

# NatureDSP Signal Library for Fusion F1

**Digital Signal Processing** 

**Library Reference** 

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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 Fusion F1 little endian targets. Call **IntegrIT** to support more targets or other CPUs.

# **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

# 1 General Library Organization

# 1.1 Headers

NatureDSP\_Signal library is supplied with several header files

```
./include/NatureDSP_types.h
./include/NatureDSP_Math.h
./include/NatureDSP_baseopXtensa.h
NatureDSP_Signal.h
```

Declarations of basic data types and compiler auto detection Prototypes of basic operations (see below) Mapping of basic operations to Xtensa-specific intrinsics Declarations of library functions

# 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
f24	24-bit fractional type	4
int16_t	16-bit signed value	2
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 Manual that constant pointers passed through function arguments point at read-only data.

Normally, £24 fractional data are stored 3 higher bytes of 32-bit words and 8 LSBs are ignored, however, few routines use packed 24-bit data where 24-bit fractional numbers allocates only 3 consecutive bytes.

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

# 1.4 Fractional Formats

Natively, Fusion F1 CPU uses a special fractional type £24 which is stored in a memory as a 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 Qm.

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
int16 t	Q3.12	-8 7 <b>,</b> 9998	2e-4	-32768	32767
int16 t	Q8.7	-256 255 <b>,</b> 992	8e-3	-32768	32767
int32 t	Q1.30	-2 1 <b>,</b> 9999999991	9e-10	-2147483648	2147483647
int32 t	Q0.31	-1 0 <b>,</b> 9999999995	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
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 <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 Q format number. Thus, in 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 Fusion F1 Instruction Set Architecture. See section 1.12 for exact target details.

# 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.), the 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 (NaN). 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 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 the two last methods.

# 1.12 Fusion F1 Supported ISA Options

Fusion F1 core has various ISA options or simply options. The library has optimizations that take advantage of these specific options. Instructions used in the library require the following Fusion F1 ISA options:

- Audio/Voice/Speech (AVS)
- 16-bit Quad MAC
- Floating point (FP)

The library is conditionalized for the Fusion F1 ISA options and the implementation of the routines are selected according to the configuration options. For floating point kernels FP option needs to be selected for the core configuration.

# 1.13 Object Model

Effective use of all Fusion F1 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 Fusion F1 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

\_

<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

functions	Block real FIR filter Complex block FIR filter Decimating block real FIR filter Interpolating block real FIR filter Circular convolution Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation Blockwise Adaptive LMS algorithm	2.1.1, 2.1.2 2.1.3 2.1.4 2.1.5 2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11 2.1.12
	Complex block FIR filter Decimating block real FIR filter Interpolating block real FIR filter Circular convolution Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.3 2.1.4 2.1.5 2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Complex block FIR filter Decimating block real FIR filter Interpolating block real FIR filter Circular convolution Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.3 2.1.4 2.1.5 2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Decimating block real FIR filter Interpolating block real FIR filter Circular convolution Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.4 2.1.5 2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Interpolating block real FIR filter Circular convolution Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.5 2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Circular convolution  Linear convolution  Circular correlation  Linear correlation  Circular autocorrelation  Linear autocorrelation	2.1.6 2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Linear convolution Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.7 2.1.8 2.1.9 2.1.9 2.1.11
	Circular correlation Linear correlation Circular autocorrelation Linear autocorrelation	2.1.8 2.1.9 2.1.9 2.1.11
	Linear correlation Circular autocorrelation Linear autocorrelation	2.1.9 2.1.9 2.1.11
	Circular autocorrelation Linear autocorrelation	2.1.9 2.1.11
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	Lattice block Real IIR	2.2.2
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_		2.3.10
<del>_</del>		2.3.11
_		2.3.12
		2.3.13
		2.3.13
	Arctangent	2.3.14, 2.3.15
scl bexp	Common exponent	2.3.16
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	This a maximum/minimum in a voctor	
	Polynomial approximation	2.3.18
scl_int2float	Integer to float conversion	2.3.19
scl_float2int	Float to integer conversion	2.3.20
scl_float2floo r	Rounding	2.3.21
scl float2ceil		
scl_complex2ma	Complex magnitude	2.3.22
g	, ,	0.000
=	Reciprocal of complex magnitude	2.3.22
viiiag		
	scl antilogn scl_sqrt scl_sqrt scl_rsqrt scl_sine scl_cosine scl_tan scl_asin scl_acos scl_atan, scl_atan2 scl_bexp  scl_int2float scl_float2floo r scl_float2ceil scl_complex2ma	Vector dot product  Vector sum  Power of a vector  Vector scaling with saturation  scl_recip Reciprocal  scl_divide Division  scl_logn Different kinds of logarithm  scl_logn Scl_antilog2 Scl_antilog10  scl_antilog1 Scl_antilogn  scl_sqrt Square root  scl_resqrt Reciprocal square root  scl_sine Sine  scl_cosine Cosine  scl_cosine Cosine  scl_asin Arcsine  scl_asin Arcsine  scl_acos Arccosine  scl_atan, scl_atan, scl_atan2  scl_bexp Common exponent  Find a maximum/minimum in a vector  Polynomial approximation  scl_int2float Integer to float conversion  scl_float2cit Float to integer conversion  scl_float2ceil scl_complex2ma g  scl_complex2in Reciprocal of complex magnitude

Vectorized version	Scalar version		Reference	
Matrix operations				
mtx_mpy		Matrix mul	Itiply	2.4.1
mtx_vecmpy		Matrix by v	vector multiple	2.4.2
mtx_add, cmtx add		Matrix add		2.4.3
mtx_sub, cmtx sub		Matrix sub	traction	2.4.3
mtx_mul, cmtx mul		Matrix mul	ltiply	2.4.3
mtx_tran, cmtx tran		Matrix tran	nspose	2.4.3
mtx_det, cmtx det		Matrix dete	erminant	2.4.3
mtx_inv, cmtx inv		Matrix inve	erse	2.4.4
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FFT				
fft_cplx		FFT on co	mplex data	2.5.1
fft_real		FFT on rea	al data	2.5.2
ifft_cplx		Inverse FF	T on complex data	2.5.3
ifft_real			T forming real data	2.5.4
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ifft_cplx_ie			T on complex data with optimized memory usage	2.5.8
			T forming real data with optimized memory usage	2.5.9
Identification				
NatureDSP_Signal_o	get_library_vers	ion	Library Version Request	2.6.1
NatureDSP Signal get library api version			Library API Version Request	2.6.2

# 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 must 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  ${\bf h}$ . The real data input is stored in vector  ${\bf x}$ . The filter output result is stored in vector  ${\bf y}$ . The filter calculates  ${\bf n}$  output samples using  ${\bf m}$  coefficients and requires last  ${\bf m}$ -1 samples in the delay line which is updated in circular manner for each new sample.

#### Precision

#### 6 versions available:

Туре	Description		
16x16 16-bit data, 16-bit coefficients, 16-bit outputs			
24x24 24-bit data, 24-bit coefficients, 24-bit outputs			
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		
f	floating point		

#### **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 bkfir16x16\_alloc (int M)
size\_t bkfir24x24\_alloc (int M)
size\_t bkfir24x24p\_alloc(int M)
size\_t bkfir32x16\_alloc (int M)
size\_t bkfir32x32\_alloc (int M)
size\_t bkfirf alloc (int M)

Туре	Name	Size	Description	
Input				
int	М		length of filter, should be a multiple of 4	

Returns: size of memory in bytes to be allocated

#### NOTE

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes
bkfir16x16 alloc	72+M*4
bkfir24x24 alloc	72+M*8

bkfir24x24p_alloc	80+ <u>M</u> *6
bkfir32x16 alloc	72+M*6
bkfir32x32_alloc	72+M*8
bkfirf alloc	72+M*8

## **Object initialization**

Туре	Name	Size	Description
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
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, f24, int32_t, float32 t	Х	N	input samples	
int	N		length of sample block	
Output				
int16_t, f24, int32_t, float32_t	У	N	output samples	

Returns: none

#### Restrictions

```
x,y-should not overlap
```

x,h - aligned on a 8-bytes boundary

 ${\tt N\,,M}\,$  - 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 versions available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
24x24	24-bit data, 24-bit coefficients, 24-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
f	floating point

## **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 bkfira16x16_alloc(int M)
size_t bkfira24x24_alloc(int M)
size_t bkfira32x16_alloc(int M)
size_t bkfira32x32_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
bkfira24x24_alloc	80+ M*8
bkfira32x32_alloc	80+ M*8
bkfiraf alloc	80+ M*8

#### **Object initialization**

Туре	Name	Size	Description
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
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, f24, int32_t, float32 t	Х	N	input samples	
int	N		length of sample block	
Output				
int16_t, f24, int32_t, float32 t	У	N	output samples	

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 n. The complex data input is stored in vector n. The filter output result is stored in vector n. The filter calculates n output samples using n coefficients, requires last n 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).

#### Precision

#### 5 versions available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
24x24	24-bit data, 24-bit coefficients, 24-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
f	floating point

#### 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)
size_t cxfir24x24_alloc(int M)
size_t cxfir32x16_alloc(int M)
size_t cxfir32x32_alloc(int M)
size t cxfirf alloc(int M)
```

Туре	Name	Size	Description
Input			
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
cxfir16x16_alloc	80+8*M
cxfir32x16 alloc	80+12*M
cxfir24x24_alloc	64+16*M
cxfir32x32_alloc	64+16*M
cxfirf alloc	64+16*M

## Object initialization

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
complex_fract32, complex_fract16, complex_float	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

Returns: handle to the object

# Update the delay line and compute filter output

Туре	Name	Size	Description	
Input		-		
<pre>complex_fract16, complex_fract32, complex_float</pre>	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  $\mathbf{h}$ . 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 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  $\mathbf{D}$  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 versions available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit outputs
24x24	24-bit data, 24-bit coefficients, 24-bit outputs
32x16	32-bit data, 16-bit coefficients, 32-bit outputs
32x32	32-bit data, 32-bit coefficients, 32-bit outputs
f	floating point

#### **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 firdec24x24_alloc(int D, int M)
size_t firdec32x16_alloc(int D, int M)
size_t firdec32x32_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
firdec16x16 alloc	40+ (M+4*D) *4+ (M+4) *2
firdec32x16_alloc	40+ (M+8*D) *4+ (M+4) *2
firdec24x24 alloc	40+ (M+8*D) *4+ (M+4) *4
firdec32x32_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
f24, 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, f24, 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					
int16_t, f24, int32_t, float32 t	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/Complex FIR Filter

#### **Description**

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

#### Precision

#### 6 versions available:

Туре	Description
16x16	16-bit real data, 16-bit coefficients, 16-bit real outputs
16x16	16-bit complex data, 16-bit coefficients, 16-bit complex outputs
24x24	24-bit real data, 24-bit coefficients, 24-bit real outputs
32x16	32-bit real data, 16-bit coefficients, 32-bit real outputs
32x32	32-bit real data, 32-bit coefficients, 32-bit real outputs
f	floating point

#### 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 cxfirinterp16x16_alloc(int D, int M)
size_t firinterp24x24_alloc(int D, int M)
size_t firinterp32x16_alloc(int D, int M)
size_t firinterp32x32_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
cxfirinterp16x16 alloc	40+ (M+8) *4+ (M+4) *D*2
firinterp32x16_alloc	40+ (M+8) *4+ (M+4) *D*2
firinterp24x24 alloc	40+ (M+8) *4+ (M+4) *D*4
firinterp32x32_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)
cxfirinterp16x16 handle t cfirinterp16x16 init(void * objmem,
                            int D, int M, const int16 t * h)
firinterp24x24 handle t firinterp24x24 init(void * objmem,
int D, int M, const f24 * 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)
firinterpf handle t firinterpf init(void * objmem,
                            int \overline{D}, int M, const float32 t * h)
```

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
f24, 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

```
void firinterp16x16 process(firinterp16x16 handle t handle,
                        int16 t * y, const int16 t * x, int N);
void cxfirinterp16x16 process(cfirinterp16x16 handle t handle,
                         complex fract16 * y, const complex fract16 * x,
                         int N);
void firinterp24x24_process(firinterp24x24_handle_t handle,
int32_t* y, const int32_t* x, int N);
void firinterp32x32 process(firinterp32x32 handle t handle,
                         int32 t* y, const int32 t* x, int N);
void firinterpf process
                         (firinterpf handle t handle,
                         float32 t * y, const float32 t * x, int N);
```

Туре	Name	Size	Description	
Input	•			
int16_t, complex_fract16, f24, int32_t, float32 t	Х	N	input samples,Q15, Q31 or floating point	
int	N		length of input sample block	
Output				
int16_t, complex_fract16, f24, int32_t, float32 t	У	N*D	output samples, Q15, Q31 or floating point	

Returns: none

#### Restrictions

x,h,y should not overlap

x, h - aligned on a 8-bytes boundary

м - multiples of 4 N - multiples of 8 D should be >1 D - 2, 3 or 4

# optimum performance

**Conditions for** 

#### 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 variants of these functions available: faster version (fir\_convol16x16, fir\_convol24x24, fir\_convol32x16, cxfir\_convol32x16, fir\_convol32x32, fir\_convolf) with some restrictions on input arguments and slower version (fir\_convola16x16, fir\_convola24x24, fir\_convola32x16, cxfir\_convola32x16, fir\_convola32x32, fir\_convolaf) for arbitrary arguments. In addition, these slower version implementations require scratch memory area. 5 versions available:

## Precision

Туре	Description
16x16	16x16-bit data, 16-bit outputs
24x24	24x24-bit data, 24-bit outputs
32x16	32x16-bit data, 32-bit outputs (both real and complex)
32x32	32x32-bit data, 32-bit outputs
f	floating point

#### Algorithm

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

#### **Prototype**

```
void fir_convol16x16 (int16_t* r,
           const int16 t * x, const int16 t * y, int N, int M);
void fir convol24x24 (f24 * r,
           const f24 * x, const f24 * y, int N, int M);
void fir convol32x16 ( int32 t * r,
           const int32_t * x, const int16_t * y, int N, int M);
void cxfir convol32x16 ( complex fract32 * r,
           const complex fract3\overline{2} * x, const complex fract16 * 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 convolf ( float32 t * r,
            const float32 t * x, const float32_t * y, int N, int M);
void fir convola16x16 (void * s,
                      int16 t * r,
                const int16_t * x, const int16_t * y, int N, int M);
void fir convola24x24 (void *
                      f24 * r,
                const f24 * x, const f24 * y, int N, int M);
void fir convola32x16 (void *
                      int32 t * r,
                const int32_t * x, const int16_t * y, int N, int M);
void cxfir convola32x16 (void * s,
                      complex_fract32 * r,
                const complex fract32 * x, const complex fract16 * 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 convolaf
                     (void
                      float32 t * r,
                const float32 t * x, const float32 t * y, int N, int M);
```

## **Arguments**

Туре	Name	Size	Description
Input	I.		
int16_t, f24,	Х	N	input data (Q15, Q31 or floating point)
int32_t, complex fract32,			
float32 t			
int32 t, f24,			
int16 t,		М	input data (O21, O15 or floating point)
complex_fract16,	У	141	input data (Q31, Q15 or floating point)
or float32_t			
int	N		length of x
int	М		length of y
Output			-
int16_t, f24,	r	N	output data, Q15, Q31 or floating point
int32_t,			
complex_fract32,			
float32 t			
Temporary			
void	s		Scratch memory,
			FIR CONVOLA16X16 SCRATCH SIZE( N, M)
			FIR CONVOLA24X24 SCRATCH SIZE( N, M )
			FIR_CONVOLA32X16_SCRATCH_SIZE( N, M )
			CXFIR_CONVOLA32X16_SCRATCH_SIZE(N,M)
			FIR_CONVOLA32X32_SCRATCH_SIZE( N, M )
			FIR_CONVOLAF_SCRATCH_SIZE( N, M )
			bytes

### Returned value

#### none

#### Restrictions

```
For slow versions (fir_convola16x16, fir_convola24x24, fir_convola32x16, cxfir_convola32x16, fir_convola32x32, fir_convolaf): x,y,r,s should not overlap s should be aligned on 8-byte boundary N>0, M>0 N>=M-1
```

```
For fast versions (fir_convol16x16, fir_convol24x24, fir_convol32x16, cxfir_convol32x16, fir_convol32x32, fir_convolf): x,y,r should not overlap x,y,r should be aligned on 8-byte boundary N>0, M>0 N,M - multiples of 4
```

#### 2.1.7 Linear Convolution

Description

Functions perform linear convolution between vectors  $\mathbf{x}$  (of length  $\mathbf{n}$ ) and  $\mathbf{y}$  (of length  $\mathbf{m}$ ) resulting in vector  $\mathbf{r}$  of length  $\mathbf{n}+\mathbf{m}-1$ .

