned on 8-byte boundary  ${\scriptscriptstyle \rm N}$  - multiple of 4

vec dot16x16 fast utilizes 32-bit saturating accumulator, so, input data should be scaled properly to avoid erroneous results especially in case of heterogenic data

# 2.5.2 Batch Computation of Vector Dot Products

# Description

These routines take a set of input vectors and compute their dot product with specific reference data. Two versions of routines are available:

- regular versions (vec\_dot\_batch8x8, vec\_dot\_batch8x16, vec\_dot\_batch16x16, vec\_dot\_batchf). They work with arbitrary arguments
- fast versions (vec\_dot\_batch8x8\_fast, vec\_dot\_batch8x16\_fast, vec\_dot\_batch16x16 fast, vec\_dot\_batchf\_fast) apply some restrictions.

#### Precision

#### 4 variants available:

Туре	Description
8x8	8x8-bit data, 16-bit output (fractional multiply Q7xQ7->Q15)
8x16	8x16-bit data, 16-bit output (fractional multiply Q7xQ15->Q15)
16x16	16x16-bit data, 16-bit output (fractional multiply Q15xQ15->Q31)
f	single precision floating point

# Algorithm

$$z = 2^{-rsh} \sum_{n=0}^{N-1} x_n y_n^{(m)}$$

# **Prototype**

```
typedef const int8 t * cint8ptr t;
typedef const int16 t * cint16ptr t;
typedef const float32 t * cfloat32ptr t;
void vec dot batch8x8 (int16 t *z, const int8 t * x,
                        const cint8ptr t * y,int rsh, int N, int M);
void vec_dot_batch8x16 (int16_t *z, const int8_t * x,
const cint16ptr_t* y,int rsh, int N, int M);
void vec_dot_batch16x16(int32_t *z, const int16_t * x,
                        const cint16ptr t * y,int rsh, int N, int M);
void vec_dot_batchf (float32_t *z, const float32_t * x,
                        const cfloat32ptr t * y, int N, int M);
void vec dot batch8x8 fast (int16 t *z, const int8 t * x,
                        const cint8ptr t * y,int rsh, int N, int M);
void vec dot batch8x16 fast (int16 t ^{*}z, const int8 t ^{*}x,
                             const cint16ptr_t * y, int rsh, int N, int M);
void vec dot batch16x16 fast(int32 t *z, const int16 t * x,
                             const cint16ptr t * y,int rsh, int N, int M);
void vec dot batchf fast (float32 t *z, const float32 t * x,
                             const cfloat32ptr t * y, int N, int M);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
<pre>int8_t int16_t, float32_t</pre>	Х	N	reference vector, Q7, Q15 or floating point
<pre>const int8_t, const int16_t, const float32_t</pre>	У	[M] [N]	pointers to M input vectors, Q7, Q15 or floating point
int	rsh		additional right shift (for fixed point only)
int	N		length of vectors
int	М		number of vectors
Output:			
int16_t, int32_t, float32_t	Z	М	dot products between reference vector and M input vector, Q15, Q31 or floating point

#### Returned value

#### none

#### Restrictions

Regular versions: none

Fast versions:

x, y[m] aligned on 8-byte boundary

 $^{\mathbb{N}}$  multiple of 8  $^{\mathbb{M}}$  multiple of 4

#### 2.5.3 Vector Sum

#### Description

This routine makes pair wise saturated summation of vectors. Two versions of routines are available: regular versions (vec add32x32, vec add16x16, vec addf) work with arbitrary arguments, faster versions (vec add32x32 fast, vec add16x16 fast) apply some restrictions.

#### Precision

#### 3 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output
16x16	16-bit inputs, 16-bit output
f	floating point. Requires VFPU/SFPU core option

#### **Algorithm**

$$z_n = x_n + y_n, n = 0...N - 1$$

## **Prototype**

```
void vec_add32x32 ( int32_t* z, const int32_t* x, const int32_t* y, int N); void vec_add16x16 ( int16_t* z, const int16_t* x, const int16_t* y, int N); void vec_addf(float32_t* z, const float32_t* x, const float32_t* y, int N);
```

#### **Arguments**

Туре	Name	Size	Description	
Input				
int32_t, int16_t or float32_t	Х	N	input data	
int32_t, int16_t or float32 t	У	N	input data	
int	N		length of vectors	
Output				
int32_t, int16_t or float32_t	Z	N	output data	

#### Returned value

#### none

#### Restrictions

Regular versions (vec add32x32, vec add16x16, vec addf): x, y, z - should not be overlapped

Faster versions (vec add32x32 fast, vec add16x16 fast): z,x,y - aligned on 8-byte boundary N - multiple of 4

#### 2.5.4 Power of a Vector

#### Description

These routines compute power of vector with scaling output result by rsh bits. Fixed point rountines make accumulation in the 64-bit wide accumulator and output may scaled down with saturation by rsh bits. So, if representation of x input is Qx, result will be represented in Q(2x-rsh) format.

Two versions of routines are available: regular versions (vec power32x32, vec power16x16, vec powerf) work with arbitrary arguments,

faster versions (vec power32x32 fast, vec power16x16 fast) apply some restrictions.

# Precision

# 3 variants available:

Type Description		Description
32x32 32x32-bit data, 64-bit ou		32x32-bit data, 64-bit output
16x16 16x16-bit data, 6		16x16-bit data, 64-bit output
	f	floating point. Requires VFPU/SFPU core option

#### **Algorithm**

$$r = \frac{1}{2^{rsh}} \sum_{n=0}^{N-1} |x_n|^2$$

#### **Prototype**

# **Arguments**

Туре	Name	Size	Description
Input			
int32_t, int16_t, float32_t	Х	N	input data, Q31, Q15 or floating point
int	rsh		right shift of result (only for fixed point routines):  for vec_power32x32(): rsh should be in range 3162  for vec_power16x16(): rsh should be in range 031
int	N		length of vector

# Returned value

Sum of squares of a vector, Q(2x-rsh) or floating point

#### Restrictions

For regular versions (vec\_power32x32, vec\_power16x16, vec\_powerf): none

For faster versions (vec\_power32x32\_fast, vec\_power16x16\_fast) x - aligned on 8-byte boundary N - multiple of 4

# 2.5.5 Vector Scaling with Saturation

#### **Description**

These routines make shift with saturation of data values in the vector by given scale factor (degree of 2). Functions vec scale() make multiplication of vector to coefficient which is not a power of 2.

Two versions of routines are available: regular versions (vec\_shift32x32, vec\_shift16x16, vec\_shiftf, vec\_scale32x32, vec\_scale16x16, vec\_scalef, vec\_scale\_sf) work with arbitrary arguments, faster versions (vec\_shift32x32\_fast, vec\_scale32x32\_fast, vec\_scale16x16\_fast) apply some restrictions.

#### For floating point:

Function  $vec\_shiftf()$  makes scaling without saturation of data values in the vector by given scale factor (degree of 2). Functions  $vec\_scalef()$  and  $vec\_scale\_sf()$  make multiplication of input vector to coefficient which is not a power of 2.  $vec\_scalef()$  makes scaling without saturations,  $vec\_scalef()$  allows to saturate results on given boundaries.

# **Precision**

#### 3 variants available:

Туре	Description		
32x32 32-bit input, 32-bit output			
16x16	16-bit input, 16-bit output		
f floating point. Requires VFPU/SFPU core option			

#### Algorithm

#### **Prototype**

#### $r = x \cdot 2^t$

$I_n - \lambda_n \cdot \lambda$	
<pre>void vec_shift32x32</pre>	( int32_t * y,
	<pre>const int32_t * x, int t, int N);</pre>
<pre>void vec_shift16x16</pre>	( int16_t * y,
	<pre>const int16_t * x, int t, int N);</pre>
void vec_shiftf	( float32_t * y,
	<pre>const float32 * x, int t, int N);</pre>
<pre>void vec_shift32x32</pre>	_fast ( int32_t * y,
	<pre>const int32_t * x, int t, int N);</pre>
<pre>void vec_shift16x16</pre>	_fast ( int16_t * y,
	const int16 t * $x$ , int t, int $N$ ):

#### **Arguments**

Туре	Name	Size	Description	
Input	•	•		
int32_t, int16_t or float32_t	Х	N	input data, Q31, Q15 or floating point	
int	t		shift count. If positive, it shifts left with saturation, if negative it shifts right	
int	N		length of vector	
Output				
int32_t, int16_t or float32 t	У	N	output data, Q31, Q15 or floating point	

#### **Prototype**

#### non-power 2 scaling

#### **Arguments**

Const intio t - A, intio t S, int N),				
Туре	Name	Size	Description	
Input				
int32_t, int16 t or	X	N	input data, Q31, Q15 or floating point	
float32_t				
int16_t, float32 t	s		scale factor, Q31, Q15 or floating point	
int	N		length of vector	
float32_t	fmin		lower bound of resulted values (for vec_scale_sf() only)	
float32_t	fmax		upper bound of resulted values (for vec_scale_sf() only)	
Output				
int32_t,	У	N	output data, Q31, Q15 or floating point	
int16 t or			J	
float32 t				

#### Returned value

#### None

#### Restrictions

```
For regular versions (vec_shift32x32, vec_shift16x16, vec_shiftf, vec_scale32x32, vec_scale16x16, vec_scalef, vec_scalesf):
x,y should not overlap

chauld be in range, 21, 21 for fived point functions and 120, 146 for floating point.
```

₺ should be in range -31...31 for fixed-point functions and -129...146 for floating point

# 2.5.6 Common Exponent

# Description

These functions determine the number of redundant sign bits for each value (as if it was loaded in a 32-bit register) and returns the minimum number over the whole vector. This may be useful for a FFT implementation to normalize data.

Floating point function returns 0-floor (log2 (max (abs (x)))). Returned result will be always in range

[-129...146].

#### Special cases

Input	Result
0	0
+/-Inf	-129
NaN	0

#### NOTES:

Faster versions of functions make the same task but in a different manner – they compute exponent of maximum absolute value in the array. It allows faster computations but not bitexact results – if minimum value in the array will be -2<sup>n</sup>, fast function returns  $\max(0,30-n)$  while non-fast function returns (31-n). Functions return zero if N<=0

#### **Precision**

#### 3 variants available:

Type Description		Description
32 32-bit inputs		
16 16-bit inputs		16-bit inputs
Ī	f	floating point inputs. Requires VFPU/SFPU core option

#### Algorithm

$$z_n = \min \left( \underset{n=0...N-1}{norm}(x_n) \right)$$
 non-fast version

$$z_n = \min \left( norm(abs(x_n)) \right)$$
 fast version

$$z_n = -floor \left( \log_2(\max_{n=0...N-1}(abs(x_n))) \right)$$
 for floating point

where norm is exponent value (maximum possible shift count) for 32-bit data.

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t, int16_t, float32_t	Х	N	input data
int	N		length of vector

#### Returned value

#### minimum exponent

#### Restrictions

non-fast functions (vec\_bexp16, vec\_bexp32, vec\_bexpf):

for fast functions (vec bexp16 fast, vec bexp32x32 fast):

x, y - aligned on 8-byte boundary

N - multiple of 4

# Scalar versions

#### Prototype

int	scl	bexp32	(int32_t	x);
int	scl	bexp16	(int16 t	x);
int	scl	bexpf	(float32	t x);

#### **Arguments**

Туре	Name	Description	
Input			
int32_t, int16 t,	X	input data	

float32 t		
	float32 +	

Returned value

result

#### 2.5.7 Vector Min/Max

Description

These routines find maximum/minimum value in a vector.

Two versions of functions available: regular version (vec\_min32x32, vec\_max32x32, vec\_max16x16, vec\_min16x16, vec\_maxf, vec\_minf) with arbitrary arguments and faster version

(vec\_min32x32\_fast, vec\_max32x32\_fast, vec\_min16x16\_fast, vec\_min16x16\_fast) that apply some restrictions

NOTE: functions return zero if  ${\scriptscriptstyle N}$  is less or equal to zero

Precision

3 variants available:

Туре	Description
32x32	32-bit data, 32-bit output
16x16	16-bit data, 16-bit output
f	floating point. Requires VFPU/SFPU core option

Algorithm

$$v = \min(x_n), n = \overline{0...N-1}$$

or

$$v = \max(x_n), n = \overline{0...N-1}$$

**Prototype** 

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t, int16_t, float32_t	Х	N	input data
int	N		length of vector

Returned value

minimum or maximum value

Restrictions

```
For regular routines (vec_min32x32, vec_max32x32, vec_max16x16, vec_min16x16, vec_maxf, vec_minf):
none
For faster routines (vec_min32x32_fast, vec_max32x32_fast, vec_min16x16_fast,
```

For faster routines ( $vec_min32x32_fast$ ,  $vec_max32x32_fast$ ,  $vec_min16x16_fast$ ): x aligned on 8-byte boundary x - multiple of 4

# 2.6 Emulated Floating Point Operations

These routines make basic operations with emulated floating point data representing in pairs 32-bit mantissa/16-bit exponent. Mantissa is represented in 2's complement format (as usual integer numbers). Input data might be normalized (positive mantissa from 0x40000000 to 0x7FFFFFFF, negative mantissa from 0x80000000 to 0xBFFFFFFF) or denormalized (mantissa in range from 0xC0000000 to 0x3FFFFFFF). Denormalized numbers are allowed on input, but may cause degraded accuracy. However, on output all functions form normalized numbers.

Exponent bias is 31 so floating point number 1 might be represented as:

- mantissa 0x7FFFFFF (1 in Q31 representation), exponent 0 (this not exact 1,0 but 0,9999999993)

- mantissa 0x40000000, (1 in Q30 representation), exponent 1
- mantissa 0x20000000, (1 in Q29 representation), exponent 2 and so on.

Special numbers are represented as following:

- zero: mantissa is o, exponent does not care. Negative zero is not representable
- positive infinity: mantissa 0x7ffffffff, exponent 0x7fff
- negative infinity: mantissa 0x80000000, exponent 0x7fff

Not a numbers (NaNs) do not have special encoding and cannot be carried by this format.

Comparing with usual single precision floating point computations, these emulation functions

- have wider dynamic range (approximately from 10-9873 to 109864)
- have better resolution (31-bit mantissa vs 23-bit in single precision floating point)
- underflowed numbers are converted to zeroes so denormalized numbers are never occurred on the output (but allowable on input)
- result of zero to infinity multiply is not defined
- rounding mode is not specified

It is important to note that similarly as for usual floating point computations, the results of emulated floating point operations may depend on the data order so (a+b)+c may not be exactly the same as a+(b+c) and (a+b)\*c may not be equal to a\*c+b\*c. The difference is explained by different sequence of rounding and normalization.

# 2.6.1 Vector Addition for Emulated Floating Point

**Description** add two vectors represented in emulated floating point format

Algorithm  $z_n = x_n + y_n, n = \overline{0...N-1}$ 

Prototype

void vec\_add\_32x16ef ( int32\_t \* zmant, int16\_t \* zexp,

const int32\_t \* xmant, const int16\_t \* xexp,

const int32\_t \* ymant, const int16\_t \* yexp,

const int32\_t \* ymant, const int16\_t \* yexp,
int N);

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t,	xmant, ymant	N	mantissa of input data, Q31
int16_t,	xexp, yexp	N	exponent of input data
int	N		length of vector
Output			
int32_t,	zmant	N	mantissa of output data, Q31
int16_t,	zexp	N	exponent of output data

Returned value

none

Scalar function

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t,	xmant, ymant		mantissa of input data, Q31
int16_t,	xexp, yexp		exponent of input data
Output			
int32_t,	zmant	1	mantissa of output data, Q31
int16_t,	zexp	1	exponent of output data

Restrictions

xmant, ymant, xexp, yexp, zmant, zexp should not overlap

# 2.6.2 Vector Multiply for Emulated Floating Point

multiply two vectors represented in emulated floating point format Description

 $z_n = x_n \cdot y_n, n = \overline{0...N - 1}$ **Algorithm** 

( int32\_t \* zmant, int16\_t \* zexp, const int32\_t \* xmant, const int16\_t \* xexp, const int32\_t \* ymant, const int16\_t \* yexp, void vec\_mul\_32x16ef ( **Prototype** 

int N);

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t,	xmant, ymant	N	mantissa of input data, Q31
int16_t,	xexp, yexp	N	exponent of input data
int	N		length of vector
Output			
int32_t,	zmant	N	mantissa of output data, Q31
int16_t,	zexp	N	exponent of output data

none Returned value

void scl mul 32x16ef ( int32\_t \* zmant, int16\_t \* zexp, **Scalar function** 

int32\_t xmant, int16\_t xexp,
int32\_t ymant, int16\_t yexp) yexp);

**Arguments** 

Туре	Name	Size	Description
Input			
int32_t,	xmant,		mantissa of input data, Q31
	ymant		
int16_t,	xexp,		exponent of input data
	yexp		
Output			
int32_t,	zmant	1	mantissa of output data, Q31
int16_t,	zexp	1	exponent of output data

Restrictions

 $\verb|xmant,ymant,xexp,yexp,zmant,zexp| | \textbf{Should not overlap} \\$ 

# Vector Multiply-Accumulate for Emulated Floating Point

make multiply-accumulate with scalar for data represented in emulated floating point format Description

 $z_n = z_n + x_n \cdot y, n = \overline{0...N - 1}$ Algorithm

none

void vec\_mac\_32x16ef ( int32\_t \* zmant, int16 t \* zexp, **Prototype** const int32 t \* xmant, const int16 t \* xexp, int32 t ymant, int16 t yexp,

int N);

**Arguments** 

THE N/,			
Туре	Name	Size	Description
Input			
int32_t,	xmant	N	mantissa of input data, Q31
int16_t,	xexp	N	exponent of input data
int32_t,	ymant		mantissa of scalar, Q31
int16_t,	yexp		exponent of scalar
int	N		length of vector
Input/Output			
int32_t,	zmant	N	mantissa of input/output data, Q31
int16_t,	zexp	N	exponent of input/output data

Returned value

void scl mac 32x16ef ( int32 t \* zmant, int16 t \* zexp, Scalar function int32 t xmant, int16\_t xexp, int32 t ymant, int16\_t yexp);

## **Arguments**

Туре	Name	Size	Description
Input			
int32_t,	xmant, ymant		mantissa of input data, Q31
int16_t,	xexp, yexp		exponent of input data
Output			
int32_t,	zmant	1	mantissa of input/output data, Q31
int16_t,	zexp	1	exponent of input/output data

Restrictions

xmant, xexp, zmant, zexp should not overlap

# **Vector Dot Product for Emulated Floating Point**

dot product of input data represented in emulated floating point format Description

Algorithm

 $z = \sum_{n=0}^{N-1} x_n \cdot y_n$  void vec\_dot\_32x16ef ( int32\_t \* zmant, int16\_t \* zexp, const int32\_t \* xmant, const int16\_t \* xexp, const int32\_t \* ymant, const int16\_t \* yexp, int\_N); Prototype

**Arguments** 

Fr.	IIIC N);				
Туре	Name	Size	Description		
Input					
int32_t,	xmant	N	mantissa of input data, Q31		
int16_t,	xexp	N	exponent of input data		
int32_t,	ymant	N	mantissa of input data, Q31		
int16_t,	yexp	N	exponent of input data		
int	N		length of vector		
Input/Output					
int32_t,	zmant	1	mantissa of input/output data, Q31		
int16_t,	zexp	1	exponent of input/output data		

Returned value

none

Restrictions

xmant, ymant, xexp, yexp, zmant, zexp should not overlap

# 2.7 Matrix Operations

# 2.7.1 Matrix Multiply

# **Description**

These functions compute the expression  $z=2^{\hline 1} + x + y$  for the matrices x and y. The column dimension of x must match the row dimension of y. The resulting matrix has the same number of rows as x and the same number of columns as y.

Transposing API allows to interpret input yt as transposed matrix y.

NOTE: lsh factor is not relevant for floating point routines.

Functions require scratch memory for storing intermediate data. This scratch memory area should be aligned on 8 byte boundary and its size is calculated by corresponding funcitons

Two versions of functions available: regular version ( $mtx_mpy[t]32x32$ ,  $mtx_mpy[t]16x16$ ,  $mtx_mpy[t]8x16$ ,  $mtx_mpy[t]8x8$ ,  $mtx[t]_mpyf$ ) with arbitrary arguments and faster version ( $mtx_mpy[t]32x32_fast$ ,  $mtx_mpy[t]16x16_fast$ ,  $mtx_mpy[t]8x16_fast$ ,  $mtx_mpy[t]8x8_fast$ ,  $mtx_mpy[t]f_fast$ ) that apply some restrictions.

#### Precision

#### 5 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output
16x16	16-bit inputs, 16-bit output
8x8	8-bit inputs, 8-bit output
8x16	8/16-bit inputs, 16-bit output
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

#### For fixed-point routines:

$$z_{m,p} = 2^{lsh} \sum_{n=0}^{N-1} x_{m,n} \cdot y_{n,p}, m = 0...\overline{M-1}, p = \overline{0...P-1}$$

For floating point routines:

$$z_{m,p} = \sum_{n=0}^{N-1} x_{m,n} \cdot y_{n,p}, m = 0...\overline{M-1}, p = \overline{0...P-1}$$

#### **Prototype**

**Prototype** 

(transposing API)

