

NatureDSP Signal Library for HiFi 1 DSP

Digital Signal Processing

Library Reference

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Document History

Revision	Date	Major changes
1.0	16 Nov 2021	First release version.
1.1	10 Oct 2022	Second release version

Preface

About This Manual

Welcome to the **NatureDSP Signal Processing Library**, or **NatureDSP Signal** or library for short. The library is a collection of a number of highly optimized DSP functions for the HiFi 1 DSP.

This source code library includes C-callable functions (ANSI-C language compatible) for general signal processing (filtering, correlation, convolution), math and vector functions specially optimized for high-density signal processing. Library supports both fixed-point and single precision floating data types. These library APIs are developed based on Nature DSP APIs developed by IntegrIT.

Supported Targets

Library supports Cadence Tensilica HiFi 1 DSP with VFPU option. HiFi 1 DSP supports only little-endian configuration. RI-2022.9 toolchain shall be used for building HiFi1e extension cores with enhanced ISA. The library still supports backward compatibility for cores with earlier toolchain.

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	1.1 () 1

IDE Integrated development environment **IFFT** Inverse Fast Fourier transform IIR Infinite impulse response IR Impulse response **LMS** Least mean squares LSB Least significant bit **ULP** Units of least precision **VFPU** Vector Floating Point Unit

XX Xtensa Xplorer

1 General Library Organization

1.1 Headers

NatureDSP_Signal library is supplied with following header files

```
./include/NatureDSP_types.h
./include/NatureDSP_Signal.h
```

Declarations of basic data types and compiler auto detection 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
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
f24	24-bit fractional type	4

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 at 3 higher bytes of 32-bit words and 8 LSBs are assumed to be 0. However, a few routines use packed 24-bit data where 24-bit fractional numbers are allocated only 3 consecutive bytes.

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

1.4 Fractional Formats

Natively, HiFi 1 CPU uses special fractional type £24 which is stored in a memory as 32-bit word keeping significant bits in bits 8 through 31. So, from that perspective it may be treated as Q31 number. But users should take into account that 8 LSB are assumed to be 0s. **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 , 999969	3e-5	-32768	32767
int16 t	Q6.9	-64 63 , 998	2e-3	-32768	32767
int16 t	Q3.12	-8 7 , 9998	2e-4	-32768	32767
int16 t	Q8.7	-256 255 , 992	8e-3	-32768	32767
int32 t	Q1.30	-2 1 , 9999999991	9e-10	-2147483648	2147483647
int32 t	Q0.31	-1 0 , 9999999995	5e-10	-2147483648	2147483647
int32 t	Q6.25	-64 63 , 999999970	3e-8	-2147483648	2147483647
int32 t	Q16.15	-65536 65535 , 99997	3e-5	-2147483648	2147483647
f24	Q1.30	-2 1 , 9999997625	2e-7	-2147483648	2147483392
f24	Q0.31	-1 0 , 9999998784	1e-7	-2147483648	2147483392
f24	Q6.25	-64 63 , 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.

For example, if a number is representing in Q31 format and have decimal representation N, its floating-point equivalent will be N^2-31.

1.5 Compiler Requirements

When building the library source files or library-dependent modules it is assumed that the target is a Tensilica processor implementing the Xtensa HiFi 1 Instruction Set Architecture with or without VFPU option. This is accomplished through checking the <code>COMPILER_XTENSA</code> preprocessor symbol performed in <code>NatureDSP_types.h</code>. This symbol is automatically defined by the compiler. The library is delivered as an Xtensa Xplorer workspace, along with a second workspace containing demonstrations and tests of the various library routines. The library is intended to be used with Xtensa Xplorer. When the workspaces are imported, all relevant compiler options are automatically set, although the user can modify them if needed.

1.6 Call Conventions

Library uses ANSI-C call conventions.

1.7 Overflow Control and Intermediate Data Format

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

In most of the fixed-point routines with operations involving summing of multiple elements (i.e. FIR, matrix multiplies, etc.), library stores intermediate values in 64-bit accumulators using Q16.47 fixed-point representation thus protecting from the overflows in the intermediate stages. Floating point routines use single precision floating point format for storing intermediate data.

The user is expected to conform to the range requirements (if specified) and take care of restricting 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 VFPU enabled bits. VFPU 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. The end implementation is expected to use that signal to control an external interrupt – thus enabling an imprecise interrupt.
- The caller could zero the status flags before a function and check them when the function returns to see if any exceptional conditions occurred.

1.9 Special Numbers

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

- negative zero is treated as usual negative number
- the result of operations under NaN is NaN
- operations with infinity return 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 Endianness

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, users 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 slightly bigger size and align pointers¹

¹ It shall be the application's responsibility to pass the pointers with required alignment.

Test examples included, uses the last two methods.

1.12 Object Model

Effective use of all HiFi 1 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 HiFi 1 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 allocated for to the object, for those 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 the object.

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

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

1.13 Brief Function List

Vectorized version Scalar version		Purpose	Reference
FIR filters and relate	d functions	•	<u> </u>
bkfir		Block real FIR filter	2.1.1, 2.1.2
cxfir		Complex block FIR filter	2.1.3
firdec		Decimating block real FIR filter	2.1.4
firinterp		Interpolating block real FIR filter	2.1.5
fir_convol, cxfir_convol		Circular convolution	2.1.6
fir_lconvol		Linear convolution	2.1.7
fir_xcorr		Circular correlation	2.1.8
fir_lxcorr		Linear correlation	2.1.9
fir_acorr		Circular autocorrelation	2.1.9
fir_lacorr		Linear autocorrelation	2.1.11
fir_blms		Blockwise Adaptive LMS algorithm	2.1.12
IIR filters			
bqriir, bqciir		Biquad Real block IIR	2.2.1
latr		Lattice block Real IIR	2.2.2
Vector mathematics		•	
vec_dot		Vector dot product	2.3.1
vec_add		Vector sum	2.3.2
vec_power		Power of a vector	2.3.3
vec_shift		Vector scaling with saturation	2.3.4
vec_scale	,		
vec_recip	scl_recip	Reciprocal	2.3.5