**Precision** 

4 versions available:

Туре	Description
16x16	16x16-bit data, 16-bit outputs
32x16	32x16-bit data, 32-bit outputs
32x32	32x32-bit data, 32-bit outputs
f	floating point

**Algorithm** 

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

**Prototype** 

#### **Arguments**

Туре	Name	Size	Description
Input		•	
<pre>int16_t, int32_t, float32 t</pre>	Х	N	input data (Q15, Q31 or floating point)
<pre>int16_t, int32_t, float32 t</pre>	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	М		length of y
Output	•		· •
<pre>int16_t, int32_t, float32 t</pre>	r	M+N-1	output data, Q15, Q31 or floating point
Temporary			
void	S		Scratch memory,  FIR LCONVOLA16X16 SCRATCH SIZE ( N, M)  FIR_LCONVOLA32X16_SCRATCH_SIZE ( N, M)  FIR_LCONVOLA32X32_SCRATCH_SIZE ( N, M)  FIR_LCONVOLAF_SCRATCH_SIZE ( N, M)  bytes

Returned value

none

Restrictions

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

s should be aligned on 8-byte boundary

N>0, M>0 N>=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  $\mathbf{n}$ ) resulting in vector  $\mathbf{r}$  of length  $\mathbf{n}$ . It is a similar to convolution but  $\mathbf{y}$  is read in opposite direction.

Two variants of these functions available: faster version (fir\_xcorr16x16, fir\_xcorr24x24, fir\_xcorr32x16, fir\_xcorr32x32, fir\_xcorrf, cxfir\_xcorrf) with some restrictions on input arguments and slower version (fir\_xcorra16x16, fir\_xcorra24x24, fir\_xcorra32x16, fir\_xcorr32x32, fir\_xcorraf, cxfir\_xcorraf) for arbitrary arguments. In addition, these slower version implementations require scratch memory area.

#### Precision

#### 5 versions available:

Туре	Description
16x16	16x16-bit data, 16-bit outputs
24x24	24x24-bit data, 24-bit outputs
32x16	32x16-bit data, 32-bit outputs
32x32	32x32-bit data, 32-bit outputs
f	floating point (both real and complex data)

#### Algorithm

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

#### **Prototype**

```
void fir xcorr16x16 ( int16 t *
              const int16 t * x, const int16 t * y, int N, int M);
                            * r,
void fir xcorr24x24 ( f24
               const f24
                            * x, const f24
                                              * y, int N, int M);
                           * r,
void fir xcorr32x16 ( int32 t
                            * x, const int16_t * y, int N, int M);
              const int32 t
* r,
                \frac{-}{\text{(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 xcorra16x16 ( void
                            * s,
                     int16 t * r,
               const int16 t *
                               x, const int16 t * y, int N, int M);
                             * s,
void fir xcorra24x24 (void
                            * r,
                    £24
                            * x, const f24
               const f24
                                               * y, int N, int M);
void fir xcorra32x16 (void
                   int32 t
                               r,
                            * x, const int16_t * y, int N, int M);
              const int32_t
void fir xcorra32x32 ( void
                               s,
                    int32 t *
                               r,
               const int32 t *
                               x, const int32 t * y, int N, int M);
void fir_xcorraf
                    (void
                              s,
                     float32 t*
                               r,
               const float32_t* x, const float32_t* y, int N, int M);
                            * s,
void cxfir xcorraf
                  (void
                    complex float *
                                   r,
               const complex float * x, const complex float * y,
               int N, int M);
```

## **Arguments**

Туре	Name	Size	Description
Input			
int16_t, f24,	Х	N	input data (Q15, Q31 or floating point)
int32_t,			
float32_t, complex float			
f24, int16 t,		+	
float32 t,	У	М	input data (Q31, Q15 or floating point)
complex float	1	11	input data (&o1, &10 of floating point)
int	N		length of x
int	М		length of y
Output		•	
int16 t, f24,	r	N	output data, Q15, Q31 or floating point
int32_t,			Traper and, are, as a meaning point
float32_t,			
complex float			
Temporary			
void	S		Scratch memory,
			FIR XCORRA16X16 SCRATCH SIZE(N, M)
			FIR XCORRA24X24 SCRATCH SIZE(N, M)
			FIR XCORRA32X32 SCRATCH SIZE(N, M)
			FIR_XCORRAF_SCRATCH_SIZE(N, M)
			CXFIR_XCORRAF_SCRATCH_SIZE(N, M)
			FIR_XCORRA32X16_SCRATCH_SIZE(N, M)
			bytes

### Returned value

#### none

#### Restrictions

For slow versions (fir\_xcorra16x16, fir\_xcorra24x24, fir\_xcorra32x16, fir\_xcorra32x32, fir\_xcorraf, cxfir\_xcorraf):
x,y,r,s should not overlap
s should be aligned on 8-byte boundary
N>0, M>0
N>=M-1

For fast versions (fir\_xcorr16x16, fir\_xcorr24x24, fir\_xcorr32x16, fir\_xcorr32x32, fir\_xcorrf, exfir\_xcorrf): 
x,y,r should not overlap 
x,y,r should be aligned on 8-byte boundary 
N>0, M>0 
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

4 versions available:

Туре	Description
16x16	16x16-bit data, 16-bit outputs
32x16	32x16-bit data, 32-bit outputs
32x32	32x32-bit data, 32-bit outputs
f	floating point

**Algorithm** 

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

**Prototype** 

## **Arguments**

Туре	Name	Size	Description
Input	•	•	
int16_t, int32_t, float32_t	Х	N	input data (Q15, Q31 or floating point)
<pre>int16_t, int32_t, float32_t,</pre>	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	M		length of y
Output	•	•	. •
<pre>int16_t, int32_t, float32_t</pre>	r	M+N-1	output data, Q15, Q31 or floating point
Temporary			
void	5		Scratch memory,  FIR LXCORRA16X16 SCRATCH SIZE (N, M)  FIR LXCORRA32X16 SCRATCH SIZE (N, M)  FIR LXCORRA32X32 SCRATCH SIZE (N, M)  FIR LXCORRAF 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 x. Returns autocorrelation of length x.

Two variants of these functions available: faster version ( $fir_acorr24x24$ ,  $fir_acorrf$ ) with some restrictions on input arguments and slower version ( $fir_acorra24x24$ ,  $fir_acorraf$ ) for arbitrary arguments. In addition, this slower version implementations require scratch memory area.

#### Precision

4 versions available:

Туре	Description
16x16	16-bit data, 16-bit outputs
24x24	24-bit data, 24-bit outputs
32x32	32-bit data, 32-bit outputs
f	floating point

#### Algorithm

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

### **Prototype**

#### **Arguments**

Туре	Name	Size	Description	
Input	•			
int16_t, f24, int32_t or	Х	N	input data (Q15, Q31 or floating point)	
float32 t				
int	N		length of x	
Output				
int16_t, f24,	r	N	output data, Q15, Q31 or floating point	
int32_t or float32 t				
Temporary	1			
void	s		Scratch memory,	
			FIR ACORRA16X16 SCRATCH SIZE( N )	
			FIR ACORRA24X24 SCRATCH SIZE( N )	
			FIR ACORRA32X32 SCRATCH SIZE ( N )	
			FIR ACORRAF SCRATCH SIZE( N )	
			bytes	

#### Returned value

none

#### Restrictions

For slow versions (fir\_acorr16x16, fir\_acorr24x24, fir\_acorr32x32, fir\_acorrf): x,r,s should not overlap N - must be non-zero s - aligned on an 8-bytes boundary

For fast versions (fir\_acorra16x16, fir\_acorra24x24, fir\_acorra32x32, fir\_acorraf): x,r should not overlap x,r should be aligned on 8-byte boundary N>0
N -multiple of 4

## 2.1.11 Linear Autocorrelation

 $\textbf{Description} \qquad \qquad \text{Functions estimate the linear auto-correlation of vector } \texttt{x.} \ \text{Returns autocorrelation of length } \texttt{N.}$ 

**Precision** 

3 versions available:

Туре	Description		
16x16	16-bit data, 16-bit outputs		
32x32	32-bit data, 32-bit outputs		
f	floating point		

Algorithm

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

**Prototype** 

```
void fir lacorral6x16 (void* s, int16 t * r, const int16 t * x, int N); void fir lacorra32x32 (void* s, int32 t * r, const int32 t * x, int N); void fir_lacorraf (void* s, float32_t* r, const float32_t* x, int N);
```

**Arguments** 

Туре	Name	Size	Description
Input		•	
<pre>int16_t, int32_t or float32 t</pre>	Х	N	input data (Q15, Q31 or floating point)
int	N		length of x
Output		•	-
<pre>int16_t, int32_t or float32 t</pre>	r	N	output data, Q15, Q31 or floating point
Temporary			
void	5		Scratch memory,  FIR LACORRA16X16 SCRATCH SIZE( N )  FIR_LACORRA32X32_SCRATCH_SIZE( N )  FIR_LACORRAF_SCRATCH_SIZE( N )  bytes

Returned value

none

Restrictions

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

N>0

 $_{\mbox{\scriptsize S}}\,\,$  - aligned on an 8-bytes boundary

## 2.1.12 Blockwise Adaptive LMS Algorithm for Real Data

#### Description

Blockwise LMS algorithm performs filtering of input samples x[N+M-1], computation of error e[N] over a block of reference data x[N] and makes a 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 input signal times  $\tt N$  the number of samples in a data block. This can be calculated, i.e., by using the  $\tt vec\ power24x24()$  or  $\tt vec\ power16x16()$  functions. In order to avoid the saturation of the normalization factor, it may be biased, i.e., shifted to the right. If that is 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  ${\tt N}$  depends on the change rate of impulse response: on static or slow varying channels convergence rate depends on selected  ${\tt mu}$  and  ${\tt M}$ , but not on  ${\tt N}$ .
- 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

#### Precision

#### 5 versions available:

Туре	Description
16x16	16-bit coefficients, 16-bit data, 16-bit output
24x24	24-bit coefficients, 24-bit data, 32-bit output
16x32	32-bit coefficients, 16-bit data, 16-bit output
32x32	32-bit coefficients, 32-bit data, 32-bit output
f	floating point

#### **Algorithm**

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

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

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

#### **Prototype**

```
void fir blms16x16 ( int16 t* e, int16 t * h,
             const int16 t * r,
const int16_t * x,
             int16 t norm, int16 t
const f24 * r,
             const f24 * x,
             f24 norm, f24 int N, int
                            mu,
                            M);
void fir blms16x32 ( int32 t * e, int32 t * h,
             const int16 t * r,
             const int16 t *
             int32_t norm,int16_t mu,
const int32 t * x,
             int32 t norm, int32 t mu,
                     N, int
              ( float32 t * e, float32_t * h, const float32_t * r,
void fir blmsf
             const float32_t * x,
             float32 t norm, float32 t mu,
                        N, int
```

# Arguments

Туре	Name	Size	Description		
Input					
int16_t, f24,	h	М	impulse response, Q15, Q31 or floating point		
int32_t,					
float32 t					
f24, int16_t,	r	N	reference (near end) data vector. First in time value is		
int32_t or			in r[0], Q31, Q15 or floating point		
float32 t					
f24, int16_t,	X	N+M-1	input (far end) data vector. First in time value is in x[0],		
int32_t or			Q31, Q15 or floating point		
float32 t			, , , , , , , , , , , , , , , , , , , ,		
int16_t, f24,	norm		normalization factor: power of signal multiplied by N,		
int32_t, float32 t			Q15, Q31 or floating point		
			1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1 1		
f24, int16_t	mu		adaptation coefficient in Q31, Q15 or floating point		
int32_t, float32 t			(LMS step)		
int	N		lawath of data black		
	IN		length of data block		
int	M		length of h		
Output					
f24, int16_t,	е	N	estimated error, Q31,Q15 or floating point		
int32_t,			, . ,		
float32_t					
f24, int16_t,	h	M	updated impulse response, Q15, Q31		
int32_t,					
float32_t					

## Returned value

none

Restrictions

 $\texttt{h,x,r,y,e} \; \; \text{-} \; \text{should not overlap}$ 

x, e, h, r - aligned on a 8-bytes boundary N, M - multiples of 8

# 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.

Filters are able to process data in following formats:

- real (just array of samples)
- 2-way or complex (interleaved real/imaginary samples)
- 3-way (stream of interleaved samples from 3 channels)

The same coefficients are used for filtering of multiple channels or real/imaginary parts and they are processed independently.

The same format have to be used both for input and output streams.