```
void mtx mpy32x32 fast ( void* pScr,
                 int32_t* z,const int32_t* x, const int32 t* y,
                 int M, int N, int P, int lsh );
void mtx mpy16x16 fast ( void* pScr,
                 int16 t* z, const int16 t* x, const int16 t* y,
                 int M, int N, int P, int lsh );
void mtx mpy8x8 fast ( void* pScr,
                 int8 t* z, const int8 t* x, const int8 t* y,
                 int \overline{M}, int N, int P, \overline{I}nt lsh );
void mtx mpy8x16 fast ( void* pScr,
                 int16 t* z, const int8 t* x, const int16 t* y,
                 int M, int N, int P, int lsh );
int M, \overline{i}nt N, int P);
void mtx mpyt16x16 ( void* pScr,
                    int16_t* z, const int16_t* x, const int16_t* yt,
                    int M, int N, int P, int lsh );
void mtx mpyt8x8 ( void* pScr,
                    int8 t* z, const int8 t* x, const int8 t* yt,
                    int \overline{M}, int N, int P, int lsh );
void mtx mpyt8x16( void* pScr,
                    int16 t* z, const int8 t* x, const int16 t* yt,
                    int M, int N, int P, int lsh );
void mtx mpyt32x32 ( void* pScr,
                    int32 t* z, const int32 t* x, const int32 t* yt,
                    int M, int N, int P, int lsh );
                    void* pScr,
void mtx mpytf (
                    float32_t* z, const float32_t* x, const float32_t* yt,
                    int M, \overline{i}nt N, int P);
void mtx mpyt16x16 fast ( void* pScr,
                    int16_t* z, const int16_t* x, const int16_t* yt,
                    int M, int N, int P, int lsh );
void mtx mpyt8x8 fast ( void* pScr,
                    int8 t* z, const int8 t* x, const int8 t* yt,
                    int \overline{M}, int N, int P, int lsh);
void mtx mpyt8x16_fast ( void* pScr,
                    int16_t* z, const int8_t* x, const int16_t* yt,
                    int M, int N, int P, int lsh );
int M, int N, int P, int lsh );
void mtx mpytf fast( void* pScr,
                    float32_t* z, const float32_t* x, const float32_t* yt,
int M, int N, int P);
```

Function name	Scratch allocation function
mtx mpy16x16	mtx mpy16x16 getScratchSize
mtx_mpy8x8	mtx_mpy8x8_getScratchSize
mtx_mpy8x16	<pre>mtx_mpy8x16_getScratchSize</pre>
mtx_mpy32x32	mtx_mpy32x32_getScratchSize
mtx_mpyf	mtx_mpyf_getScratchSize
mtx_mpy16x16_fast	mtx_mpy16x16_fast_getScratchSize
mtx_mpy8x8_fast	<pre>mtx_mpy8x8_fast_getScratchSize</pre>
mtx mpy8x16 fast	mtx mpy8x16 fast getScratchSize
mtx_mpy32x32_fast	mtx_mpy32x32_fast_getScratchSize
mtx_mpyf_fast	mtx_mpyt_fast_getScratchSize
mtx_mpyt16x16	<pre>mtx_mpyt16x16_getScratchSize</pre>
mtx_mpyt8x8	<pre>mtx_mpyt8x8_getScratchSize</pre>
mtx_mpyt8x16	mtx_mpyt8x16_getScratchSize
mtx_mpyt32x32	mtx_mpyt32x32_getScratchSize
mtx_mpytf	mtx_mpytf_getScratchSize
mtx_mpyt16x16_fast	<pre>mtx_mpyt16x16_fast_getScratchSize</pre>
mtx_mpyt32x32_fast	<pre>mtx_mpyt32x32_fast_getScratchSize</pre>
mtx_mpytf_fast	mtx_mpytf_fast_getScratchSize

# **Arguments**

Туре	Name	Size	Description
Input		•	•
int8_t,	Х	M*N	input matrix,Q31,Q15, Q7 floating point
int16_t,			
int32_t,			
float32_t		N*P	
int8_t, int16 t,	У	N^P	input matrix y. Q31,Q15, Q7 floating point.
int32 t,			
float32 t			
int8 t,	yt	P*N	transposed input matrix y. Q31,Q15, Q7 floating
$int1\overline{6}$ t,	_		· · · · ·
int32_t,			point. (for transposing API only)
float32_t			
int	M		number of rows in matrix $\mathbf{x}$ and $\mathbf{z}$
int	N		number of columns in matrix x and number of rows in
			matrix <sub>V</sub>
int	P		number of columns in matrices y and z
int	lsh		left shift applied to the result (applied to the fixed-
			point functions only)
0.44			point functions only)
Output		T	
int8_t,	Z	M*P	output matrix z,Q31,Q15, Q7 floating point
int16_t, int32 t,			
float32 t			
Temporary			
void*	pScr		Scratch memory area with size in bytes defined by
	1		
			corresponding functions

#### Returned value

#### none

#### Restrictions

For regular routines (mtx\_mpy[t]32x32, mtx\_mpy[t]16x16, mtx\_mpy[t]8x8, mtx\_mpy[t]8x16, mtx\_mpy[t]f):

x, y, z should not overlap

For faster routines ( $mtx_mpy[t]32x32_fast$ ,  $mtx_mpy[t]16x16_fast$ ,  $mtx_mpy[t]8x8_fast$ ,  $mtx_mpy[t]8x16_fast$ ,  $mtx_mpy[t]f_fast$ ):

x, y, z should not overlap

x, y, z aligned on 8-byte boundary

M, N, P multiplies of 4

-31...31 for mtx\_mpy[t]32x32, mtx\_mpy[t]32x32\_fast

-15...15 for  $mtx_mpy[t]$ 16x16,  $mtx_mpy[t]$ 16x16\_fast,  $mtx_mpy[t]$ 8x8,  $mtx_mpy[t]$ 8x8\_fast,  $mtx_mpy[t]$ 8x16,  $mtx_mpy[t]$ 8x16\_fast

# 2.7.2 Matrix by Vector Multiply

#### Description

These functions compute the expression  $z = 2^{lsh} * x * y$  for the matrices x and vector y. NOTE: lsh factor is not relevant for floating point routines.

Two versions of functions available: regular version (mtx\_vecmpy32x32, mtx\_vecmpy16x16, mtx\_vecmpy8x8, mtx\_vecmpy8x16, mtx\_vecmpyf) with arbitrary arguments and faster version (mtx\_vecmpy32x32\_fast, mtx\_vecmpy16x16\_fast, mtx\_vecmpy8x8\_fast, mtx\_vecmpy8x16\_fast, mtx\_vecmpyf\_fast) that apply some restrictions.

#### Precision

#### 5 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output
16x16	16-bit inputs, 16-bit output
8x8	8-bit inputs, 8-bit output
8x16	8/16-bit inputs, 16-bit output
f	floating point. Requires VFPU/SFPU core option

**Algorithm** 

For fixed-point routines:

$$z_n = 2^{lsh} \sum_{m=0}^{M-1} x_{n,m} \cdot y_m, n = 0...\overline{N-1}$$

For floating-point routines:

$$z_{n} = \sum_{m=0}^{M-1} x_{n,m} \cdot y_{m}, n = 0...\overline{N-1}$$

**Prototype** 

```
void mtx_vecmpy32x32 ( int32_t* z,
               const int32 t* x,
               const int32_t* y,
               int M, int N, int lsh);
void mtx_vecmpy16x16 ( int16 t* z,
               const int16_t* x,
               const int16 t* y,
               int M, int \overline{N}, int lsh);
void mtx vecmpy8x8 ( int8 t* z,
               const int8_t* x,
               int M, int N, int lsh);
void mtx_vecmpy8x16( int16_t* z,
               const int8 \pm * x,
               const int16_t* y,
               int M, int \overline{N}, int lsh);
const float32 t* y,
               int M, int N);
void mtx vecmpy32x32 fast ( int32 t* z,
               const int32 t* x,
               const int32_t* y,
               int M, int \overline{N}, int lsh);
void mtx_vecmpy16x16_fast ( int16_t* z,
               const int16_t* x,
               const int16 t* y,
               int M, int N, int lsh);
void mtx vecmpy8x8 fast ( int8 t*
               const int8_t* x,
               const int8_t* y, int M, int N, int lsh);
void mtx vecmpy8x16 fast (int16 t* z,
               const int8_t * x,
const int16_t* y,
               int M, int \overline{N}, int lsh);
void mtx_vecmpyf_fast ( float32_t* z,
               const float32 t* x,
               const float32 t* y,
               int M, int N);
```

#### Arguments

Туре	Name	Size	Description
Input	•	•	
int32_t, int16_t, int8_t, float32 t	Х	M*N	input matrix,Q31, Q15, Q7 or floating point
int32_t, int16_t, int8_t, float32 t	У	N	input vector,Q31, Q15, Q7 or floating point
int	М		number of rows in matrix x
int	N		number of columns in matrix x
int	lsh		left shift applied to the result (applied to the fixed-point functions only)

int32_t, int16_t, int8_t,	Z	М	output vector,Q31, Q15, Q7 or floating point
float32_t			

#### Returned value

None

#### Restrictions

For regular routines (mtx\_vecmpy32x32, mtx\_vecmpy16x16, mtx\_vecmpy8x8, mtx\_vecmpy8x16, mtx\_vecmpyf):

x, y, z should not overlap

For faster routines (mtx\_vecmpy32x32\_fast, mtx\_vecmpy16x16\_fast, mtx\_vecmpy8x8\_fast, mtx\_vecmpy8x16\_fast, mtx\_vecmpyf\_fast):

x, y, z should not overlap

x, y, z aligned on 8-byte boundary

N, M multiples of 4

-31...31 for mtx\_vecmpy32x32, mtx\_vecmpy32x32\_fast

-15...15 for  $mtx\_vecmpy16x16$ ,  $mtx\_vecmpy16x16\_fast$ ,  $mtx\_vecmpy8x8\_fast$ ,  $mtx\_vecmpy8x16$  fast

# 2.7.3 Matrix Transpose

#### Description

These functions transpose matrices.

#### Precision

4 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit output
16x16	16-bit inputs, 16-bit output
8x8	8-bit inputs, 8-bit output
f	floating point

#### Algorithm

$$y_{n,m} = x_{m,n}, m = 0...\overline{M-1}, n = 0...\overline{N-1}$$

#### **Prototype**

```
void mtx_transpose32x32 (int32_t* y, const int32_t* x, int M, int N);
void mtx_transpose16x16 (int16_t* y, const int16_t* x, int M, int N);
void mtx_transpose8x8 (int8_t * y, const int8_t * x, int M, int N);
void mtx_transposef (float32_t* y, const float32_t * x, int M, int N);
void mtx_transpose32x32_fast (int32_t* y, const int32_t* x, int M, int N);
void mtx_transpose16x16_fast (int16_t* y, const int16_t* x, int M, int N);
void mtx_transpose8x8_fast (int8_t * y, const int8_t * x, int M, int N);
void mtx_transposef_fast (float32_t* y, const float32_t * x, int M, int N);
```

#### **Arguments**

Туре	Name	Size	Description		
Input					
<pre>int32_t, int16_t, int8_t, float32_t</pre>	X	M*N	input matrix,Q31, Q15, Q7 or floating point		
int	M		number of rows in matrix x		
int	N		number of columns in matrix x		
Output					
<pre>int32_t, int16_t, int8_t, float32_t</pre>	У	M*N	output vector,Q31, Q15, Q7 or floating point		

#### Returned value

None

#### Restrictions

For regular routines (mtx\_transpose\_32x32, mtx\_transpose\_16x16, mtx\_transpose\_8x8, mtx\_transposef):

x, y should not overlap

For faster routines (mtx\_transpose 32x32\_fast, mtx\_transpose 16x16\_fast, mtx\_transpose\_8x8\_fast, mtx\_transposef\_fast):

x, y should not overlap

x, y aligned on 8-byte boundary

N, M multiples of 4

# 2.8 Matrix Decomposition/Inversion

#### 2.8.1 Gauss-Jordan Matrix Inverse

Description

These functions implement in-place matrix inversion by Gauss elimination with full pivoting.

NOTE: user may detect "invalid" or "divide-by-zero" exception in the CPU flags which MAY indicate that inversion results are not accurate. Also it's responsibility of the user to provide valid input matrix for

inversion.

Fixed point version takes representation of input matrix and forms representation of output matrix with proper scaling.

Precision

2 variants available:

Туре	Description	
f	floating point. Requires VFPU/SFPU core option	
32x32	32-bit input, 32-bit output	

**Algorithm** 

 $y = x^{-1}$ 

**Prototype** 

Floating point API:

void rinvf (void\* pScr, float32\_t \*x);

Fixed point API:

int rinv32x32 (void\* pScr, int32\_t \*x, int qX);
int cinv32x32 (void\* pScr, complex\_fract32 \*x, int qX);

N	Function	API	Scratch allocation function
2	mtx_inv2x2f	rinvf	mtx_inv2x2f_getScratchSize
3	mtx_inv3x3f	rinvf	mtx_inv3x3f_getScratchSize
4	mtx_inv4x4f	rinvf	mtx_inv4x4f_getScratchSize
6	mtx_inv6x6f	rinvf	mtx_inv6x6f_getScratchSize
8	mtx_inv8x8f	rinvf	mtx_inv8x8f_getScratchSize
10	mtx_inv10x10f	rinvf	mtx_inv10x10f_getScratchSize
2	mtx_inv2x2_32x32	rinv32x32	mtx_inv2x2_32x32_getScratchSize
3	mtx_inv3x3_32x32	rinv32x32	mtx_inv3x3_32x32_getScratchSize
4	mtx_inv4x4_32x32	rinv32x32	mtx_inv4x4_32x32_getScratchSize
6	mtx_inv6x6_32x32	rinv32x32	mtx_inv6x6_32x32_getScratchSize
8	mtx_inv8x8_32x32	rinv32x32	mtx_inv8x8_32x32_getScratchSize
10	mtx_inv10x10_32x32	rinv32x32	mtx_inv10x10_32x32_getScratchSize
2	cmtx_inv2x2_32x32	cinv32x32	cmtx_inv2x2_32x32_getScratchSize
3	cmtx_inv3x3_32x32	cinv32x32	cmtx_inv3x3_32x32_getScratchSize
4	cmtx_inv4x4_32x32	cinv32x32	cmtx_inv4x4_32x32_getScratchSize
6	cmtx_inv6x6_32x32	cinv32x32	cmtx_inv6x6_32x32_getScratchSize
8	cmtx_inv8x8_32x32	cinv32x32	cmtx_inv8x8_32x32_getScratchSize
10	cmtx_inv10x10_32x32	cinv32x32	cmtx_inv10x10_32x32_getScratchSize

# **Arguments**

Туре	Name	Size	Description
Input		•	
float32_t, int32_t, complex fract32	Х	N*N	input matrix
int	dΧ		input matrix representation (for fixed point API only)
Output	•		
float32_t, int32_t, complex_fract32	Х	N*N	output inverted matrix
Temporary	•		•
void	pScr		scratch memory. Size in bytes is defined by corresponding scratch allocation function

Returned value

floating functions return none, fixed point functions return fixed-point representation of inverted matrix

Restrictions

none

# 2.8.2 Gauss-Jordan Matrix Equation Solver

Description

These functions implement solution of matrix equation by Gauss elimination with full pivoting. Fixed point functions take representation of input matrix and vector and form representation of output vector with proper scaling.

Precision

1 variant available:

Туре	Description	
32x32	32-bit input, 32-bit output	

Algorithm