Vectorized version	Scalar version	Purpose	Reference	
vec_divide	scl_divide	Division	2.3.6	
vec_logn	scl_logn	Different kinds of logarithm	2.3.7	
vec_log2	scl_log2	g		
vec_logn	scl_logn			
vec_antilog2 vec_antilog10	scl_antilog2 scl_antilog10	Different kinds of antilogarithm	2.3.8	
vec_antilogio	scl_antilog10			
vec sqrt	scl sqrt	Square root	2.3.9	
vec rsqrt	scl rsqrt	Reciprocal square root	2.3.10	
vec sine	scl sine	Sine	2.3.11	
vec cosine	scl cosine	Cosine	2.3.11	
=	scl tan		0.0.40	
vec_tan	_	Tangent	2.3.12	
vec_asin	scl_asin	Arcsine	2.3.13	
vec_acos	scl_acos	Arccosine	2.3.13	
vec_atan, vec atan2	scl_atan, scl_atan2	Arctangent	2.3.14, 2.3.15	
vec_bexp	scl_bexp	Common exponent	2.3.16	
vec_min,		Find a maximum/minimum in a vector	2.3.17	
vec_max				
vec_poly		Polynomial approximation	2.3.18	
vec_int2float	scl_int2float	Integer to float conversion	2.3.19	
vec_float2int	scl_float2int	Float to integer conversion	2.3.20	
vec_float2floor	scl_float2floo r	Rounding	2.3.21	
vec float2ceil	scl float2ceil	1		
vec_complex2mag	scl_complex2ma	Complex magnitude	2.3.22	
vec complex2inv	scl complex2in	Reciprocal of complex magnitude	2.3.22	
mag	vmag	Treespread of complex magnitude	2.0.22	
Matrix operations		That is the second of the seco	1044	
mtx_mpy		Matrix multiply	2.4.1	
mtx_vecmpy		Matrix by vector multiple	2.4.2	
mtx_add, cmtx_add		Matrix addition	2.4.3	
mtx_sub, cmtx_sub		Matrix subtraction	2.4.3	
mtx_mul, cmtx mul		Matrix multiply	2.4.3	
mtx_tran, cmtx tran		Matrix transpose	2.4.3	
mtx_det, cmtx det		Matrix determinant	2.4.3	
mtx_inv,		Matrix inverse	2.4.4	
cmtx_inv q2rot		Quaternion to Rotation Matrix Conversion	2.4.5	
FFT				
fft_cplx		FFT on complex data	2.5.1	
fft real		FFT on real data	2.5.2	
ifft cplx		Inverse FFT on complex data	2.5.3	
ifft real		Inverse FFT forming real data	2.5.4	
dct		Discrete cosine transform	2.5.5	
fft cplx ie				
fft real ie		FFT on complex data with optimized memory usage 2.5.6		
		FFT on real data with optimized memory usage	2.5.7	
ifft_cplx_ie		Inverse FFT on complex data with optimized memory usage	2.5.8	
ifft_real_ie	- Interior is the state of the			
Identification				
NatureDSP Signal	get_library_vers	ion Library Version Request	2.6.1	
NatureDSP_Signal_			2.6.2	

2 Reference

2.1 FIR Filters and Related Functions

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

2.1.1 Block Real FIR Filter

Description

Computes a real FIR filter (direct-form) using IR stored in vector ${\tt h}$. The real data input is stored in vector ${\tt x}$. The filter output result is stored in vector ${\tt y}$. The filter calculates ${\tt N}$ output samples using ${\tt M}$ coefficients and requires last ${\tt M-1}$ 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	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 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	72+M*8
bkfir32x16 alloc	72+M*6
bkfir32x32 alloc	72+M*8
	, 2 . 11 0
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

```
void bkfir16x16 process (
            bkfir16x16_handle_t handle,
            int16 t * y, const int16 t * x, int N)
void bkfir24x24 process (
            bkfir24x24 handle t handle,
            f24 * y, const f\overline{2}4 * x, int N )
void bkfir24x24p process(
            bkfir24x24p handle t handle,
            f24* y, const f24* x, int N)
void bkfir32x16 process (
            bkfir32x16 handle t handle,
bkfir32x32 handle t handle,
            int32_t * y, const int32_t * y, int N)
void bkfirf process (
            bkfirf handle t handle,
            float32_t * y, const float32_t * x, int N);
```

Туре	Name	Size	Description		
Input	•	•	·		
int16_t, f24,	Х	N	input samples		
int32 t,					
float32 t					
int	N		length of sample block		
Output					
int16_t, f24,	У	N	output samples		
int32_t,			- Samples		
float32 t					

Returns: none

Restrictions

```
x, y – should not overlap x, handle - aligned on 8-byte boundary N, M - multiples of 4
```

2.1.2 Block Real FIR Filter with Arbitrary Parameters

Description

These functions implement FIR filter described in previous chapter with no limitation on size of data block, alignment and length of impulse response for the cost of performance.

Precision

5 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 but internally 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, but internally 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	Input					
int	M		length of filter			

Returns: size of memory in bytes to be allocated

NOTE:

Approximate amount of requested memory is listed below

Function	Approximate memory requirements, bytes	
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
<pre>complex_fract32, complex_fract16, complex_float</pre>	h	М	complex filter coefficients; h[0] is to be multiplied with the newest sample, Q31, Q15 or floating point

int	M	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, handle - aligned on 8-byte 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 but internally 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	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

 ${\tt x}$, handle - aligned on 8-byte 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 but internally 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 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*□

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			·	
<pre>int16_t, complex_fract16, f24, int32_t, float32 t</pre>	Х	N	input samples,Q15, Q31 or floating point	
int	N		length of input sample block	
Output				
<pre>int16_t, complex_fract16, f24, int32_t, float32 t</pre>	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

Conditions for optimum performance

D - 2, 3 or 4

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 \overline{M});
void cxfir convol32x16 (complex fract32 * r,
                         const complex_fract32 * 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 \overline{M});
void fir convolf (float32 t * r,
                   const float32_t * x,
                   const float32 t * y,
                   int N, int M);
```

```
const int16 t * y,
                      int N, int \overline{M});
const f24 * x,
const f24 * y,
                       int N, int M);
                       (void * s,
int32_t * r,
void fir convola32x16 (void
                       const int32_t * x,
                       const int16 t * y,
                       int N, int \overline{M});
void cxfir convola32x16 (void * s,
                        complex_fract32 * r,
const complex_fract32 * x,
                        const complex_fract16 * y,
                        int N, int M);
const int32_t * x,
const int32_t * y,
                      int N, int \overline{M});
                      (void * s, float32_t * r,
void fir_convolaf
                      (void
                       const float32_t * x,
                      const float32_t * y,
int N, int M);
```

Arguments

Туре	Name	Size	Description
Input	II.	I.	
int16_t, f24, int32_t, complex_fract32, float32_t	х	N	input data (Q15, Q31 or floating point)
<pre>int32_t, f24, int16_t, complex_fract16, or float32_t</pre>	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	М		length of y
Output	•		•
int16_t, f24, int32_t, complex_fract32, float32_t	r	N	output data, Q15, Q31 or floating point
Temporary			
void	S		Scratch memory, FIR CONVOLA16X16_SCRATCH_SIZE(N, M) FIR 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	•	•	
int16_t, int32_t, float32_t	X	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			-
int16_t, int32_t, float32_t	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 \mathtt{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 $\mathbf M$) and $\mathbf y$ (of length $\mathbf M$) resulting in

vector ${\tt r}$ of length ${\tt N}$. It is similar to convolution but ${\tt y}$ 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,

 $\texttt{fir_xcorr32x32, fir_xcorraf, cxfir_xcorraf)} \ \textbf{for arbitrary arguments. In addition, these slower}$

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 * r, const int16_t * x, const int16_t * y,
                  int N, int \overline{M});
int N, int M);
int N, int \overline{M});
int N, int \overline{M});
void fir xcorrf (float32 t* r,
               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
                   (void     * s,
    int16_t     * r,
    const int16_t     * x, const int16_t     * y,
                   int N, int \overline{M});
int N, int M);
void fir xcorra32x16 (void
                            * s,
                   int32_t * r,
const int32_t * x, const int16_t * y,
                   int N, int \overline{M});
                   (void * s,
int32_t * r,
void fir xcorra32x32 (void
                   const int32 t * x, const int32 t * y,
                   int N, int M);
               (void * s, float32_t* r,
void fir xcorraf (void
                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);
```

```
void fir xcorra16x16 (void
                        int16_t * r,
const int16_t * x, const int16_t * y,
                        int N, int \overline{M});
                                  * s,
void fir xcorra24x24 (void
                        (void * s,
f24 * r,
const f24 * x, const f24 * y,
                        int N, int M);
                        (void     * s,
   int32_t     * r,
   const int32_t     * x, const int16_t     * y,
void fir xcorra32x16 (void
                        int N, int M);
void fir xcorra32x32 (void
                        int32_t * r,
                        const int32 t * x, const int32_t * y,
                        int N, int \overline{M});
                       (void * s, float32_t* r,
void fir_xcorraf
                         const float32_t* x, const float32_t* y, int N, int M);
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			
<pre>int16_t, f24, int32_t, float32_t, complex_float</pre>	Х	N	input data (Q15, Q31 or floating point)
f24, int16_t, float32_t, complex float	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	М		length of y
Output	•		
int16_t, f24, int32_t, float32_t, complex float	r	N	output data, Q15, Q31 or floating point
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, cxfir_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	u .	· ·	
int16_t, int32_t, float32_t	Х	N	input data (Q15, Q31 or floating point)
int16_t, int32_t, float32_t,	У	М	input data (Q31, Q15 or floating point)
int	N		length of x
int	М		length of y
Output		•	
int16_t, int32_t, float32 t	r	M+N-1	output data, Q15, Q31 or floating point
Temporary			
void	S		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

N>=M-1

Restrictions

x,y,r,s should not overlap s should be aligned on 8-byte boundary N>0, M>0

2.1.10 Circular Autocorrelation

Description

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

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 float32 t	Х	N	input data (Q15, Q31 or floating point)
int	N		length of x
Output			
int16_t, f24, int32_t or float32_t	r	N	output data, Q15, Q31 or floating point
Temporary			
void	S		Scratch memory, FIR_ACORRA16X16_SCRATCH_SIZE(N) FIR_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

Description

Functions estimate the linear auto-correlation of vector ${\tt x}.$ Returns autocorrelation of length ${\tt N}.$

Precision

3 versions available:

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

Algorithm

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

Prototype