- NOTES:
- 1. 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 coef\_g[M] 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 nonnegative. It is specified through the total gain shift amount parameter gain of each filter initialization function.
- 2. 16x16 filters may suffer more from accumulation of the roundoff errors, so filters should be properly designed to match noise requirements

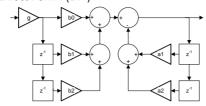
## Precision 10 versions available:

Туре	Description
16x16	16-bit data, 16-bit coefficients, 16-bit intermediate stage outputs (DF I, DF II), real data
16x16	16-bit data, 16-bit coefficients, 16-bit intermediate stage outputs (DF I, DF II), 3-way data
24x24	32-bit data, 24-bit coefficients, 32-bit intermediate stage outputs (DF I, DF II), real data
32x16	32-bit data, 16-bit coefficients, 32-bit intermediate stage outputs (DF I, DF II), real data
32x16	32-bit data, 16-bit coefficients, 32-bit intermediate stage outputs (DF I, DF II), 3-way data
32x32	32-bit data, 32-bit coefficients, 32-bit intermediate stage outputs (DF I, DF II)
32x32	32-bit data, 32-bit coefficients, 32-bit intermediate stage outputs (DF I, DF II) 3-way data
f	floating point (DF I, DF II and DF IIt)
f	floating point (DF I), 2-way (complex) data
f	floating point (DF I, DF II) 3-way data

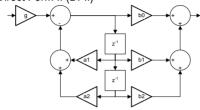
## **Algorithm**

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:

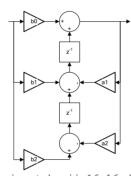
Direct Form I (DFI)



Direct Form II (DFII)



Direct Form II transposed (DF IIt)



## **Object allocation**

```
size_t bqriir16x16_df1_alloc(int M)
size_t bqriir16x16_df2_alloc(int M)
size_t bq3iir16x16_df1_alloc(int M)
size_t bq3iir16x16_df2_alloc(int M)
size_t bqriir24x24_df1_alloc(int M)
size_t bqriir24x24_df1_alloc(int M)
size_t bqriir32x16_df1_alloc(int M)
size_t bqriir32x16_df1_alloc(int M)
size_t bq3iir32x16_df1_alloc(int M)
size_t bq3iir32x16_df1_alloc(int M)
size_t bq3iir32x32_df1_alloc(int M)
size_t bq7iir32x32_df1_alloc(int M)
size_t bq7iir32x32_df1_alloc(int M)
size_t bq7iir32x32_df1_alloc(int M)
size_t bq7iir32x32_df1_alloc(int M)
size_t bq7iir32x32_df2_alloc(int M)
size_t bq7iirf_df1_alloc(int M)
size_t bq7iirf_df2_alloc(int M)
size_t bq7iirf_df1_alloc(int M)
size_t bq3iirf_df1_alloc(int M)
size_t bq3iirf_df1_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**

```
bqriir16x16 dfl handle t bqriir16x16 dfl init(void * objmem, int M,
           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);
bq3iir16x16_df1_handle_t bq3iir16x16_df1 init(void * objmem, int M,
const int16 t * coef_sos, const int16 t * coef_g, int16 t gain );
bq3iir16x16_df2_handle_t bq3iir16x16_df2_init(void * objmem, int M,
           const int16 t * coef sos, const int16 t * coef g, int16 t gain);
bqriir24x24_df1_handle_t bqriir24x24_df1_init(void * objmem, int M,
           const f24
                          * coef sos, const int16 t * coef g, int16 t gain );
bqriir24x24 df2 handle t bqriir24x24 df2 init(void * objmem, int M,
                         * coef sos, const int16 t * coef g, int16 t gain);
           const f24
bgriir32x16 df1 handle t bgriir32x16 df1 init(void * objmem, int M,
           const int16 t * coef sos, const int16 t * coef g, int16 t gain);
bgriir32x16 df2 handle t bgriir32x16 df2 init(void * objmem, int M,
           const int16_t * coef_sos, const int16_t * coef_g, int16_t gain);
bq3iir32x16 df1 handle t bq3iir32x16 df1 init(void * objmem, int M,
           const int16 t * coef sos, const int16_t * coef_g, int16_t gain);
bq3iir32x16 df2 handle t bq3iir32x16 df2 init(void * objmem, int M,
const int16_t * coef_sos, const int16_t * coef_g, int16_t gain);
bgriir32x32 df1 handle t bgriir32x32 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 q, int16 t gain)
bq3iir32x32_df1_handle_t bq3iir32x32_df1_init(void * objmem, int M,
           const int32 t * coef sos, const int16 t * coef g, int16 t gain)
bq3iir32x32_df2_handle_t bq3iir32x32_df2_init(void * objmem, int M,
           const int32 t * coef sos, const int16 t * coef g, int16 t gain)
bgriirf dfl handle t bgriirf dfl 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 dfl handle t bqciirf dfl init(void * objmem, int M,
           const float32 t * coef sos, int16 t gain);
bq3iirf_df1_handle_t bq3iirf_df1_init(void * objmem, int M,
const float32_t* coef_sos, int16_t gain );
bq3iirf_df2_handle_t bq3iirf_df2_init(void * objmem, int M,
           const float32 t * coef sos, int16 t gain);
```

Туре	Name	Size	Description
Input			
void*	objmem		allocated memory block
int	М		number of bi-quad sections
f24, 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 funcions, fixed point format of filter coefficients is Q1.14 for 32x16, or Q1.30 for 32x16 and 24x24 (in the latter case 8 LSBs are actually ignored).
int16_t	coef_g	М	scale factor for each section, Q15 (for fixed-point functions only). Please note that 24x24 DFI implementation internally truncates scale factors to Q7 values.
int16_t	gain		total gain shift amount, -4815

Returns: handle to the object

# Update the delay line and compute filter output

```
void bgriir16x16 df1(bgriir16x16 df1 handle t bgriir,
           void * s,int16 t * r,const int16 t *x, int N);
void bqriir16x16 df2(bqriir16x16 df2 handle t bqriir,
           void * s,int16 t * r,const int16 t *x, int N);
void bq3iir16x16_df2(bq3iir16x16_df2_handle_t _bqriir,
           void * s,int16 t * r,const int16 t *x, int N);
void bgriir24x24 df1(bgriir24x24_df1_handle_t _bgriir,
           void *
                 s,int32 t * r,const int32 t *x, int N);
void bqriir24x24 df2 (bqriir24x24 df2 handle t bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriir32x16 df2 (bqriir32x16 df2 handle t bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bq3iir32x16 df1(bq3iir32x16 df1 handle t
                                         bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bq3iir32x16_df2(bq3iir32x16_df2_handle_t _bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriir32x32 df1(bqriir32x32_df1_handle_t _bqriir,
           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 bq3iir32x32_df1(bq3iir32x32_df1_handle_t _bqriir,
           void * s,int32 t * r,const int32 t *x, int N);
void bqriirf_dfl (bqriirf_dfl_handle_t,
               float32_t
                          * r, const float32 t * x, int N);
void bgriirf df2 (bgriirf df2 handle t,
               float32 t
                         * r, const float32 t * x, int N);
float32_t * r, con void bqciirf_df1 (bqciirf_df1_handle_t,
               complex float* r, const complex float * x, int N);
void bq3iirf df1 (bq3iirf df1 handle t,
float32_t * r, con
void bq3iirf_df2 (bq3iirf_df2_handle_t,
                          * r, const float32 t * x, int N);
               float32 t
                          * r, const float32 t * x, int N);
```

Functon	Scratch memory, bytes
bqriir16x16 df1	BQRIIR16X16 DF1 SCRATCH SIZE(M)
bqriir16x16 df2	BQRIIR16X16 DF2 SCRATCH SIZE(M)
bq3iir16x16 df1	BQ3IIR16X16 DF1 SCRATCH SIZE(M)
bq3iir16x16 df2	BQ3IIR16X16 DF2 SCRATCH SIZE(M)
bqriir24x24 df1	BQRIIR24X24 DF1 SCRATCH SIZE(M)
bqriir24x24 df2	BQRIIR24X24 DF2 SCRATCH SIZE(M)
bqriir32x16 df1	BQRIIR32X16 DF1 SCRATCH SIZE(M)
bqriir32x16 df2	BQRIIR32X16 DF2 SCRATCH SIZE(M)
bq3iir32x16 df1	BQ3IIR32X16 DF1 SCRATCH SIZE(M)
bq3iir32x16 df2	BQ3IIR32X16 DF2 SCRATCH SIZE(M)
bqriir32x32 df1	BQRIIR32X32 DF1 SCRATCH SIZE(M)
bqriir32x32_df2	BQRIIR32X32_DF2_SCRATCH_SIZE (M)
bq3iir32x32 df1	BQ3IIR32X32 DF1 SCRATCH SIZE(M)
bq3iir32x32 df2	BQ3IIR32X32 DF2 SCRATCH SIZE(M)

Туре	Name	Size	Description
Input	- II	l.	
<pre>int16_t, int32_t, float32_t, complex float</pre>	Х	N	input samples, Q31, Q15 or floating point.  For 3-way functions (bq3iirxxx), N is a number of triplets, so array size should be 3*N.
int	N		length of input sample block. For 3-way functions (bq3iirxxx), N is a number of triplets
Output			
<pre>int16_t, int32_t, float32_t, complex_float</pre>	r	N	output data, Q31, Q15 or floating point.  For 3-way functions (bq3iirxxx), N is a number of triplets, so array size should be 3*N.
Temporary			
void*	Ø		scratch memory area (for fixed-point functions only), Minimum number of bytes depends on selected filter structure and precision. see table above If a particular macro returns zero, then the corresponding IIR doesn't require a scratch area and parameter s may hold zero

## Returned value

none

## Restrictions

 $\begin{array}{c} \mathtt{x,r,s,coef\_g,coef\_sos} \ \text{must not overlap} \\ \mathtt{N} \quad \text{- must be a multiple of 2} \end{array}$ 

 $_{\mbox{\scriptsize s}}\,$  - 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  $\mathbf{k}$ . The real data input are stored in vector  $\mathbf{x}$ . The filter output result is stored in vector  $\mathbf{r}$ . Input scaling is done before the first cascade for normalization and overflow protection.

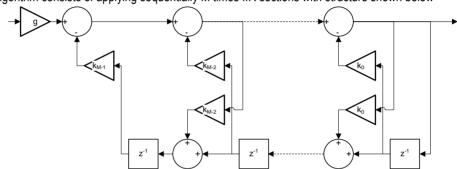
#### Precision

5 versions available:

Туре	Description
16x16	16-bit data, 16-bit coefficients
24x24	24-bit data, 24-bit coefficients
32x16	32-bit data, 16-bit coefficients
32x32	32-bit data, 32-bit coefficients
f	floating point

## Algorithm

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



## Object allocation

```
size_t latr16x16_alloc(int M);
size_t latr24x24_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			
int□	М		number of sections

Returns: size of memory in bytes to be allocated

## **Object initialization**

```
latr16x16 handle t latr16x16 init
  (void * objmem, int M, const int16_t
                                  * k, int16 t
                                                  scale);
latr24x24 handle t latr24x24 init
  (void * objmem, int M, const f24
                                    * k, f24
                                                  scale);
latr32x16_handle_t latr32x16_init
  (void * objmem, int M, const int16 t * k, int16 t
                                                  scale);
latr32x32 handle t latr32x32 init
  scale);
latrf handle t latrf init
  (void * objmem, int M, const float32 t * k, float32 t scale);
```

Туре	Name	Size	Description
Input		•	
void*	objmem		allocated memory block
int	М		number of sections
f24, int16_t, int32_t or float32 t	k	М	reflection coefficients, Q31, Q15 or floating point
f24, 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, f24,	Х	N	input samples, Q15, Q31 or floating point	
int32 t or			mparamphas, and, acres maning pann	
float32 t				
int	N		length of input sample block	
Output				
int16 t, f24,	r	N	output data, Q15, Q31 or floating point	
int32_t or			output data, and, as it is issuing point	
float32 t				

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 Vector Mathematics

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

```
scl_log2f, scl_lognf, scl_log10f, scl_sinef, scl_cosinef, scl_tanf,
scl_atanf, scl_atan2f, scl_antilog2f, scl_antilognf, scl_antilog10f,
scl_asinf, scl_acosf, scl_sqrtf
```

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
  errno 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 errno, 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 errno and may generate exceptions in bit different manner to minimize the overhead.

## 2.3.1 Vector Dot Product

## **Description**

These routines take two vectors and calculates their dot product. Two versions of routines are available: regular versions (vec\_dot24x24, vec\_dot32x16, vec\_dot32x32, vec\_dot16x16, vec\_dotf) work with arbitrary arguments, faster versions (vec\_dot24x24\_fast, vec\_dot32x16\_fast, vec\_dot32x32\_fast, vec\_dot16x16\_fast) apply some restrictions.