```
y = Ax
```

**Prototype** 

Fixed point API:

N	Function	API	Scratch allocation function
2	mtx_gjelim2x2_32x32	r32x32	mtx_gjelim2x2_32x32_getScratchSize
3	mtx_gjelim3x3_32x32	r32x32	mtx_gjelim3x3_32x32_getScratchSize
4	mtx_gjelim 4x4_32x32	r32x32	mtx_gjelim4x4_32x32_getScratchSize
6	mtx_gjelim 6x6_32x32	r32x32	mtx_gjelim6x6_32x32_getScratchSize
8	mtx_gjelim 8x8_32x32	r32x32	mtx_gjelim8x8_32x32_getScratchSize
10	mtx_gjelim10x10_32x32	r32x32	mtx_gjelim10x10_32x32_getScratchSize
2	cmtx_gjelim 2x2_32x32	c32x32	cmtx_gjelim2x2_32x32_getScratchSize
3	cmtx_gjelim 3x3_32x32	c32x32	cmtx_gjelim3x3_32x32_getScratchSize
4	cmtx_gjelim 4x4_32x32	c32x32	cmtx_gjelim4x4_32x32_getScratchSize
6	cmtx_gjelim 6x6_32x32	c32x32	cmtx_gjelim6x6_32x32_getScratchSize
8	cmtx_gjelim 8x8_32x32	c32x32	cmtx_gjelim8x8_32x32_getScratchSize
10	cmtx_gjelim10x10_32x32	c32x32	cmtx_gjelim10x10_32x32_getScratchSize

#### **Arguments**

Туре	Name	Size	Description			
Input						
int32_t, complex_fract32	A	N*N	input matrix, representation is defined by parameter qA			
int32_t, complex_fract32	Х	N	input rigth side of equation, representation is defined by parameter qX			
int	qA		input matrix representation			
int	qX		input vector representation			
Output						
int32_t, complex_fract32	У	N	output vector			
Temporary						
void	pScr		scratch memory. Size in bytes is defined by corresponding scratch allocation function			

Returned value

fixed point functions return fixed-point representation of output vector

Restrictions

none

# 2.8.3 Cholesky MMSE Solver

#### Description

Compute the MMSE solution for a system of linear equations  $A^*x=B$ , where A is an MxN real (complex) matrix with M>=N and xank (A) ==N, x is an Nx1 vector of unknowns, and B is an Mx1 right hand side vector. This task is accomplished in 3 steps:

Cholesky decomposition is applied to the matrix of normal equations system, which results in an
upper triangular matrix R<sub>NxN</sub> with real and positive numbers on the main diagonal, such that

$$R_{NxN}^* \cdot R_{NxN} \cong A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN}$$

- Forward substitution step: solve  $R_{NxN}^* \cdot y_{NxP} = A_{MxN}^* B_{MxP}$  for N×1 vector y.
- Backward substitution step: solve the system  $R_{NxN} \cdot x_{NxP} = y_{NxP}$  for the Nx1 vector of unknowns x.

For a single  $M \times N$  matrix A, these 3 steps may be done simultaneously for P variants of  $M \times 1$  right hand side column vectors b gathered into an  $M \times P$  input matrix B. MMSE solution is computed independently for each of P columns, with resulting column vectors forming the solution matrix E of size E

Fixed point API requires explicit setting of fixed point representation of input/output matrices as well as for internal temporary matrices such as  $\mathbb{R}$  (Cholesky decomposition) and  $\mathbb{Y}$ .

#### Precision

#### 2 variants available:

Туре	Description			
f	floating point input, floating point output			
32x32	32-bit fixed point			

#### Algorithm

$$\begin{aligned} & \text{find } R_{NxN} = chol(A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN}) \colon \ R_{NxN}^* \cdot R_{NxN} \cong A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN} \\ & \text{solve } R_{NxN}^* \cdot y_{NxP} = A_{MxN}^* B_{MxP} \\ & \text{solve } R_{NxN} \cdot x_{NxP} = y_{NxP} \\ & \text{void rmmse} \quad \text{(void * pScr.)} \end{aligned}$$

# Prototype, floating point

N, M	Function	API	Scratch allocation function
	Real data		
4	matcholmmsesolver4x4f	rmmse	matcholmmsesolver4x4f_getScratchSize
6	matcholmmsesolver6x6f	rmmse	matcholmmsesolver6x6f_getScratchSize
8	matcholmmsesolver8x8f	rmmse	matcholmmsesolver8x8f_getScratchSize
10	matcholmmsesolver10x1 Of	rmmse	<pre>matcholmmsesolver10x10f_getScratchSiz e</pre>
	Complex data		
4	cmatcholmmsesolver4x4 f	cmmse	cmatcholmmsesolver4x4f_getScratchSize
6	cmatcholmmsesolver6x6 f	cmmse	cmatcholmmsesolver6x6f_getScratchSize
8	cmatcholmmsesolver8x8 f	cmmse	cmatcholmmsesolver8x8f_getScratchSize
10	cmatcholmmsesolver10x 10f	cmmse	<pre>cmatcholmmsesolver10x10f_getScratchSi ze</pre>

#### Prototype, fixed point

```
complex_fract32 * x,
const complex_ fract32 * A,
const complex_ fract32 * B,
   int32_t sigma2, int qRA, int qYBRA, int qXYR);
```

N, M	Function	API	Scratch allocation function
	Real 32-bit data		
4	matcholmmsesolver4x4_32x32	rmmse32	<pre>matcholmmsesolver4x4_32x32_get ScratchSize</pre>
6	matcholmmsesolver6x6_32x32	rmmse32	<pre>matcholmmsesolver6x6_32x32_get ScratchSize</pre>
8	matcholmmsesolver8x8_32x32	rmmse32	<pre>matcholmmsesolver8x8_32x32_get ScratchSize</pre>
10	<pre>matcholmmsesolver10x10_32x3 2</pre>	rmmse32	<pre>matcholmmsesolver10x10_32x32_g etScratchSize</pre>
	Complex 32-bit data		
4	cmatcholmmsesolver4x4_32x32	cmmse32	<pre>cmatcholmmsesolver4x4_32x32_ge tScratchSize</pre>
6	cmatcholmmsesolver6x6_32x32	cmmse32	<pre>cmatcholmmsesolver6x6_32x32_ge tScratchSize</pre>
8	cmatcholmmsesolver8x8_32x32	cmmse32	<pre>cmatcholmmsesolver8x8_32x32_ge tScratchSize</pre>
10	cmatcholmmsesolver10x10_32x 32	cmmse32	<pre>cmatcholmmsesolver10x10_32x32_ getScratchSize</pre>

Туре	Name	Size	Description
Parameters		•	
int	М		Dimensional parameter: number of rows in input matrix A
int	N		Dimensional parameter: number of columns in input
			matrix A
int	P		Number of columns in right-side matrix B
Input:	•	•	•
float32_t, int32_t	sigma2		Regularization term For fixed point, the representation should be $Q(2*qA-30)$
float32_t, complex_float, int32_t, complex_fract32	А	[M*N]	matrix ${\tt A}$ . For fixed point API, the representation is $Q({\tt qA})$
float32_t, complex_float, int32_t, complex fract32	В	[M*P]	Original right-side matrix B. For fixed point, the representation should be $Q({}_{\rm QB})$
int	qRA		qR-qA (for the fixed point API only); difference between fixed point representations of $R$ and $A$ matrices (for the fixed point API only). Should be equal or less than 0 (typically -2)
int	qYBRA		$q_{Y-q_B+q_R-q_A}$ , combination of fixed point representations of matrices $y$ , $p$ , $p$ , $p$ , and $p$ (for the fixed point API only)
int	qXYR		combination of fixed point representation (matrices $R$ , $X$ and $Y$ ) $QXYR=QX-QY+QR$
Output			
float32_t,	Х	[N*P]	Decision matrix x. For fixed point, the representation is
<pre>complex_float, int32 t,</pre>			Q(qx)
complex fract32			
Temporary			
,	pScr		Scratch data
			1

# Returned value

Restrictions

none

N = M = 4, 6, 8, 10P = 1

All matrices should not overlap and be aligned on 8-bytes boundary

# 2.8.4 Cholesky Decomposition for Pseudo-inversion

#### Description

Apply the Cholesky decomposition to the matrix of normal equations system associated with a complex-valued least squares problem:  $A_{MxN} \cdot X_{NxP} = B_{MxP}$ ,  $M \ge N$ .

The decomposition results in an upper triangular complex matrix  $R_{NxN}$  with real and positive numbers on the main diagonal, such that  $R^* \cdot R = A^* \cdot A + \sigma^2 \cdot I_N$ , where < ... > \* denotes the conjugate transpose of a matrix, and  $\sigma^2 \cdot I_N$  is the  $N \times N$  identity matrix multiplied with the regularization term.

#### Precision

#### 2 variants available:

Туре	Description	
f	floating point input, floating point output	
32x32	32-bit fixed point	

#### **Algorithm**

# Prototype, floating point

```
\begin{split} R_{NxN} &= chol(A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN}) \colon \ R_{NxN}^* \cdot R_{NxN} \cong A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN} \\ \text{void cchol (void* pScr,} & \text{complex_float *R,} & \text{complex_float *D,} \\ \text{const complex_float *A,} & \text{float32\_t sigma2);} \\ \text{void rchol (void* pScr,} & \text{float32\_t *R,} & \text{float32\_t *D,} \\ \text{const float32\_t *A,} & \text{float32\_t *A,} & \text{float32\_t sigma2);} \end{split}
```

N, M	Function	API	Scratch allocation function
	Real data		
4	matcholdecomp4x4f	rchol	matcholdecomp4x4f_getScratchSize
6	matcholdecomp6x6f	rchol	matcholdecomp6x6f_getScratchSize
8	matcholdecomp8x8f	rchol	matcholdecomp8x8f_getScratchSize
10	matcholdecomp10x10f	rchol	matcholdecomp10x10f_getScratchSize
	Complex data		
4	cmatcholdecomp4x4f	cchol	cmatcholdecomp4x4f_getScratchSize
6	cmatcholdecomp6x6f	cchol	cmatcholdecomp6x6f_getScratchSize
8	cmatcholdecomp8x8f	cchol	cmatcholdecomp8x8f_getScratchSize
10	cmatcholdecomp10x10 f	cchol	cmatcholdecomp10x10f_getScratchSize

# Prototype, floating point

N, M	Function	API	Scratch allocation function
	Real 32-bit data		
4	matcholdecomp4x4_32x32	rchol32	matcholdecomp4x4_32x32_getScratchSiz e
6	matcholdecomp6x6_32x32	rchol32	<pre>matcholdecomp6x6_32x32_getScratchSiz e</pre>
8	matcholdecomp8x8_32x32	rchol32	matcholdecomp8x8_32x32_getScratchSiz e
10	matcholdecomp10x10_32x 32	rchol32	<pre>matcholdecomp10x10_32x32_getScratchS ize</pre>
	Complex 32-bit data		

4	cmatcholdecomp4x4_32x3	cchol32	cmatcholdecomp4x4_32x32_getScratchSi
	2		ze
6	cmatcholdecomp6x6 32x3	cchol32	cmatcholdecomp6x6 32x32 getScratchSi
	2		ze
8	cmatcholdecomp8x8 32x3	cchol32	cmatcholdecomp8x8 32x32 getScratchSi
	2		ze
10	cmatcholdecomp10x10 32	cchol32	cmatcholdecomp10x10 32x32 getScratch
	x32		Size

Туре	Name	Size	Description
Parameters		•	
int	М		Dimensional parameter: number of rows in input matrix A
int	N		Dimensional parameter: number of columns in input matrix A, number of rows and columns in upper triangular matrix R
Input			
float32_t, complex_float, int32_t, complex_fract32	A	[M*N]	matrix ${\mathbb A}$ . For fixed point API, the representation is $Q({}_{{\mathbb Q}{\mathbb A}})$
float32_t, int32_t	sigma2		Regularization term For fixed point, the representation should be $Q(2*qA-30)$
int	qRA		$_{\mathbb{Q}\mathbb{R}^-\mathbb{Q}\mathbb{A}}$ (for the fixed point API only); difference between fixed point representations of $\mathbb{R}$ and $\mathbb{A}$ matrices (for the fixed point API only). Should be equal or less than 0 (typically -2)
Output			
float32_t, complex_float, int32_t, complex fract32	R	((N+1)*N)/2	upper triangular matrix ${\tt R}$ . For fixed point API, the representation is $Q({\tt qR})$
float32_t, complex_float, complex_fract32	D	[N]	reciprocals of the main diagonal NOTE: for the fixed point API, these data are stored internally in special format with separate mantissa and exponent for better accuracy and dynamic range control. So, even for the real data, they stored as pairs of 2 integers and packed to the complex_fract32 format
Temporary	1	T	1
	pScr		Scratch memory

Returned value

none

Restrictions

N = M = 4, 6, 8, 10

All matrices and scratch memory should not overlap and be aligned on 8-bytes boundary

# 2.8.5 Cholesky Forward Substitution for Pseudo-inversion

Description

These functions make forward recursion stage of pseudo-inversion. They use Cholesky decomposition  $R_{N\times N}$  of original matrices  $A_{M\times N}$ .

**Precision** 

2 variants available:

Туре	Description	
f	floating point input, floating point output	
32x32	32-bit fixed point	

Algorithm

Solution of equations

$$R_{NxN}^* \cdot y_{NxP} = A_{MxN}^* B_{MxP}$$

# Prototype, floating point

N, M	Function	API	Scratch allocation function
	Real data		
4	matcholfwdsubst4x4f	rfwd	matcholfwdsubst4x4f_getScratchSize
6	matcholfwdsubst6x6f	rfwd	matcholfwdsubst6x6f_getScratchSize
8	matcholfwdsubst8x8f	rfwd	matcholfwdsubst8x8f_getScratchSize
10	matcholfwdsubst10x10f	rfwd	matcholfwdsubst10x10f_getScratchSize
	Complex data		
4	cmatcholfwdsubst4x4f	cfwd	cmatcholfwdsubst4x4f_getScratchSize
6	cmatcholfwdsubst6x6f	cfwd	cmatcholfwdsubst6x6f_getScratchSize
8	cmatcholfwdsubst8x8f	cfwd	cmatcholfwdsubst8x8f_getScratchSize
10	cmatcholfwdsubst10x10 f	cfwd	cmatcholfwdsubst10x10f_getScratchSize

#### Prototype, fixed point

N, M	Function	API	Scratch allocation function
	Real 32-bit data		
4	matcholfwdsubst4x4_32x32	rfwd32	<pre>matcholfwdsubst4x4_32x32_getScrat chSize</pre>
6	matcholfwdsubst6x6_32x32	rfwd32	<pre>matcholfwdsubst6x6_32x32_getScrat chSize</pre>
8	matcholfwdsubst8x8_32x32	rfwd32	<pre>matcholfwdsubst8x8_32x32_getScrat chSize</pre>
10	matcholfwdsubst10x10_32x3 2	rfwd32	<pre>matcholfwdsubst10x10_32x32_getScr atchSize</pre>
	Complex 32-bit data		
4	cmatcholfwdsubst4x4_32x32	cfwd32	cmatcholfwdsubst4x4_32x32_getScratchSize
6	cmatcholfwdsubst6x6_32x32	cfwd32	cmatcholfwdsubst6x6_32x32_getScratchSize
8	cmatcholfwdsubst8x8_32x32	cfwd32	cmatcholfwdsubst8x8_32x32_getScratchSize
10	cmatcholfwdsubst10x10_32x 32	cfwd32	<pre>cmatcholfwdsubst10x10_32x32_getSc ratchSize</pre>

Туре	Name	Size	Description
Parameters	•		
int	M		Matrix dimension (number of rows in matrix A)
int	N		Matrix dimension (number of columns in matrix
int	P		Number of columns in right-side matrix B
Input	•		-
float32_t, complex_float, int32_t, complex fract32	R	[((N+1)*N)/2]	upper triangular matrix ${_{\rm R}}.$ For fixed point, representation is Q( ${_{\rm QR}})$
float32_t, complex_float, int32_t, complex_fract32	A	[M*N]	matrix ${\tt A}.$ For fixed point, representation is ${\tt Q}({\tt q}{\tt A})$
float32_t, complex_float, int32_t, complex_fract32	В	[M*P]	original right-side matrix ${_{\rm B}}.\;$ For fixed point, representation is Q( ${_{\rm QB}})$
float32_t, complex_float, complex_fract32	D	[N]	reciprocals of main diagonal.  NOTE: for the fixed point API, these data are stored internally in special format with separate mantissa and exponent for better accuracy and dynamic range control. So, even for the real data, they stored as pairs of 2 integers and packed to the complex_fract32 format
int	qYBRA		$q_{Y-q_B+q_R-q_A}$ , combination of fixed poin representations of matrices $_{Y,B,\ R}$ and $_{A}$ (for the fixed point API only)
Output			
float32_t, complex_float, int32_t, complex fract32	У	[N*P]	Decision matrix $y$ . For fixed point, representation is $Q(qy)$
Temporary	•		
,	pScr		Scratch memory

Returned value

none

Restrictions

N = M = 4, 6, 8, 10

P = 1

All matrices and scratch memory should not overlap and be aligned on 8-bytes boundary

# 2.8.6 Cholesky Backward Substitution for Pseudo-inversion

Description

These functions make backward substitution stage of pseudo-inversion. They use Cholesky decomposition of original matrices and results of forward substitution.

Precision

2 variants available:

Туре	Description	
f	floating point input, floating point output	
32x32	32-bit fixed point	

#### Algorithm

Solution of equations

$$\begin{split} R_{NxN} \cdot x_{NxP} &= y_{NxP} \\ \text{void cbkw (void * pScr,} \end{split}$$

Prototype, floating point

complex\_float \* x, const complex\_float \* R,

N	Function		Scratch allocation function	
	Real data			
4	matcholbkwsubst4x4f	rbkw	matcholbkwsubst4x4f_getScratchSize	
6	matcholbkwsubst6x6f	rbkw	matcholbkwsubst6x6f_getScratchSize	
8	matcholbkwsubst8x8f	rbkw	matcholbkwsubst8x8f_getScratchSize	
10	10 matcholbkwsubst10x10f		matcholbkwsubst10x10f_getScratchSize	
	Complex data			
4	4 cmatcholbkwsubst4x4f cbkw		cmatcholbkwsubst4x4f_getScratchSize	
6	6 cmatcholbkwsubst6x6f cb		cmatcholbkwsubst6x6f_getScratchSize	
8	8 cmatcholbkwsubst8x8f cbkw		cmatcholbkwsubst8x8f_getScratchSize	
10	cmatcholbkwsubst10x10 f	cbkw	cmatcholbkwsubst10x10f_getScratchSize	

## Prototype, fixed point

N	Function	API	Scratch allocation function
	Real 32-bit data		
4	matcholbkwsubst4x4_32x32	rbkw32	matcholbkwsubst4x4_32x32_getScratchS ize
6	matcholbkwsubst6x6_32x32	rbkw32	<pre>matcholbkwsubst6x6_32x32_getScratchS ize</pre>
8	matcholbkwsubst8x8_32x32	rbkw32	matcholbkwsubst8x8_32x32_getScratchS ize
10	matcholbkwsubst10x10_32x3 2	rbkw32	<pre>matcholbkwsubst10x10_32x32_getScratc hSize</pre>
	Complex 32-bit data		
4	cmatcholbkwsubst4x4_32x32	cbkw32	cmatcholbkwsubst4x4_32x32_getScratch Size
6	cmatcholbkwsubst6x6_32x32	cbkw32	cmatcholbkwsubst6x6_32x32_getScratch Size
8	cmatcholbkwsubst8x8_32x32	cbkw32	<pre>cmatcholbkwsubst8x8_32x32_getScratch Size</pre>
10	cmatcholbkwsubst10x10_32x 32	cbkw32	<pre>cmatcholbkwsubst10x10_32x32_getScrat chSize</pre>

#### **Arguments**

Туре	Name	Size	Description
Parameters			
int	N		Matrix dimension (number of columns in matrix A)
int	P		Number of columns in right-side matrix B
Input			
<pre>float32_t, complex_float, int32 t,</pre>	R	[((N+1)*N)/2]	Cholesky upper-triangle matrix $\tt R$ For fixed point API, the representation is $Q(\tt QR)$

complex fract32			
float32_t, complex_float, int32_t, complex_fract32	У	[N*P]	Results of forward substitution stage. For fixed point API, the representation is $Q(qY)$
float32_t, complex_float	D	[N]	reciprocals of main diagonal.  NOTE: for the fixed point API, these data are stored internally in special format with separate mantissa and exponent for better accuracy and dynamic range control. So, even for the real data, they stored as pairs of 2 integers and packed to the complex fract32 format
Output			·
float32_t, complex_float, complex fract32	Х	[N*P]	Decision matrix $\times$ . For fixed point API, the representation is Q( $_{\mathbb{Q}^X}$ )
int	qXYR		combination of fixed point representation (matrices $R$ , $x$ and $y$ ) $qxyR=qx-qy+qR$ (for fixed point API only)
Temporary			
	pScr		Scratch memory

Returned value

none

Restrictions

N = 4, 6, 8, 10

All matrices and scratch should not overlap and be aligned on 8-bytes boundary

# 2.8.7 Cholesky Presudoinversion

Description

Obtain Left Inverse of a matrix using Cholesky Decomposition

**Precision** 

2 variants available:

Type Description	
f	floating point input, floating point output
32x32	32-bit fixed point

Algorithm

$$\begin{array}{l} \text{find } R_{NxN} = chol(A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN}) \colon \ R_{NxN}^* \cdot R_{NxN} \cong A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN} \\ \text{solve } R_{NxN}^* \cdot x_{NxM} = A_{MxN}^* \end{array}$$

Prototypes, floating point

N, M	Function	API	Scratch allocation function
Real data			
4	matcholpseudoinv4x4f	rinv	matcholpseudoinv4x4f_getScratchSize
6	matcholpseudoinv6x6f	rinv	matcholpseudoinv6x6f_getScratchSize
8 matcholpseudoinv8x8f rinv matcholpseudoinv		matcholpseudoinv8x8f_getScratchSize	
10 matcholpseudoinv10x10 rinv ma		matcholpseudoinv10x10f_getScratchSize	
Complex data			
4 cmatcholpseudoinv4x4f		cinv	cmatcholpseudoinv4x4f_getScratchSize
6 cmatcholpseudoinv6x6f cinv cmatcholpse		cmatcholpseudoinv6x6f_getScratchSize	
8	cmatcholpseudoinv8x8f	cinv	cmatcholpseudoinv8x8f_getScratchSize

10	cmatcholpseudoinv10x1	cinv	cmatcholpseudoinv10x10f_getScratchSiz
	0f		е

# Prototypes, fixed point

N, M	Function	API	Scratch allocation function	
	Real data			
4	matcholpseudoinv4x4f	rinv	matcholpseudoinv4x4f_getScratchSize	
6	matcholpseudoinv6x6f	rinv	matcholpseudoinv6x6f_getScratchSize	
8	matcholpseudoinv8x8f	rinv	matcholpseudoinv8x8f_getScratchSize	
10 matcholpseudoinv10x10 f		rinv	matcholpseudoinv10x10f_getScratchSiz	
	Complex data			
4	cmatcholpseudoinv4x4f	cinv	cmatcholpseudoinv4x4f_getScratchSize	
6	cmatcholpseudoinv6x6f	cinv	cmatcholpseudoinv6x6f_getScratchSize	
8	cmatcholpseudoinv8x8f	cinv	cmatcholpseudoinv8x8f_getScratchSize	
10	cmatcholpseudoinv10x1 0f	cinv	<pre>cmatcholpseudoinv10x10f_getScratchSiz e</pre>	

#### **Arguments**

Туре	Name	Size	Description	
Parameters		•		
int	М		Matrix dimension (number of rows in matrix A)	
int	N		Matrix dimension (number of columns in matrix A)	
Input				
float32_t, complex float	А	[M] [N]	input matrix	
float32_t	sigma2		regularization term	
Output			·	
<pre>float32_t, complex_float</pre>	Х	[N] [M]	pseudoinverse matrix	
Temporary				
•	pScr		Scratch memory	

Returned value

none

Restrictions

N = 4, 6, 8, 10

All matrices and scratch should not overlap and be aligned on 8-bytes boundary

# 2.8.8 Cholesky Preprocessing

Description

Preprocessing for Least Square Solutions

**Precision** 

2 variants available:

Туре	Description	
f	floating point input, floating point output	
32x32	32-bit fixed point	

**Algorithm** 

 $Z_{NxN} = A_{MxN}^* \cdot A_{MxN} + \sigma^2 I_{NxN}$ 

# Prototypes, floating point