```
void fir_lacorra16x16 (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		•		
int16_t, int32_t or float32 t	Х	N	input data (Q15, Q31 or floating point)	
int	N		length of x	
Output				
int16_t, int32_t or float32 t	r	N	output data, Q15, Q31 or floating point	
Temporary				
void	Ø		Scratch memory, FIR_LACORRA16X16_SCRATCH_SIZE(N) FIR_LACORRA32X32_SCRATCH_SIZE(N) FIR_LACORRAF_SCRATCH_SIZE(N) bytes	

Returned value

none

Restrictions

x,r,s should not overlap

N>0

s - aligned on an 8-bytes boundary

2.1.12 Blockwise Adaptive LMS Algorithm for Real Data

Description

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

- 1. The algorithm must be provided with the normalization factor, which is the power of the 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 it's the case, then the adaptation coefficient must be also shifted to the right by the same number of bit positions.
- 2. This algorithm consumes less CPU cycles per block than single sample algorithm at similar convergence rate.
- 3. Right selection of N depends on the change rate of impulse response: on static or slow varying channels convergence rate depends on selected mu and M, but not on 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
                                                mıı.
                      int N, int M);
void fir blms24x24 (f24 * e, f24 * h,
                      const f24 * r,
                      f24 norm, f24 mu, int N, int M);
void fir blms16x32 (int32 t * e, int32 t * h,
                      const int16_t * r,
const int16_t * x,
                      int32_t norm,int16_t mu,
                                 Ν,
void fir blms32x32 (int32 t * e, int32 t * h,
                      const int32_t * r, const int32_t * x,
                      int32_t norm, int32_t mu,
                      int N, int M);
                      (float32_t * e, float32_t * h, const float32_t * r,
void fir_blmsf
                       const \overline{\text{loat32}} t * x,
                       float32_t norm, float32_t mu,
                       int
                                 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,	×	N+M-1			
int32 t or	Α	1111111	input (far end) data vector. First in time value is in x[0], Q31, Q15 or floating point		
float32_t			Q31, Q15 or lloating point		
int16_t, f24, int32 t,	norm		normalization factor: power of signal multiplied by N,		
float32 t			Q15, Q31 or floating point		
f24, int16_t	mu		adaptation coefficient in Q31, Q15 or floating point		
int32_t, float32 t			(LMS step)		
int	N		length of data block		
int	M		length of h		
Output					
f24, int16 t,	е	N	estimated error, Q31,Q15 or floating point		
int32_t,			Joseph Strategy Company Comments of the Commen		
float32_t	h		045 004		
f24, int16_t, int32 t,	n	М	updated impulse response, Q15, Q31		
float32_t					

Returned value

none

Restrictions

h,x,r,y,e - should not overlap x,e,h,r - aligned on a 8-bytes boundary $_{\rm N,M}$ - 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 non-negative. 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

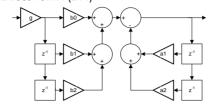
10 versions available:

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

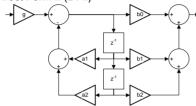
Algorithm

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:

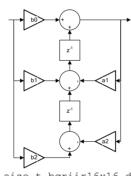
Direct Form I (DFI)



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 bq3iir16x16_df2_alloc(int M)
size_t bqriir24x24_df1_alloc(int M)
size_t bqriir32x16_df1_alloc(int M)
size_t bqriir32x16_df2_alloc(int M)
size_t bq3iir32x16_df2_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 bqriir32x32_df2_alloc(int M)
size_t bq3iir32x32_df1_alloc(int M)
size_t bq3iir32x32_df1_alloc(int M)
size_t bqxiirf_df1_alloc(int M)
size_t bqriirf_df1_alloc(int M)
size_t bqriirf_df2_alloc(int M)
size_t bqciirf_df1_alloc(int M)
size_t bq3iirf_df1_alloc(int M)
size_t bq3iirf_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 df1 handle t bqriir16x16 df1 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);
bqriir32x32_df1_handle_t bqriir32x32_df1_init(void * objmem, int M,
            const int32 t * coef sos, const int16 t * coef g, int16 t gain)
bqriir32x32_df2_handle_t bqriir32x32_df2_init(void * objmem, int M,
            const int32 t * coef sos, const int16 t * coef 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_df2_handle_t bqriirf_df2_init(void * objmem, int M,
            const float32 t * coef sos, int16 t gain);
bqriirf_df2t_handle_t bqriirf_df2t_init(void * objmem, int M,
            const float32_t * coef_sos, int16_t gain);
bqciirf 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 functions, 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 bqriir24x24_df1(bqriir24x24_df1_handle_t _bqriir,
            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 bgriir32x16 df2 (bgriir32x16 df2 handle t bgriir,
            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 bqriirf_df2 (bqriirf_df2_handle_t _bqriir,
                float32 t
                            * z, const float32 t * x, int N);
void bqriirf_df2t (bqriirf_df2t_handle_t _bqriir,
float32 t * r, con void bqciirf df1 (bqciirf df1 handle t,
                            * r, const float32 t * x, int N);
                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);
```

Function	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		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 the 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 the number of triplets, so array size should be 3*N.
Temporary			
void*	s		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 k. The real data input is stored in vector k. The filter output result is stored in vector k. Input scaling is done before the first cascade for normalization and overflow protection.

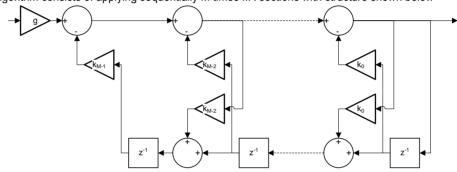
Precision

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

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

