

5 DSP LIBRARY FOR ADSP-2116X PROCESSORS

Overview

The run-time library for ADSP-2116x processors contains a collection of functions that provide services commonly required by DSP applications; these functions are in addition to the C/C++ run-time library functions that are described in Chapter 3. The services provided by the DSP library functions include support for interrupt handling, signal processing, and access to hardware registers. All these services are Analog extensions to ANSI standard C.

For more information on the algorithms on which many of the C library's math functions are based, see Cody, W.J. and W. Waite, *Software Manual for the Elementary Functions*, Englewood Cliffs, New Jersey: Prentice Hall, 1980.

The sections of this chapter present the following information on the compiler:

- “[DSP Run-Time Library Guide](#)” (starting on [page 5-2](#)) contains introductory information about the ADI special header files and built-in functions that are included with this release of the cc21k compiler.
- “[DSP Run-Time Library Reference](#)” (starting on [page 5-15](#)) contains the complete reference of the DSP run-time functions included with this release of the cc21k compiler.

DSP Run-Time Library Guide

The C++ library reference information in HTML format is included on the software distribution CD-ROM. To access the reference files from VisualDSP++, see the procedure described in [“Related Documents” on page 1-5](#). Select the *C++ Run-Time Library Reference* from the list of documents.



You can also manually access the HTML files using a web browser.

DSP Run-Time Library Guide

The DSP run-time library contains routines that you can call from your source program. This section describes how to use the library and provides information on the following topics:

- [“Linking DSP Library Functions” on page 5-3](#)
- [“Working With Library Source Code” on page 5-3](#)
- [“DSP Header Files” on page 5-4](#)
- [“Built-In DSP Functions” on page 5-12](#)
- [“Pitfalls Using SIMD Mode” on page 5-13](#)

For information on the contents of the DSP library, see [“DSP Run-Time Library Reference” on page 5-15](#) and also on-line Help.

Linking DSP Library Functions

When your C code calls a DSP run-time library function, the call creates a reference that the linker resolves when linking your program. One way to direct the linker to the location of the DSP library is to use the default Linker Description File (ADSP-21<your_target>.ldf). The default Linker Description File will automatically direct the linker to the library `libdsp160.dlb` in the `21xxx\lib` subdirectory of your VisualDSP installation. If not using the default LDF file, then either add `libdsp160.dlb` to the LDF used for your project, or alternatively use the compiler's `-ldsp160` switch to specify that `libdsp160.dlb` is to be added to the link line.

Working With Library Source Code

The source code for the functions and macros in the DSP run-time library is provided with your VisualDSP software. By default, the installation program copies the source code to a subdirectory of the directory where the run-time libraries are kept, named `...\21xxx\lib\src`. The directory contains the source for the C run-time library, for the DSP run-time library, and for the I/O run-time library, as well as the source for the main program start-up functions. If you do not intend to modify any of the run-time library functions, you can delete this directory and its contents to conserve disk space.

The source code is provided so you can customize any particular function for your own needs. To modify these files, you need proficiency in ADSP-21xxx assembly language and an understanding of the run-time environment, as explained in [“C/C++ Run-Time Model” on page 2-121](#). Before you make any modifications to the source code, copy the source code to a file with a different filename and rename the function itself. Test the function before you use it in your system to verify that it is functionally correct. Note that Analog Devices supports the run-time library functions only as provided.

DSP Header Files

The following header files are supplied with this release of the cc21k compiler.

21160.h — ADSP-2116x DSP Functions

The `21160.h` header file includes the ADSP-2116x processor-specific functions of the DSP library, such as `poll_flag_in()`, `timer_set()`, and `idle()`. The `timer_set()`, `timer_on()`, and `timer_off()` functions are also available as in-line functions.

asm_sprt.h — Mixed C/Assembly Support

The `asm_sprt.h` header file consists of ADSP-21xxx family assembly language macros, not C functions. They are used in your assembly routines that interface with C functions. For more information on this header file, see [“Using Mixed C/C++ and Assembly Support Macros” on page 2-153](#).

comm.h — A-law and μ -law Companders

The `comm.h` header file includes the voice-band compression and expansion communication functions as supported by the DSP library for ADSP-2106x DSP processors. However, the functions defined by this header file have not been optimized for the ADSP-2116x architecture. Versions of these functions that have been optimized for the ADSP-2116x architecture are available in the header file `filter.h`. Since these two sets of functions have different arguments, it is important that the user program includes the appropriate header file.

complex.h — Basic Complex Arithmetic Functions

The `complex.h` header file contains the type definition and some basic functions for `complex_float` variables.