```
void rprep(void* pScr,
    float32_t *Z,
    const float32_t * A,
    float32_t sigma2);

void cprep(void* pScr,
    complex_float *Z,
    const complex_float * A,
    float32_t sigma2);
```

N, M	Function	API	Scratch allocation function
	Real data, floating point		
4	matcholpreprocess4x4f	rprep	matcholpreprocess4x4f_getScratchSize
6	matcholpreprocess6x6f	rprep	matcholpreprocess6x6f_getScratchSize
8	matcholpreprocess8x8f	rprep	matcholpreprocess8x8f_getScratchSize
10	matcholpreprocess10x1 Of	rprep	matcholpreprocess10x10f_getScratchSiz e
	Complex data, floating point		
4	cmatcholpreprocess4x4 f	cprep	cmatcholpreprocess4x4f_getScratchSize
6	cmatcholpreprocess6x6 f	cprep	cmatcholpreprocess6x6f_getScratchSize
8	cmatcholpreprocess8x8 f	cprep	cmatcholpreprocess8x8f_getScratchSize
10	cmatcholpreprocess10x 10f	cprep	<pre>cmatcholpreprocess10x10f_getScratchSi ze</pre>

# Prototypes, fixed point

N, M	Function	API	Scratch allocation function
	Real 32-bit data		
4	matcholpreprocess4x4_32x32	rprep32	<pre>matcholpreprocess4x4_32x32     _getScratchSize</pre>
6	matcholpreprocess6x6_32x32	rprep32	<pre>matcholpreprocess6x6_32x32     _getScratchSize</pre>
8	matcholpreprocess8x8_32x32	rprep32	<pre>matcholpreprocess8x8_32x32     _getScratchSize</pre>
10	matcholpreprocess10x10_32x32	rprep32	<pre>matcholpreprocess10x10_32x 32_getScratchSize</pre>
	Complex 32-bit data		
4	cmatcholpreprocess4x4_32x32	cprep32	<pre>cmatcholpreprocess4x4_32x3 2 getScratchSize</pre>
6	cmatcholpreprocess6x6_32x32	cprep32	<pre>cmatcholpreprocess6x6_32x3 2 getScratchSize</pre>
8	cmatcholpreprocess8x8_32x32	cprep32	cmatcholpreprocess8x8_32x3 2_getScratchSize
10	cmatcholpreprocess10x10_32x32	cprep32	cmatcholpreprocess10x10_32 x32_getScratchSize

Туре	Name	Size	Description	
Parameters	•	•		
int	М		Matrix dimension (number of rows in matrix A)	
int	N		Matrix dimension (number of columns in matrix A)	
Input				
float32_t, complex_float, int32_t, complex_fract32	А	[M] [N]	input matrix, For the fixed point, the representation is $Q({\tt q\mathbb{A}})$	
float32_t, int32_t	sigma2		Regularization term For fixed point, the representation should be Q(2*qA-30)	
int	qRA		$_{\mathrm{QR-QA}}$ (for the fixed point API only); difference between fixed point representations of $_{\mathrm{R}}$ and $_{\mathrm{A}}$ matrices (for the fixed point API only). Should be equal or less than 0 (typically -2)	
Output				
float32_t, complex_float, int64_t, complex_fract64	Z	[N] [N]	output matrix. For the fixed poiont, the representation is $\mathbf{Q}(2*_{\mathbf{QR}-4})$	
Temporary				
	pScr		Scratch memory	

Returned value

none

Restrictions

N = 4, 6, 8, 10

All matrices and scratch should not overlap and be aligned on 8-bytes boundary

# 2.9 Fitting/Interpolation

# 2.9.1 Polynomial Approximation

Description

Functions calculate polynomial approximation for all values from given vector. Fixed point functions take polynomial coefficients in Q31 precision.

NOTE:

approximation is calculated like Taylor series that is why overflow may potentially occur if cumulative sum of coefficients given from the last to the first coefficient is bigger that 1. To avoid this negative effect for fixed point routines, all the coefficients may be scaled down and result will be shifted left after all intermediate computations. Amount of this left shift is controlled by lsh argument.

Precision

2 variants available:

Туре	Description
32x32	32-bit inputs, 32-bit coefficients, 32-bit output.
f	floating point. Requires VFPU/SFPU core option

**Algorithm** 

$$z_n = \sum_{m=0}^{M} c_m x_n^m, n = \overline{0...N-1}$$

**Prototype** 

Туре	Name	Size	Description
Input		•	
int32_t, float32_t	Х	N	input data, Q31 or floating point
int32_t, float32_t	С	5 or 9	coefficients (5 coefficients for vec_poly4_xxx and 9 coefficients for vec_poly8_xxx), Q31 or floating point
int	lsh		additional left shift for result (for fixed point routines only)
int	N		length of vectors
Output			
int32_t, float32_t	Z	N	result, Q31 or floating point

Returned value

None

Restrictions

x, c, z should not overlap

**Conditions for** optimum performance

 $\mathtt{x}$  , c , z  $\phantom{x}$  - aligned on 8-byte boundary  $\mathtt{N}\phantom{x}$  - multiple of 2

# 2.10 Fast Fourier Transforms

FFT functions make floating point, 32x32, 32x16, 16x16-bit scaling fast Fourier transforms for complex/real data. Also, they use bit-reversal permutations so spectral data appear in the usual order. They normally use in-place transformations so **input data may be damaged**.

Different types if data scaling are provided by FFT functions. For all types of scaling, the internal representation of the data is the same as the input/output data. For these functions, the internal representation of the data is complex fract32.

#### **Basic scaling modes:**

- no scaling (scalingOption = 0), provides the best cycles, but worse accuracy on data with small magnitude and may suffer from overflow in the intermediate stages. In this mode, the FFT never scales the intermediate data. So, this mode is used only in very specific cases where this behavior is acceptable, i.e. in 2D DCT for JPEG, see 2.10.12
- **dynamic scaling** (scalingOption = 2), provides the best accuracy, but has less performance comparing with static scaling;
- static scaling (scalingOption = 3), has more performance but worse accuracy than dynamic scaling.

With dynamic scaling (scalingOption = 2), the input data are normalized in the first phase of the FFT, but so that there is no overflow. In subsequent phases, the data are automatically shifted to the right, so that there is no overflow. The function returns a total shift count, which can be negative under certain conditions (i.e. weak input signals).

With static scaling (scalingOption = 3), the data are shifted to the right before each FFT phase, the amount of shift is independent of the input data and is chosen so that there is no overflow for any input data.

FFT/IFFT functions family with improved memory efficiency (fft\_cplxprec>\_ie, fft\_realprec>\_ie) as well as floating point FFT functions² expose smaller program- and constant data memory footprint. They differ from regular FFT/IFFT functions in the following aspects:

- cycles performance is compromised in favor of memory efficiency
- twiddle factor tables are provided by user. A single table may be shared between FFTs/IFFTs of varying size (see para 4.2)

All fixed-point FFT functions (including scaling and non-scaling) return total number of right shifts (t) occurred during all stages. Floating point FFTs do not make additional scaling so they always return 0 to indicate this fact. So, FFT/IFFT output will be scaled by  $2^{\rm t}$ . Library functions from 2.5.5 help to convert results to desired scale or Q-representation. In these computations you have to take into account the fact that FFT $\rightarrow$ IFFT chain amplifies signal by the length of FFT N for complex transforms and by N/2 for real transforms.

For example, consider processing chain:

 $y=FFT(x) \rightarrow w=some\_processing(y) \rightarrow z=IFFT(w)$  where N is the length of FFT, FFT returns total shift amount  $t_{FFT}$  and IFFT returns  $t_{IFFT}$ .

To move z to the same scale as x you have to shift it by:

<sup>&</sup>lt;sup>2</sup> Floating point FFT available only with improved memory efficiency API

```
t_{FFT} + t_{IFFT} + \log_2(N) \equiv t_{FFT} + t_{IFFT} + 31 - \text{scl bexp32}(N)
```

Alternatively, you may treat it as changing Q-representation. For example, DCT functions (with length 32) always return total number of shifts equals to  $log_2(32) = 5$ . So, if its input is Q31, output will be in Q26.

The table below summarizes how number of right shifts depends on selected scaled option.

Scaling option	FFT functions family	Returned number of right shifts
0	2D DCT	0
2	all FFT functions	depends on input data
3	FFT/IFFT, DCT	log2(N)+1

There are limited combinations of precision, scaling options and restrictions on the dynamic range of the input signal available:

Precision	Scaling options	Restrictions on the dynamic range of the input signal
FFT/IFFT		·
cplx32x16	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
cplx32x32	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
cplx16x16	2 – 16-bit dynamic scaling	None
	3 – fixed scaling before each stage	
cplx16x16_ie	2 – 16-bit dynamic scaling	None
cplx32x16_ie	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
cplx32x32_ie	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
real32x16	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
real32x32	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
real16x16	2 – 16-bit dynamic scaling	None
	3 – fixed scaling before each stage	
real16x16_ie	2 – 16-bit dynamic scaling	None
real32x16_ie	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
real32x32_ie	2 – 32-bit dynamic scaling	None
	3 – fixed scaling before each stage	
DCT		·
dct_32x16,	3 – fixed scaling before each stage	None
dct_32x32, dct 16x16,		
dct_16x16, dct4 32x16,		
dct4 32x32,		
mdct_32x16,		
mdct_32x32, imdct 32x16,		
imdct_32x10,		
dct2d_8x16	0 – no scaling	None
idct2d_16x8	0 – no scaling	None

# 2.10.1 FFT on Complex Data

#### Description

These functions make FFT on complex data.

NOTES:

- 1. Bit-reversing permutation is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call
- 3. 32x32, 32x16,16x16 FFT supports mixed radix transforms

#### Precision

#### 3 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles
32x32	32-bit input/outputs, 32-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

#### Algorithm

## y = FFT(x)

## **Prototype**

#### FFT handles:

N	32x16	32x16	32x32
16	cfft16_16	cfft16_16	cfft32_16
32	cfft16_32	cfft16_32	cfft32_32
64	cfft16_64	cfft16_64	cfft32_64
128	cfft16_128	cfft16_128	cfft32_128
256	cfft16_256	cfft16_256	cfft32_256
512	cfft16_512	cfft16_512	cfft32_512
1024	cfft16_1024	cfft16_1024	cfft32_1024
2048	cfft16_2048	cfft16_2048	cfft32_2048
4096	cfft16_4096	cfft16_4096	cfft32_4096

# FFT handles for mixed radix transforms (for 16x16, 32x16 only):

N	16x16	N	32x16
160	cnfft16_160	160	cnfft32x16_160
192	cnfft16_192	192	cnfft32x16_192
240	cnfft16_240	240	cnfft32x16_240
320	cnfft16_320	320	cnfft32x16_320
384	cnfft16_384	384	cnfft32x16_384
480	cnfft16_480	480	cnfft32x16_480

## FFT handles for mixed radix transforms (for 32x32 only):

N	32x32	N	32x32	N	32x32
12	cnfft32_12	144	cnfft32_144	360	cnfft32_360
24	cnfft32_24	160	cnfft32_160	384	cnfft32_384
36	cnfft32_36	180	cnfft32_180	400	cnfft32_400
48	cnfft32_48	192	cnfft32_192	432	cnfft32_432
60	cnfft32_60	200	cnfft32_200	480	cnfft32_480
72	cnfft32_72	216	cnfft32_216	540	cnfft32_540
80	cnfft32_80	240	cnfft32_240	576	cnfft32_576
96	cnfft32_96	288	cnfft32_288	600	cnfft32_600
100	cnfft32_100	300	cnfft32_300	768	cnfft32_768
108	cnfft32_108	320	cnfft32_320	960	cnfft32_960

120	cnfft32_120	324	cnfft32_324	

, where N - FFT size

#### **Arguments**

Туре	Name	Size	Description	
Input				
int32_t or int16_t	х	2*N	complex input signal. Real and imaginary data are interleaved and real data goes first	
fft_handle_t	h		handle to specific FFT tables	
int	scalingOption		scaling option (see table in para 2.10)	
Output			<u> </u>	
int32_t or int16_t	У	2*N	output spectrum. Real and imaginary data are interleaved and real data goes first	

Returned value

total number of right shifts occurred during scaling procedure

Restrictions

 $\mathtt{x}$  ,  $\mathtt{y}$  should not overlap

x, y aligned on a 8-bytes boundary

#### 2.10.2 FFT on Real Data

#### **Description**

These functions make FFT on real data forming half of spectrum

NOTES:

- 1. Bit-reversing reordering is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call.
- 3. Real data FFT function calls  $fft_{cplx}()$  to apply complex FFT of size N/2 to input data and then transforms the resulting spectrum.
- 4. 32x32, 32x16,16x16 FFT supports mixed radix transforms

#### Precision

#### 3 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles
32x32	32-bit input/outputs, 32-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

#### Algorithm

# y = FFT(real(x))

#### **Prototype**

#### FFT handles:

N	32x16	32x16	32x32
32	rfft16_32	rfft16_32	rfft32_32
64	rfft16_64	rfft16_64	rfft32_64
128	rfft16_128	rfft16_128	rfft32_128
256	rfft16_256	rfft16_256	rfft32_256
512	rfft16_512	rfft16_512	rfft32_512
1024	rfft16_1024	rfft16_1024	rfft32_1024
2048	rfft16_2048	rfft16_2048	rfft32_2048
4096	rfft16_4096	rfft16_4096	rfft32_4096
8192	rfft16_8192	rfft16_8192	rfft32_8192

FFT handles for mixed radix transforms (for 16x16, 32x16 only):

N	16x16	N	32x16
160	rnfft16_160	160	rnfft32x16_160

192	rnfft16_192	192	rnfft32x16_192
240	rnfft16_240	240	rnfft32x16_240
320	rnfft16_320	320	rnfft32x16_320
384	rnfft16_384	384	rnfft32x16_384
480	rnfft16_480	480	rnfft32x16_480

FFT handles for mixed radix transforms (for 32x32 only):

N	32x32	N	32x32	N	32x32
12	rnfft32_12	160	rnfft32_160	480	rnfft32_480
24	rnfft32_24	180	rnfft32_180	540	rnfft32_540
30	rnfft32_30	192	rnfft32_192	576	rnfft32_576
36	rnfft32_36	216	rnfft32_216	720	rnfft32_720
48	rnfft32_48	240	rnfft32_240	768	rnfft32_768
60	rnfft32_60	288	rnfft32_288	960	rnfft32_960
72	rnfft32_72	300	rnfft32_300	1152	rnfft32_1152
90	rnfft32_90	320	rnfft32_320	1440	rnfft32_1440
96	rnfft32_96	324	rnfft32_324	1536	rnfft32_1536
108	rnfft32_108	360	rnfft32_360	1920	rnfft32_1920
120	rnfft32_120	384	rnfft32_384		
144	rnfft32_144	432	rnfft32_432		

<sup>,</sup> where N - FFT size

#### **Arguments**

Туре	Name	Size	Description	
Input				
int32_t or int16 t	Х	N	input signal	
fft_handle_t	h		handle to specific FFT tables	
int	scalingOpt		scaling option (see table in para 2.10)	
Output				
int32_t or int16_t	У	(N/2+1)*2	output spectrum (positive side). Real and imaginary data are interleaved and real data goes first	

Returned value

total number of right shifts occurred during scaling procedure

Restrictions

Arrays should not overlap

 $_{\rm X}$  ,  $_{\rm Y}$  - aligned on a 8-bytes boundary

# 2.10.3 Inverse FFT on Complex Data

Description

These functions make inverse FFT on complex data.

NOTES:

- 1. Bit-reversing reordering is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after call
- 3. 32x32, 32x16,16x16 FFT supports mixed radix transforms

**Precision** 

3 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles
32x32	32-bit input/outputs, 32-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

Algorithm

 $y = FFT^{-1}(x)$ 

**Prototype** 

int32\_t \* y, int32\_t\* x, fft\_handle\_t h, int scalingOption) int ifft cplx16x16(

int16\_t\* y, int16\_t\* x, fft\_handle\_t h, int scalingOption)

#### FFT handles:

N	32x16	32x16	32x32
16	cifft16_16	cifft16_16	cifft32_16
32	cifft16_32	cifft16_32	cifft32_32
64	cifft16_64	cifft16_64	cifft32_64
128	cifft16_128	cifft16_128	cifft32_128
256	cifft16_256	cifft16_256	cifft32_256
512	cifft16_512	cifft16_512	cifft32_512
1024	cifft16_1024	cifft16_1024	cifft32_1024
2048	cifft16_2048	cifft16_2048	cifft32_2048
4096	cifft16_4096	cifft16_4096	cifft32_4096

FFT handles for mixed radix transforms (for 16x16, 32x16 only):

N	16x16	N	32x16
160	cinfft16_160	160	cinfft32x16_160
192	cinfft16_192	192	cinfft32x16_192
240	cinfft16_240	240	cinfft32x16_240
320	cinfft16_320	320	cinfft32x16_320
384	cinfft16_384	384	cinfft32x16_384
480	cinfft16_480	480	cinfft32x16_480

FFT handles for mixed radix transforms (for 32x32 only):

N	32x32	N	32x32	N	32x32
12	cinfft32_12	144	cinfft32_144	360	cinfft32_360
24	cinfft32_24	160	cinfft32_160	384	cinfft32_384
36	cinfft32_36	180	cinfft32_180	400	cinfft32_400
48	cinfft32_48	192	cinfft32_192	432	cinfft32_432
60	cinfft32_60	200	cinfft32_200	480	cinfft32_480
72	cinfft32_72	216	cinfft32_216	540	cinfft32_540
80	cinfft32_80	240	cinfft32_240	576	cinfft32_576
96	cinfft32_96	288	cinfft32_288	600	cinfft32_600
100	cinfft32_100	300	cinfft32_300	768	cinfft32_768
108	cinfft32_108	320	cinfft32_320	960	cinfft32_960
120	cinfft32_120	324	cinfft32_324		

, where N - IFFT size

# **Arguments**

Туре	Name	Size	Description
Input			
int32_t or int16_t	Х	2*N	input spectrum. Real and imaginary data are interleaved and real data goes first
fft_handle_t	h		handle to specific FFT tables
int	scalingOpt		scaling option (see table in para 2.10)
Output			
int32_t or int16_t	У	2*N	complex output signal. Real and imaginary data are interleaved and real data goes first

Returned value

total number of right shifts occurred during scaling procedure

Restrictions

x, y - should not overlap

 $_{\rm X}$  ,  $_{\rm Y}$  - aligned on 8-bytes boundary

# 2.10.4 Inverse FFT Forming Real Data

#### Description

These functions make inverse FFT on half spectral data forming real data samples NOTES:

- 1. Bit-reversing reordering is done here.
- 2. IFFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after call.
- 3. Inverse FFT function for real signal transforms the input spectrum and then calls  $ifft_{cplx}()$  with FFT size set to N/2.
- 4. 32x32, 32x16,16x16 FFT supports mixed radix transforms

#### Precision

#### 3 variants available:

	Туре	Description	
	32x16	32-bit input/outputs, 16-bit twiddles	
	32x32	32-bit input/outputs, 32-bit twiddles	
Ī	16x16	16-bit input/outputs, 16-bit twiddles	

## Algorithm

# Prototype