```
void latr16x16_process
    (latr16x16_handle_t handle, int16_t * r, const int16_t * x, int N);
void latr24x24_process
    (latr24x24_handle_t handle, f24 * r, const f24 * x, int N);
void latr32x16_process
    (latr32x16_handle_t handle, int32_t * r, const int32_t * x, int N);
void latr32x32_process
    (latr32x32_handle_t handle, int32_t * r, const int32_t * x, int N);
void latrf_process
    (latrf_handle_t handle, float32_t * r, const float32_t * x, int N);
```

Туре	Name	Size	Description	
Input				
int16_t, f24, int32_t or float32 t	Х	N	input samples, Q15, Q31 or floating point	
int	N		length of input sample block	
Output				
int16_t, f24, int32_t or float32 t	r	N	output data, Q15, Q31 or floating point	

Returns: none

Returned value

none

Restrictions

x, r, k should not overlap

Conditions for optimum performance

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

2.3 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 f\overline{2}4 * 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_dot32x32
  (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	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

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

Restrictions

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 Addition

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

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

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);
```

Arguments

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

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 routines 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 N - multiple of 4

2.3.4 Vector Scaling with Saturation

Description

These routines implement shift with saturation of data values in the vector, by the 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.

Function 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_scale16x16 fast) apply some restrictions.

For floating point:

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

Precision

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

$r_n = x_n \cdot 2^t$

n ildot n =	
<pre>void vec_shift24x24</pre>	
	const f24 * x,
	int t,
	int N);
void vec shift32x32	
_	const int32 t * x,
	int t,
	int N);
<pre>void vec_shift16x16</pre>	(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,
= -	const f24 * x,
	int t,
	int N);
void vec shift32x32	fast (int32 t * y,
void vee_bhile52232_	const int32 t * x,
	int t,
ida abif+1616	int N);
void sec_sullflexie	_fast (
	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

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 *
                   const int32_t * x,
                   f24 s,
                   int N);
void vec scale24x24 ( f24 * y,
                   const f24 * x,
                   f24 s,
                   int N);
void vec_scale16x16 ( int16_t * y,
                   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 t should be in range -31...31 x,y - aligned on 8-byte boundary 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 functions, 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	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 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 functions, 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

$$frac_n \cdot 2^{exp_n} = x_n / y_n, n = \overline{0...N-1}$$
 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, int16_t, float32 t	Х	N	nominator,Q31, Q15, floating point

f24, int32_t, int16_t, float32 t	У	N	denominator,Q31, Q15, floating point
int	N		length of vectors
Output			
f24, int32_t or int16_t	frac	N	fractional parts of result, Q(31-exp) or Q(15-exp) (for 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,		3 pr
float32_t		
int32_t, f24,	У	denominator, Q31, Q15, floating point
int16_t,		
float32_t		

Returned value

packed value (for fixed point functions):

scl_divide24x24(),scl_divide32x32():

bits $\overline{2}3...0$ 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 fixedpoint 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:

- 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

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

Prototypes

```
float32_t * z, const float32_t * x, int N);
float32_t * z, const float32_t * x, int N);
 void vec log10f (
                     float32 t * z, const float32 t * x, int N);
```

Arguments

Туре	Name	Size	Description
Input	•		
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,	У	N	Q6.25 (32 or 24-bit functions), Q3.12 (16-bit functions) or floating point
float32 t			

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);
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);
float32_t scl_log2f (float32_t x);
float32_t scl_lognf (float32_t x);
float32_t scl_log10f (float32_t x);

Arguments

Туре	Name	Description
Input	•	•
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:

- 1. 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_32x32 (int32_t * y, const float32_t* x, int N);
void vec_antilog10 (float32_t * y, const float32_t* x, int N);
void vec_antilog10 (float32_t * y, const float32_t* x, int N);
void vec_antilog10 (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			
<pre>int16_t, f24,int32_t, float32_t</pre>	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 Conditions for optimum performance

```
x, y – should not overlap x, y - aligned on 8-byte boundary n - multiple of 2
```

Scalar versions

Prototypes

```
int16_t scl_antilog2_16x16 (int16_t x);
int16_t scl_antilogn_16x16 (int16_t x);
int16_t scl_antilog10_16x16(int16_t x);
f24 scl_antilog2_24x24 (f24 x);
f24 scl_antilogn_24x24 (f24 x);
```

```
f24 scl_antilog10_24x24(f24 x);
int32_t scl_antilog2_32x32 (int32_t x);
int32_t scl_antilogn_32x32 (int32_t x);
int32_t scl_antilog10_32x32 (int32_t x);
float32_t scl_antilog2f (float32_t x);
float32_t scl_antilognf (float32_t x);
float32_t scl_antilog10f(float32_t x);
```

Arguments

Туре	Name	Description
Input		
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

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 sgrtf	(float32	t x);

Arguments

Туре	Туре		Name	Description	
Input					
f24 c	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 $y_n = 1/\sqrt{x_n}$

Prototype void vec_rsqrtf (float32_t* y, const float32_t* x, int N);

float32_t y N output data

none

Restrictions x, y – should not overlap

Scalar versions

Returned value

Prototypes float32_t scl_rsqrtf (float32_t x);

Arguments

Туре	Name	Description	
Input			
float32_t	Х	input data	

2.3.11 Sine/Cosine

Description

Fixed-point functions calculate $\sin(pi*x)$ or $\cos(pi*x)$ 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 ($vec_sine16x16$, $vec_sine24x24$, $vec_cosine24x24$, $vec_sine32x32$, $vec_cosine32x32$, $vec_sine32x32$, $vec_sine32x32$, $vec_sine32x32$, $vec_sine24x24$ fast, $vec_sine24x24$ fast, $vec_sine32x32$ fast, $vec_sine32x32$ fast) that apply some

NOTE:

restrictions.