The following structure is used to represent complex numbers in rectangular coordinates:

```
typedef struct {  
    float re;  
    float im;  
} complex_float ;
```

Additional support for `complex_float` is available via the `cvector.h` header file.

cvector.h — Complex Vector Functions

The `cvector.h` header file contains functions for basic arithmetical operations on vectors of type `complex_float`. Support is provided for the dot product operation, as well as for adding, subtracting, and multiplying a vector by either a scalar or vector.

def21160.h — ADSP-21160 Bit Definitions

The `def21160.h` header file includes macro definitions to enable usage of symbolic names for the system register bits for the ADSP-21160 processors. It also contains macro definitions for the IOP register addresses and bit fields.

def21161.h — ADSP-21161 Bit Definitions

The `def21161.h` header file includes macro definitions to enable usage of symbolic names for the system register bits for the ADSP-21161 processors. It also contains macro definitions for the IOP register addresses and bit field.

dma.h — DMA Support Functions

The `dma.h` header file provides definitions and setup, status, enable, and disable functions for DMA operations.

filter.h — DSP Filters and Transformations

The `filter.h` header file contains filters used in digital signal processing. It also includes the A-law and μ -law companders that are used by voice-band compression and expansion applications.

Additionally, the header file contains the Fast Fourier Transform (FFT) functions of the DSP library. The various forms of the FFT functions provided are `cfftN`, `ifftN`, `rfftN`, and `rftN_2`. Each form is represented by N separate functions, where N represents the number of points supported by the FFT. For example, `cfftN` stands for the functions `cfft65536`, `cfft32768`, and so forth.

The prototypes defined by this header file have been designed to allow the functions to exploit the parallelism within the ADSP-2116x architecture. Equivalent functions that have not been optimized for the ADSP-2116x architecture are supported and are documented in [Chapter 4 DSP Library for ADSP-2106x Processors](#). These functions are defined in separate header files (`comm.h` for the companding functions, `filters.h` for the filters, and `trans.h` for the FFTs). Since these two sets of functions have different arguments, it is important that the user program includes the appropriate header file.

The FFT functions are optimized to take advantage of the SIMD (Single Instruction Multiple Data) execution model of the ADSP-2116x SHARC DSP, and the function prototypes have been adjusted accordingly.

Complex arguments and complex results must be defined using the `complex_float` data type (defined in the header file `complex.h`). Using the `complex_float` data type ensures that the real and imaginary parts of a complex value are interleaved in memory and therefore are accessible in a single cycle using the wider data bus of the processor.

The FFT functions have also been optimized with respect to memory usage. In general, Fast Fourier Transform algorithms require access to a temporary working buffer. All the FFT functions defined by the `filter.h` header file assume that the input buffer may be corrupted and therefore may be used as a temporary area. This assumption allows the functions to be more efficient both in terms of memory usage and processor speed.

The `cfftN` functions compute the fast Fourier transform of their N-point complex input signal. The `ifftN` functions compute the inverse fast Fourier transform of their N-point complex input signal. The input to each of these functions is a `complex_float` array of N elements. The routines output an N-element array of type `complex_float`. If you wish only to input the real part of a signal, ensure that the imaginary components of the input array are set to zero before calling the function.

The functions first bit-reverse the input arrays and then process them with an optimized block-floating-point FFT (or IFFT) routine.

The `rfftN` functions work like `cfftN` functions, except they operate only on input arrays of real data. This type of operation is equivalent to a `cfftN` function whose imaginary input component is set to zero. The `rfft2_N` functions are similar to the `rfftN` functions except that they operate on an interleaved real array and return two complex FFT results.

filters.h — DSP Filters

The `filters.h` header file includes the digital signal processing filter functions as supported by the DSP library for ADSP-2106x DSP processors. However, the functions defined by this header file have not been optimized for the ADSP-2116x architecture. Versions of these functions that have been optimized for the ADSP-2116x architecture are available in the `filter.h` header file. Since these two sets of functions have different arguments, it is important that the user program includes the appropriate header file.

macros.h – Circular Buffers

The `macro.h` header file consists of ADSP-21xxx family assembly language macros, not C functions. Some are used to manipulate the circular buffer features of the ADSP-21xxx family processors.

math.h — Math Functions

The standard math functions defined in `math.h` have been augmented by implementations for the `float` data type and some additional functions that are Analog Devices extensions to the ANSI standard. [Table 5-1](#) provides a summary of the additional library functions defined by the header file.

Table 5-1. Math Library - Additional Functions

Description	Prototype
sign copy	<code>double copysign (double x, double y);</code> <code>float copysignf (float x, float y);</code>
cotangent	<code>double cot (double x);</code> <code>float cotf (float x);</code>
average	<code>double favg (double x, double y);</code> <code>float favgf (float x, float y);</code>
clip	<code>double fclip (double x, double y);</code> <code>float fclipf (float x, float y);</code>
maximum	<code>double fmax (double x, double y);</code> <code>float fmaxf (float x, float y);</code>
minimum	<code>double fmin (double x, double y);</code> <code>float fminf (float x, float y);</code>
reciprocal of square root	<code>double rsqrt (double x, double y);</code> <code>float rsqrtf (float x, float y);</code>