```
y = real(FFT^{-1}(x))
```

#### FFT handles:

N	32x16	32x16	32x32			
32	rifft16_32	rifft16_32	rifft32_32			
64	rifft16_64	rifft16_64	rifft32_64			
128	rifft16_128	rifft16_128	rifft32_128			
256	rifft16_256	rifft16_256	rifft32_256			
512	rifft16_512	rifft16_512	rifft32_512			
1024	rifft16_1024	rifft16_1024	rifft32_1024			
2048	rifft16_2048	rifft16_2048	rifft32_2048			
4096	rifft16_4096	rifft16_4096	rifft32_4096			
8192	rifft16_8192	rifft16_8192	rifft32_8192			

#### FFT handles for mixed radix transforms (for 16x16, 32x16 only):

N	16x16	N	32x16
160	rinfft16_160	160	rinfft32x16_160
192	rinfft16_192	192	rinfft32x16_192
240	rinfft16_240	240	rinfft32x16_240
320	rinfft16_320	320	rinfft32x16_320
384	rinfft16_384	384	rinfft32x16_384
480	rinfft16_480	480	rinfft32x16_480

#### FFT handles for mixed radix transforms (for 32x32 only):

N	32x32	N	32x32	N	32x32
12	rinfft32_12	160	rinfft32_160	480	rinfft32_480
24	rinfft32_24	180	rinfft32_180	540	rinfft32_540
30	rinfft32_30	192	rinfft32_192	576	rinfft32_576
36	rinfft32_36	216	rinfft32_216	720	rinfft32_720
48	rinfft32_48	240	rinfft32_240	768	rinfft32_768
60	rinfft32_60	288	rinfft32_288	960	rinfft32_960
72	rinfft32_72	300	rinfft32_300	1152	rinfft32_1152
90	rinfft32_90	320	rinfft32_320	1440	rinfft32_1440
96	rinfft32_96	324	rinfft32_324	1536	rinfft32_1536

108	rinfft32_108	360	rinfft32_360	1920	rinfft32_1920
120	rinfft32_120	384	rinfft32_384		
144	rinfft32_144	432	rinfft32_432		

#### , where N - IFFT size

#### **Arguments**

, where N - HIII	5120		
Туре	Name	Size	Description
Input			
int32_t or int16_t	Х	(N/2+1)*2	input spectrum. Real and imaginary data are interleaved and real data goes first
fft_handle_t	h		handle to specific FFT tables
int	scalingOpt		scaling option (see table in para 2.10)
Output			
int32_t or int16_t	У	N	real output signal

Returned value

total number of right shifts occurred during scaling procedure

Restrictions

x, y should not overlap

x, y - aligned on 8-bytes boundary

# 2.10.5 FFT on Complex Data with Optimized Memory Usage

#### Description

These functions make FFT on complex data with optimized memory usage NOTES:

- 1. Bit-reversing permutation is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call
- 3. FFT of size N may be supplied with constant data (twiddle factors) of a larger-sized FFT = N\*twdstep.
- 4. stereo FFTs accept inputs from and provides outputs in interleaved order: left complex sample, right complex sample

#### Precision

4 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles. Ordinary (single channel) variant and stereo
32x32	32-bit input/outputs, 32-bit twiddles. Ordinary (single channel) variant and stereo
16x16	16-bit input/outputs, 16-bit twiddles. Ordinary (single channel) variant and stereo
f	floating point. Requires VFPU/SFPU core option. Ordinary variant (single channel) and
	stereo

#### **Algorithm**

# y = FFT(x)

int stereo fft cplx32x32 ie(

int stereo\_fft\_cplx16x16\_ie(

# Prototype (single channel variants)

```
int fft cplx32x16 ie(
              complex fract32* y, complex fract32* x,
              const complex_fract16* twd,
              int twdstep, int N, int scalingOpt);
int fft cplx32x32 ie(
              complex_fract32* y, complex_fract32* x,
              const complex fract32* twd,
              int twdstep, int N, int scalingOpt);
int fft cplx16x16 ie(
              complex fract16* y, complex fract16* x,
              const complex fract16* twd,
              int twdstep, int N, int scalingOpt);
int fft_cplxf_ie (
             complex float * y, complex_float * x,
             const complex float* twd,
             int twdstep, int N );
int stereo_fft_cplx32x16_ie(
              complex fract32* y, complex fract32* x,
              const complex_fract16* twd,
```

# Prototype (stereo variants)

const complex\_fract32\* twd,

int twdstep, int N, int scalingOpt);

int twdstep, int N, int scalingOpt);

complex fract32\* y, complex fract32\* x,

Туре	Name	Size	Description
Input			
<pre>complex_fract16, complex_fract32, complex_float</pre>	х	N*S	complex input signal. Real and imaginary data are interleaved and real data goes first
<pre>complex_fract32, complex_fract16, complex_float</pre>	twd	N*3/4*twdstep	twiddle factor table of a complex-valued FFT of size N*twdstep
int	twdstep		twiddle step
int	N		FFT size
int	scalingOpt		scaling option (see table in para 2.10), not applicable to the floating point function
Parameter			
	S		stereo option, 1 for ordinary (single channel) FFT, 2 - for stereo input/outputs.  Note: it does not come in parameter list of functions, but just specifies the total size of input/output data
Output			
<pre>complex_fract16, complex_fract32, complex_float</pre>	У	N*S	output spectrum. Real and imaginary data are interleaved and real data goes first

#### Returned value

total number of right shifts occurred during scaling procedure. Floating function always return 0

#### Restrictions

x, y should not overlap

x, y - aligned on a 8-bytes boundary

## 2.10.6 FFT on Real Data with Optimized Memory Usage

#### Description

These functions make FFT on real data forming half of spectrum with optimized memory usage NOTES:

- 1. Bit-reversing reordering is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call.
- FFT functions use input and output buffers for temporary storage of intermediate 32-bit data, so FFT functions with 24-bit packed I/O (Nx3-byte data) require that the buffers are large enough to keep Nx4-byte data.
- 4. FFT of size  $_{\mathbb{N}}$  may be supplied with constant data (twiddle factors) of a larger-sized FFT =  $\mathbb{N}^*$ twdstep 4 variants available:

#### Precision

# Type Description 32×16 32-bit input/outputs, 16-bit twiddles 32×32 32-bit input/outputs, 32-bit twiddles 16×16 16-bit input/outputs, 16-bit twiddles f floating point. Requires VFPU/SFPU core option

#### Algorithm

y = FFT(real(x))

**Prototype** 

Туре	Name	Size	Allocated Size	Description
Input				
int16_t, int32_t, float32_t	Х	N	N	input signal
complex_fract32, complex_fract16, complex_float	twd	N*3/4 *twdstep		twiddle factor table of a complex-valued FFT of size N*twdstep
int	twdstep			twiddle step
int	N			FFT size
int	scalingOpt			scaling option (see table in para 2.10), not applicable to the floating point function
Output				
complex_fract16, complex_fract32, complex_float	У	N/2+1	N/2+1	output spectrum (positive side). Real and imaginary data are interleaved and real data goes first

Returned value

total number of right shifts occurred during scaling procedure. Floating function always return 0

Restrictions

Arrays should not overlap

x, y - aligned on a 8-bytes boundary

## 2.10.7 Inverse FFT on Complex Data with Optimized Memory Usage

#### **Description**

These functions make inverse FFT on complex data with optimized memory usage NOTES:

- 1. Bit-reversing permutation is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call
- 3. FFT of size N may be supplied with constant data (twiddle factors) of a larger-sized FFT = N\*twdstep.
- 4. stereo FFTs accept inputs from and provides outputs in interleaved order: left complex sample, right complex sample

#### Precision

4 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles. Ordinary (single channel) variant and stereo
32x32	32-bit input/outputs, 32-bit twiddles. Ordinary (single channel) variant and stereo
16x16	16-bit input/outputs, 16-bit twiddles. Ordinary (single channel) variant and stereo
f	floating point. Requires VFPU/SFPU core option. Ordinary (single channel) variant and
	stereo

#### Algorithm

# Prototype (single channel variant)

```
const complex fract16* twd,
              int twdstep, int N, int scalingOpt);
int ifft cplxf ie(
              complex float* y, complex float * x,
              const complex float * twd,
              int twdstep, \overline{i}nt N);
int stereo ifft cplx32x16 ie(
              complex fract32* y, complex fract32* x,
              const complex_fract16* twd,
              int twdstep, int N, int scalingOpt);
int stereo ifft cplx32x32 ie(
             complex fract32* y, complex fract32* x,
              const complex fract32* twd,
              int twdstep, int N, int scalingOpt);
int stereo ifft cplx16x16 ie(
              complex fract16* y, complex fract16* x,
              const complex fract16* twd,
              int twdstep, int N, int scalingOpt);
int stereo ifft cplxf ie(
              complex float* y, complex_float * x,
              const complex float * twd,
              int twdstep, int N);
```

Prototype (streo

variants)

Туре	Name	Size	Description
Input	•	•	
<pre>complex_fract16, complex_float complex_fract32, complex_float</pre>	х	N*S	complex input signal. Real and imaginary data are interleaved and real data goes first
<pre>complex_fract32, complex_fract16, complex_float</pre>	twd	N*3/4*twdstep	twiddle factor table of a complex- valued FFT of size N*twdstep
int	twdstep		twiddle step
int	N		FFT size
int	scalingOpt		scaling option (see table in para 2.10) , not applicable to the floating point function
Parameter			
	S		stereo option, 1 for ordinary (single channel) FFT, 2 - for stereo input/outputs.  Note: it does not come in parameter list of functions, but just specifies the total size of input/output data
Output		·	·
<pre>complex_fract16, complex_fract32, complex_float</pre>	У	N*S	output spectrum. Real and imaginary data are interleaved and real data goes first

#### Returned value

total number of right shifts occurred during scaling procedure. Floating function always return 0

#### Restrictions

x, y should not overlap

x, y - aligned on a 8-bytes boundary

#### 2.10.8 Inverse FFT on Real Data with Optimized Memory Usage

#### Description

These functions make inverse FFT on real data from half of spectrum with optimized memory usage NOTES:

- 1. Bit-reversing reordering is done here.
- 2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call.
- 3. FFT functions use input and output buffers for temporary storage of intermediate 32-bit data, so FFT functions with 24-bit packed I/O (Nx3-byte data) require that the buffers are large enough to keep

Nx4-byte data.

4. FFT of size N may be supplied with constant data (twiddle factors) of a larger-sized FFT = N\*twdstep 4 variants available:

#### Precision

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles
32x32	32-bit input/outputs, 32-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

```
int i
```

# **Prototype**

#### **Arguments**

Туре	Name	Size	Allocat ed Size	Description
Input				
complex_fract16, complex_fract32, complex_float	х	N/2+1	N/2+1	input spectrum (positive side). Real and imaginary data are interleaved and real data goes first
complex_fract32, complex_fract16, complex_float	twd	N*3/4* twdstep		twiddle factor table of a complex-valued FFT of size N*twdstep
int	twdstep			twiddle step
int	N			FFT size
int	scalingOpt			scaling option (see table in para 2.10), not applicable to the floating point function
Output				
int16_t, int32_t, float32_t	У	N	N	output real signal

#### Returned value

total number of right shifts occurred during scaling procedure. Floating function always return 0

#### Restrictions

Arrays should not overlap

x, y - aligned on a 8-bytes boundary

#### 2.10.9 Power Spectrum

#### Description

These functions compute a normalized power spectrum from the output signal generated by an FFT function.

The  $\mathbb{N}$  argument specifies the size of the FFT and must be a power of 2.

The mode argument is used to specify the type of FFT function used to generate the x array. If the x array has been generated from a frequency-domain complex input signal (output of complex FFT function), the mode argument must be set to 0. Otherwise the mode argument must be set to 1 to signify that the x array has been generated from a frequency-domain real input signal (output of real FFT function).

The block\_exponent argument is used to control the normalization of the power spectrum. It will usually be set to the block\_exponent that is returned by corresponding FFT functions. If the input array was generated by some other means, then the value specified for the block exponent argument will depend

upon how the FFT was calculated. If the function used to calculate the FFT did not scale the intermediate results at any of the stages of the computation, then set block\_exponent to zero; if the FFT function scaled the intermediate results at each stage of the computation, then set block\_exponent to -1; otherwise set block\_exponent to the sum of negated base-2 logarithm of all scaling factors applied to data at intermediate FFT stages. This value will be in the range 0 to log2 (N).

fft\_spectrum functions write the power spectrum to the output array y. If mode is set to 0, then the length of the power spectrum will be N. If mode is set to 1, then the length of the power spectrum will be (N/2+1)

Precision

#### 3 variants available:

Туре	Description
16x32	16-bit inputs, 32-bit outputs
32x32	32-bit inputs/outputs
f	floating point inputs/outputs. Requires VFPU/SFPU core option

**Algorithm** 

For mode = 0 (cfft-generated input data):

$$y_n = \frac{\sqrt{x_n \cdot x_n^*}}{N}, n = \overline{0...N - 1}$$

For mode = 1 (rfft-generated input data):

$$y_n = \frac{2 \cdot \sqrt{x_n \cdot x_n^*}}{N}, n = \overline{0...N/2}$$

**Prototype** 

## **Arguments**

Туре	Name	Size	Description		
Input					
complex_fract16, complex_fract32, complex_float	x	N (for mode==0) N/2+1 (for mode==1)	input spectrum		
int	N		FFT length		
int	block_exponent		power spectrum normalization control		
int	mode		power spectrum mode: 0 – complex signal 1 – real signal		
Output					
complex_fract16, complex_fract32, complex_float	У	N (for mode==0) N/2+1 (for mode==1)	output power spectrum		

Returned value

none

Restrictions

Arrays should not overlap

 $_{\rm X}$  ,  $_{\rm Y}$  - aligned on a 8-bytes boundary

## 2.10.10 Discrete Cosine Transform

Description

These functions apply DCT (Type II, Type IV) to input NOTE:

#### DCT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call.

#### Precision

#### 4 variants available:

Туре	Description
32x16	32-bit input/outputs, 16-bit twiddles
32x32	32-bit input/outputs, 32-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles
f	floating point. Requires VFPU/SFPU core option

#### Algorithm

DCT Type II: 
$$y_{k=0..N-1} = \sum_{n=0}^{N-1} x_n \cdot \cos\left(\frac{\pi}{N} \cdot (n+0.5) \cdot k\right), n = \overline{0...N-1}$$

DCT Type IV: 
$$y_{k=0..N-1} = \sum_{n=0}^{N-1} x_n \cdot \cos\left(\frac{\pi}{N} \cdot (n+0.5) \cdot [k+0.5]\right), n = \overline{0...N-1}$$

# Prototype (DCT Type II)