- 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_cosine6 (float32_t* y, const float32_t* x, int N);
void vec_sine24x24_fast (f24 * y, const f24 * 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);
```

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, f24, int32_t, float32_t	У	N	Result, Q15, Q31 or floating point		

Returned value

None

Restrictions

Regular versions (vec_sine16x16, vec_cosine16x16, vec_sine24x24, vec_cosine24x24, vec_sine32x32, vec_sine32x32, 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

 $_{
m N}$ - multiple of 2

Scalar versions

Prototypes

Arguments

Туре	Name	Description
Input		
int16_t, f24, int32_t,	Х	input data, Q15, Q31 or floating 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

Conditions for optimum performance

x, y - aligned on 8-byte boundary

 $_{\rm N} \quad$ - multiple of 2

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,		The same of the sa
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

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

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

exception, assign EDOM to errno and return NaN..

Precision 1 version available:

Туре	Description
f	floating point. Accuracy: 2 ULP

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

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

Arguments Type Name Size Description

float32_t z N result

Returned value None

Restrictions x, z should not overlap

Scalar versions

Arguments

Prototype float32_t scl_asinf (float32_t x); float32_t scl_acosf (float32_t x);

 Type
 Name
 Description

 Input
 float32_t
 x
 input data

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

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

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, f24, int32_t, float32 t	Z	N	result, Q15, Q31 or floating point

Returned value

None

Restrictions Conditions for optimum performance

x, z should not overlap

x, z aligned on 8-byte boundary

multiple of 2

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,	X	input data, Q15, Q31 or floating point
int32_t,		The same of the sa
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				
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		γ, α		

int16 t.	V	input data, Q15 or floating point
· · · · · · · · · · · · · · · · · · ·	2	input data, Q15 or iloating point
float32 t		
1100032_0		

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 the 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 bit exact 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

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

Prototype

where norm is exponent value (maximum possible shift count) for 32-bit data. int vec bexp32 (const int32 t * x, int N);

int	vec bexp32	(const	: int32	2 t *	х,	int 1	(V		
int	vec_bexp24	(const	f24	*	х,	int 1	N);		
int	vec_bexp16	(const	int1	6_t *	х,	int 1	N);		
int	vec_bexpf	(const	float	t32_t	* X	, int	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):

 $_{\rm X}$, $_{\rm Y}~$ - aligned on 8-byte boundary $_{\rm N}~$ - multiple of 4

Scalar versions

Prototype	int scl_bexp32	! (int32_t x);
Fiolotype	int scl_bexp24	(f24 x);
	int scl bexp16	(int16 t x);
	int scl_bexpf	$(float32_t x);$

Arguments

Туре	Name	Description	
Input			
f24, int32_t, int16_t,	Х	input data	
int16_t,		Freeze	
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_max24x24_fast, vec_min16x16_fast, vec_min16x16_fast) that apply some restrictions

NOTE: functions return zero if N is less or equal to zero

Precision

4 versions available:

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

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

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

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

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
Conditions for
Optimum
Performance

x,c,z should not overlap

x,c,z - aligned on 8-byte boundary N - multiple of 2

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

Algorithm $y_n = x_n \cdot 2^t, n = \overline{0...N-1}$ Prototype $\begin{array}{c} \text{void} & \text{vec_int2float} \\ \text{(float32_t * y,} \\ \text{const int32_t * int t, int N);} \\ \end{array}$

Arguments

Туре	Name	Size	Description			
Input	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	X	input data	

Returned value

result, floating point

Restrictions

t should be in range -126...126

2.3.20 Float to Integer Conversion

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

Precision 1 version available:

Туре	Description
f	floating point input, 32-bit output

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

Prototype void vec_float2int

(int32_t * y,
 const float32_t * x,
 int t, int N);

Arguments

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

Returned value

None

Restrictions

t should be in range -126...126

Scalar version

Arguments

Туре	Name	Description
Input		
float32_t	Х	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				
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

Description Routines compute complex magnitude or its reciprocal

Precision 1 version available:

Туре	Description
f	floating point input, 32-bit output

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

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

Prototype void vec_complex2mag

Arguments

Туре	Name	Size	Description			
Input	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

Scalar version

Prototype float32 t scl complex2mag (complex float x); float32_t scl_complex2invmag (complex_float x);

Arguments

Туре	Name	Description	
Input			
complex_float	X	input data	

Returned value

result, floating point

Restrictions

None

None

2.4 Matrix Operations

2.4.1 Matrix Multiply

Description

These functions compute the expression $z=2^{lsh} * x * y$ 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 <code>scratch_mtx_mpy32x32(M,N,P)</code>, <code>scratch_mtx_mpy16x16(M,N,P)</code>

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

Arguments

Туре	Name	Size	Description
Input	•		
int32_t, f24,	X	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 \mathbb{N} vectors of size \mathbb{P} , Q31 or Q15. For floating point, this is just a matrix of size $\mathbb{N} \times \mathbb{P}$.
int	М		number of rows in matrix x and z
int	N		number of columns in matrix x and number of rows in matrix $_{\rm Y}$
int	P		number of columns in matrices y and z
int	lsh		left shift applied to the result (applied to the fixed- point functions only)
Output	· ·	1	
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 * 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* z,
              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 N, int lsh);
const f24* y,
              int M, int N, int lsh);
void mtx vecmpy16x16 fast ( int16 t* z,
              const int16 t* x,
              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	М		number of rows in matrix x
int	N		number of columns in matrix x
int	lsh		left shift applied to the result (applied to the fixed-point functions only)

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
Ish should be in range:
-31...31 for mtx_vecmpy32x32, mtx_vecmpy32x32_fast, mtx_vecmpy24x24, mtx_vecmpy24x24_fast;
-15...15 for mtx vecmpy16x16, mtx vecmpy16x16 fast

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

$$\begin{split} &z_l = x_l + y_l \\ &z_l = x_l - y_l \\ &z_l = x_l \cdot y_l \\ &z_l = x_l^T \\ &d_l = \det(x_l) \cdot 2^{-rsh} \\ &l = \overline{0...L - 1} \,, \; x_l, y_l, z_l \text{ - matrices of size NXN} \end{split}$$

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	<u> </u>	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 mul3x3f	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:

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	ч		
<pre>int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float</pre>	х	[L][N*N]	⊥ 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)
int16_t, int32_t, float32_t, complex_fract16, complex_fract32, complex_float	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_inv2x2f(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			
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
Ī	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 I-th quaternion

Prototype

Arguments

Туре	Name	Size	Description
Input			
<pre>int16_t, int32_t float32_t</pre>	ď	[L][4]	ь quaternions
int	L		number of matrices
Output			
int16_t, int32_t float32 t	r	[L][3*3]	L rotation matrices

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**.

Different types of data scaling options are provided by FFT functions. For all types of scaling, the internal representation of the data is the same as the input/output data, except for $^{*24\times24}$ _ie_24p, $^{*32\times16}$ _ie_24p functions. For these functions, the internal representation of the data is complex fract32.

Basic scaling modes:

- **dynamic scaling** (scalingOption = 2), provides the best accuracy, but has less performance compared to static scaling;
- static scaling (scalingOption = 3), has better performance but lesser accuracy than dynamic scaling.

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

HiFi 1 provides specialized instructions to facilitate dynamic scaling in FFT. The AE_SAR register contains guard-bits information and shift-amount information, which is used for implementation of dynamic scaling. The AE_S16X4RNG* instructions scan the 16-bit data stream to detect the guard-bits available and preserve the information in the AE_SAR register in addition to store operation. AE_CALCRNG3 is the helper instruction to translate the guard-bits information to meaningful shift values and updates the shift-amount information in AE_SAR register. The intrinsics AE_ADDANDSUBRNG16RAS_S1 and AE_ADDANDSUBRNG16RAS_S2 use the shift-amount information in AE_SAR for fused addition and subtraction with arithmetic-right shift operation. The instruction AE_CALCRNG3 is meant to be used after a series of AE_S16X4RNG* instructions, and before AE_ADDANDSUBRNG16RAS_S* intrinsics.

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

FFTs 24x24 have additional scaling modes:

- No scaling (scalingOption = 0), provides the highest performance, but the worst accuracy. To avoid overflow, the input data must be prescaled by the user, so that the maximum and minimum values of the samples in the input array have at least 2 + log2 (N) spare (signed) bits
- 24-bit scaling(scalingOption = 1) phase there is no normalization of the input signal, this gives a small increase in performance in comparison with scalingOption = 2. This mode is recommended for normalized input data. If input signal is small than quality will degrade.

Example of prescaling data for scalingOpt = 0:

FFT/IFFT functions family with improved memory efficiency (fft_cplx_ie, fft_real_ie, fft_cplx_ie_24p, fft_real_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 where applicable

All fixed-point FFT functions (including scaling and non-scaling) return total number of right shifts (t) occurred during all stages. Floating point FFTs do not make additional scaling so they always return 0 to indicate this fact. So, FFT/IFFT output will be scaled by 2^{t} . Library functions $\text{vec_shift}()/\text{vec_shift}32()$ helps to convert results to desired scale or Q-representation. In these computations you have to take into account the fact that FFT \rightarrow IFFT chain amplifies signal by the length of FFT $_{\text{N}}$ for complex transforms and by $_{\text{N}/2}$ for real transforms.

For example, consider processing chain:

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

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