The following functions are only available if `double` is the same size as `float`:

```
copysign
favg
fclip
fmax
fmin
```

matrix.h — Matrix Functions

The `matrix.h` header file contains functions for operating on real matrices; specifically it defines functions for adding and subtracting two matrices and for multiplying a matrix by either a scalar or a matrix.

saturate.h — Saturation Mode Arithmetic

The `saturate.h` header file defines the interface for the saturated arithmetic operations. See [“Saturated Arithmetic” on page 2-87](#) for further information.

sport.h — Serial Port Support Functions

The `sport.h` header file provides definitions and setup, enable, and disable functions for the ADSP-2116x SHARC serial ports.

stats.h — Statistical Functions

The `stats.h` header file includes various statistics functions of the DSP library, such as `mean()` and `autocorr()`.

sysreg.h — Register Access

The `sysreg.h` header file defines a set of built-in functions that provide efficient access to the SHARC system registers from C. The supported functions are fully described in the section [Chapter 2 Compiler](#).

trans.h — Fast Fourier Transforms

The `trans.h` header file includes Fast Fourier Transform (FFT) functions as supported by the DSP library for 2106x DSP processors. However, the functions defined by this header file have not been optimized for the ADSP-2116x architecture. Versions of these functions that have been optimized for the ADSP-2116x architecture are available in the header file `filter.h`. Since these two sets of functions have different arguments, it is important that the user program includes the appropriate header file.

vector.h — Vector Functions

The `vector.h` header file contains functions for operating on vectors of type `float`. Support is provided for the dot product operation and well as for adding, subtracting, and multiplying a vector by either a scalar or vector.

window.h — Window Generators

The `window.h` header file contains various functions to generate windows based on various methodologies. The functions defined in the `window.h` header file are listed in [Table 5-2](#).

For all window functions, a stride parameter `a` can be used to space the window values. The window length parameter `n` equates to the number of elements in the window. Therefore, for a “stride `a`” of 2 and a “length `n`” of 10, an array of length 20 is required, where every second entry is untouched.

Table 5-2. Window Generator Functions

Description	Prototype
generate bartlett window	void gen_bartlett (float w[], int a, int n)
generate blackman window	void gen_blackman (float w[], int a, int n)
Generate gaussian Window	void gen_gaussian (float w[], float alpha, int a, int n)
generate hamming window	void gen_hamming (float w[], int a, int n)
generate hanning window	void gen_hanning (float w[], int a, int n)
generate harris window	void gen_harris (float w[], int a, int n)
generate kaiser window	void gen_kaiser (float w[], float beta, int a, int n)
generate rectangular window	void gen_rectangular (float w[], int a, int n)
generate triangle window	void gen_triangle (float w[], int a, int n)

Built-In DSP Functions

The C/C++ compiler supports built-in functions (also known as intrinsic functions) that enable efficient use of hardware resources. Knowledge of these functions is built into the compiler. Your program uses them via normal function call syntax. The compiler notices the invocation and replaces a call to a DSP library function with one or more machine instructions—just as it does for normal operators like “+” and “*”.

Built-in functions are declared in system header files and have names which begin with double underscores, `__builtin`.

 Identifiers beginning with “`__`” are reserved by the C standard, so these names do not conflict with user defined identifiers.

These functions are specific to individual architectures. The built-in DSP library functions supported at this time on the ADSP-2116x architectures are listed in [Table 5-3](#). Refer to [“Using the Compiler’s Built-In C library Functions” on page 3-18](#) for further information on this topic.

Table 5-3. Built-in DSP Functions

<code>copysign¹</code>	<code>copysignf</code>
<code>favg¹</code>	<code>favgf</code>
<code>fmax¹</code>	<code>fmaxf</code>
<code>fmin¹</code>	<code>fminf</code>

¹ These functions will only be compiled as a built-in function if `double` is the same size as `float`.

 Use the `-no-builtin` compiler switch to disable this feature.

The compiler also supports a set of built-in functions for which no in-line machine instructions are substituted. This set of built-in functions is characterized by defining one or more pointers in their argument list.

For this set of built-in functions, the compiler relaxes the normal rule whereby any pointer that is passed to a library function must address Data Memory (DM). The compiler recognizes when certain pointers address Program Memory (PM) and generates a call to an appropriate version of the run-time library function. The following is a list of library functions that may be called with pointers that address Program Memory:

```
matadd
matmul
matscalmult
matsub
```



Use the `-no-builtin` compiler switch to disable this feature.

Pitfalls Using SIMD Mode

The DSP run-time library for the ADSP-2116x DSP makes extensive use of the DSP's SIMD capabilities. This feature is described under [“SIMD Support Annotation \(#pragma SIMD_for\)” on page 2-88](#). In essence, when running in SIMD mode, data contained in memory is always accessed as two 32-bit words, starting at an even word boundary. Therefore, it is essential that any array that is passed to a DSP library function be allocated on a double word (even word) boundary.