#### Precision

#### 5 versions available:

Туре	Description
16x16	16x16-bit data, 64-bit output for regular version and 32-bit for fast version
24x24	24x24-bit data, 64-bit output
32x16	32x16-bit data, 64-bit output
32x32	32x32-bit data, 64-bit output
f	floating point

#### Algorithm

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

## **Prototype**

```
int64 t vec dot24x24
   (const f24 * x, const f24 * y, int N);
int64 t vec dot32x16
  (const int32 t * x, const int16_t * y, int N);
int64 t vec dot16x16
   (const int16_t * x, const int16_t * y, int N);
int64_t vec dot \overline{3}2x32
  (const int32 t * x, const int32 t * y, int N);
float32 t vec dotf
   (const float32 t * x, const float32 t * y, int N);
int64 t vec dot24x24 fast
   (const f\overline{2}4 * x, const f24 * y, int N);
int64 t vec dot32x16 fast
  (const int32 t * x, const int16 t * y, int N);
int64 t vec dot32x2 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			
f24, int32_t, int16_t, float32 t	Х	N	input data, Q31, Q15 or floating point
f24,int16_t, float32_t	У	N	input data, Q31, Q15 or floating point
int	N		length of vectors

# Returned value Restrictions

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

Regular versions ( $vec_dot24x24$ ,  $vec_dot32x16$ ,  $vec_dot32x32$ ,  $vec_dot16x16$ ,  $vec_dotf$ ): None

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.

## 2.3.2 Vector Sum

#### Description

This routine makes pair wise saturated summation of vectors. Two versions of routines are available: regular versions (vec\_add32x32, vec\_add24x24, vec\_add16x16, vec\_addf) work with arbitrary arguments, faster versions (vec\_add32x32\_fast, vec\_add24x24\_fast, vec\_add16x16\_fast) apply some restrictions.

#### Precision

#### 4 versions available:

Туре	Description
32x32	32-bit inputs, 32-bit output
24x24	24-bit inputs, 24-bit output
16x16	16-bit inputs, 16-bit output
f	floating point

## Algorithm

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

## Prototype

```
void vec_add32x32 ( int32_t* z, const int32_t* x, const int32_t* y, int N);
void vec_add24x24 ( f24 * z, const f24 * x, const f24 * 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);
```

```
void vec_add32x32_fast(int32_t* z, const int32_t* x, const int32_t* y,int N);
void vec_add24x24_fast(f24 * z, const f24 * x, const f24 * y, int N);
void vec_add16x16_fast(int16_t* z, const int16_t* x, const int16_t* y,int N);
```

## **Arguments**

Туре	Name	Size	Description
Input			
f24, int32_t,	Х	N	input data
int16_t or			
float32_t			
f24, int32_t,	У	N	input data
int16 t or			
float32 t			
int	N		length of vectors
Output			
f24, int32 t,	Z	N	output data
int16 t or			
float32 t			

## Returned value

#### none

#### Restrictions

Regular versions ( $vec_add32x32$ ,  $vec_add24x24$ ,  $vec_add16x16$ ,  $vec_addf$ ): x,y,z - should not be overlapped

Faster versions (vec\_add32x32\_fast, vec\_add24x24\_fast, vec\_add16x16\_fast): z,x,y - aligned on 8-byte boundary N - multiple of 4

## 2.3.3 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 power24x24, vec power32x32,

vec power16x16, vec powerf) work with arbitrary arguments.

faster versions (vec\_power24x24\_fast, vec\_power32x32\_fast, vec\_power16x16\_fast) apply some restrictions.

#### Precision

#### 4 versions available:

Туре	Description
24x24	24x24-bit data, 64-bit output
32x32	32x32-bit data, 64-bit output
16x16	16x16-bit data, 64-bit output
f	floating point

#### **Algorithm**

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

## **Prototype**

#### **Arguments**

Туре	Name	Size	Description
Input			
f24, int32_t, int16_t, float32_t	Х	N	input data, Q31, Q15 or floating point
int	rsh		right shift of result:  for vec_power32x32():rsh should be in range 3162  for vec_power24x24():rsh should be in range 1546  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)

#### Restrictions

For regular versions (vec\_power24x24, vec\_power32x32, vec\_power16x16, vec\_powerf): none

For faster versions (vec\_power24x24\_fast, vec\_power32x32\_fast, vec\_power16x16\_fast) x - aligned on 8-byte boundary

 ${\scriptscriptstyle N}$  - multiple of 4

# 2.3.4 Vector Scaling with Saturation

#### Description

These routines make shift with saturation of data values in the vector by given scale factor (degree of 2). 24-bit routine works with f24 data type and faster while 32-bit version keep all 32-bits and slower. 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\_shift24x24, vec\_shift32x32, vec\_shift16x16, vec\_shiftf, vec\_scale32x24, vec\_scale24x24, vec\_scale16x16, vec\_scalef, vec\_scale\_sf) work with arbitrary arguments, faster versions (vec\_shift24x24\_fast, vec\_shift32x32\_fast, vec\_shift16x16\_fast, vec\_scale32x24\_fast, vec\_scale24x24\_fast, vec\_scale24x24\_fast, vec\_scale16x16\_fast) apply some restrictions.

#### For floating point:

Fuction 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**

#### 4 versions available:

Туре	Description
24x24	24-bit input, 32-bit output
32x32	32-bit input, 32-bit output
16x16	16-bit input, 16-bit output
f	floating point

## Algorithm

## Prototype

# $r_n = x_n \cdot 2^t$

$r_n - x_n = z$	
<pre>void vec_shift24x24</pre>	
	const f24 * x,
	int t,
	int N);
<pre>void vec_shift32x32</pre>	
	const int32_t * x,
	int t,
	int N);
void vec shift16x16	( int16 t * y,
_	const int16 t * x,
	int t,
	int N);
void vec shiftf	( float32 t * y,
	const float32_t * x,
	int t,
	int N);
void vec shift24x24	fast ( f24 * y,
void vee_biiii ez inz i_	const f24 * x,
	int t,
	int N);
id ahif+3030	
void vec_siiiit32x32_	_fast (
	const int32_t * x,
	int t,
11 116.16.16	int N);
voia vec_shiftl6x16_	_fast ( int16_t * y,
	const int16_t * x,
	int t,
	int N);

## **Arguments**

Туре	Name	Size	Description	
Input				
f24, 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				

f24, int32_t, int16_t or	У	N	output data, Q31, Q15 or floating point
float32 t			

#### **Prototype**

```
non-power 2 scaling
void vec scale32x24 (
                          int32 t * y,
                    const int32_t * x,
                    f24 s,
                    int N);
void vec scale24x24 ( f24 * y,
                    const f24 * x,
                    f24 s,
                    int N);
                        int16 t * y,
void vec scale16x16 (
                    const int16 t * x,
                    int16 t s,
                    int N);
                    ( float32 t * y,
void vec scalef
                    const float32 t * x,
                    float32_t s,
                    int N);
void vec scale sf
                   ( float32 t * restrict y,
                    const float32_t * restrict x, float32_t s, float32_t fmin, float32_t fmax,
                    int N);
void vec scale32x24 fast (
                               int32 t * y,
                    const int32_t * x,
                    f24 s,
                    int N);
void vec scale24x24_fast (
                               f24 * y,
                    const f24 * x,
                    f24 s.
                    int N);
void vec scale16x16 fast (
                               int16 t * y,
                    const int16 t * x,
                    int16 t s,
                    int N);
```

#### **Arguments**

Туре	Name	Size	Description	
Input	•	•		
f24, int32_t, int16_t or float32_t	Х	N	input data, Q31, Q15 or floating point	
f24, 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			· · · · · · · · · · · · · · · · · · ·	
f24, int32_t, int16_t or float32_t	У	N	output data, Q31, Q15 or floating point	

#### Returned value

#### None

#### Restrictions

```
For regular versions (vec_shift24x24, vec_shift32x32, vec_shift16x16, vec_shiftf, vec_scale32x24, vec_scale24x24, vec_scale16x16, vec_scalef, vec_scalesf): x,y should not overlap t should be in range -31...31 for fixed-point functions and -129...146 for floating point
```

For faster versions (vec\_shift24x24\_fast, vec\_shift32x32\_fast, vec\_shift16x16\_fast, vec\_scale32x24\_fast, vec\_scale24x24\_fast, vec\_scale16x16\_fast): x,y should not overlap

 $_{x,\,y}$  should not overlap  $_{t}$  should be in range -31...31  $_{x,\,y}$  - aligned on 8-byte boundary  $_{\rm N}$  - multiple of 4

## 2.3.5 Reciprocal

#### Description

Fixed point routines return the fractional and exponential portion of the reciprocal of a vector  $\mathbf{x}$  of Q31 or Q15 numbers. Since the reciprocal is always greater than 1, it returns 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.

For fixed point finctions, mantissa accuracy is 1 LSB, so relative accuracy is:

vec_recip16x16,	scl_recip16x16	6.2e-5
vec_recip24x24,	scl_recip32x32, scl_recip24x24	2.4e-7
vec_recip32x32		9.2e-10

Floating point routines operate with standard floating point numbers. Functions return +/-infinity on zero or denormalized input and provide accuracy of 1 ULP.

#### Precision

#### 4 versions available:

Туре	Description
32x32	32-bit input, 32-bit output.
24x24	24-bit input, 24-bit output.
16x16	16-bit input, 16-bit output.
f	floating point

## Algorithm

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

$$y_n = 1/x_n, n = \overline{0...N-1}$$
 for floating point functions

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description	
Input				
f24, int32_t, int16_t, float32 t	Х	N	N input data, Q31,Q15 or floating point	
int	N		length of vectors	
Output				
f24, int32_t or int16_t	frac	N	fractional part of result, Q(31- $\exp$ ) or Q(15- $\exp$ ) (fixed point functions)	
int16_t	exp	N	exponent of result (fixed point functions)	
float32_t	У	N	result (floating point function)	

## Returned value

#### None

#### Restrictions

x, frac, exp should not overlap

#### Scalar versions

## Prototype

```
uint32_t scl_recip32x32 (int32_t x)
uint32_t scl_recip24x24 (f24 x)
uint32_t scl_recip16x16 (int16_t x)
float32_t scl_recipf (float32_t x)
```

#### **Arguments**

Туре	Name	Description		
Input				
f24, int32_t or int16 t	Х	input data, Q31, Q15, floating point		

## Returned value

packed value for fixed-point functions:  $scl_recip24x24()$ ,  $scl_recip32x32()$ : bits 23...0 fractional part

bits 31...24 exponent scl\_recip16x16(): bits 15...0 fractional part bits 31...16 exponent

#### 2.3.6 Division

#### Description

Fixed point routines perform pair-wise division of vectors written in 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(31- $\exp$ ) or in Q(15- $\exp$ ) format and exponent exp so true division result in the Q0.31 may be found by shifting fractional part left by exponent value. For division to 0, the result is not defined

For fixed point finctions, mantissa accuracy is 2 LSB, so relative accuracy is:

vec_divide16x16, scl_divide16x16	1.2e-4
vec_divide24x24, scl_divide32x32, scl_divide24x24	4.8e-7
vec_divide32x32	1.8e-9

Floating point routines operate with standard floating point numbers. Functions return +/-infinity in case of overflow and provide accuracy of 2 ULP.

Two versions of routines are available: regular versions (vec\_divide32x32, vec\_divide24x24, vec\_divide16x16, vec\_dividef) work with arbitrary arguments, faster versions (vec\_divide32x32\_fast, vec\_divide24x24\_fast, vec\_divide16x16\_fast) apply some restrictions.

#### Precision

#### 4 versions available:

Туре	Description
32x32	32-bit inputs, 32-bit output.
24x24	24-bit inputs, 24-bit output.
16x16	16-bit inputs, 16-bit output.
f	floating point

## Algorithm

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

$$z_n = x_n / y_n, n = \overline{0...N-1}$$
 for floating point functions

#### **Prototype**

## **Arguments**

Туре	Name	Size	Description
Input			
f24, int32_t,	Х	N	nominator,Q31, Q15, floating point
int16_t,			
float32 t			

f24, int32_t, int16_t, float32_t	У	N	denominator,Q31, Q15, floating point
int	N		length of vectors
Output			
f24, int32_t or	frac	N	fractional parts of result, Q(31-exp) or Q(15-exp) (for
int16_t			fixed point functions)
int16_t	exp	N	exponents of result (for fixed point functions)
float32_t	Z	N	result (for floating point function)

#### Returned value

#### none

#### Restrictions

For regular versions (vec\_divide32x32, vec\_divide24x24, vec\_divide16x16, vec\_dividef): x,y,frac,exp,z should not overlap

For faster versions (vec\_divide32x3\_fast, vec\_divide24x24\_fast, vec\_divide16x16\_fast): x,y,frac,exp should not overlap x, y, frac to be aligned by 8-byte boundary N - multiple of 4.

#### Scalar versions

## **Prototype**

```
uint32_t scl_divide32x32 (int32_t x, int32_t y)
uint32_t scl_divide24x24 (f24 x, f24 y)
uint32_t scl_divide16x16 (int16_t x, int16_t y)
float32_t scl_dividef (float32_t x, float32_t y)
```

## **Arguments**

Туре	Name	Description
Input		
int32_t, f24,	Х	nominator, Q31, Q15, floating point
int16_t,		<b>31</b>
float32_t		
int32_t, f24,	У	denominator, Q31, Q15, floating point
int16_t,		<b>3</b> F F F F F F F F F F F F F F F F F F F
float32_t		

## Returned value

## packed value (for fixed point functions):

scl\_divide24x24(),scl\_divide32x32():

bits 23...0 fractional part bits 31...24 exponent scl\_divide16x16(): bits 15...0 fractional part bits 31...16 exponent

# 2.3.7 Logarithm

#### Description

Different kinds of logarithm (base 2, natural, base 10). 32 and 24-bit fixed point functions interpret input as Q16.15 and represent results in Q25 format or return 0x80000000 on negative of zero input. 16-bit fixed-point functions interpret input as Q8.7 and represent result in Q3.12 or return 0x8000 on negative of zero input

#### Accuracy:

16x16 functions	2 LSB
vec_log2_32x32,scl_log2_32x32 ,	730 (2.2e-5)
vec log2 24x24,scl log2 24x24	
vec logn 32x32,scl logn 32x32 ,	510 (1.5e-5)
vec_logn_24x24,scl_logn_24x24	
vec log10 32x32,scl log10 32x32,	230 (6.9e-6)
vec_log10_24x24,scl_log10_24x24	
floating point	2 ULP

#### NOTES:

- 1. Although 32- and 24-bit functions provide the same accuracy, 32-bit functions have better input/output resolution (dynamic range)
- 2. Floating point functions are compatible with standard ANSI C routines and set errno and exception flags accordingly.
- 3. 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

#### 4 versions available:

Туре	Description
16x16	16-bit inputs, 16-bit outputs
24x24	24-bit inputs, 24-bit outputs
32x32	32-bit inputs, 32-bit outputs
f	floating point

## Algorithm

## **Prototypes**

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

void vec log2 16x16 (	<pre>int16 t * y, const int16 t * x, int N);</pre>
void vec_logn_16x16 (	int16_t * y, const int16_t * x, int N);
<pre>void vec_log10_16x16(</pre>	<pre>int16_t * y, const int16_t * x, int N);</pre>
<pre>void vec_log2_24x24 (</pre>	f24 * y, const $f24 * x$ , int N);
<pre>void vec_logn_24x24 (</pre>	f24 * y, const f24 * x, int N);
<pre>void vec_log10_24x24(</pre>	f24 * y, const $f24 * x$ , int N);
void vec_log2_32x32 (	int32_t * y, const int32_t * x, int N);
void vec_logn_32x32 (	int32_t * y, const int32_t * x, int N);
void vec_log10_32x32(	int32_t * y, const int32_t * x, int N);
void vec_log2f (	<pre>float32_t * y, const float32_t * x, int N);</pre>
void vec lognf (	<pre>float32 t * y, const float32 t * x, int N);</pre>
void vec_log10f (	<pre>float32_t * y, const float32_t * x, int N);</pre>

#### **Arguments**

Туре	Name	Size	Description
Input	•		
int16_t, f24, int32_t, float32 t	х	N	input data, Q16.15 (32 or 24-bit functions), Q8.7 (16-bit functions) or floating point
int	N		length of vectors
Output			
int16_t, f24, int32_t, float32 t	У	N	Q6.25 (32 or 24-bit functions), Q3.12 (16-bit functions) or floating point

Returned value none

**Restrictions** x, y - should not overlap

## Scalar versions

## **Prototypes**