```
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 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, real24x24	0 – no scaling 1 – 24-bit scaling 2 – 32-bit scaling on the first stage and 24-bit	Input signal < 2^23/(2*N), N - FFT size None None

² Floating point FFT available only with improved memory efficiency API

Precision	Scaling options	Restrictions on the dynamic range of the input signal	
	scaling later		
	3 – fixed scaling before each stage	None	
cplx32x16	3 – fixed scaling before each stage	None	
cplx32x32	2 – 32-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
cplx16x16	2 – 16-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
cplx16x16_ie	2 – 16-bit dynamic scaling	None	
cplx24x24_ie	3 – fixed scaling before each stage	None	
cplx32x16_ie	3 – fixed scaling before each stage	None	
cplx32x32_ie	2 – 32-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
real32x16	3 – fixed scaling before each stage	None	
real32x32	2 – 32-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
real16x16	2 – 16-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
real16x16_ie	2 – 16-bit dynamic scaling	None	
real32x16_ie	3 – fixed scaling before each stage	None	
real32x32_ie	2 – 32-bit dynamic scaling	None	
	3 – fixed scaling before each stage		
real24x24_ie	3 – fixed scaling before each stage	None	
real32x16_ie_24		None	
real24x24_ie_24		None	
cplxf_ie	•		
realf_ie			
DCT	•	-	
dct_16x16,	3 – fixed scaling before each stage	None	
dct_24x24,			
dct_32x16, dct_32x32,			
dctf			

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.

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(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	16x16
16	cfft32_16	cfft24_16	cfft16_16	cfft16x16_16
32	cfft32_32	cfft24_32	cfft16_32	cfft16x16_32
64	cfft32_64	cfft24_64	cfft16_64	cfft16x16_64
128	cfft32_128	cfft24_128	cfft16_128	cfft16x16_128
256	cfft32_256	cfft24_256	cfft16_256	cfft16x16_256
512	cfft32_512	cfft24_512	cfft16_512	cfft16x16_512
1024	cfft32_1024	cfft24_1024	cfft16_1024	cfft16x16_1024
2048	cfft32_2048	cfft24_2048	cfft16_2048	cfft16x16_2048
4096	cfft32_4096	cfft24_4096	cfft16_4096	cfft16x16_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 8-byte 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	16x16
32	rfft32_32	rfft24_32	rfft16_32	rfft16x16_32
64	rfft32_64	rfft24_64	rfft16_64	rfft16x16_64
128	rfft32_128	rfft24_128	rfft16_128	rfft16x16_128
256	rfft32_256	rfft24_256	rfft16_256	rfft16x16_256
512	rfft32_512	rfft24_512	rfft16_512	rfft16x16_512
1024	rfft32_1024	rfft24_1024	rfft16_1024	rfft16x16_1024
2048	rfft32_2048	rfft24_2048	rfft16_2048	rfft16x16_2048
4096	rfft32_4096	rfft24_4096	rfft16_4096	rfft16x16_4096
8192	rfft32_8192	rfft24_8192	rfft16_8192	rfft16x16_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

Description

These functions make inverse FFT on complex data.

NOTES:

1. Bit-reversing reordering is done here.

2. FFT runs in-place algorithm, so input data will appear damaged after call

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

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

Prototype

FFT handles:

N	32x32	24x24	32x16	16x16
16	cifft32_16	cifft24_16	cifft16_16	cifft16x16_16
32	cifft32_32	cifft24_32	cifft16_32	cifft16x16_32
64	cifft32_64	cifft24_64	cifft16_64	cifft16x16_64
128	cifft32_128	cifft24_128	cifft16_128	cifft16x16_128
256	cifft32_256	cifft24_256	cifft16_256	cifft16x16_256
512	cifft32_512	cifft24_512	cifft16_512	cifft16x16_512
1024	cifft32_1024	cifft24_1024	cifft16_1024	cifft16x16_1024
2048	cifft32_2048	cifft24_2048	cifft16_2048	cifft16x16_2048
4096	cifft32_4096	cifft24_4096	cifft16_4096	cifft16x16_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

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

 $_{\rm X}$, $_{\rm Y}$ - 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

```
int ifft_real32x32(int32_t* y, int32_t* x, fft_handle_t h, int scalingOpt) int ifft_real24x24(f24  * y, f24  * x, fft_handle_t h, int scalingOpt) int ifft_real32x16(int32_t* y, int32_t* x, fft_handle_t h, int scalingOpt) int ifft_real16x16(int16_t* y, int16_t* x, fft_handle_t h, int scalingOpt) FFT handles:
```

N	32x32	24x24	32x16	16x16
32	rifft32_32	rifft24_32	rifft16_32	rifft16x16_32
64	rifft32_64	rifft24_64	rifft16_64	rifft16x16_64
128	rifft32_128	rifft24_128	rifft16_128	rifft16x16_128
256	rifft32_256	rifft24_256	rifft16_256	rifft16x16_256
512	rifft32_512	rifft24_512	rifft16_512	rifft16x16_512
1024	rifft32_1024	rifft24_1024	rifft16_1024	rifft16x16_1024
2048	rifft32_2048	rifft24_2048	rifft16_2048	rifft16x16_2048
4096	rifft32_4096	rifft24_4096	rifft16_4096	rifft16x16_4096
8192	rifft32_8192	rifft24_8192	rifft16_8192	rifft16x16_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

These functions apply DCT (Type II) to input Description

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

y = DCT(x)Algorithm

int dct_32x32(int32_t * y, int32_t * x,int N, int scalingOpt); **Prototype**

 $(float \overline{3}2 t * y, float \overline{3}2 t * x, int N$

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
- $_{
 m N}~$ 32 for fixed point functions, 32 or 64 for floating point functions.