The cc21k compiler normally aligns arrays properly in memory. However, the compiler cannot control the allocation of all arrays that are used as arguments to DSP library functions. For example, the alignment of the array `&a[i]` is controlled by the value of the scalar `i`. If the value of the scalar is odd, then the library function will return incorrect results. A variant of this example involves the use of pointers to arrays. If the variable `ptr` is initialized using `ptr=&a[i]` and the value of the scalar `i` is odd, then you cannot use `ptr` to pass an array to a DSP library function.

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A limited number of DSP library functions, whose arguments involve the use of arrays, do not use the SIMD feature of the ADSP-2116x due to the nature of their algorithm. These library functions are:

```
histogram  
iir  
zero_cross  
matmul
```

DSP Run-Time Library Reference

The DSP run-time library is a collection of functions that you can call from your C programs. This section lists the functions in alphabetical order. Note the following items that apply to all the functions in the library.

Notation Conventions. An interval of numbers is indicated by the minimum and maximum, separated by a comma, and enclosed in two square brackets, two parentheses, or one of each. A square bracket indicates that the endpoint is included in the set of numbers; a parenthesis indicates that the endpoint is not included.

Function Benchmarks and Specifications. All functions have been timed from setup, to invocation, to results storage of returned value. This includes all register storing, parameter passing, etc. Most functions execute slightly faster if you pass constants as arguments instead of variables.

Restrictions. When polymorphic functions are used and the function returns a pointer to program memory, cast the output of the function to pm. For example:

```
(char pm *)
```

Reference Format. Each function in the library has a reference page. These pages follow the following format:

Name and Purpose of the function

Synopsis—Required header file and functional prototype

Description—Function specification

Error Conditions—Method function uses to indicate an error

Example—Typical function usage

See Also—Related functions

a_compress

A-law compression

Synopsis

```
#include <filter.h>
int *a_compress (const int dm input[],
                 int dm output[],
                 int length);
```

Description

The `a_compress` function takes an array of linear 13-bit signed speech samples and compresses them according to CCITT recommendation G.711. The array returned contains 8-bit samples that can be sent directly to an A-law codec. This function returns a pointer to the `output` data array.

The function uses serial port 0 to perform companding. Therefore, serial port 0 must not be in use when this routine is called.

Error Conditions

The `a_compress` function does not return an error condition.

Example

```
#include <filter.h>
int data[50], compressed[50];

a_compress (data, compressed, 50);
```

See Also

[a_expand](#), [mu_compress](#)

a_expand

A-law expansion

Synopsis

```
#include <filter.h>
int *a_expand (const int dm input[],
              int dm output[],
              int length);
```

Description

The `a_expand` function takes an array of 8-bit compressed speech samples and expands them according to CCITT recommendation G.711 (A-law definition). The array returned contains linear 13-bit signed samples. This function returns a pointer to the `output` data array.

The function uses serial port 0 to perform companding. Therefore, serial port 0 must not be in use when this routine is called.

Error Conditions

The `a_expand` function does not return an error condition.

Example

```
#include <filter.h>
int expanded_data[50], compressed_data[50];

a_expand (compressed_data, expanded_data, 50);
```

See Also

[a_compress](#), [mu_expand](#)

autocoh

autocoherence

Synopsis

```
#include <stats.h>
float *autocoh (float dm out[],
               const float dm in[],
               int samples,
               int lags);
```

Description

The `autocoh` function computes the autocoherence of the floating-point input, `in[]`. The autocoherence of an input signal is its autocorrelation minus its mean. The function returns a pointer to the output array, `out[]` of length `lags`.

Error Conditions

The `autocoh` function does not return an error condition.

Example

```
#include <stats.h>
#define SAMPLES 1024

float excitation[SAMPLES], response[16];
int    lags = 16;

autocoh (response, excitation, SAMPLES, lags);
```

See Also

[autocorr](#), [crosscoh](#), [crosscorr](#)

autocorr

autocorrelation

Synopsis

```
#include <stats.h>
float *autocorr (float dm out[],
                const float dm in[],
                int samples,
                int lags);
```

Description

The `autocorr` function performs an autocorrelation of a signal. Autocorrelation is the cross-correlation of a signal with a copy of itself. It provides information about the time variation of the signal. The signal to be autocorrelated is given by the `in[]` input array. The number of samples of the autocorrelation sequence to be produced is given by `lags`. The length of the input sequence is given by `samples`. This function returns a pointer to the `out[]` output data array of length `lags`.

The `autocorr` function is used in digital signal processing applications such as speech analysis.

Error Conditions

The `autocorr` function does not return an error condition.

Example

```
#include <stats.h>
float r[10], s[160];

autocorr (r, s, 160, 10);
/* compute first 10 autocorr coefficients of array s */
```

See Also

[autocoh](#), [crosscoh](#), [crosscorr](#)

cabsf

complex absolute value

Synopsis

```
#include <complex.h>
float cabsf (complex_float z);
```

Description

The `cabsf` function returns the floating-point absolute value of its complex input.