```
int16_t scl_log2_16x16 (int16_t x);
int16_t scl_logn_16x16 (int16_t x);
int16_t scl_log10_16x16 (int16_t x);
f24 scl_log2_24x24 (f24 x);
f24 scl_logn_24x24 (f24 x);
f24 scl_log10_24x24 (f24 x);
int32_t scl_log2_32x32 (int32_t x);
int32_t scl_logn_32x32 (int32_t x);
int32_t scl_log10_32x32 (int32_t x);
f10at32_t scl_log2f (f1oat32_t x);
float32_t scl_lognf (float32_t x);
float32_t scl_log10f (float32_t x);
float32_t scl_log10f (float32_t x);
```

## **Arguments**

Туре	Name	Description
Input		
int16_t, f24, int32_t, float32_t	Х	input data, Q16.15 (32 or 24-bit functions), Q8.7 (16-bit functions) or floating point

#### Returned value

result, Q6.25 (32 or 24-bit functions), Q3.12 (16-bit functions) or floating point

## 2.3.8 Antilogarithm

#### Description

These routines calculate antilogarithm (base2, natural and base10). 32- and 24-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. 16-bit fixed-point functions accept inputs in Q3.12 and form outputs in Q8.7 format and return 0x7FFF in case of overflow and 0 in case of underflow.

#### NOTES:

- Although 32- and 24-bit functions provide the similar accuracy, 32-bit functions have better input/output resolution (dynamic range).
- 2. Floating point functions are compatible with standard ANSI C routines and set errno and exception flags accordingly.

#### **Precision**

#### 4 versions available:

Туре	Description
16x16	16-bit inputs, 16-bit outputs. Accuracy: 2 LSB
24x24	24-bit inputs, 24-bit outputs. Accuracy: 8e-6*y+1LSB
32x32	32-bit inputs, 32-bit outputs. Accuracy: 8e-6*y+1LSB
f	floating point. Accuracy: 2 ULP

#### **Algorithm**

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

#### **Prototype**

```
void vec_antilog2_16x16 (int16_t * y, const int16_t * x, int N);
void vec_antilogn_16x16 (int16_t * y, const int16_t * x, int N);
void vec_antilog10_16x16(int16_t * y, const int16_t * x, int N);
void vec_antilog2_24x24 (f24 * y, const f24* x, int N);
void vec_antilogn_24x24 (f24 * y, const f24* x, int N);
void vec_antilog10_24x24(f24 * y, const f24* x, int N);
void vec_antilog2_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_antilog10_32x32(int32_t * y, const int32_t* x, int N);
void vec_antilog10_f(float32_t * y, const float32_t* x, int N);
void vec_antilognf (float32_t * y, const float32_t* x, int N);
void vec_antilognf (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	•		
int16_t, f24,int32_t, float32 t	Х	N	input data,Q6.25 (for 32- and 24-bit functions), Q3.12 (for 16-bit functions) or floating point
int	N		length of vectors
Output			
int16_t, f24,int32_t, float32_t	У	N	output data,Q16.15 (for 32- and 24-bit functions), Q8.7 (for 16-bit functions) or floating point

#### Returned value

#### none

#### Restrictions

x, y - should not overlap

#### Scalar versions

#### **Prototypes**

```
int16_t scl_antilog2_16x16 (int16_t x);
int16_t scl_antilog10_16x16 (int16_t x);
int16_t scl_antilog10_16x16(int16_t x);
f24 scl_antilog2_24x24 (f24 x);
f24 scl_antilog0_24x24 (f24 x);
f24 scl_antilog10_24x24 (f24 x);
int32_t scl_antilog2_32x32 (int32_t x);
int32_t scl_antilog0_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		
int16_t, f24, int32_t, float32_t	х	input data, Q6.25 (for 32 and 24-bit functions), Q3.12 (for 16-bit functions) or floating point

## Returned value

result, Q16.15 (for 32 and 24-bit functions), Q8.7 (for 16-bit functions) or floating point

# 2.3.9 Square Root

#### **Description**

These routines calculate square root.

#### NOTES:

- 1. Fixed point functions return 0x80000000 (for 24 and 32-bit functions), 0x8000 (for 16-bit functions) on negative argument
- 2. For floating point function, whenever an input value is negative, functions raise the "invalid" floating-point exception, assign EDOM to errno and set output value to NaN. Negative zero is considered as a valid input, the result is also -0

Two versions of functions available: regular version (vec\_sqrt16x16, vec\_sqrt24x24, vec\_sqrt32x32, vec\_sqrtf) with arbitrary arguments and faster version (vec\_sqrt24x24\_fast, vec\_sqrt32x32 fast) that apply some restrictions.

#### Precision

#### 4 versions available:

Туре	Description
16x16	16-bit inputs, 16-bit output. Accuracy: (2 LSB)
24x24	24-bit inputs, 24-bit output. Accuracy: (2.6e-7*y+1LSB)
32x32	32-bit inputs, 32-bit output. Accuracy: (2.6e-7*y+1LSB)
f	floating point. Accuracy 1 ULP

#### Algorithm

$$y_n = \sqrt{x_n}$$

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description
Input			
int16_t, f24 int32_t, float32 t	Х	N	input data,Q31, Q15 or floating point
int	N		length of vectors
Output			
int16_t, f24 or int32_t, float32_t	У	N	output data,Q31, Q15 or floating point

#### Returned value

#### none

## Restrictions

Regular versions (vec\_sqrt16x16, vec\_sqrt24x24, vec\_sqrt32x32):

x, y - should not overlap

Faster versions (vec\_sqrt24x24\_fast, vec\_sqrt32x32\_fast): x,y - should not overlap

x, y - aligned on 8-byte boundary

 $_{\mathrm{N}}$  - multiple of 2

#### Scalar versions

#### **Prototypes**

int16 t	scl	sqrt16x16	(int16	t	x);	
f24	scl	sqrt24x24	(f24 x)	;		
int32_t	scl	sqrt32x32	(int32_	t	x);	
float32 t	scl	sartf	(float3	2	t x	; (2

## **Arguments**

Туре	Name	Description
Input		
f24 or int32 t, float32 t	X	input data Q31 Q15 or floating point

#### Returned value

result, Q31, Q15 or floating point

# 2.3.10 Reciprocal Square Root

**Description** These routines compute reciprocals of positive square root.

zero" floating-point exception, and assign ERANGE to errno.

Precision 1 version available:

Туре	Description
f	floating point. Accuracy 2 ULP

Algorithm  $y_n = 1/\sqrt{x_n}$ 

Restrictions x,y - should not overlap

Scalar versions

**Arguments** 

Туре	Name	Description
Input		
float32_t	Х	input data

#### 2.3.11 Sine/Cosine

#### Description

Fixed-point functions calculate  $\sin{(\text{pi}^*x)}$  or  $\cos{(\text{pi}^*x)}$  for numbers written in Q31 or Q15 format. Return results in the same format. Floating point functions compute  $\sin{(x)}$  or  $\cos{(x)}$ . Two versions of functions available: regular version ( $\text{vec}\_\sin{16x16}$ ,  $\text{vec}\_\cos\sin{16x16}$ ,  $\text{vec}\_\sin{16x16}$ ,  $\text{ve$ 

#### 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. I

#### **Precision**

#### 4 versions available:

Туре	Description
16x16	16-bit inputs, 16-bit output. Accuracy: 2 LSB
24x24	24-bit inputs, 24-bit output. Accuracy: 74000(3.4e-5)
32x32	32-bit inputs, 32-bit output. Accuracy: 1700 (7.9e-7)
f	floating point. Accuracy 2 ULP

#### **Algorithm**

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

#### **Prototypes**

```
void vec_sine16x16 (int16_t * y, const int16_t * x, int N);
void vec_cosine16x16 (int16_t * y, const int16_t * x, int N);
void vec_sine24x24 (f24 * y, const f24 * x, int N);
void vec_cosine24x24 (f24 * y, const f24 * x, int N);
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_sine6 (float32_t * y, const float32_t * x, int N);
void vec_sine24x24_fast (f24 * y, const float32_t * x, int N);
void vec_cosine24x24_fast (f24 * y, const f24 * 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);
void vec_cosine32x32_fast (int32_t * y, const int32_t * x, int N);
```

#### **Arguments**

Туре	Name	Size	Description	
Input				
int16_t, f24,	X	N	input data, Q15, Q31 or floating point	
int32 t,			<b>3</b> process, at a process <b>3</b> process	
float32_t				
int	N		length of vectors	
Output				
int16 t, f24,	У	N	Result, Q15, Q31 or floating point	
int32 t,			,, <u></u> ,,	
float32 t				

#### Returned value

None

## Restrictions

Regular versions (vec sine16x16, vec cosine16x16, vec sine24x24, vec cosine24x24, vec sine32x32, vec cosine32x32, vec sinef, vec cosinef):

x, y - should not overlap

Faster versions (vec\_sine24x24\_fast, vec\_cosine24x24\_fast, vec\_sine32x32\_fast,

vec cosine32x32 fast): x,y-should not overlap

x, y - aligned on 8-byte boundary

 $_{\mathrm{N}}$  - multiple of 2

#### Scalar versions

## **Prototypes**

int16\_t scl\_sine16x16 (int16\_t x);
int16\_t scl\_cosine16x16 (int16\_t x); f24 scl\_sine24x24 (f24 x); f24 scl\_cosine24x24 (f24 x); int32\_t scl\_sine32x32 (int (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		
int16_t, f24,	Х	input data, Q15, Q31 or floating point
int32_t,		mper and, and, as a meaning point
float32 t		

#### Returned value

result, Q15, Q31 or floating point

# 2.3.12 Tangent

## **Description**

Fixed point functions calculate tan(pi\*x) for number written in Q15 or 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**

#### 4 versions available:

Туре	Description
16x16	16-bit inputs (Q15), 16-bit outputs (Q8.7). Accuracy: 1 LSB
24x24	24-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< td=""></pi*0.4776<>
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

#### Algorithm

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

#### **Prototype**

## **Arguments**

Туре	Name	Size	Description
Input			
int16_t, f24, int32_t, float32_t	Х	N	input data, Q15, Q31 or floating point
int	N		length of vectors
Output			
int16_t, int32_t, float32 t	У	N	result, Q8.7 (16-bit function), Q16.15 (24 or 32-bit functions) or floating point

#### Returned value

#### none

#### Restrictions

x, y - should not overlap

#### Scalar versions

#### **Prototype**

```
int16_t scl_tan16x16 (int16_t x);
int32_t scl_tan24x24 (f24 x);
int32_t scl_tan32x32 (int32_t x);
float32_t scl_tanf (float32_t x);
```

#### **Arguments**

Туре	Name	Description
Input		
int16_t, f24,	Х	input data, Q15, Q31 or floating point
int32 t,		5 process, at 2, and 2
float32 t		

### Returned value

result, Q8.7 (16-bit function), Q16.15 (24 or 32-bit functions) or floating point

## 2.3.13 Arc Sine/Cosine

The arc sine/cosine functions return the arc sine/cosine of x. Output is in radians. For floating-point Description

routines, input value should belong to [-1,1], otherwise the functions raise the "invalid" floating-point

exception, assign EDOM to errno and return NaN..

1 version available: Precision

Туре	Description
f	floating point. Accuracy: 2 ULP

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

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

(float32 t \* z, const float32 t \* x, int N ); (float32\_t \* z, const float32\_t \* x, int N ); void vec asinf **Prototype** 

void vec acosf

Z

**Arguments** Type Name Size **Description** Input float32\_t Х Ν input data int Ν length of vectors Output float32\_t Ν

None Returned value

x, z should not overlap Restrictions

Scalar versions

**Arguments** 

float32 t scl asinf (float32 t x); **Prototype** float32 t scl acosf (float32 t x);

Type Name Description Input float32 t input data

result

Returned value

result

## 2.3.14 Arctangent

#### **Description**

Functions calculate arctangent of number. Fixed point functions scale output by pi which corresponds to the real phases +pi/4 and represent input and output in Q15 or Q31

#### NOTE:

1. Scalar floating point function is compatible with standard ANSI C routines and sets errno and exception flags accordingly

## **Precision**

#### 4 versions available:

Туре	Description
16x16	16-bit inputs, 16-bit output. Accuracy: 2 LSB
24x24	24-bit inputs, 24-bit output. Accuracy: 74000 (3.4e-5)
32x32	32-bit inputs, 32-bit output. Accuracy: 42 (2.0e-8)
f	floating point. Accuracy: 2 ULP

## Algorithm

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

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description		
Input	Input				
int16_t, f24, int32_t, float32_t	Х	N	input data, Q15, Q31 or floating point		
int	N		length of vectors		
Output			-		
int16_t, f24, int32_t, float32_t	Z	N	result, Q15, Q31 or floating point		

## Returned value

#### None

#### Restrictions

x, z should not overlap

#### Scalar versions

## **Prototype**