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. 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(x)
```

Prototype

Arguments

Туре	Name	Size	Description		
Input					
complex_fract32, complex_float	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					
complex_fract32, complex_float	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

 $_{ imes}$, $_{ imes}$ - aligned on 8-byte 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

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	1
complex_fract32, complex_fract16, complex_float	twd	N*3/4 *twdstep		twiddle factor table of a complex-valued FFT of size N*twdstep
int	twdstep			twiddle step
int	N			FFT size
int	scalingOpt			scaling option (see table in para 2.5), not applicable to the floating point function
Output				
<pre>complex_fract32, complex_float</pre>	·	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 ${\tt N}$ may be supplied with constant data (twiddle factors) of a larger-sized FFT = ${\tt N*twdstep}$.

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

Precision

```
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

x, y should not overlap 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.

Prototype

void NatureDSP_Signal_get_library_version (char *version_string);

Arguments

Туре	Name	Size	Description
Output			
char	version_string	>=30	buffer to store version information

Returned value

None

Restrictions

version_string must point 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.

Prototype void NatureDSP_Signal_get_library_api_version (char *version_string);

Arguments Type Name Size Description

 Output

 char
 version_string
 >=30
 buffer to store version information

Returned value None

Restrictions version_string must point to a buffer large enough to hold up to 30 characters

3 Test Environment and Examples

3.1 Supported Use Environment, Configurations and Targets

NatureDSP library and corresponding testdriver supported to be built and tested using Xtensa Xplorer IDE running under Windows, or Linux operating system.

Library is compatible with HiFi 1 DSP cores having HiFi 1 Vector FPU. Floating-Point kernels require VFPU to be part of the configuration.

3.2 Building the NatureDSP Signal Library and the Testdriver

3.2.1 Importing the workspaces in Xtensa Xplorer

NatureDSP Libraries for HiFi 1 are provided as two workspaces:

- Library workspace HiFi1_lib.xws
 - This workspace contains <code>HiFil_library</code> project with optimized kernels and modules required for demo workspace.
- Demo workspace HiFil demo.xws
 - This contains the HiFil demo demo project.

Import these two workspaces (.xws) in Xtensa Xplorer 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

Building the library: In Xtensa Xplorer, select the <code>HiFil_library</code> project, and <code>Debug</code> or <code>Release</code> target, and build. By default the software will be built with xt-clang compiler.

Building the test bench: In Xtensa Xplorer, select the <code>HiFil_demo</code> project, select Debug or Release target, and build.

To run the test bench, select <code>HiFil_demo</code> project, and Run. This will execute each routine of the <code>HiFil_library</code> in cycles performance (MIPS) mode.

Use --turbo as runtime argument to test library for functional correctness

3.2.3 Command-line Options

You may wish to launch a separate test by passing command-line options to the executable: Executing the testdriver without options performs performance testing of library with vectors -full.

Functional testing is performed with the explicit command-line option -func, with either -brief or -full vectors. Additionally, it may collect statistics and generate validation report showing the number of calls of each specific library function, amount of data passed to/from, sorts of specific tests performed, etc.

Running performance tests for all library functions or for specific category is controlled by command line option -mips. In that case, functional testing is not performed, and validation report will be empty. Performance data is always executed with -mips -full. There is no -mips -brief option.

You may wish to launch a separate test by passing command-line options to the executable:

-help or -h	List of available options		
-mips	Performance test		
-func	Functional test		
-full	Use full test vector set (if available in the directory vectors_full) for deeper validation		
-brief	Use brief vector set (if available in the directory vectors_brief) for functional validation		
-sanity	Use sanity vector set (if available in the directory vectors_sanity) for functional validation		
-dct	DCT tests		
-fft	FFT tests		
-cfft	Complex floating point FFT and fixed point FFT with memory improved usage tests		
-rfft	Real floating point FFT and fixed point FFT with memory improved usage tests		
-fir	FIR tests		
-iir	IIR tests		
-vec	Vector operations		
-math	Vector mathematics		
-mtx	Matrix operations		
-mtxinv	Matrix inversions		
-phase1	Test fixed-point routines		
-phase2	Test floating-point routines		

3.2.4 Other Supported Environments

The library and testdriver project might be built under following toolchains and operating systems:

os	Language	Environment	Tool
Linux	С	xt-clang	make
Windows	С	xt-clang	xt-make

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 bgriir24x24 df1 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir24x24_df1 function)
% parameters:
% SOS,G - SOS matrix and gain vector G
         - sample rate
% Fs
          - FFT length for analisys
% nfft
% output:
% coef - vector with coefficients, Q30
% gain
          - biquad gains, Q15
         - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir24x24 df1(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
f = (0:nfft-1)/nfft*(Fs/2):
tf0=sos2freqz(SOS,G,nfft);
coef=[];
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    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));
tf=sos2freqz(sos,g,nfft);
plot(f,20*loq10(abs(tf)),f,20*loq10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
on;
\mbox{\%} convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
```

```
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
    [b,a]=sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

4.1.2 bgriir32x16 df1 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16 df1 function)
% parameters:
% SOS,G
           - SOS matrix and gain vector G
% Fs
           - sample rate
           - FFT length for analisys
% nfft
% output:
           - vector with coefficients, 030
% coef
          - biquad gains, Q15
% gain
         - final scale factor (amount of left shifts)
% scale
function [coef, gain, scale] = cvtsos bqriir32x16 df1(SOS, G, Fs, nfft)
sz=size(SOS):
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for intermediate output <=0.5
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax=max(abs(tf));
    G(m) = min(1, 0.5/tfmax);
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);
\mbox{\ensuremath{\$}} check results and plot final filter response
sos=reshape(double(coef),5,M).'/16384;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
plot(f, 20*log10(abs(tf)), f, 20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid
\mbox{\%} convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
    [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

4.1.3 bqriir24x24_df2 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir24x24_df1 function)
% parameters:
% SOS,G
          - SOS matrix and gain vector G
% Fs
          - sample rate
           - FFT length for analisys
% nfft
% output:
% coef
           - vector with coefficients, 030
          - biquad gains, Q15
% gain
         - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir24x24 df2(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for
% intermediate outputs <=0.5
% note: for DF2 structure we have to check 2 points:
% b0,b1,b2,a1,a2 and 1,0,0,a1,a2
for m=1:M
    tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
    tfmax0=max(abs(tf));
    tf=sos2freqz([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1],nfft);
    tfmax1=max(abs(tf));
    tfmax = max(tfmax0,tfmax1);
    G(m) = min(1, 0.5/tfmax);
    % 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=sos2freqz(sos,q,nfft);
plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid(-80 0)
on;
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1):
[b,a] = sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
    [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
    tf=tf.*freqz(b,a,nfft);
end
```

4.1.4 bqriir32x16_df2 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16_df2 function)
% parameters:
% SOS,G
                   - SOS matrix and gain vector G
                      - sample rate
% Fs
                       - FFT length for analisys
% nfft
% output:
% coef
                       - vector with coefficients, 014
                     - biquad gains, Q15
% gain
                  - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir32x16 df2(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for
% intermediate outputs <=0.5
% note: for DF2 structure we have to check 2 points:
% b0,b1,b2,a1,a2 and 1,0,0,a1,a2
for m=1:M
        tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
         tfmax0=max(abs(tf));
         tf=sos2freqz([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1],nfft);
        tfmax1=max(abs(tf));
        tfmax = max(tfmax0,tfmax1);
        G(m) = min(1,0.5/tfmax);
end
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b,a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int16(round(16384.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
% check results and plot final filter response
sos=reshape(double(coef),5,M).'/16384;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
\verb|plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid('transfer function, 
on;
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
         [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
         tf=tf.*freqz(b,a,nfft);
end
```