The absolute value of a complex number is evaluated with the following formula:

$$y = \sqrt{(\operatorname{Re}(x))^2 + (\operatorname{Im}(x))^2}$$

where `x` is the `complex_float` input and `y` is the `float` output.

Error Conditions

The `cabsf` function does not return an error condition.

Example

```
#include <complex.h>
complex_float cnum;
float answer;

cnum.re = 12.0;
cnum.im = 5.0;

answer = cabsf (cnum);      /* answer = 13.0 */
```

See Also

[fabs](#), [fabsf](#), [labs](#)

cexpf

complex exponential

Synopsis

```
#include <complex.h>
complex_float cexpf (complex_float z);
```

Description

The `cexpf` function computes the complex exponential value e to the power of the first argument.

The exponential of a complex value is evaluated with the following formula:

$$\operatorname{Re}(y) = \exp(\operatorname{Re}(x)) * \cos(\operatorname{Im}(x));$$

$$\operatorname{Im}(y) = \exp(\operatorname{Re}(x)) * \sin(\operatorname{Im}(x));$$

where x is the `complex_float` input and y is the `complex_float` output.

Error Conditions

For underflow errors the `cexpf` function returns zero.

Example

```
#include <complex.h>
complex_float cnum;
complex_float answer;

cnum.re = 1.0;
cnum.im = 0.0;

answer = cexpf (cnum);      /* answer = (2.7182 + 0i) */
```

See Also

[pow](#), [powf](#), [log](#), [logf](#)

cfftN

N-point complex input fast Fourier transform

Synopsis

```
#include <filter.h>
complex_float *cfft65536 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *cfft32768 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *cfft16384 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *cfft8192 (complex_float dm input[],
                         complex_float dm output[]);

complex_float *cfft4096 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *cfft2048 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *cfft1024 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *cfft512 (complex_float dm input[],
                       complex_float dm output[]);

complex_float *cfft256 (complex_float dm input[],
                       complex_float dm output[]);

complex_float *cfft128 (complex_float dm input[],
                       complex_float dm output[]);

complex_float *cfft64 (complex_float dm input[],
                      complex_float dm output[]);
```

```
complex_float *cfft16 (complex_float dm input[],
                      complex_float dm output[]);

complex_float *cfft8 (complex_float dm input[],
                     complex_float dm output[]);
```

Description

These functions are optimized to take advantage of the SIMD execution model of the ADSP-2116x SHARC DSP, and requires complex arguments to ensure that the real and imaginary parts are interleaved in memory and are therefore accessible in a single cycle using the wider data bus of the processor.

Each of these 14 `cfftN` functions computes the N-point radix-2 fast Fourier transform (CFFT) of its complex input (where N is 8, 16, 32, 64, 128, 256, 512, 1024, 2048, 4096, 8192, 16384, 32768 or 65536).

There are fourteen distinct functions in this set. They all perform the same function with the same type and number of arguments. The only difference between them is the size of the arrays they operate on. Call a particular function by substituting the number of points for N, as in

```
cfft8 (input, output);
```

The input to `cfftN` is a floating-point array of N points. If there are fewer than N actual data points, you must pad the array with zeros to make N samples. Better results occur with less zero padding, however. The input data should be windowed (if necessary) before calling the function because no preprocessing is performed on the data.

`cfftN()` returns a pointer to the `output` array.

Error Conditions

The `cfftN` functions do not return any error conditions.

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Example

```
#include <filter.h>
#define N 2048
complex_float input[N], output[N];

/* input array is filled from a converter or other source */

cfft2048 (input, output);
/* Array is filled with FFT data */
```

See Also

[ifftN](#), [rfftN](#), [rfft2_N](#)



The `cfftN` functions use the input array as an intermediate workspace. If the input data is to be preserved it must first be copied to a safe location before calling these functions.

copysign, copysignf

copy the sign of the floating-point operand (IEEE arithmetic function)

Synopsis

```
#include <math.h>
double copysign (double x, double y);
float copysignf (float x, float y);
```

Description

The `copysign` and `copysignf` functions copy the sign of the second argument `y` to the first argument `x` without changing either its exponent or mantissa. The `copysignf` function is a built-in function which is implemented with an `Fn=Fx COPYSIGN Fy` instruction.

Error Conditions

This function does not return an error code.

Example

```
#include <math.h>
double x;
float y;

x = copysign (0.5, -10.0); /* x = -0.5*/
y = copysignf (-10.0, 0.5f); /* y = 10.0*/
```

See Also

No references to this function.



The double precision function `copysign` is only available under `-double-size-32` and actually calls the single precision function `copysignf`.

cot, cotf

cotangent

Synopsis

```
#include <math.h>
double cot (double x);
float cotf (float x);
```

Description

The `cot` and `cotf` functions return the cotangent of their argument. The input is interpreted as radians.