```
int16_t scl_atan16x16 (int16_t x);
f24 scl_atan24x24 (f24 x);
int32 t scl atan32x32 (int32 t x);
float32_t scl_atanf (float32_t x);
```

## **Arguments**

Туре	Name	Description
Input		
int16_t, f24,	Х	input data, Q15, Q31 or floating point
int32_t,		γ
float32 t		

#### Returned value

result, Q15, Q31 or floating point

# 2.3.15 Full Quadrant Arctangent

#### **Description**

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

#### NOTE:

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

Special cases for floating point

у	x	result	extra conditions
+/-0	-0	+/-pi	
+/-0	+0	+/-0	
+/-0	Х	+/-pi	x<0
+/-0	Х	+/-0	x>0
у	+/-0	-pi/2	y<0
у	+/-0	pi/2	y>0
+/-y	-inf	+/-pi	finite y>0
+/-y	+inf	+/-0	finite y>0
+/-inf	Х	+/-pi/2	finite x
+/-inf	-inf	+/-3*pi/4	
+/-inf	+inf	+/-pi/4	

#### Precision

#### 2 versions available:

Туре	Description
16x16	16-bit input, 16-bit output. Accuracy: 2 LSB
f	floating point. Accuracy: 2 ULP

#### Algorithm

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

## **Prototype**

void vec atan2 16x16 (int16 t \* z, const int16 t \* y, const int16 t \* x,int N); void vec\_atan2f (float32\_t \* z, const float32\_t \* y, const float32\_t \* x,int N);

## **Arguments**

Туре	Name	Size	Description			
Input	Input					
int16_t, float32 t	х	N	input data, Q15 or floating point			
int16_t, float32_t	У	N	input data, Q15 or floating point			
int	N		length of vectors			
Output						
int16_t, float32 t	Z	N	result, Q15 or floating point			

## Returned value

#### None

## Restrictions

x, y, z should not overlap

#### Scalar versions

#### **Prototype**

int16 t scl atan2 16x16(int16 t y, int16 t x);
float32\_t scl\_atan2f (float32\_t y, float32\_t x);

# Arguments

Туре	Name	Description		
Input				
int16_t,	Х	input data, Q15 or floating point		
float32 t		, ,		
int16_t,	У	input data, Q15 or floating point		
float32_t		F,		

## Returned value

result, Q15 or floating point

## 2.3.16 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

24-bit version is approximately 1.5 times faster but does not use lower 8 bits of numbers. 32-bit version use all 32-bits and delivers better dynamic range.

#### 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  $-2^n$ , fast function returns  $\max(0,30-n)$  while non-fast function returns (31-n). Functions return zero if  $N\le 0$ 

#### **Precision**

#### 4 versions available:

Туре	Description
32	32-bit inputs
24	24-bit inputs
16	16-bit inputs
f	floating point inputs

## Algorithm

$$\begin{split} &z_n = \min \left( \underset{n = \overline{0...N-1}}{norm}(x_n) \right) \quad \text{non-fast version} \\ &z_n = \min \left( \underset{n = \overline{0...N-1}}{norm}(\underset{n = \overline{0...N-1}}{abs}(x_n)) \right) \quad \text{fast version} \\ &z_n = -floor \left( \underset{n = \overline{0...N-1}}{\log_2}(\underset{n = \overline{0...N-1}}{\max}(\underset{n = \overline{0...N-1}}{abs}(x_n))) \right) \quad \text{for floating point} \end{split}$$

## **Prototype**

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

ınt	vec bexp32	(const	int32	2 t *	х,	ınt	N);	
int	vec_bexp24	(const	f24	*	х,	int	N);	
int	vec_bexp16	(const	int16	5_t *	х,	int	N);	
int	vec_bexpf	(const	float	:32_t	* ×	k, in	t N)	;
int	vec_bexp32_	fast (	const	int32	_t *	х,	int	N);
int	vec_bexp24	fast (	const	f24	*	х,	int	N);
int	vec_bexp16	fast (	const	int16	_t *	х,	int	N);

## **Arguments**

Туре	Name	Size	Description	
Input				
f24, 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\_bexp24, vec\_bexp32, vec\_bexpf):
none

for fast functions (vec\_bexp16\_fast, vec\_bexp24x24\_fast, vec\_bexp32x32\_fast): x,y - aligned on 8-byte boundary N - multiple of 4

## Scalar versions

#### 

## **Arguments**

Туре	Name	Description
Input		
f24, int32_t, int16_t,	X	input data
int16_t,		Fr
float32_t		

## Returned value

result

#### 2.3.17 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\_min24x24, vec\_max24x24, vec\_max16x16, vec\_min16x16, vec\_maxf, vec\_minf) with arbitrary arguments and faster version (vec\_min32x32\_fast, vec\_max32x32\_fast, vec\_min24x24\_fast, vec\_min24x24\_fast, vec\_min32x32\_fast, vec\_min24x24\_fast, vec\_min32x32\_fast, vec\_min32x32\_fast, vec\_min24x24\_fast, vec\_min32x32\_fast, vec\_min32x32\_fast,

vec\_max24x24\_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**

Туре	Description		
32x32	32-bit data, 32-bit output		
24x24	24-bit data, 24-bit output		
16x16	16-bit data, 16-bit output		
f	floating point		

## Algorithm

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

4 versions available:

or

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

#### **Prototype**

#### **Arguments**

Туре	Name	Size	Description	
Input				
f24,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_min24x24, vec_max24x24, vec_max16x16, vec_min16x16, vec_maxf, vec_minf): none For faster routines (vec_min32x32_fast, vec_max32x32_fast, vec_min24x24_fast, vec_max24x24_fast, vec_min16x16_fast, vec_min16x16_fast): x aligned on 8-byte boundary N - multiple of 4
```

## 2.3.18 Polynomial Approximation

Description

Functions calculate polynomial approximation for all values from given vector. Fixed point functions take polynomial coefficients in Q15 or 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 than 1. To avoid this negative effect, 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

4 versions available:

Туре	Description			
16x16	16-bit inputs, 16-bit coefficients, 16-bit output.			
24x24	24-bit inputs, 24-bit coefficients, 24-bit output.			
32x32	32-bit inputs, 32-bit coefficients, 32-bit output.			
f	floating point			

Algorithm

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

**Prototype** 

## **Arguments**

Туре	Name	Size	Description	
Input				
int16_t, f24, int32_t, float32_t	Х	N	input data, Q15, Q31 or floating point	
int16_t, f24 or int32_t	С	5 or 9	coefficients (5 coefficients for vec_poly4_xxx and 9 coefficients for vec_poly8_xxx), Q15, Q31 or floating point	
int	lsh		additional left shift for result	
int	N		length of vectors	
Output				
int16_t, f24, int32_t, float32 t	Z	N	result, Q15, Q31 or floating point	

Returned value

None

Restrictions

x,c,z should not overlap

# 2.3.19 Integer to Float Conversion

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

Precision 1 version available:

Туре	Description			
f	32-bit input, floating point output			

 $\begin{aligned} & \mathbf{y}_n = \mathbf{x}_n \cdot 2^t, n = \overline{0...N-1} \\ & \text{Prototype} \end{aligned} \qquad \begin{aligned} & \mathbf{y}_n = \mathbf{x}_n \cdot 2^t, n = \overline{0...N-1} \\ & \text{void} & \underset{\text{(float32\_t * y, const int32\_t * x, int t, int $\overline{\mathbf{N}}$);} \end{aligned}$ 

**Arguments** 

Туре	Name	Size	Description	
Input				
int32_t	Х	N	input data	
int	t		scale factor	
int	N		length of vectors	
Output				
float32_t	У	N	scaled floating point values	

Returned value

None

Restrictions

t should be in range -126...126

Scalar version

**Prototype** 

float32\_t scl\_int2float (int32\_t x, int t);

**Arguments** 

Туре	Name	Description	
Input			
int32_t	Х	input data	

Returned value

result, floating point

Restrictions

t should be in range -126...126

# 2.3.20 Float to Integer Conversion

Routine scales floating point input down by 2<sup>t</sup> and converts it to integer with saturation Description

1 version available: Precision

Туре	Description
f	floating point input, 32-bit output

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

**Prototype** 

**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	scaled floating point values	

None Returned value

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

Scalar version

int32\_t scl\_float2int (float32\_t x, int t); **Prototype** 

**Arguments** 

Туре	Name	Description		
Input				
float32_t	X	input data, floating point		

Returned value

result

Restrictions

t should be in range -126...126

# 2.3.21 Rounding

Description

Routines make floating point round to integral value. Input data are rounded up/down to the nearest integral value but maintained in the same floating point format.

Precision

1 version available:

Туре	Description		
f	floating point input/output		

**Algorithm** 

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

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

**Prototype** 

**Arguments** 

Туре	Name	Size	Description			
Input	Input					
float32_t	Х	N	input data, floating point			
int	N		length of vectors			
Output						
float32_t	У	N	rounded floating point values			

Returned value

None

Restrictions

x, y should not overlap

#### Scalar version

**Prototype** 

```
float32 t scl float2floor (float32 t x);
float32_t scl_float2ceil (float32_t x);
```

**Arguments** 

Туре	Name	Description		
Input				
float32 t	X	input data, floating point		

Returned value

result

Restrictions

none

# 2.3.22 Complex Magnitude

Routines compute complex magnitude or its reciprocal Description

1 version available: Precision

Туре	Description	
f	floating point input, 32-bit output	

Algorithm

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

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

vec complex2mag **Prototype** 

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

(float32\_t \* y,
 const complex\_float \* x, int N);

**Arguments** 

Туре	Name	Size	Description			
Input						
complex_float	Х	N	input data			
int	N		length of vectors			
Output						
float32_t	У	N	magnitude or its reciprocal			

Returned value

Restrictions

None None

## Scalar version

scl complex2mag (complex float x); float32 t **Prototype** float32 t scl complex2invmag (complex float x);

**Arguments** Type

Name Description Input complex\_float Х input data

Returned value

result, floating point

Restrictions

None

# 2.4 Matrix Operations

# 2.4.1 Matrix Multiply

# Description

These functions compute the expression  $z=2^{\sl sh} * x * y$  for the matrices x and y. The columnar 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.

NOTES:

In the fixed-point routines, rows of matrices z and y may be stored in non-consecutive manner. Matrix x will have all the elements in contiguous memory locations.

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 macros  $SCRATCH\_MTX\_MPY32X32(M,N,P)$ ,  $SCRATCH\_MTX\_MPY24X24(M,N,P)$ ,  $SCRATCH\_MTX\_MPY16X16(M,N,P)$ 

Two versions of functions available: regular version (mtx\_mpy32x32, mtx\_mpy24x24, mtx\_mpy16x16, mtx\_mpyf) with arbitrary arguments and faster version (mtx\_mpy32x32\_fast, mtx\_mpy24x24\_fast, mtx mpy16x16 fast, mtx mpyf fast) that apply some restrictions.

#### Precision

#### 4 versions available:

Туре	Description
32x32	32-bit inputs, 32-bit output
24x24	24-bit inputs, 24-bit output
16x16	16-bit inputs, 16-bit output
f	floating point

## Algorithm

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

#### **Prototype**

```
void mtx mpy32x32 ( void* pScr,
                       int32 t** z,
                const int32 t* x,
                const int32 t** y,
                int M, int N, int P, int lsh );
void mtx mpy24x24 ( void* pScr,
                       f24** z,
                const f24* x,
                const f24** y, int M, int N, int P, int lsh );
void mtx_mpy16x16 ( void* pScr,
                       int16 t** z,
                const int16_t* x, const int16_t** y,
                int M, int \overline{N}, int P, int lsh );
                ( float32_t* z, const float32 t* x,
void mtx mpyf (
                const float32 t*
                int M, int N, int P);
```

# **Arguments**

Туре	Name	Size	Description	
Input	•			
int32_t, f24,	Х	M*N	input matrix,Q31 or Q15	
int16_t, float32 t				
int32_t, f24, int16_t, float32_t	У	N*P	input matrix $_{Y}$ . For fixed point routines, these are $_{N}$ vectors of size $_{P}$ , Q31 or Q15. For floating point, this is just a matrix of size $_{N\times P}$ .	
int	М		number of rows in matrix x and z	
int	N		number of columns in matrix x and number of rows in matrix y	
int	Р		number of columns in matrices y and z	
int	lsh		left shift applied to the result (applied to the fixed- point functions only)	
Output	-1	<b>I</b>	7/	
int32_t, f24, int16_t, float32_t	Z	M*P	output matrix z. For fixed point routines, these are M vectors of size P Q31 or Q15. For floating point, this is single matrix of size MxP	
Temporary		•		
void*	pScr		Scratch memory area with size in bytes defined by macros  SCRATCH_MTX_MPY32X32,  SCRATCH_MTX_MPY24X24,  SCRATCH_MTX_MPY16X16	

#### Returned value

#### none

## Restrictions

```
For regular routines (mtx_mpy32x32, mtx_mpy24x24, mtx_mpy16x16, mtx_mpyf): x,y,z should not overlap
```

# 2.4.2 Matrix by Vector Multiply

#### Description

These functions compute the expression  $z = 2^{lsh} * x * y$  for the matrices x and vector y.

Two versions of functions available: regular version (mtx\_vecmpy32x32, mtx\_vecmpy24x24, mtx\_vecmpy16x16, mtx\_vecmpyf) with arbitrary arguments and faster version (mtx\_vecmpy32x32\_fast, mtx\_vecmpy24x24\_fast, mtx\_vecmpy16x16\_fast, mtx\_vecmpyf\_fast) that apply some restrictions.

# **Precision**

#### 4 versions available:

Туре	Description
32x32	32-bit inputs, 32-bit output
24x24	24-bit inputs, 24-bit output
16x16	16-bit inputs, 16-bit output
f	floating point

# Algorithm

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

# **Prototype**

```
void mtx_vecmpy32x32 ( int32 t* z,
               const int32 t \times x,
               const int32 t * y,
               int M, int \overline{N}, int lsh);
void mtx vecmpy24x24 ( f24* z,
               const f24* x,
               const f24* y,
               int M, int N, int lsh);
void mtx vecmpy16x16 ( int16 t*
               const int16 t* x,
               const int16_t* y,
               int M, int \overline{N}, int lsh);
void mtx vecmpyf ( float32 t* z,
               const float32 t* x,
               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_vecmpy24x24_fast ( f24* z,
               const f24*
               const f24* y,
               int M, int N, int lsh);
const int16 t* y,
               int M, int \overline{N}, int lsh);
void mtx vecmpyf fast ( float32 t*
               const float32 t* x,
               const float32 t* y,
               int M, int N);
```

# **Arguments**

Туре	Name	Size	Description
Input	•	•	
int32_t, f24, int16_t, float32 t	Х	M*N	input matrix, Q31, Q15 or floating point
int32_t, f24, int16_t, float32 t	У	N	input vector, Q31, Q15 or floating point
int	M		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)

Output			
int32_t, f24, int16_t, float32_t	Z	М	output vector, Q31, Q15 or floating point

#### Returned value

None

# Restrictions

For regular routines (mtx\_vecmpy32x32, mtx\_vecmpy24x24, mtx\_vecmpy16x16, mtx\_vecmpyf) x,y,z should not overlap

For faster routines (mtx\_vecmpy32x32\_fast, mtx\_vecmpy24x24\_fast, mtx\_vecmpy16x16\_fast, mtx\_vecmpyf\_fast) 
x,y,z should not overlap 
x,y aligned on 8-byte boundary 
N and M are multiples of 4

# 2.4.3 Operations with Small Matrices

# Description

These functions implement basic operations under the sequence of small square matrices. Fixed point data are interpreted as Q15 or Q31 and results might be saturated.

#### NOTE

Determinant is computed recursively via minors of submatrices. So, in the fixed-point routines, intermediate results might be saturated although final result is in range. To avoid this saturation, right shift might be applied at the first stage of computations. It means that final result would be represented in Q(15-rsh) or Q(31-rsh) respectively. Ad-hoc formula for rsh is rsh>=N-2 for real matrices and rsh>=N-1 for complex matrices.