```
int dct_32x16(int32_t* y, int32_t* x, dct_handle_t h, int scalingOpt);
int dct_32x32(int32_t* y, int32_t* x, dct_handle_t h, int scalingOpt);
int dct_16x16(int16_t* y, int16_t* x, dct_handle_t h, int scalingOpt);
int dctf (float32_t * y,float32_t * x, dct_handle_t h);
```

#### DCT-II handles:

N	32x32	N	32x16, 16x16	N	floating point
32	dct2_32_32	32	dct2_16_32	32	dct2_f_32
64	dct2_32_64	64	dct2_16_64	64	dct4_f_64

, where N - DCT size

#### **Arguments**

Туре	Name	Size	Description
Input			
int16_t,	Х	N	input signal
int32_t, float32 t			
dct_handle_t	h		DCT-II handle
int	scalingOpt		scaling option (see table in para 2.10), not applicable to the floating point function
Output			
int16_t,	У	N	output of transform
int32_t, float32_t			

# Prototype (DCT Type IV)

int dct4\_32x16(int32\_t\* y, int32\_t\* x, dct\_handle h, int scalingOpt);
int dct4\_32x32(int32\_t\* y, int32\_t\* x, dct\_handle h, int scalingOpt);

#### DCT-IV handles:

N	32x32	N	32x16
32	dct4_32_32	32	dct4_16_32
64	dct4_32_64	64	dct4_16_64
128	dct4_32_128	128	dct4_16_128
256	dct4_32_256	256	dct4_16_256
512	dct4_32_512	512	dct4_16_512

, where  $\mbox{N}$  - DCT size

#### **Arguments**

Туре	Name	Size	Description
Input			
int32_t	Х	N	input signal
dct_handle_t	h		DCT-IV handle

int	scalingOpt		scaling option (see table in para 2.10), not applicable to the floating point function
Output			
int16_t, int32_t, float32_t	У	N	output of transform

#### Returned value

total number of right shifts occurred during scaling procedure (0 for floating point function)

#### Restrictions

- x, y should not overlap
- x,y aligned on 8-bytes boundary

#### 2.10.11 Modified Discrete Cosine Transform

#### Description

These functions apply Modified DCT to input (convert 2N real data to N spectral components) and make inverse conversion forming 2N numbers from N inputs.

NOTE:

MDCT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after the call.

#### Precision

#### 2 variants available:

Туре	Description				
32x16	32-bit input/outputs, 16-bit twiddles				
32x32	32-bit input/outputs, 32-bit twiddles				

#### Algorithm

MDCT: 
$$y_{k=0..N-1} = \sum_{n=0}^{N-1} x_n \cdot \cos\left(\frac{\pi}{N} \cdot (n+0.5) \cdot k\right), n = \overline{0...N-1}$$

#### Prototype (direct trasform)

int mdct\_32x16(int32\_t\* y, int32\_t\* x, dct\_handle\_t h, int scalingOpt); int mdct 32x32(int32\_t\* y, int32\_t\* x, dct\_handle\_t h, int scalingOpt);

#### MDCT/IMDCT handles:

N	32x32	N	32x16
32	mdct_32_32	32	mdct_16_32
64	mdct_32_64	64	mdct_16_64
128	mdct_32_128	128	mdct_16_128
256	mdct_32_256	256	mdct_16_256
512	mdct_32_512	512	mdct_16_512

where N - MDCT/IMDCT size

#### **Arguments**

Туре	Name	Size	Description			
Input						
int32_t	Х	N	input signal			
dct_handle_t	h		MDCT handle			
int	scalingOpt		scaling option (see table in para 2.10)			
Output						
int32_t	У	2*N	output of transform			

#### Algorithm

**Arguments** 

Inverse MDCT: 
$$y_{k=0..N-1} = \sum_{n=0}^{N-1} x_n \cdot \cos\left(\frac{\pi}{N} \cdot (n+0.5) \cdot (k+0.5)\right), n = \overline{0...N-1}$$

## Prototype (inverse trasform)

int	imdct	32x16(int32	t*	У,	int32	t*	х,	dct	handle	h,	int	<pre>scalingOpt);</pre>
int	imdct	32x32(int32	_t*	У,	int32	t*	х,	dct	handle	h,	int	<pre>scalingOpt);</pre>

Туре	Name	Size	Description
Input			

int32_t	Х	2*N	input signal	
dct_handle_t	h		IMDCT handle	
int	scalingOpt		scaling option (see table in para 2.10)	
Output				
int32_t	У	N	output of transform	

Returned value

total number of right shifts occurred during scaling procedure (0 for floating point function)

Restrictions

x, y should not overlap

x, y - aligned on 8-bytes boundary

#### 2.10.12 2D Discrete Cosine Transform

Description

These functions apply DCT (Type II) to the series of L input blocks of NxN pixels.

Precision

1 variant available:

Туре	Description
8x16	8-bit unsigned input, 16-bit signed output

Algorithm

Algorithm uses ITU-T T.81 (JPEG compression) DCT-II definition with bias 128 and left-to-right, top-to-bottom orientation.

$$y_{n,m}^{(l)} = \frac{1}{4} C_m C_n \sum_{i=0}^{N-1} \sum_{j=0}^{N-1} (x_{j,i}^{(l)} - 128) \cos \frac{(2i+1)m\pi}{2N} \cos \frac{(2j+1)n\pi}{2N}, l = \overline{0...L-1}$$

$$C_k = \begin{cases} 1, k \neq 0 \\ 1/\sqrt{2}, k \equiv 0 \end{cases}$$

$$x_{i,i}^{(l)} = x[l \cdot N \cdot N + (N \cdot j + i)]$$
 - pixel from j-th row, i-th column from l-th NxN block

**Prototype** 

int dct2d\_8x16(int16\_t\* y, uint8\_t \* x, dct\_handle\_t h, int L, int scalingOpt);
2D-DCT handles:

N	8x16
8	dct2d_16_8

, where N - DCT size

#### **Arguments**

Туре	Name	Size	Description	
Input				
uint8_t	X	N*N*L	input pixels: L NXN blocks	
dct_handle_t	h		DCT handle	
int	L		number of input blocks	
int	scalingOpt		scaling option (see table in para 2.10), should be 0	
Output				
int16_t	У	N*N*L	output of transform: L NXN blocks	

Returned value

U

Restrictions

x, y should not overlap

x, y - aligned on 8-bytes boundary

#### 2.10.13 2D Inverse Discrete Cosine Transform

Description

These functions apply inverse DCT (Type II) to the series of  ${\tt L}$  input blocks of  ${\tt N}{\tt X}{\tt N}$  pixels.

**Precision** 

1 variant available:

Туре	Description	
16x8	16-bit signed input, 8-bit unsigned output	

#### Algorithm

Algorithm uses ITU-T T.81 (JPEG compression) IDCT-II definition with bias 128 and left-to-right, top-to-bottom orientation.

$$y_{j,i}^{(l)} = 128 + \frac{1}{4} \sum_{m=0}^{N-1} \sum_{n=0}^{N-1} C_m C_n x_{n,m}^{(l)} \cos \frac{(2i+1)m\pi}{2N} \cos \frac{(2j+1)n\pi}{2N}, l = \overline{0...L-1}$$

$$C_k = \begin{cases} 1, k \neq 0 \\ 1/\sqrt{2}, k \equiv 0 \end{cases}$$

$$x_{n,m}^{(l)} = x[l \cdot N \cdot N + (N \cdot n + m)]$$
 - sample from n-th row, m-th column from l-th 8x8 block

#### **Prototype**

int idct2d\_16x8(uint8\_t \* y, int16\_t\* x, dct\_handle\_t h, int L, int scalingOpt);

#### 2D-IDCT handles:

N	16x8	
8	idct2d_16_8	

, where N - IDCT size

#### **Arguments**

Туре	Name	Size	Description	
Input				
int16_t	Х	N*N*L	input data: L NXN blocks	
dct_handle_t	h		IDCT handle	
int	L		number of input blocks	
int	scalingOpt		scaling option (see table in para 2.10), should be 0	
Output				
uint8_t	У	N*N*L	pixels: L NXN blocks	

#### Returned value

0

Restrictions

x, y should not overlap

x, y - aligned on 8-bytes boundary

# 2.11 Mel-Frequency Cepstral Coefficients

Mel-Frequency Cepstral Coefficients (MFCC) is an industry-standard representation of audio data for advanced audio processing, such as speech recognition software.

The MFCC package of the Library comprises the MFCC features extractor and its subsidiary component, the Log-scale Mel-frequency (LogMel) filterbank. The interconnection of these components and their operation are depicted in the figure below:

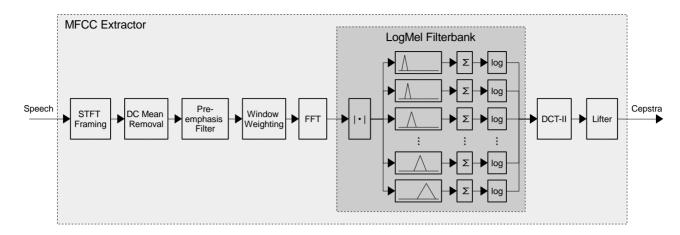


Figure 1. Block Diagram of the MFCC Extractor

Cepsral coefficient computation includes the following stages:

- 1. Input speech signal is optionally passed through a pre-emphasis filter (first order FIR).
- 2. Filtered signal is subject to short-time Fourier transform (STFT) followed by magnitude spectrum computation.
- 3. A set of filters is applied to the magnitude spectrum, with triangular weight functions constructed in such a way that the prescribed frequency range is divided into overlapping bands of equal mel-frequency width.
- 4. Log-scaled filterbank energies are decorrelated via a Discrete Cosine Transform Type II (DCT-II) to form cepstrum coefficients.
- 5. In the last step a sine lifter is optionally applied to cepstra to align coefficient magnitudes.

In general, the computation procedure follows the MFCC features extraction algorithm adopted in the Hidden Markov Models Toolkit (HTK), as descibed in:

S. Young, G. Evermann, M. Gales, T. Hain, D. Kershaw, X. Liu, G. Moo re, J. Odell, D. Ollason, D. Povey, V. Valtchev, P. Woodland, The HTK Book (for HTK version 3.4), Cambridge University Engineering Department, 2009. http://htk.eng.cam.ac.uk/docs/docs.shtml

In addition, a number of options provide an ability to emulate the operation of another popular package for speech analysis:

The Auditory Toolbox for MATLAB by Malcolm Slaney, Version 2, Interval Research Corporation <a href="https://engineering.purdue.edu/~malcolm/interval/1998-010/">https://engineering.purdue.edu/~malcolm/interval/1998-010/</a>

Besides of the primary use of the LogMel filterbank as an integral part of the MFCC extractor, it may be utilized on its own by means of a dedicated API, as shown below.

# 2.11.1 Compute Log Mel Filterbank Energies

Description

In the initialization stage, split the specified frequency range into 1/2 overlapping bands of equal melfrequency width, and compute triangular weighting function for each band.

Data processing functions are applied to Fourier image of real signal, specified through the input argument spectra[fftsize/2+1]. Fourier image is converted to magnitude spectrum. For every mel-frequency band, magnitude samples are multiplied by the corresponding triangular weighting function, and summed together to form filterbank energies (FBEs). Finally, log-scaled FBEs are stored to the output argument

logFbe[mfbBandNum].

Precision

2 variants available:

Туре	Description	
32x32	32-bit fixed-point input/output data	
f	Single precision floating-point input/output data	

#### Algorithm

 $logFbe = log(coeffMatrix \cdot | spectra|)$ 

# Parameters of log mel filterbank operation

typedef struct logmel\_params\_tag

Туре	Name	Description	
int	Fs	Sampling rate, Hz	
int	fftSize	FFT size	
fract32	mfbLowFreqQ8	Lowest band's left frequency edge, Hz (Q8).	
fract32	mfbUppFreqQ8	Uppermost band's right frequency edge, Hz (Q8).	
int	mfbBandNum	Number of mel filterbank spectral bands.	
int	opt	Options to control various aspects of MFCC features extraction, ORed	
		combination of LOGMEL OPT <> flags (see the table below).	

} logmel params t;

#### Option flags:

Name	Value	Description
LOGMEL_OPT_MELSCALE_HTK	0<<0	Use HTK mapping between linear and mel-scale frequencies.
LOGMEL_OPT_MELSCALE_AUDITORY	1<<0	Use Auditory Toolbox mapping between linear and mel-scale frequencies.
LOGMEL_OPT_FBELOG_NATURAL	0<<1	Compute base-e logarithm of filterbank energies (HTK).
LOGMEL_OPT_FBELOG_BASE10	1<<1	Compute base-10 logarithm of filterbank energies (Auditory).
LOGMEL_OPT_FBNORM_NONE	0<<2	No normalization of filterbank weight functions, peak at 1.0 (HTK).
LOGMEL_OPT_FBNORM_AREA	1<<2	Perform area normalization of filterbank weight functions (Auditory).

#### **Object allocation**

Type Name Description		Description
Input		
logmel_params_t	params	Parameters of log mel filterbank operation

Returns: size of memory to be allocated for an instance object, in bytes, or zero if parameters do not satisfy the restrictions.

#### **Object initialization**

Туре	Name	Description	
Input			
void *	objmem	Memory block allocated for the instance object	
logmel_params_t	params	Parameters of log mel filterbank operation	

Returns: handle to the object, or  $\mathtt{NULL}$  if initialization failed.

# Compute log mel filterbank energies

```
float32_t * restrict logFbe,
const complex_float * restrict spectra,
int scaleExp );
```

Туре	Name	Size	Description
Input	•	•	
complex_fract32, complex_float	spectra	fftSize/2+1	Fourier image of real signal, positive frequencies only; Q31 for 32x32
int	scaleExp		Exponent value to scale the Fourier image by a factor of 2scaleExp:  32x32 For full-scale Q31 real signal the scale exponent should be set to 15 plus the sum of bit shifts applied to data throughout the real-to-complex FFT transform, as indicated by the respective FFT routine.  f For real signal varying in the range [-1,1] the scale exponent should be set to 15.
Output			
fract32, float32_t	logFbe	mfbBandNum	Log-scaled filterbank energies; Q6.25 for 32x32.
Temporary			
void*	pScr		Scratch memory area for the processing function. To determine the scratch area size, use the respective helper function:  logmel<32x32 f>_getScratchSize().

## Determine size of the scratch memory area

```
size_t logmel32x32_getScratchSize( const logmel_params_t * params );
size_t logmelf_getScratchSize ( const logmel_params_t * params );
```

Туре	Name	Description	
Input			
logmel_params_t	params	Parameters of log mel filterbank operation	

Returns: size of scratch memory area, in bytes.

#### Restrictions

## 2.11.2 Compute Mel-Frequency Cepstrum Coefficients

### Description

In the initialization stage, perform preliminary computations that allow efficient processing of audio stream in real-time environment:

- 1. Setup an internal instance of log mel filterbank.
- 2. Optionally invoke a user-supplied callback function to calculate STFT weighting window.
- 3. Compute a DCT Type II transform matrix.
- 4. Optionally pre-compute the sine lifter coefficients.

Audio stream is split into chunks of stftHopLen speech samples, specified through the input argument speech[stftHopLen]. Processing of a speech chunk encompasses the following steps:

- 1. Update the STFT sliding frame.
- 2. Estimate and subtract the DC mean from the source signal (optional).
- 3. Perform pre-emphasis filtering (optional).
- 4. Apply the STFT weighting window (optional).
- 5. Invoke a user-supplied callback function to perform the real-to-complex FFT.
- Pass the resulting Fourier image to log mel filterbank to obtain log-scaled filterbank energies (LogFBEs).
- 7. Decorrelate LogFBEs through the DCT Type II transform, this forms the cepstral coefficients.
- 8. Align magnitudes of cepstral coefficients by the sine lifter (optional).

## Precision

Resulting cepstral coefficients are strored to the output argument cepstralcepstrallems. 2 variants available:

Туре	Description	
32x32	32-bit fixed-point input/output data	
f	Single precision floating-point input/output data	

## Parameters of MFCC features extraction

typedef struct mfcc\_params\_tag

Туре	Name	Default Value	Description
int	Fs	16000	Sampling rate, Hz
int	scaleExp	15	Specifies the scaling factor applied to speech signal: 2 <sup>scaleExp</sup> .
fract16	preemph	31785	Pre-emphasis filter coefficient, Q15; set to 0 to disable the filter.
int	fftSize	512	FFT size
int	stftWinLen	400	Short-time Fourier transform window length, must not
			exceed fftSize.
int	stftHopLen	160	Number of audio samples between successive
			windows, must not exceed stftWinLen.
fract32	mfbLowFreqQ8	0	Lowest band's left frequency edge, Hz (Q8).
fract32	mfbUppFreqQ8	4000*256	Uppermost band's right frequency edge, Hz (Q8).
int	mfbBandNum	20	Number of mel filterbank spectral bands.
int	cepstraNum	12	Number of cepstral coefficients to compute, including the 0th coefficient.
int	lifter	22	Cepstral lifter parameter; zet to zero to disable the lifter.
int	opt	0	Options to control various aspects of MFCC features extraction, ORed combination of MFCC_OPT_<>
			and LOGMEL_OPT_<> flags (see the table below).

<sup>}</sup> mfcc\_params\_t;

## Option flags:

Name	Value	Description
LOGMEL_OPT_MELSCALE_HTK	0<<0	Use HTK mapping between linear and mel-scale frequencies.
LOGMEL_OPT_MELSCALE_AUDITORY	1<<0	Use Auditory Toolbox mapping between linear and mel-scale frequencies.
LOGMEL_OPT_FBELOG_NATURAL	0<<1	Compute base-e logarithm of filterbank energies (HTK).
LOGMEL_OPT_FBELOG_BASE10	1<<1	Compute base-10 logarithm of filterbank energies (Auditory).
LOGMEL_OPT_FBNORM_NONE	0<<2	No normalization of filterbank weight functions, peak at 1.0 (HTK).
LOGMEL_OPT_FBNORM_AREA	1<<2	Perform area normalization of filterbank weight functions (Auditory).
MFCC_OPT_REMOVE_DC_MEAN	0<<3	For every STFT frame, evaluate and subtract the DC mean (HTK).
MFCC_OPT_DONT_REMOVE_DC_MEAN	1<<3	Do not remove DC mean (Auditory).
MFCC_OPT_PREEMPH_FRAMEBYFRAME	0<<4	Pre-emphasis filter state is reset between STFT frames (HTK).
MFCC_OPT_PREEMPH_CONTINUOUS	1<<4	Pre-emphasis filter state is reset once, during initialization (Auditory).
MFCC_OPT_DCT_NORMALIZED	0<<5	Multiply the DCT-II matrix by $\sqrt{2/N}$ , where N is the
		transform size (HTK).

MFCC_OPT_DCT_ORTHOGONAL	1<<5	As above, and multiply the first row by $\frac{1}{\sqrt{2}}$
		(Auditory).

# STFT weighting window generator callback function

Туре	Name	Size	Description
Input			
void*	host		User-supplied host handle
int	len		Length of the weighting window
Output			
fract32, float32_t	window	len	Generated weighting window samples; Q31 for 32x32; aligned by 8-bytes

## Real-to-complex FFT callback function

Туре	Name	Size	Description	
Input				
void*	host		User-supplied host handle	
int	fftSize		Transform size	
fract32, float32_t	Х	fftSize	Time-domain input signal	
Output				
<pre>complex_fract32, complex_float</pre>	У	fftSize/2+1	Positive-frequency spectrum samples	

32x32 variant of the function must return the block exponent of the Fourier image, which is the sum of bit shifts applied to data throughout the transform. Block exponent is a signed quantity, where positive values render shifting data to the right.

#### Notes:

- 1. Input and output arguments x[fftSize] and y[fftSize/2+1] do not overlap, and are aligned by 8-bytes.
- 2. The FFT function is allowed to re-use the input argument x[fftsize] for temporal storage of intermediate data.

Callbacks for fixedpoint MFCC extractor (32x32).