The `cot` and `cotf` functions return a value that is accurate to 20 bits of the mantissa. This accuracy corresponds to a maximum relative error of 2^{-20} over its input range.

Error Conditions

The `cot` and `cotf` functions do not return an error condition.

Example

```
#include <math.h>
double x, y;
float v, w;

y = cot (x);
v = cotf (w);
```

See Also

[tan, tanf](#)

crosscoh

cross-coherence

Synopsis

```
#include <stats.h>
float *crosscoh (float dm out[],
                const float dm x[],
                const float dm y[],
                int samples,
                int lags);
```

Description

The `crosscoh` function computes the cross-coherence of two floating point inputs, `x[]` and `y[]`. The cross-coherence is the cross-correlation minus the product of the mean of `x` and the mean of `y`. The length of the input arrays is given by `samples`. This function returns a pointer to the output data array, `out[]`, of length `lags`.

Error Conditions

The `crosscoh` function does not return an error condition.

Example

```
#include <stats.h>
#define SAMPLES 1024

float excitation[SAMPLES], response[16], y[SAMPLES];
int lags = 16;

crosscoh (response, excitation, y, SAMPLES, lags);
```

See Also

[autocoh](#), [autocorr](#), [crosscorr](#)

CROSSCORR

cross-correlation

Synopsis

```
#include <stats.h>
float *crosscorr (float dm out[],
                 const float dm x[],
                 const float dm y[],
                 int samples,
                 int lags);
```

Description

The `crosscorr` function performs a cross-correlation between two signals. The cross-correlation is the sum of the scalar products of the signals in which the signals are displaced in time with respect to one another. The signals to be correlated are given by input `x[]` and `y[]` arrays. The length of the input arrays is given by `samples`. This function returns a pointer to the output data array, `out[]`, of length `lags`.

The `crosscorr` function is used in digital signal processing applications such as speech analysis.

Error Conditions

The `crosscorr` function does not return an error condition.

Example

```
#include <stats.h>
float r[10], s[160];
float p[160];

crosscorr (r, s, p, 160, 10);
```

See Also

[autocoh](#), [autocorr](#), [crosscorr](#)

cvecdot

complex vector dot product

Synopsis

```
#include <cvector.h>
complex_float cvecdot (const complex_float dm x_input[],
                      const complex_float dm y_input[],
                      int samples);
```

Description

The `cvecdot` function computes the complex dot product of the complex vectors, `x_input` and `y_input`, which are `samples` in size. The scalar result is returned by the function.

Error Conditions

The `cvecdot` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer, x[100], y[100];

answer = cvecdot (x, y, 100);
```

See Also

[vecdot](#)

cvecsadd

complex vector scalar addition

Synopsis

```
#include <cvector.h>
complex_float *cvecsadd (const complex_float dm input[],
                        complex_float scalar,
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecsadd` function computes the sum of each element of the complex vector, `input`, added to the complex scalar. Both the input and output vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecsadd` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y;

cvecsadd (x, y, answer, 50);
```

See Also

[vecsadd](#)

cvecsmlt

complex vector scalar multiply

Synopsis

```
#include <cvector.h>
complex_float *cvecsmlt (const complex_float dm input[],
                        complex_float scalar,
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecsmlt` function computes the product of each element of the complex vector, `input`, multiplied by the complex scalar. Both the input and output vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecsmlt` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y;

cvecsmlt (x, y, answer, 50);
```

See Also

[vecsmlt](#)

cvecssub

complex vector scalar subtraction

Synopsis

```
#include <cvector.h>
complex_float *cvecssub (const complex_float dm input[],
                        complex_float scalar,
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecssub` function computes the difference of each element of the complex vector, `input`, minus the complex scalar. Both the input and output vector are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecssub` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y

cvecssub (x, y, answer, 50);
```

See Also

[vecssub](#)

cvecvadd

complex vector addition

Synopsis

```
#include <cvector.h>
complex_float *cvecvadd (const complex_float dm x_input[],
                        const complex_float dm y_input[],
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecvadd` function computes the sum of each of the elements of the complex vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecvadd` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y[50];

cvecvadd (x, y, answer, 50);
```

See Also

[vecvadd](#)

cvecvmlt

complex vector multiply

Synopsis

```
#include <cvector.h>
complex_float *cvecvmlt (const_complex_float dm x_input[],
                        const_complex_float dm y_input[],
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecvmlt` function computes the product of each of the elements of the complex vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecvmlt` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y[50];

cvecvmlt (x, y, answer, 50);
```

See Also

[vecvmlt](#)

cvecvsub

complex vector subtraction

Synopsis

```
#include <cvector.h>
complex_float *cvecvsub (const complex_float dm x_input[],
                        const complex_float dm y_input[],
                        complex_float dm output[],
                        int samples);
```

Description

The `cvecvsub` function computes the difference of each of the elements of the complex vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `cvecvsub` function does not return an error condition.

Example