#### Precision

# 3 versions available:

Туре	Description
16x16	16-bit input, 16-bit output (real and complex)
32x32	32-bit input, 32-bit output (real and complex)
f	floating point (real and complex)

#### **Algorithm**

$$z_{l} = x_{l} + y_{l}$$

$$z_{l} = x_{l} - y_{l}$$

$$z_{l} = x_{l} \cdot y_{l}$$

$$z_{l} = x_{l}^{T}$$

$$d_t = \det(x_t) \cdot 2^{-rsh}$$

$$l = \overline{0...L - 1}$$
 ,  $x_l$  ,  $y_l$  ,  $z_l$  - matrices of size NXN

# Prototypes (addition, subtraction, multiply)

#### Real matrices:

Data type	Function name			
	N=2	N=3	N=4	
Matrix addition				
int16 t	mtx add2x2 16x16	mtx add3x3 16x16	mtx add4x4 16x16	
int32 t	mtx add2x2 32x32	mtx add3x3 32x32	mtx add4x4 32x32	
float32 t	mtx add2x2f	mtx add3x3f	mtx add4x4f	
complex fract16	cmtx add2x2 16x16	cmtx add3x3 16x16	cmtx add4x4 16x16	
complex fract32 cmtx add2x2 32x32		cmtx add3x3 32x32	cmtx add4x4 32x32	
complex_float	cmtx_add2x2f	cmtx_add3x3f	cmtx_add4x4f	
Matrix subtraction				
int16 t	mtx sub2x2 16x16	mtx sub3x3 16x16	mtx sub4x4 16x16	
int32 t	mtx sub2x2 32x32	mtx sub3x3 32x32	mtx sub4x4 32x32	
float32 t	mtx sub2x2f	mtx sub3x3f	mtx sub4x4f	
complex fract16	cmtx sub2x2 16x16	cmtx sub3x3 16x16	cmtx sub4x4 16x16	
complex_fract32	cmtx_sub2x2_32x32	cmtx_sub3x3_32x32	cmtx_sub4x4_32x32	
complex float cmtx sub2x2f		cmtx sub3x3f	cmtx sub4x4f	

## Prototypes (multiply)

#### Real matrices:

#### Complex matrices:

Data type	Function name				
	N=2	N=3	N=4		
Matrix multiply	1	1	•		
int16 t	mtx mul2x2 16x16	mtx mul3x3 16x16	mtx mul4x4 16x16		
int32 t	mtx mul2x2 32x32	mtx mul3x3 32x32	mtx mul4x4 32x32		
float32 t	mtx mul2x2f	mtx mul3x3f	mtx mul4x4f		
complex_fract16	cmtx_mul2x2_16x16	cmtx_mul3x3_16x16	cmtx_mul4x4_16x16		
complex fract32	cmtx_mul2x2_32x32	cmtx_mul3x3_32x32	cmtx_mul4x4_32x32		
complex float	cmtx mul2x2f	cmtx mull3x3f	cmtx mul4x4f		

# Prototypes (transpose)

#### Real matrices:

```
void fun(int16_t    *z, const int16_t    *x, int L);
void fun(int32_t    *z, const int32_t    *x, int L);
void fun(float32_t *z, const float32_t *x, int L);
```

#### Complex matrices:

Data type		Function name	
	N=2	N=3	N=4
int16 t	mtx tran2x2 16x16	mtx tran3x3 16x16	mtx tran4x4 16x16
int32_t	mtx_tran2x2_32x32	mtx_tran3x3_32x32	mtx_tran4x4_32x32
float32 t	mtx tran2x2f	mtx tran3x3f	mtx tran4x4f
complex_fract16	cmtx_tran2x2_16x16	cmtx_tran3x3_16x16	cmtx_tran4x4_16x16
complex fract32	cmtx tran2x2 32x32	cmtx tran3x3 32x32	cmtx tran4x4 32x32
complex_float	cmtx_tran2x2f	cmtx_tran3x3f	cmtx_tran4x4f

# Prototypes (determinant)

#### Real matrices:

```
void fun(int16_t     *d, const int16_t     *x, int rsh, int L);
void fun(int32 t     *d, const int32 t     *x, int rsh, int L);
void fun(float32_t *d, const float32_t *x, int L);
```

# Complex matrices:

Data type	Function name				
	N=2	N=3	N=4		
int16 t	mtx det2x2 16x16	mtx det3x3 16x16	mtx det4x4 16x16		
int32_t	mtx_det2x2_32x32	mtx_det3x3_32x32	mtx_det4x4_32x32		
float32 t	mtx det2x2f	mtx det3x3f	mtx det4x4f		
complex fract16	cmtx det2x2 16x16	cmtx det3x3 16x16	cmtx det4x4 16x16		
complex_fract32	cmtx_det2x2_32x32	cmtx_det3x3_32x32	cmtx_det4x4_32x32		
complex_float	cmtx_det2x2f	cmtx_det3x3f	cmtx_det4x4f		

# **Arguments**

Туре	Name	Size	Description
Input	1.		,
int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float	х	[L][N*N]	L input matrices
<pre>int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float</pre>	У	[L][N*N]	⊥ input matrices (for addition, subtraction, multiply functions)
int	rsh		right shift for fixed-point multiply and determinant functions
int	L		number of matrices
Output	•		
<pre>int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float</pre>	Z	[L][N*N]	L output matrices (for addition, subtraction, multiply, transpose functions)
<pre>int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float</pre>	d	L	determinants for L matrices (for determinant functions)

Returned value

none

Restrictions

rsh should be in range 0..15 x, y, z should not overlap

# 2.4.4 Matrix Inverse

Description

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

**Precision** 

1 version available:

Туре	)	Description
	f	floating point (real and complex)

Algorithm

 $y = x^{-1}$ 

**Prototype** 

```
void mtx_inv2x2f(float32_t *x);
void mtx_inv3x3f(float32_t *x);
void mtx_inv4x4f(float32_t *x);
void cmtx_inv4x4f(float32_t *x);
void cmtx_inv3x3f(complex_float *x);
void cmtx_inv3x3f(complex_float *x);
void cmtx_inv4x4f(complex_float *x);
```

Matrix dimension, N	Function		
2	mxt_inv2x2f, cmxt_inv2x2f		
3	mxt_inv3x3f, cmxt_inv3x3f		
4	mxt_inv4x4f, cmxt_inv4x4f		

#### **Arguments**

Туре	Name	Size	Description			
Input	Input					
float32_t, complex float	х	N*N	input matrix			
Output						
float32_t, complex_float	Х	N*N	output inverted matrix			

Returned value

none

Restrictions

none

# 2.4.5 Quaternion to Rotation Matrix Conversion

**Description** 

These functions convert sequence of unit quaternions to corresponding rotation matrices,

**Precision** 

3 versions available:

Туре	Description
16x16	16-bit input, 16-bit output
32x32	32-bit input, 32-bit output
f	floating point

# Algorithm

$$R_{l} = \begin{bmatrix} q_{0}^{2} + q_{1}^{2} - q_{2}^{2} - q_{3}^{2} & 2q_{1}q_{2} + 2q_{0}q_{3} & 2q_{1}q_{3} - 2q_{0}q_{2} \\ 2q_{1}q_{2} - 2q_{0}q_{3} & q_{0}^{2} - q_{1}^{2} + q_{2}^{2} - q_{3}^{2} & 2q_{2}q_{3} + 2q_{0}q_{1} \\ 2q_{1}q_{3} + 2q_{0}q_{2} & 2q_{2}q_{3} - 2q_{0}q_{1} & q_{0}^{2} - q_{1}^{2} - q_{2}^{2} + q_{3}^{2} \end{bmatrix}, l = \overline{0....L-1}$$

where

 $q_{0\dots 3}$  - elements of l-th quaternion

**Prototype** 

void q2rot\_16x16(int16\_t \*r, const int16\_t \*q, int L);
void q2rot 32x32(int32 t \*r, const int32 t \*q, int L);
void q2rotf (float32\_t \*r, const float32\_t \*q, int L);

#### **Arguments**

Туре	Name	Size	Description		
Input					
int16_t, int32_t	q	[L][4]	⊥ quaternions		
float32 t					
int	L		number of matrices		
Output					
int16_t,	r	[L][3*3]	⊥ rotation matrices		
int32_t					
float32 t					

Returned value

none

Restrictions

q,r should not overlap

# 2.5 Fast Fourier Transforms

FFT functions make floating point, 32x16, 24x24, 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 radix-4/radix-2 in-place transformations so **input data may be damaged**.

Intelligent scaling the data in the fixed-point FFT is done before each stage automatically and FFT will return number of resulted right shifts after all stages. However, at the first butterfly stage scaling may be performed either with 32-bit or 24-bit accuracy. On the other stages all computations are done in 24-bit domain and scaling is done with 24-bit data. 32-bit scaling on the first stage is useful if input data are represented in true Q31 with full 32-bit width. It takes a little bit more cycles but provides wider dynamic range. If input data are represented as f24 data scaling may be done with 24-bit accuracy. This way guarantees no overflow at the any stage and the best noise level.

Scaling may be omitted that saves computation time but the user should downscale inputs properly before FFT/IFFT to avoid overflows.

At the first scaling stage data may be shifted left, on other stages they are always shifting right. That is why FFT functions may return negative shift count on weak input signals.

Non-scaling FFT functions are faster than scaling versions however they worse in terms of signal-to-noise, dynamic range and less flexible.

There is an extra scaling option (3) which allows scaling down the data before each stage. Scale factor depends on FFT size and specific FFT stage, but not depends on the data itself. This may be used as a reasonable compromise between non-scaling version and intelligent scaling because it is faster than intelligent scaling routine and more accurate than non-scaling. However, it should be taken into account that the lowest noise with this scaling option is achieved if input data are full-scaled on input. So, if you expecting the wide dynamic range on FFT input you should either prescale them before FFT or use intelligent scaling options.

FFT/IFFT functions family with improved memory efficiency (fft\_cplx<prec>\_ie, fft\_real<prec>\_ie, fft\_real<prec>\_ie, fft\_real<prec>\_ie\_24p, fft\_real<prec>\_ie\_24p) 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
- scaling method selection is fixed at a single option
- twiddle factor tables are provided by user. A single table may be shared between FFTs/IFFTs of varying size
- 24-bit packed format is used for input/output/temporary data storage were applicable

All fixed-point FFT functions (including scaling and non-scaling) return the 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  $2^t$ . Library functions  $vec\_shift()/vec\_shift32()$  helps to convert the results to desired scale or Q-representation. In these computations, you have to take into account the fact that FFT $\rightarrow$ IFFT chain amplifies the signal by the length of FFT N for complex transforms and by N/2 for real transforms.

For example, consider the processing chain:

 $y = \texttt{FFT} \ (x) \ \ \, \to \ \, \texttt{w} = \texttt{some\_processing} \ (y) \ \ \, \to \ \, \texttt{z} = \texttt{IFFT} \ (\texttt{w}) \ \, \text{where} \ \, \texttt{N} \ \, \text{is the length of FFT, FFT returns total shift amount} \ \, \texttt{t}_{\texttt{IFFT}}.$ 

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

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

```
t_{FFT} + t_{IFFT} + log_2(N) \equiv t_{FFT} + t_{IFFT} + 31 - 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 the number of right shifts depends on selected scaled option.

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

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	•		
cplx24x24,	0 – no scaling	Input signal < 2^23/(2*N), N - FFT size	
real24x24	1 – 24-bit scaling	None	
	2 – 32-bit scaling on the first stage and 24-bit	None	
	scaling later		
	3 – fixed scaling before each stage	None	
cplx32x16	3 – fixed scaling before each stage	None	
cplx32x32	3 – fixed scaling before each stage	None	
cplx16x16	3 – fixed scaling before each stage	None	
cplx24x24_ie	3 – fixed scaling before each stage	None	
cplx32x16_ie	3 – fixed scaling before each stage	None	
real32x16	3 – fixed scaling before each stage	None	
real32x32	3 – fixed scaling before each stage	None	
real16x16	3 – fixed scaling before each stage	None	
real32x16_ie	3 – fixed scaling before each stage	None	
real24x24_ie	3 – fixed scaling before each stage	None	
real32x16_ie_24p	3 – fixed scaling before each stage	None	
rea24x24_ie_24p	1 – 24-bit scaling	None	
cplxf_ie			
DCT	1	'	
dct_24x24	3 – fixed scaling before each stage	None	
dct_32x16	3 – fixed scaling before each stage	None	
dct_32x32	3 – fixed scaling before each stage	None	

# 2.5.1 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. FFT does not make scaling of input data and it should be done externally to avoid possible overflows.

4 versions available: Precision

Туре	Description
32x32	32-bit input/outputs, 32-bit twiddles
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

# Algorithm

y = FFT(x)

# **Prototype**

```
int fft_cplx32x32(int32_t* y, int32_t * x, fft_handle_t h, int scalingOption)
int fft_cplx24x24(f24  * y, f24  * x, fft_handle_t h, int scalingOption)
int fft_cplx32x16(int32_t* y, int32_t * x, fft_handle_t h, int scalingOption)
int fft_cplx16x16(int16_t* y, int16_t * x, fft_handle_t h, int scalingOption)
```

# FFT handles:

N	32x32	24x24	32x16	32x16
16	cfft32_16	cfft24_16	cfft16_16	cfft16_16
32	cfft32_32	cfft24_32	cfft16_32	cfft16_32
64	cfft32_64	cfft24_64	cfft16_64	cfft16_64
128	cfft32_128	cfft24_128	cfft16_128	cfft16_128
256	cfft32_256	cfft24_256	cfft16_256	cfft16_256
512	cfft32_512	cfft24_512	cfft16_512	cfft16_512
1024	cfft32_1024	cfft24_1024	cfft16_1024	cfft16_1024
2048	cfft32_2048	cfft24_2048	cfft16_2048	cfft16_2048
4096	cfft32_4096	cfft24_4096	cfft16_4096	cfft16_4096

# **Arguments**

Туре	Name	Size	Description
Input			
f24, 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 2.5)
Output			
f24, 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

x, y should not overlap

x, y aligned on a 8-bytes boundary

# 2.5.2 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.

#### Precision

4 versions available:

Туре	Description	
32x32	32-bit input/outputs, 32-bit twiddles	
24x24	24-bit input/outputs, 24-bit twiddles	
32x16	32-bit input/outputs, 16-bit twiddles	
16x16	16-bit input/outputs, 16-bit twiddles	

# Algorithm

# y = FFT(real(x))

# **Prototype**