```
typedef struct mfcc32x32_callback_tag
{
```

Туре	Name	Description
void*	host	User-supplied host handle to be passed as the first input argument of a callback function.
<pre>mfcc32x32_genWindow_cbfxn_t *</pre>	genWindow	Optional STFT weighting window generator function. If NULL, then MFCC assumes the rectangular window, otherwise it invokes the user-supplied function during initialization.
mfcc32x32_rfft_cbfxn_t*	rfft	Real-to-complex FFT function, is called once per STFT sliding window update.

} mfcc32x32\_callback\_t;

## Callbacks for floatingpoint MFCC extractor (single precision).

typedef struct mfccf\_callback\_tag

Туре	Name	Description
void*	host	User-supplied host handle to be passed as the first input argument of a callback function.
mfccf_genWindow_cbfxn_t *	genWindow	Optional STFT weighting window generator

		function. If NULL, then MFCC assumes the rectangular window, otherwise it invokes the user-supplied function during initialization.
mfccf_rfft_cbfxn_t*	rfft	Real-to-complex FFT function, is called once per STFT sliding window update.

} mfccf callback t;

# Fill the parameters structure with default values

void mfcc getDefaultParams( mfcc params t \* params );

Туре	Name	Description	
Output			
mfcc_params_t	params	Default parameters of MFCC features extraction.	

#### **Object allocation**

Type Name		Description
Input		
mfcc_params_t	params	Parameters of MFCC features extraction.

Returns: size of memory to be allocated for an instance object, in bytes, or zero if parameters do not satisfy the restrictions (see below).

### Object initialization

Туре	Name	Description
Input		
void *	objmem	Memory block allocated for the instance object
mfcc_params_t	params	Parameters of MFCC features extraction.
<pre>mfcc32x32_callback_t, mfccf_callback_t</pre>	callback	User-supplied callback functions

Returns: handle to the object, or NULL if initialization failed.

### Perform the sliding window STFT analysis and extract the MFCC features

Туре	Name	Size	Description
Input			
fract32, float32 t	speech	stftHopLen	Speech samples; Q31 for 32x32.
Output		•	
fract32, float32_t	cepstra	cepstraNum	Cepstral coefficients; the number of fractional bits for 32x32 is defined by MFCC_CEPSTRA_FRACT_BITS macro constant.
Temporary			
void*	pScr		Scratch memory area for the processing function. To determine the scratch area size, use the respective helper function:  mfcc<32x32 f> getScratchSize().

## 2.12 Image Processing

This set of routines allows to process image data. Most routines support three formats:

- 8-bit unsigned grayscale data (suffix \_gu8 in the API)
- 8-bit signed grayscale data (suffix \_gs8 in the API)
- 16-bit signed grayscale data (suffix \_gs16 in the API)

For 16-bit format all negative values are saturated to zero.

All images are represented by width, height and stride between successive rows. Pixel order is from left to right and from upside to downside. For performance reasons, all functions are divided into 2 categories:

- generic functions which process images with arbitrary parameters
- 'fast' functions which assume that the pointer for each row is aligned on 8-byte boundary. This also means that the pointer to the image is aligned and stride is a multiple of 8 for 8-bit images and multiple of 4 for 16-bit images.

As a general rule, all image transforms do not touch pixels outside output image area defined by width and height which means that output would be directly mapped existing image of bigger size.

typedef struct {

Туре	Name	Description	
unsigned int	width	width of image	
unsigned int	height	height of image	
unsigned int	stride	stride between consecutive rows, in pixels. Should be not less than width	

<sup>}</sup> imgsize\_ t;

## 2.12.1 Image Padding/Cropping

3 variants available:

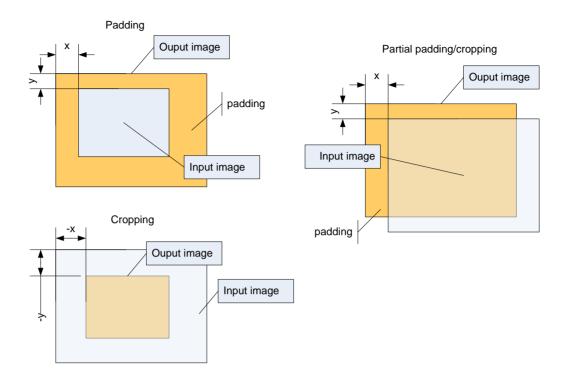
#### Description

Padding/cropping functions make left/right/top/bottom padding for input image. They copy input image to the output at specific position and fills borders either with specified color or by duplication of pixels from corresponding edges.

Region to be copies is set by position of left/upper corner of input in the coordinates of output image. If this region is outside the boundaries of output image, then function will crop required region from the input with padding of unused regions

## Image formats

Туре	Description			
gu8	8-bit unsigned grayscale data			
gs8	8-bit signed grayscale data			
gs16	16-bit signed grayscale data			



### **Prototypes**

Туре	Name	Description	
imgsize_t	in	input image size	
imgsize_t	out	output image size	
int	х, у	position of left upper corner of input image on the output if positions of input image corners are outside the output image, function will crop required region from the input	
int	fill	fill color or special constant IMGPAG_EDGE to use edge pixels	

## } imgpad\_params\_t;

#### **Arguments**

Туре	Name	Size	Description			
Input	Input					
	inImg		input image data			
imgpad_params_t	params	1	parameters			
Output						
	outImg		output mage data			
Temporary						
	pScr		scratch memory, size is requested by corresponding			
			xxx_getScrtactSize () function			

Returned value none

**Restrictions** see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.2 Image Histogram

#### Description

Functions compute the histogram of input image. First bin corresponds minimum intensity (0 for signed 16-bit and unsigned 8-bit data, -128 for signed 8-bit) and the last bin corresponds to the maximum intensity (32767 for signed 16-bit, 255 for unsigned 8-bit data, 127 for signed 8-bit data)

#### Image formats

3 variants available:

Туре	Description			
gu8	8-bit unsigned grayscale data			
gs8	8-bit signed grayscale data			
gs16	16-bit signed grayscale data			

### **Prototypes**

#### typedef struct {

Туре	Name	Description	
int32_t	mean	mean, Q15.16	
int32_t	var	variance. Q31.0 (integral with no fractional bits)	
int	М	number of bins	
int32_t*	h	histogram values	

#### } imghist\_t;

#### **Arguments**

Туре	Name	Size	Description		
Input					
	inImg		input image data		
imgsize_t	SZ	1	image size		
int32_t*	res->h	М	pointer to the array of histogram values		
int	М		number of bins		
Output					
imghist_t	res	1	resulted statistics		

#### Returned value

none

Restrictions

see general definitions in 2.12

## 2.12.3 Image Normalization

#### Description

Function normalize the intensity of pixels to the given range.

Linear normalization takes min/max values from the image and linearly normalizes them to the given range.

Nonlinear method maps input pixel values to the output range using tabulated transform with interpolation

#### Image formats

#### 3 variants available:

Туре	Description			
gu8	8-bit unsigned grayscale data			
gs8	8-bit signed grayscale data			
gs16	16-bit signed grayscale data			

#### **Prototypes**

#### Linear method

```
void imgnorm gu8
                      ( void * outImg, const void * inImg, const imgsize t* sz,
                       int minInt, int maxInt);
void imgfastnorm gu8 ( void * outImg, const void * inImg, const imgsize t* sz,
                       int minInt, int maxInt);
void imgnorm gs8
                     ( void * outImg, const void * inImg, const imgsize t* sz,
                       int minInt, int maxInt);
void imgfastnorm gs 8( void * outImg, const void * inImg, const imgsize t* sz,
                       int minInt, int maxInt);
                     ( void * outImg, const void * inImg, const imgsize t* sz,
void imgnorm gs16
                       int minInt, int maxInt);
void imgfastnorm gs16( void * outImg, const void * inImg, const imgsize t* sz,
                       int minInt, int maxInt);
```

#### Non-linear method

```
void imgnorm gu8 nonlinear
                                 ( void * outImg, const void * inImg,
                                 const imgsize t* sz, const int16 t * tbl);
                                ( void * outImg, const void * inImg,
void imgfastnorm gu8 nonlinear
                                 const imgsize t* sz, const int16 t * tbl);
void imgnorm gs8 nonlinear
                              ( void * outImg, const void * inImg,
                                 const imgsize_t* sz, const int16 t * tbl);
void imgfastnorm gs8 nonlinear ( void * outImg, const void * inImg,
                                  const imgsize t* sz, const int16 t * tbl);
void imgnorm gs16 nonlinear
                                ( void * outImg, const void * inImg,
                                 const imgsize t* sz, const int16 t * tbl);
void imgfastnorm gs16 nonlinear ( void * outImg, const void * inImg,
                                 const imgsize t* sz, const int16 t * tbl);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImg		input image data
imgsize_t	SZ	1	image size
int	minInt, maxInt		output range (for linear method)
int16_t	tbl	64	64-points tabulated transform (for nonlinear method), see example in 4.3
Output			
	outImg		output image data

Returned value

none

Restrictions

see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.4 RGB/YUV Conversion

Description

Functions make conversion between RGB and YUV formats. RGB/YUV data go in separate planes

Image formats

2 variants available:

Туре	Description
------	-------------

	8-bit unsigned data
16	16-bit signed data

#### Algorithm

```
Y = c_0 R + c_1 G + c_2 B
U = c_3 R + c_4 G + c_5 B + bias
V = c_6 R + c_7 G + c_8 B + bias
YUV > RGB
R = Y + c_9 (V - bias)
G = Y + c_{10} (U - bias) + c_{11} (V - bias)
B = Y + c_{12} (U - bias)
```

where bias is 128 for 8-bit data, 16384 for 16-bit data

Transform coefficients are selected to fit appropriate encoding standard

## **Prototypes**

#### From RGB to YUV:

RGB->YUV

```
void * outImgV,
                        const void * inImgR,
                        const void * inImgG,
                        const void * inImgB,
                        const int32 t *c,
                        const imgsize t* sz);
void imgfastconvert rgbyuv
                      ( void * outImgY,
                        void * outImgU,
                        void * outImgV,
                        const void * inImgR,
                        const void * inImgG,
                        const void * inImgB,
                        const int32 t *c,
                        const imgsize t* sz);
void imgconvert_rgbyuv16( void * outImgY,
                        void * outImgU,
                        void * outImgV,
                        const void * inImgR,
                        const void * inImgG,
                        const void * inImgB,
                        const int32 t *c,
                        const imgsize t* sz);
void imgfastconvert rgbyuv16
                      ( void * outImgY,
                        void * outImgU,
                        void * outImgV,
                        const void * inImgR,
                       const void * inImgG,
const void * inImgB,
                        const int32 t *c,
                        const imgsize t* sz);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImgR, inImgG, inImgB		input image data for RGB color planes
int32_t	С	13	transform coefficients, Q29
imgsize_t	SZ	1	image size
Output			_
	outImgY, outImgU, outImgV		output image data for YUV color planes

#### **Prototypes**

#### From YUV to RGB:

```
void * outImgB,
                         const void * inImgY,
const void * inImgU,
                         const void * inImgV,
                         const int32_t *c,
                         const imgsize t* sz);
void imgfastconvert yuvrgb
                       ( void * outImgR,
                         void * outImgG,
                         void * outImgB,
                         const void * inImgY,
                         const void * inImgU,
                         const void * inImgV,
                         const int32 t *c,
                         const imgsize t* sz);
 void imgconvert_yuvrgb16
                       ( void * outImgR,
   void * outImgG,
                         void * outImgB,
                         const void * inImgY,
                         const void * inImgU,
                         const void * inImgV,
                         const int32 t *c,
                         const imgsize_t* sz);
void imgfastconvert_yuvrgb16
                       ( void * outImgR,
                         void * outImgG,
                         void * outImgB,
                         const void * inImgY,
const void * inImgU,
                         const void * inImgV,
                         const int32 t *c,
                         const imgsize_t* sz);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	<pre>inImgY, inImgU, inImgV</pre>		input image data for YUV color planes
int32_t	С	13	transform coefficients, Q29
imgsize_t	SZ	1	image size
Output			
	outImgR, outImgG, outImgB		output image data for RGB color planes

Returned value

none

Restrictions

see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.5 Image Interleave/Deinterleave

## Description

Functions make conversion between packed and planar image representation. In the packed representation pixels from different color planes are interleaved so they are grouped in 24/48-bit chunks. It is assumed that the width and the stride of packed image are 3 times bigger than for input image 2 variants available:

#### Image formats

Туре	Description			
	8-bit signed or unsigned data			
16	16-bit signed data			

### **Prototypes**

#### From planar to packed:

```
void imginterleave
                            void * restrict outImg,
                        const void * restrict inImgR,
                        const void * restrict inImgG,
                        const void * restrict inImgB,
                        const imgsize t* sz);
                       ( void * restrict outImg,
 const void * restrict inImgR,
void imgfastinterleave(
                        const void * restrict inImgG,
                        const void * restrict inImgB,
                        const imasize t* sz);
                              void * restrict outImg,
void imginterleave16 (
                        const void * restrict inImgR,
                        const void * restrict inImgG,
                        const void * restrict inImgB,
                        const imgsize t* sz);
void imgfastinterleave16 ( void * restrict outImg,
                        const void * restrict inImgR,
                        const void * restrict inImgG,
                        const void * restrict inImgB,
                        const imgsize t* sz);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImgR, inImgG, inImgB		input image data for RGB color planes
imgsize_t	SZ	1	input image size
Output			
	outImg		output packed image data

#### **Prototypes**

### From packed to planar:

```
void imgdeinterleave (
                               void * outImgR,
                               void * outImgG,
                               void * outImgB,
                         const void * inImg,
                         const imgsize t* sz);
void imgfastdeinterleave(
                               void * outImgR,
                               void * outImgG,
                               void * outImgB,
                         const void * restrict inImg,
                         const imgsize t* sz);
void imgdeinterleave16(
                               void * outImgR,
                               void * outImgG,
                               void * outImgB,
                         const void * inImg,
                         const imgsize_t* sz);
                               void * outImgR,
void * outImgG,
void imgfastdeinterleavel6(
                               void * outImgB,
```

```
const void * restrict inImg,
const imgsize t* sz);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImg		input packed image data
imgsize_t	SZ	1	output image size
Output			
	outImgR,		output image data for RGB color planes
	outImgG,		
	outImgB		

#### Returned value

none

#### Restrictions

see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.6 Image Resize

### Description

These functions resize images. There are 3 methods for transformation: neighbor, bilinear and bicubic. Resize process is divided onto initialization and processing phases. On the initialization, function returns the size of memory requesting for persistent data (i.e. coefficients) needed for downsampling and upsampling operations. In the processing phase, input image is converted to the output image according to the transformation parameters. Additionally, image resize requires some amount of temporary (scratch) data which may not be kept between successive calls of processing function and might be reused.

#### Image formats

### 3 variants available:

Туре	Description		
gu8	8-bit unsigned grayscale data		
gs8	8-bit signed grayscale data		
gs16	16-bit signed grayscale data		

#### **Prototypes**

#### typedef struct {

Туре	Name	Description
imgsize_t	in	size of input image
imgsize_t	out	size of output image
<pre>imgresize_method_t</pre>	method	resize method: one of 3 possible constants: imgresize method nearest imgresize method bilinear imgresize method bicubic

<sup>}</sup> imgresize params t;

#### Allocation functions:

#### Initialization functions:

```
imgresize_handle_t imgresize_gu8_init
   ( void * objmem, const imgresize_params_t * params );
imgresize_handle_t imgresize_gs8_init
   ( void * objmem, const imgresize_params_t * params );
imgresize_handle_t imgresize_gs16_init
   ( void * objmem, const imgresize_params_t * params );
imgresize_handle_t imgfastresize_gu8_init
   ( void * objmem, const imgresize_params_t * params );
imgresize_handle_t imgfastresize_gs8_init
   ( void * objmem, const imgresize_params_t * params );
imgresize_handle_t imgfastresize_gs16_init
   ( void * objmem, const imgresize_params_t * params );
```

#### Processing functions:

#### requesting the scratch size

```
size_t imgresize_gu8_getScratchSize ( const imgresize_params_t * params );
size_t imgresize_gs8_getScratchSize ( const imgresize_params_t * params );
size_t imgresize_gs16_getScratchSize ( const imgresize_params_t * params );
size_t imgfastresize_gu8_getScratchSize ( const imgresize_params_t * params );
size_t imgfastresize_gs8_getScratchSize( const imgresize_params_t * params );
size_t imgfastresize_gs16_getScratchSize( const imgresize_params_t * params );
```

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImg		input image data
imgresize_params_t	params	1	input image size
Output			
	outImg		output image data

#### Returned value

none

#### Restrictions

see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.7 Image Rotation

#### Description

Image rotation functions rotate input image counter clockwise about their center. They use:

- dedicated variants for 90, 180 and -90 degree rotations
- 3-shear steps for small angles in range -pi/4...pi/4
- proper combinations of above variants for other angles

Since all of these algorithms show very different computational speed, overall transformation performance depends not only from the image size, but from the rotation angle as well.

Except for the rotation on 90, 180 and -90 degrees, the size of resulted image depends on the rotation angle and all unused portions of image will be filled with the fill color. Required output size should be requested by <code>xxx\_getOutSize()</code> API, however user may change the stride of output image to make possible direct mapping to existing image framebuffer.