```
#include <cvector.h>
complex_float answer[50], x[50], y[50];

cvecvsub (x, y, answer, 50);
```

See Also

[vecvsub](#)

favg, favgf

return mean of two values

Synopsis

```
#include <math.h>
double favg (double x, double y);
float favgf (float x, float y);
```

Description

The `favg` and `favgf` functions return the mean of its two arguments. The `favgf` function is a built-in function which is implemented with an `Fn=(Fx+Fy)/2` instruction.

Error Conditions

The `favg` and `favgf` functions do not return an error code.

Example

```
#include <math.h>
float x;

x = favgf (10.0f, 8.0f); /* returns 9.0f */
```

See Also

[avg](#), [lavg](#)



The double precision function `favg` is only available under `-double-size-32` and actually calls the single precision function `favgf`.

fclip, fclipf

clip x by y

Synopsis

```
#include <math.h>
double fclip (double x, double y);
float fclipf (float x, float y);
```

Description

The `fclip` and `fclipf` functions return the first argument if it is less than the absolute value of the second argument, otherwise it returns the absolute value of the second argument if the first is positive, or minus the absolute value if the first argument is negative. The `fclipf` function is a built-in function which is implemented with an `Fn=CLIP Fx BY Fy` instruction.

Error Conditions

The `fclip` and `fclipf` functions do not return an error code.

Example

```
#include <math.h>
float y;

y = fclipf (5.1f, 8.0f); /* returns 5.1f */
```

See Also

[clip](#), [lclip](#)



The double precision function `fclip` is only available under `-double-size-32` and actually calls the single precision function `fclipf`.

fft_mag

compute FFT magnitude

Synopsis

```
#include <filter.h>
float fft_mag (const complex_float dm input[],
              float dm output[],
              int samples);
```

Description

The `fft_mag` function computes the normalized power spectrum from the complex results of an FFT operation. The complex input array is `samples` in size and the result is `samples/2` in size.

The result evaluates as:

$$\text{magnitude}(z) = \frac{((\text{Re}(z))^2 + (\text{Im}(z))^2)}{((\text{samples})/2)^2}$$

The `fft_mag` function returns a pointer to the output array.

Error Conditions

The `fft_mag` function does not return an error condition.

Example

```
#include <filter.h>
complex_float fft_data[1024];
float spectra[512];

fft_mag (fft_data, spectra, 1024);
```

See Also

[cfftN](#), [rfftN](#), [rfft2_N](#)

fir

finite impulse response (FIR) filter

Synopsis

```
#include <filter.h>
float *fir (const float dm input[],
           float dm output[],
           const float pm coeffs[],
           float dm state[],
           int samples,
           int taps);
```

Description

This function is optimized to take advantage of the SIMD execution model of the ADSP-2116x SHARC DSP.

The `fir` function implements a finite impulse response (FIR) filter defined by the coefficients and delay line that are supplied in the call of `fir`. The function produces the filtered response of its input data and stores the result in the vector `output`. This FIR filter is structured as a sum of products. The characteristics of the filter (passband, stop band, etc.) are dependent on the coefficient values and the number of taps supplied by the calling program.

The floating-point input array to the filter is `samples` in length. The integer `taps` indicates the length of the filter, which is also the length of the array `coeffs`. The `coeffs` array holds one FIR filter coefficient per element. The coefficients are stored in reverse order; for example, `a_coeffs[0]` holds the `taps - 1` (the last coefficient). The `coeffs` array must be located in program memory data space so that the single-cycle dual-memory fetch of the processor can be used.

The `state` array contains a pointer to the delay line as its first element, followed by the delay line values. The length of the `state` array is therefore 1 greater than the number of taps.

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Each filter has its own `state` array, which should not be modified by the calling program, only by the `fir` function. The state array should be initialized to zeros before the `fir` function is called for the first time. The function returns a pointer to the output vector.

Error Conditions

The `fir` function does not return an error condition.

Example

```
#include <filter.h>
float x[100], y[100];
float pm coeffs[10];      /* coeffs array must be */
                          /* initialized and in    */
                          /* PM memory           */

float state[11];
int i;

for (i = 0; i < 11; i++)
    state[i] = 0;        /* initialize state array*/

fir (x, y, coeffs, state, 100, 10);
/* y holds the filtered output */
```

See Also

[iir](#)

fmax, fmaxf

return larger of two values

Synopsis

```
#include <math.h>
double fmax (double x, double y);
float fmaxf (float x, float y);
```

Description

The `fmax` and `fmaxf` functions return the larger of its two arguments. The `fmaxf` function is a built-in function which is implemented with an `Fn=MAX(Fx,Fy)` instruction.

Error Conditions

The `fmax` and `fmaxf` functions do not return an error code.

Example

```
#include <math.h>
float y;

y = fmaxf (5.1f, 8.0f); /* returns 8.0f */
```

See Also

[fmin](#), [fminf](#), [lmax](#), [lmin](#), [max](#), [min](#)



The double precision function `fmax` is only available under `-double-size-32` and actually calls the single precision function `fmaxf`.

fmin, fminf

return smaller of two values

Synopsis