```
int fft_real32x32(int32_t* y, int32_t* x, fft_handle_t h, int scalingOpt)
int fft_real24x24(f24  * y, f24  * x, fft_handle_t h, int scalingOpt)
int fft_real32x16(int32_t* y, int32_t* x, fft_handle_t h, int scalingOpt)
int fft_real16x16(int16_t* y, int16_t* x, fft_handle_t h, int scalingOpt)
```

#### FFT handles:

N	32x32	24x24	32x16	32x16
32	rfft32_32	rfft24_32	rfft16_32	rfft16_32
64	rfft32_64	rfft24_64	rfft16_64	rfft16_64
128	rfft32_128	rfft24_128	rfft16_128	rfft16_128
256	rfft32_256	rfft24_256	rfft16_256	rfft16_256
512	rfft32_512	rfft24_512	rfft16_512	rfft16_512
1024	rfft32_1024	rfft24_1024	rfft16_1024	rfft16_1024
2048	rfft32_2048	rfft24_2048	rfft16_2048	rfft16_2048
4096	rfft32_4096	rfft24_4096	rfft16_4096	rfft16_4096
8192	rfft32_8192	rfft24_8192	rfft16_8192	rfft16_8192

# **Arguments**

Туре	Name	Size	Description
Input			
f24, 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 2.5)
Output			
int32_t, f24 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

x, y - aligned on a 8-bytes boundary

# 2.5.3 Inverse FFT on Complex Data

These functions make inverse FFT on complex data. **Description** 

NOTES:

1. Bit-reversing reordering is done here.

2. FFT runs in-place algorithm so INPUT DATA WILL APPEAR DAMAGED after call

4 versions available:

Туре	Description
32x32	32-bit input/outputs, 32-bit twiddles
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

Algorithm

Precision

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

**Prototype** 

#### FFT handles:

N	32x32	24x24	32x16	32x16
16	cifft32_16	cifft24_16	cifft16_16	cifft16_16
32	cifft32_32	cifft24_32	cifft16_32	cifft16_32
64	cifft32_64	cifft24_64	cifft16_64	cifft16_64
128	cifft32_128	cifft24_128	cifft16_128	cifft16_128
256	cifft32_256	cifft24_256	cifft16_256	cifft16_256
512	cifft32_512	cifft24_512	cifft16_512	cifft16_512
1024	cifft32_1024	cifft24_1024	cifft16_1024	cifft16_1024
2048	cifft32_2048	cifft24_2048	cifft16_2048	cifft16_2048
4096	cifft32_4096	cifft24_4096	cifft16_4096	cifft16_4096

# **Arguments**

Туре	Name	Size	Description
Input			
f24, 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 2.5)
Output	•		
f24, 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

x, y - aligned on 8-bytes boundary

# 2.5.4 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.

#### Precision

4 versions available:

Туре	Description
32x32	32-bit input/outputs, 32-bit twiddles
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles

# **Algorithm**

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

# **Prototype**

	•		1	1
N	32x32	24x24	32x16	32x16
32	rifft32_32	rifft24_32	rifft16_32	rifft16_32
64	rifft32_64	rifft24_64	rifft16_64	rifft16_64
128	rifft32_128	rifft24_128	rifft16_128	rifft16_128
256	rifft32_256	rifft24_256	rifft16_256	rifft16_256
512	rifft32_512	rifft24_512	rifft16_512	rifft16_512
1024	rifft32_1024	rifft24_1024	rifft16_1024	rifft16_1024
2048	rifft32_2048	rifft24_2048	rifft16_2048	rifft16_2048
4096	rifft32_4096	rifft24_4096	rifft16_4096	rifft16_4096
8192	rifft32_8192	rifft24_8192	rifft16_8192	rifft16_8192

# **Arguments**

Туре	Name	Size	Description
Input			
f24, 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 2.5)
Output			
f24, int32_t or int16_t	У	N	real output signal

# Returned value

total number of right shifts occurred during scaling procedure

Restrictions

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

x, y - aligned on 8-bytes boundary

# 2.5.5 Discrete Cosine Transform

**Description** These functions apply DCT (Type II) to input

NOTES:

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

Precision

5 versions available:

Туре	Description
32x32	32-bit input/outputs, 32-bit twiddles
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
16x16	16-bit input/outputs, 16-bit twiddles
f	floating point

Algorithm y = DCT(x)

# **Arguments**

Туре	Name	Size	Description
Input			
int32_t, f24, int16_t, float32 t	х	N	input signal
int	N		DCT size (32 for fixed point functions, 32 or 64 for floating point function)
int	scalingOpt		scaling option (see table in para 2.5), not applicable to the floating point function
Output			
int32_t, f24, int16_t, float32 t	У	N	output of transform

Returned value

total number of right shifts occurred during scaling procedure (always 5 for fixed point functions and 0 for floating point function)

Restrictions

x, y should not overlap

x,y - aligned on 8-bytes boundary

N - 32 for fixed point functions, 32 or 64 for floating point function)

# 2.5.6 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.

#### Precision

#### 3 versions available:

Туре	Description
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
f	floating point

# Algorithm

# y = FFT(x)

# Prototype

```
int fft cplx24x24 ie(
               complex_fract32* y, complex_fract32* x,
               const complex fract32* twd,
               int twdstep, int N, int scalingOpt);
int fft_cplx32x16 ie(
               complex fract32* y, complex_fract32* 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 );
```

## **Arguments**

Туре	Name	Size	Description					
Input	Input							
<pre>complex_fract32, complex_float</pre>	х	N	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 2.5), not applicable to the floating point function					
Output								
<pre>complex_fract32, complex_float</pre>	У	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. Floating function always return 0

#### Restrictions

x, y should not overlap

x, y - aligned on a 8-bytes boundary

# 2.5.7 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 N may be supplied with constant data (twiddle factors) of a larger-sized FFT = N\*twdstep 5 versions available:

## **Precision**

o reference available.	
Туре	Description
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
24x24_ie_24p	24-bit packed input/outputs, 24-bit data, 24-bit twiddles
32x16_ie_24p	24-bit packed input/outputs, 32-bit data, 16-bit twiddles
f	floating point

# **Algorithm**

# y = FFT(real(x))

# **Prototype**

# **Arguments**

Туре	Name	Size	Allocated Size	Description
Input			•	
f24, int32_t, float32 t	х	N	N	input signal
uint8_t		3*N	4*N+8	
<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 2.5), not applicable to the floating point function
	Output			
complex_fract32, complex float		N/2+1	N/2+1	output spectrum (positive side). Real and imaginary data are
uint8_t	У	3* (N+2)	4*N+8	interleaved and real data goes

# 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

# 2.5.8 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. 3 versions available:

Precision

Туре	Description
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
f	floating point

**Algorithm** 

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

**Prototype** 

# **Arguments**

Туре	Name	Size	Description				
Input							
<pre>complex_fract32, complex_float complex_fract32, complex_float</pre>	х	N	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 2.5), not applicable to the floating point function				
Output							
<pre>complex_fract32, complex_float</pre>	У	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. Floating function always return 0

Restrictions

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

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

# 2.5.9 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 5 versions available:

# Precision

Туре	Description
24x24	24-bit input/outputs, 24-bit twiddles
32x16	32-bit input/outputs, 16-bit twiddles
24x24_ie_24p	24-bit packed input/outputs, 24-bit data, 24-bit twiddles
32x16_ie_24p	24-bit packed input/outputs, 32-bit data, 16-bit twiddles
f	floating point

# Algorithm

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

# **Prototype**

## **Arguments**

Туре	Name	Size	Allocat ed Size	Description
Input				
<pre>complex_fract32, complex float</pre>	×	N/2+1	N/2+1	input spectrum (positive side). Real and imaginary data are
uint8_t	A	3* (N+2)	4*N+8	interleaved and real data goes
<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 2.5), not applicable to the floating point function
Output				
f24, int16_t, float32_t	У	N	N	output real signal
uint8_t		3*N	4*N+8	

# 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

# 2.6 Identification Routines

# 2.6.1 Library Version Request

**Description** This function returns library version information.

Arguments Type 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

# 2.6.2 Library API Version Request

**Description** This function returns library API version information.

Arguments Type 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

# 3 Test Environment and Examples

# 3.1 Supported Use Environment, Configurations and Targets

Here, the NatureDSP Signal library and corresponding testdriver is supported to be built and test using Xtensa Xplorer IDE running under Windows, or Linux operating system.

# 3.2 Building the NatureDSP Signal Library

# 3.2.1 Importing the workspaces in Xtensa Xplorer

NatureDSP Signal Library for Fusion F1 is provided as two workspaces:

- Library workspace fusion\_lib.xws
  The workspace contains library project which has optimized library kernel routines.
- Demo workspace fusion\_demo.xws
   This contains the demo testbench project fusion\_demo.

Import the .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 fusion\_demo project, select Debug or Release target, and build.

To run the test-bench, select fusion\_demo project, and Run.

This will execute cycles performance test for each routine in the Fusion F1 library.

# 3.3 Other Environments Supported

Library and testdriver project can also be built under the following platforms:

os	Language	Environment	Tool
Linux	С	xcc	make
Linux	C++	gcc	make
Windows	С	xcc	xt-make
Windows	C++	MSVC	Visual Studio

# 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 bqriir24x24\_df1 Conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir24x24 df1 function)
% parameters:
         - SOS matrix and gain vector G
% datatype - 'double', 'single', int16' or 'int32'
% output:
          - vector with coefficients, Q30
% coef
% gain
          - biquad gains, Q15
         - final scale factor (amount of left shifts)
% scale
function [coef, gain, scale] = cvtsos bgriir24x24 df1(SOS, G)
sz=size(SOS);
M=sz(1);
coef=[];
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
    [b,a] = sos2tf(SOS(1:m,:), [G(1:m) 1]);
    tf=freqz(b,a);
    tfmax=max(abs(tf));
    G(m) = min(1, 0.5/tfmax);
    % round to nearest Q7 value
    G(m) = min(127, round(G(m)*128))/128;
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=(floor(double(gain)/128)*128)/32768;
g(M+1) = pow2(1, double(scale));
[b,a]=sos2tf(sos,g);
tf=freqz(b,a);
plot(20*log10(abs(tf))); ylim([-80 0]); title('transfer function, dB'); grid on;
```

# 4.1.2 bqriir32x16\_df1 Conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16_df1 function)
% parameters:
% SOS,G
         - SOS matrix and gain vector G
% datatype - 'double', 'single', int16' or 'int32'
% output:
% coef
           - vector with coefficients, Q30
          - biguad gains, Q15
% gain
         - final scale factor (amount of left shifts)
% scale
&_____
function [coef,gain,scale]=cvtsos bqriir32x16 df1(SOS,G)
sz=size(SOS);
M=sz(1);
coef=[];
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
    [b,a]=sos2tf(SOS(1:m,:),[G(1:m) 1]);
    tf=freqz(b,a);
    tfmax=max(abs(tf));
    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));
[b,a]=sos2tf(sos,g);
tf=freqz(b,a);
plot(20*log10(abs(tf))); ylim([-80 0]); title('transfer function, dB'); grid on;
```

# 4.1.3 bqriir24x24\_df2 Conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir24x24 df1 function)
% parameters:
        - SOS matrix and gain vector G
% datatype - 'double', 'single', int16' or 'int32'
          - vector with coefficients, Q30
% coef
          - biquad gains, Q15
% gain
        - final scale factor (amount of left shifts)
% scale
function [coef,gain,scale]=cvtsos bqriir24x24 df2(SOS,G)
sz=size(SOS);
M=sz(1);
coef=[];
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
    [b,a] = sos2tf(SOS(1:m,:), [G(1:m) 1]);
    tf=freqz(b,a);
    tfmax0=max(abs(tf));
    [b,a]=sos2tf([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1]);
    tf=freqz(b,a);
    tfmax1=max(abs(tf));
    tfmax = max(tfmax0, tfmax1);
    G(m) = min(1,0.5/tfmax);
    % round to nearest 07 value
    G(m) = min(127, round(G(m)*128))/128;
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=(floor(double(gain)/128)*128)/32768;
g(M+1) = pow2(1, double(scale));
[b,a]=sos2tf(sos,g);
tf=freqz(b,a);
plot(20*log10(abs(tf))); ylim([-80 0]); title('transfer function, dB'); grid on;
```

# 4.1.4 bgriir32x16 df2 Conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16 df2 function)
% parameters:
% SOS,G - SOS matrix and gain vector G
% datatype - 'double', 'single', int16' or 'int32'
% output:
          - vector with coefficients, Q14
% coef
          - biquad gains, Q15
% gain
         - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir32x16 df2(SOS,G)
sz=size(SOS);
M=sz(1);
coef=[];
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
    [b,a] = sos2tf(SOS(1:m,:),[G(1:m) 1]);
    tf=freqz(b,a);
    tfmax0=max(abs(tf));
    [b,a]=sos2tf([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1]);
    tf=freqz(b,a);
    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)/dq;
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));
[b,a]=sos2tf(sos,g);
tf=freqz(b,a);
plot(20*log10(abs(tf))); ylim([-80 0]); title('transfer function, dB'); grid on;
```

# 4.1.5 bgriir32x32 df2 Conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16 df2 function)
% parameters:
         - SOS matrix and gain vector G
% SOS,G
% datatype - 'double', 'single', int16' or 'int32'
% output:
% coef
          - vector with coefficients, Q30
          - biquad gains, Q15
% gain
% scale
          - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir32x32 df2(SOS,G)
sz=size(SOS);
M=sz(1);
coef=[];
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
    [b,a]=sos2tf(SOS(1:m,:),[G(1:m) 1]);
    tf=freqz(b,a);
    tfmax0=max(abs(tf));
    [b,a]=sos2tf([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1]);
    tf=freqz(b,a);
    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));
[b,a]=sos2tf(sos,g);
tf=freqz(b,a);
plot(20*loq10(abs(tf))); ylim([-80 0]); title('transfer function, dB'); grid on;
```

# 4.2 Matlab Code for Generation the Twiddle Tables

FFT with optimized memory usage requires external twiddle tables. The Matlab code below shows how to generate twiddles for different functions.

# 4.2.1 Twiddles for fft\_cplx24x24\_ie, ifft\_cplx24x24\_ie, fft\_real24x24\_ie, ifft\_real24x24\_ie

```
function [twd]=twd24x24_ie(N)
twd = exp(-2j*pi*[1;2;3]*(0:N/4-1)/N);
twd=twd.';
twd = reshape([real(twd(:).');imag(twd(:).')],1,2*numel(twd));
twd = int32(round(pow2(twd,31)));
```

# 4.2.2 Twiddles for fft\_cplx32x16\_ie, ifft\_cplx32x16\_ie, fft\_real32x16\_ie, ifft\_real32x16\_ie

```
function [twd]=twd32x16_ie(N)
twd = exp(-2j*pi*[1;2;3]*(0:N/4-1)/N);
twd=twd.';
twd = reshape([imag(twd(:).');real(twd(:).')],1,2*numel(twd));
twd = int16(round(pow2(twd,15)));
```

# 4.2.3 Twiddles for fft\_cplxf\_ie, ifft\_cplxf\_ie, fft\_realf\_ie, ifft\_realf\_ie

```
function [twd]=twdf_ie(N)
twd = exp(-2j*pi*[1;2;3]*(0:N/4-1)/N);
twd = reshape([real(twd(:).');imag(twd(:).')],1,2*numel(twd));
```

# **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.