4.1.5 bgriir32x32 df2 conversion

```
% convert SOS+G to coefficients of IIR filter
% (bqriir32x16_df2 function)
% parameters:
% SOS,G
                   - SOS matrix and gain vector G
% Fs
                      - sample rate
                       - FFT length for analisys
% nfft
% output:
% coef
                       - vector with coefficients, 030
                     - biquad gains, Q15
% gain
                  - final scale factor (amount of left shifts)
function [coef,gain,scale]=cvtsos bqriir32x32 df2(SOS,G,Fs,nfft)
sz=size(SOS);
M=sz(1);
coef=[];
f=(0:nfft-1)/nfft*(Fs/2);
tf0=sos2freqz(SOS,G,nfft);
Gtotal=prod(G);
G=ones(1,M+1);
% define gain for each stage to provide maximim of tf for
% intermediate outputs <=0.5
% note: for DF2 structure we have to check 2 points:
% b0,b1,b2,a1,a2 and 1,0,0,a1,a2
for m=1:M
        tf=sos2freqz(SOS(1:m,:),[G(1:m) 1],nfft);
         tfmax0=max(abs(tf));
        tf=sos2freqz([SOS(1:m-1,:);1 0 0 1 SOS(m,5:6)],[G(1:m) 1],nfft);
        tfmax1=max(abs(tf));
        tfmax = max(tfmax0,tfmax1);
        G(m) = min(1, 0.5/tfmax);
end
% define last stage shift
dg=Gtotal/prod(G);
scale=ceil(log2(dg));
% correct coefficient of the last stage
d=pow2(1,scale)/dg;
G(M) = G(M) / d;
% output b, a
coef=SOS; coef(:,4)=[]; coef=reshape(coef.',1,numel(coef));
% and convert coefficients to given format
coef = int32(round(1073741824.*coef));
gain = int16(round(32768.*G(1:M)));
scale= int16(scale);
% check results and plot final filter response
sos=reshape(double(coef),5,M).'/1073741824;
sos=[sos(:,1:3) ones(M,1) sos(:,4:5)];
g=double(gain)/32768;
g(M+1) = pow2(1, double(scale));
tf=sos2freqz(sos,g,nfft);
\verb|plot(f,20*log10(abs(tf)),f,20*log10(abs(tf0))); ylim([-80 0]); title('transfer function, dB'); grid('transfer function, 
on;
% convert SOS/G to frequency response
function [tf]=sos2freqz(SOS,G,nfft)
sz=size(SOS);
M=sz(1);
[b,a]=sos2tf(SOS(1,:),[G(1) G(end)]);
tf=freqz(b,a,nfft);
for m=2:M
         [b,a] = sos2tf(SOS(m,:),[G(m) 1]);
         tf=tf.*freqz(b,a,nfft);
end
```

4.2 Matlab Code for Generating the Twiddle Tables

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

4.2.1 Twiddles for fft_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_reaB2x16_ie, ifft_reaB2x16_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 = \operatorname{reshape}([\operatorname{real}(\operatorname{twd}(:).');\operatorname{imag}(\operatorname{twd}(:).')],1,2*\operatorname{numel}(\operatorname{twd}));
```

4.3 Accuracy of Mathematical Functions

Some math functions such as sine, cosine, etc. specify accuracy of computation. General approach is to provide accuracy of 2 LSB (+/-2) for 16x16 routines and 2 ULP for single precision floating point routines. The 24x24 and 32x32 routines are normally more accurate than 16x16, but the level of accuracy varies from function to function.

Accuracy for functions is guaranteed over defined domain i.e. for <code>vec_sinef()</code> it is 2 ULP for all values in range <code>[-102940.0, 102940.0]</code>. For most fixed-point functions, domain is a whole range from minimum to maximum number, for example, <code>-32768</code> to <code>32767</code> for 16x16 routines. Since mathematical functions are defined for continuous values while the processor works with discrete numbers, the method of measuring accuracy requires clarification.

Accuracy for function y=f(x) is defined as a maximum difference between digitized output of function y and its digitized ideal value f(x). Input for function is digitized as well. In practice, seqfiles contain input data for function and allowable upper/lower bounds for its output. Test harness tool just passes input data to function under the test and compares output with upper/lower bounds from the seq-file.

As an example, look for the test data from scl sine16x16.seq:

```
13 2; test case # and type (small and negative N) 2 ; N -9076 -8055 ; x[2] -25051 -22866 ; ylo[2] (lower bound)
```

```
-25047 -22862 ; yhi[2] (upper bound)
```

Here we have 2 inputs for sine function. Inputs and outputs are in Q15 and input is scaled by pi, so expected outputs might be computed by Matlab code:

```
round(pow2(sin(pow2([-9076 -8055],-15)*pi),15))
result is [-25049, -22864].
```

So, lower and higher bounds differ from expected result by 2.

```
Similar case from scl_sinef()
2 1; test case # and type (known data)
1 ; N
47154fc2 ; x[1]
367ald9c ; ylo[1] (lower bound)
367alda0 ; yhi[1] (upper bound)
```

Here all the single precision data are shown in their hex machine representation as they appear in the processor.

In natural representation, input is

```
x=(typecast(int32(hex2dec('47154fc2')),'single'))
38223.7578125
```

Right sine value is 3.7270143e-006 or

num2hex(single(sin(double(x))))=367a1d9e in machine representation

So, we see that upper/lower bounds in seq-file are 2 ULP apart from this ideal value.

For 24x24, the situation is bit different because 24-bit data are stored in 3 most significant bytes of 32-bit data. So, for input data, lower 8 bits should be ignored, and ideal output should be computed using digitized inputs with cleared 8 LSBs. Comparison with lower/upper bounds should ignore these 8 LSBs, this is achieved by small modification of bounds - 8 LSB for lower bound is cleared and 8 LSB for upper bound are set.

```
Here is the part of scl_sine24x24.seq:
```

```
22 2; test case # and type (small and negative N)
4 ; N
1519943108 1365826688 2070175853 2062996324 ; x[4]
1705915136 1954328576 242278400 264675584 ; ylo[4] (lower bound)
1706063359 1954476799 242426623 264823807 ; yhi[4] (upper bound)
```

So, for first input x[0]=1519943108 we have digitized value $1519943108\&\sim255=1519942912$. The ideal sine value for it is y=round(pow2(sin(pow2(x,-31)*pi),31))=1705989149. Function provides accuracy 74000, so lower bound is $(y-74000)\&\sim255=1705915136$ and upper bound is (y+74000)|255=1706063359

Fixed point functions usually indicate decimal accuracy as well as its floating-point equivalent. The relationship between these quantities depends on fractional format which is specific for each function. For example, $vec_sine24x24$ () specifies decimal accuracy 74000. Output is represented in Q31, so floating point accuracy will be $74000*2^-31=3.4e-5$. See chapter 1.4 for details.