Rotation API consists of 5 categories:

- allocation function returning amount of memory for persistent data (i.e. interpolation arrays) required for rotation of image of given size to the given angle
- initialization function which compute and fill all persistent data
- requesting the scratch size required for transform
- requesting the size of output image
- processing function for rotation a single image

Note: if user processes images of the same size and with the same rotation angle, the next allocation, initialization functions might be omitted and only processing function will be called

#### Image formats

#### 3 variants available:

o varianto avalia	varianto avaliable.				
Туре	Description				
gu8	8-bit unsigned grayscale data				
gs8	8-bit signed grayscale data				

#### **Prototypes**

gs16	16-bit signed grayscale data

typedef struct {

Туре	Name	Description
imgsize_t	in	size of input image
int16_t	angleQ15	angle scaled by pi, counterclockwise, in Q15, (so 90 degrees counter clockwise corresponds to 0x4000, 180 degrees - 0x8000, 270 degrees -0xc000)
int	fill	fill color for unused image parts

} imgrotate params t;

#### Allocation functions:

```
size_t imgrotate_gu8_alloc ( const imgrotate_params_t * params );
size_t imgfastrotate_gu8_alloc ( const imgrotate_params_t * params );
size_t imgrotate_gs8_alloc ( const imgrotate_params_t * params );
size_t imgrotate_gs8_alloc ( const imgrotate_params_t * params );
size_t imgrotate_gs16_alloc ( const imgrotate_params_t * params );
size_t imgfastrotate_gs16_alloc ( const imgrotate_params_t * params );
```

#### Initialization functions:

```
imgrotate handle t imgrotate gu8 init
                                            (void * objmem,
                                            const imgrotate params t * params);
imgrotate handle t imgfastrotate gu8 init
                                           (void * objmem,
                                            const imgrotate params t * params);
imgrotate handle t imgrotate gs8 init
                                          (void * objmem,
                                            const imgrotate_params_t * params);
imgrotate handle t imgfastrotate gs8 init (void * objmem,
                                            const imgrotate params t * params);
imgrotate handle t imgrotate gs16 init
                                         (void * objmem,
                                            const imgrotate params t * params );
imgrotate handle t imgfastrotate gs16 init(void * objmem,
                                             const imgrotate params t * params);
```

#### Request output image size:

#### Processing functions:

#### Requesting the scratch size:

```
size_t imgrotate_gu8_getScratchSize ( const imgrotate_params_t * params );
size_t imgfastrotate_gu8_getScratchSize ( const imgrotate_params_t * params );
size_t imgrotate_gs8_getScratchSize ( const imgrotate_params_t * params );
size_t imgrotate_gs8_getScratchSize ( const imgrotate_params_t * params );
size_t imgrotate_gs16_getScratchSize ( const imgrotate_params_t * params );
```

size t imgfastrotate gs16 getScratchSize( const imgrotate params t \* params );

#### **Arguments**

Туре	Name	Size	Description
Input			
	inImg		input image data
imgresize_params_t	params	1	input image size
imgsize_t	outSz	1	width, height and stride of output image.  NOTE: width and height should be set by  xxx_getOutSize() API, but the stride might be changed by user
Output			
	outImg		output image data

#### Returned value

none

#### Restrictions

see general definitions in 2.12

Input and output image buffers can not overlapping

## 2.12.8 Image 2D FFT

#### Description

Image 2D-FFT makes special kind of Fourier transform over image data. It supports a number of resolutions. For processing of arbitrary sizes, user might pad original data to the closest available dimension.

#### Important notes:

- Accuracy of the fixed point 2D FFT suffers from the presence of unknown overall image brightness so for the optimal results, mean pixel value should be precomputed (i.e. by histogram function) and passed on the input of FFT.
- FFT processes the data with autoscaling similarly as it is done in regular FFT routines so it returns the scaled transform and number of right shifts done during scaling procedure. Inverse FFT uses this scale factor to recover original data range and saturates results to 8-bit unsigned/signed or 16-bit signed data.
- IFFT ignores zero-th tap of transform. Instead, it uses dedicated parameter to set mean intensity of the output image
- 4. FFT accepts input in desired format and form output in form of 2D array of complex 16-bit numbers
- IFFT makes in-place transform so input spectrum will be damaged after the call

### Supported dimensions:

64, 96, 128, 144, 176, 240, 256, 288, 320, 352, 384, 480, 512, 576, 640

## **Precision**

#### 3 variants available:

Туре	Description
gu8	8-bit unsigned grayscale data
gs8	8-bit signed grayscale data
gs16	16-bit signed grayscale data

#### Prototype (FFT)

#### Request the scratch size:

```
size_t imgfft_gu8_getScratchSize (const imgsize_t* sz);
size_t imgfft_gs8_getScratchSize (const imgsize_t* sz);
size_t imgfft_gs16_getScratchSize (const imgsize_t* sz);
```

#### **Arguments**

Туре	Name	Size	Description	
Input				
uint8_t, int16_t	img		input image	
imgsize_t	SZ	1	size, see list of supported dimensions	
int32_t	mean		mean value (average intensity) over the image,	
			Q15.16	
Output				
complex_fract16	У	2*w*h	output spectrum. Real and imaginary data are	
			interleaved and real data goes first. w and h -	
			width and height of original image	

## Returned value

total number of right shifts occurred during scaling procedure

## Prototype (IFFT)

#### Inverse transform:

#### Request the scratch size:

```
size_t imgifft_gu8_getScratchSize (const imgsize_t* sz);
size_t imgifft_gs8_getScratchSize (const imgsize_t* sz);
size t imgifft gs16 getScratchSize (const imgsize t* sz);
```

#### **Arguments**

Туре	Name	Size	Description
Input			
complex_fract16	У	2*w*h	input spectrum. $\mathbf{w}$ and $\mathbf{h}$ - width and height of output image
int32_t	mean		desired mean value (average intensity) over the output image, Q15.16
int	fftscale		scale factor returned by FFT
imgsize_t	SZ	1	output image size
Output			
uint8_t, int16_t	img		output image

#### Returned value

none

Restrictions

 ${\tt img,y}$   $\;$  should be aligned on a 8-bytes boundary

image dimensions should be selected from the list of supported dimensions image stride should be a multiple of 8 for 8-bit images and 4 for 16-bit images

## 2.13 Identification Routines

## 2.13.1 Library Version Request

**Description** This function returns library version information.

**Prototype** 

void NatureDSP\_Signal\_get\_library\_version(char \*version\_string);

**Arguments** 

Туре	Name	Size	Description
------	------	------	-------------

Output			
char	version_string	>=30	buffer to store version information

Returned value

None

Restrictions

version\_string must points to a buffer large enough to hold up to 30 characters

Conditions for optimum performance:

None

## 2.13.2 Library API Version Request

**Description** 

This function returns library API version information.

**Prototype** 

void NatureDSP\_Signal\_get\_library\_api\_version(char \*version\_string);

**Arguments** 

Туре	Name	Size	Description	
Output				
char	version_string	>=30	buffer to store version information	

Returned value

None

Restrictions

version\_string must points to a buffer large enough to hold up to 30 characters

## 2.13.3 Library API Capability Request

## **Description**

This function returns non-zero if given function (by its address) is supported by specific processor capabilities (i.e. VFPU/SFPU option).

#### NOTE:

- in the gcc/xcc environment, calls of this function are not necessary if function pointer is nonzero it means it is supported. VisualStudio linker does not support section removal so this function might be used for running library under MSVC environment
- 2. Very few library functions may disable their capabilities dynamically (only for particular combination of input parameters). Behavior for such situation is defined in the description of those functions.

#### **Prototype**

int NatureDSP Signal isPresent(NatureDSP Signal funptr fun)

#### **Arguments**

Туре	Name	Description
Input:		
NatureDSP_Signal_funptr	fun	one of library functions

#### **Returned Value**

non-zero if function is supported by library

## 3 Test Environment and Examples

## 3.1 Supported Use Environment, Configurations and Targets

NatureDSP Signal library and corresponding testdriver is supported to be built and test using Xtensa Xplorer IDE running under Windows, or Linux operating system.

Library is compatible with HiFi cores having following options:

- HiFi4 base ISA
- HiFi4 Vector/Scalar FP
- NSA/NSAU ISA option
- MIN/MAX ISA option
- Boolean Registers ISA option
- Little endian target

## 3.2 Building the NatureDSP Signal Library and the Testdriver

## 3.2.1 Importing the workspaces in Xtensa Xplorer

NatureDSP Signal Library for HiFi4 VFPU is provided as two workspaces:

- Library workspace HiFi4 VFPU library.xws
  - This workspace contains optimized kernels, modules as required for demo workspaces
- Demo workspace HiFi4 VFPU demo.xws
  - This contains the HiFi4 VFPU demo demo project.

Import these two workspaces (.xws) in Xtensa Xplorer (XX) as "Xtensa Xplorer workspace".

Make sure that the library workspace is imported first. This is because the project in the demo workspace has a dependency on the library projects, and the dependency is not correctly set if the library projects are not present when the demo workspace is imported.

## 3.2.2 Building and Running Tests

To build the library: In XX, select the desired library project to build, and Debug or Release target, and build.

To build the test bench: In XX, select the  $\mbox{HiFi4}_{\tt VFPU\_demo}$  project, select Debug or Release target, and build.

To run the test bench, select <code>HiFi4\_VFPU\_demo</code> project, and Run. This will execute each routine in the <code>HiFi4\_VFPU</code> library in cycles performance (MIPS) mode.

For performance measurement with memory modeling, build the test bench with MEM\_MODEL=1 (set in build properties) and use --mem\_model as runtime argument for execution.

Use --turbo as runtime argument to test library for functional correctness

## 3.2.3 Command-line Options

You may wish to launch a separate test by passing command-line options to the executable: Running the Testdriver without options performs functional testing of library. Additionally, it may collect statistics and generate validation report showing the number of calls of each specific library function, amount of data passed to/from, sorts of specific tests performed, etc.

Running performance tests for all library functions or for specific category is controlled by command line option <code>-mips</code>. In that case, functional testing is not performed and validation report will be empty.

Brief performance data are formed with -mips -verbose. Detailed performance data are prepared with -mips -full.

You may wish to launch a separate test by passing command-line options to the executable:

Package	API	Meaning	Option
		List of available options	-help or -h
		Performance test	-mips
		Functional tests	-func
		Generate validation report and statistics after completion of functional testing	-vreport
		test fixed point functions only	-phase1
		test floating point functions only	-phase2
		For functional tests, this switch instructs to use bigger data vectors from directory vectors_full instead of vectors_brief (test time might be 3 to 5 times longer). For performance test, it controls amount of tests performed for each library function. With this switch the test tool forms detailed testing using bigger set of function parametes, if not - it just makes brief performance data.	-full
		This switch controls amount of tests executed (in contrary with – full)	-brief
		Verbose reporing. For functional tests, it controls verbosilty of test reports (i.e. shows number, type of tests performed and detailed error statistics if some test is failed). For performance tests, it adds textual description of each function under the test to the performance log.	-verbose
FIR Filters and Related	2.1	all FIR filters	-fir
Functions		filtering	-firblk
		decimation	-firdec
		interpolation	-firint
		correlation, convolution, dispreading, LMS	-firother
		2D-convolution	-conv2d
IIR Filters	2.1.13	all IIR filters	-iir
		biquad filters	-iirbq
		biquad filters without algorithmic delay	-iirbqnd
		lattice filters	-iirlt
Math Functions	2.3	all math functions	-math
		vectorized math	-mathv
		vectorized math (fast variants)	-mathvf
		scalar math	-maths
Complex Math	2.4	all complex functions	-complex
Functions		vectorized complex math	-complexv
		scalar complex math	-complexs
Vector Operations	2.5	vector operations tests	-vector
Emulated Floating Point	2.6	emulated floating point tests	-ef

Package	API	Meaning	Option
Operations			
Matrix Operations	2.7	all matrix operations tests	-matop
Matrix Decomposition and Inversion Functions	2.8	all matrix decomposition and inversion	-matinv
		Gauss-Jordan	-gs
		Cholesky	-chol
FFT Routines	2.10	all FFT and DCT	-fft
		complex FFT	-cfft
		real FFT	-rfft
		mixed radix complex FFT	-cnfft
		mixed radix real FFT	-rnfft
		complex FFT with optimized memory	-cfftie
		real FFT with optimized memory	-rfftie
		power spectrum	-spectrum
		DCT	-dct
MFCC	2.11	MFCC feature extraction	-mfcc
Image processing	2.12	all image processing functions	-img
		image resize	-imgresize
		image rotate	-imgrotate
		image 2D FFT	-imgfft
		normalization, histogram, conversion, interleaving,	-imgmisc
		padding/cropping	
Fitting and Interpolation	2.9	all fitting tests	-fit
Routines		polynomial fitting	-pfit

## 4 Appendix

# 4.1 Matlab Code for Conversion of SOS Matrix to Coefficients of IIR Functions

Below is example Matlab code to simplify conversion of SOS+G matrices given from the filter design tools into the format of IIR filtering functions.

## 4.1.1 bqriir16x16\_df1, bqriir32x16\_df1 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16_df1 function)
% parameters:
% SOS,G - SOS matrix and gain vector G
          - sample rate
% Fs
          - FFT length for analisys
% nfft
% output:
% coef - vector with coefficients, Q30
% gain
          - biquad gains, Q15
         - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir32x16 df1(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
coef=[];
f = (0:nfft-1)/nfft*(Fs/2):
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax=max(abs(tf));
    G(m) = min(1, 0.5/tfmax);
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b,a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int16(round(16384.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
% check results and plot final filter response
sos=reshape (double (coef), 5, M). '/16384;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
plot(f, 20*log10(abs(tf)), f, 20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
```

```
tf=freqz(b,a,nfft);
for m=2:M
    [b,a]=sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

## 4.1.2 bqriir16x16\_df2, bqriir32x16\_df2 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16 df2 function)
% parameters:
% SOS,G
                      - SOS matrix and gain vector G
% Fs
                       - sample rate
% nfft
                        - FFT length for analisys
% output:
                        - vector with coefficients, Q14
% coef
                      - biquad gains, Q15
% gain
% scale
                   - final scale factor (amount of left shifts)
function [coef, gain, scale] = cvtsos bqriir32x16 df2(SOS, G, Fs, nfft)
sz=size(SOS):
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for
% intermediate outputs <=0.5
% note: for DF2 structure we have to check 2 points:
% b0,b1,b2,a1,a2 and 1,0,0,a1,a2
for m=1:M
         tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
         tfmax0=max(abs(tf));
         tf=sos2freqz([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1],nfft);
        tfmax1=max(abs(tf));
         tfmax = max(tfmax0,tfmax1);
        G(m) = min(1,0.5/tfmax);
end
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1, scale)/dg;
G(M) = G(M) / d;
% output b,a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int16(round(16384.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
% check results and plot final filter response
sos=reshape(double(coef),5,M).'/16384;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
\verb|plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid('transfer function, 
on;
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a] = sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
```

```
[b,a]=sos2tf(SOS(m,:),[G(m) 1]);
tf=tf.*freqz(b,a,nfft);
end
```

## 4.1.3 bqriir32x32\_df1 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x32 df1 function)
% parameters:
         - SOS matrix and gain vector G
% SOS,G
% Fs
           - sample rate
% nfft
           - FFT length for analisys
% output:
% coef
           - vector with coefficients, Q30
% gain
           - biquad gains, Q30
          - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos_bqriir32x32_df1(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax=max(abs(tf));
    G(m) = min(1,0.5/tfmax);
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b,a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int32(round(1073741824.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
\mbox{\ensuremath{\$}} check results and plot final filter response
sos=reshape(double(coef),5,M).'/1073741824;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
    [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

## 4.1.4 bqriir32x32\_df2 conversion

```
%-----% convert SOS+G to coefficients of IIR filter
```

```
% (bgriir32x32 df2 function)
% parameters:
% SOS,G
          - SOS matrix and gain vector G
           - sample rate
% Fs
% nfft
           - FFT length for analisys
% output:
           - vector with coefficients, Q30
% coef
           - biquad gains, Q15
% gain
          - final scale factor (amount of left shifts)
% scale
function [coef,gain,scale]=cvtsos bqriir32x32_df2(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1):
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for
% intermediate outputs <=0.5</pre>
% note: for DF2 structure we have to check 2 points:
% b0,b1,b2,a1,a2 and 1,0,0,a1,a2
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax0=max(abs(tf));
    tf=sos2freqz([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1],nfft);
    tfmax1=max(abs(tf));
    tfmax = max(tfmax0, tfmax1);
    G(m) = min(1,0.5/tfmax);
end
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b, a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int32(round(1073741824.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
% check results and plot final filter response
sos=reshape(double(coef),5,M).'/1073741824;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
    [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

### 4.1.5 bgriirf df1, bgriirf df2, bgriirf df2t conversion

```
% SOS,G
         - SOS matrix and gain vector G
        sample rateFFT length for analisys
응 Fs
% nfft
% output:
% coef
           - vector with coefficients
function [coef,scale]=cvtsos iir(SOS,G,Fs,nfft)
% convert SOS+G to coefficients of IIR filter
sz=size(SOS);
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax=max(abs(tf));
    G(m) = min(1.0.5/tfmax);
for m=1:M
    SOS(m, 1:3) = SOS(m, 1:3) *G(m);
end
% define last stage shift
dg=Gtotal/prod(G);
scale=round(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b,a
coef=SOS'; coef(4,:)=[]; coef=reshape(coef,1,numel(coef));
scale= int32(scale);
% check results and plot final filter response
sos=reshape(double(coef),5,M);
sos=sos';
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g(1:M) = ones(1,M);
q(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
plot(f,20*loq10(abs(tf)),f,20*loq10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
\mbox{\%} convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS):
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
    [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

# 4.2 Matlab Code for Generation the Twiddle Tables

FFT with optimized memory usage require external twiddle tables. Matlab code below shows how to generate twiddles for different functions.

4.2.1 Twiddles for fft\_cplx32x16\_ie, ifft\_cplx32x16\_ie, fft\_real32x16\_ie, ifft\_real32x16\_ie, fft\_cplx16x16\_ie, ifft\_cplx16x16\_ie, ifft\_real16x16\_ie, ifft\_real16x16\_ie

```
function [twd]=twd32x16_ie(N)

twd = \exp(-2j*pi*[1;2;3]*(0:N/4-1)/N);
```

## 4.3 Matlab Code for Nonlinear Image Normalization

Below is example Matlab code to generate tables for 8-bit and 16-bit nonlinear image normalization using sigmoid transformation.

#### 8-bit variant:

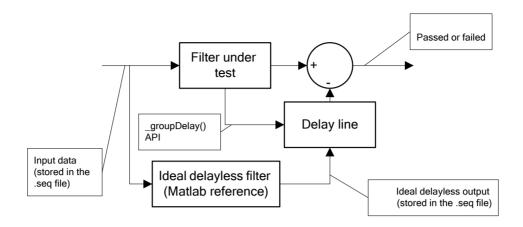
```
x=((0:63)+0.5)/64;
alpha=10; % density factor
y=1./(1+exp(-(x*alpha-alpha/2)));
tbl=uint8(round(y*256));

16-bit variant:

x=((0:63)+0.5)/64;
alpha=10; % density factor
y=1./(1+exp(-(x*alpha-alpha/2)));
tbl=int16(round(y*32768));
```

## 4.4 About IIR filter validation

Biquad filters may introduce algorithmic delay and amount of this delay is requesting by corresponding API for each specific filter. However, for the testing of all kinds of biquad filters we use **non-delayed** reference data taken from Matlab ideal filters. In fact, the same data are used both for filters with and without algorithmic delay. In the test harness the groupDelay() API is called and manages the extra time shift at the output of ideal output. This shifted reference are comparing with filter under test output and test harness tool decides if test passed or not by checking against threshold. The picture below illustrates this process:



## **5 Customer Support**

If you have questions, want to report problems or suggestions regarding the **NatureDSP Signal** library or want to port this library to another platforms, contact **IntegrIT** Ltd. at <a href="mailto:support@integrit.com">support@integrit.com</a>. Visit <a href="www.integrit.com">www.integrit.com</a> to get more information about products and services.