```
#include <math.h>
double fmin (double x, double y);
float fminf (float x, float y);
```

Description

The `fmin` and `fminf` functions return the smaller of their two arguments. The `fminf` function is a built-in function which is implemented with an `Fn=MIN(Fx,Fy)` instruction.

Error Conditions

The `fmin` and `fminf` functions do not return an error code.

Example

```
#include <math.h>
float y;

y = fminf (5.1f, 8.0f); /* returns 5.1f */
```

See Also

[fmax](#), [fmaxf](#), [lmax](#), [lmin](#), [max](#), [min](#)



The double precision function `fmin` is only available under `-double-size-32` and actually calls the single precision function `fminf`.

gen_bartlett

generate bartlett window

Synopsis

```

#include <window.h>
void gen_bartlett(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );

```

Description

This function generates a vector containing the Bartlett window. The length is specified by parameter *N*. Note that this window is similar to the Triangle window but has the following properties that differ from the Triangle window:

- The Bartlett window always returns a window with two zeros on either end of the sequence. Therefore, for odd *n*, the center section of a *N*+2 Bartlett window equals an *N* Triangle window.
- For even *n*, the Bartlett window is the convolution of two rectangular sequences. There is no standard definition for the Triangle window for even *n*; the slopes of the Triangle window are slightly steeper than those of the Bartlett window.

The algorithm used is

$$w[n] = 1 - \left| \frac{n - \frac{N-1}{2}}{\frac{N-1}{2}} \right|$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_bartlett` function does not return an error condition.

See Also

[gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#), [gen_harris](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_blackman

generate blackman window

Synopsis

```
#include <window.h>
void gen_blackman(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );
```

Description

This function generates a vector containing the Blackman window. The length is specified by parameter N .

The algorithm used is

$$w[n] = 0.42 - 0.5 \cos\left(\frac{2\pi n}{N-1}\right) + 0.08 \cos\left(\frac{4\pi n}{N-1}\right)$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_blackman` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#), [gen_harris](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_gaussian

generate gaussian window

Synopsis

```
#include <window.h>
void gen_gaussian(
float w[], /* Window vector */
float alpha, /* Gaussian alpha parameter */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );
```

Description

This function generates a vector containing the Gaussian window. The length is specified by parameter N.

The algorithm used is

$$w(n) = \exp\left[-\frac{1}{2}\left(\alpha \frac{n - N/2 - 1/2}{N/2}\right)^2\right]$$

where $n = \{0, 1, 2, \dots, N-1\}$ and a is an input parameter

The domain supported by the function is $a > 0$; $N > 0$; $\alpha > 0.0$

Error Conditions

The `gen_gaussian` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_hamming](#), [gen_hanning](#), [gen_harris](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_hamming

generate hamming window

Synopsis

```

#include <window.h>
void gen_hamming(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );

```

Description

This function generates a vector containing the Hamming window. The length is specified by parameter *N*.

The algorithm used is

$$w[n] = 0.54 - 0.46 \cos\left(\frac{2\pi n}{N-1}\right)$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_hamming` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hanning](#), [gen_harris](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_hanning

generate hanning window

Synopsis

```
#include <window.h>
void gen_hanning(w,a,N)
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */);
```

Description

This function generates a vector containing the Hanning window. The length is specified by parameter N . This window is also known as the Cosine window.

The algorithm used is

$$w[n] = 0.5 - 0.5 \cos\left(\frac{2\pi n}{N+1}\right)$$

where $n = \{1, 1, 2, \dots, N+1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_hanning` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_harris](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_harris

generate harris window

Synopsis

```

#include <window.h>
void gen_harris(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );

```

Description

This function generates a vector containing the Harris window. The length is specified by parameter N . This window is also known as the Blackman-Harris window.

The algorithm used is

$$w[n] = 0.35875 - 0.48829 * \cos\left(\frac{2\pi n}{N-1}\right) + 0.14128 * \cos\left(\frac{4\pi n}{N-1}\right) + 0.01168 * \cos\left(\frac{6\pi n}{N-1}\right)$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_harris` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#),
[gen_kaiser](#), [gen_rectangular](#), [gen_triangle](#)

gen_kaiser

generate kaiser window

Synopsis

```
#include <window.h>
void gen_kaiser(
float w[], /* Window vector */
float beta, /* Kaiser beta parameter */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */);
```

Description

This function generates a vector containing the Kaiser window. The length is specified by parameter N . The β value is specified by parameter b .

The algorithm used is

$$w[n] = \frac{I_0 \left[\beta \left(1 - \left[\frac{n - \alpha}{\alpha} \right]^2 \right)^{1/2} \right]}{I_0(\beta)}$$

where $n = \{0, 1, 2, \dots, N-1\}$, $a = (N - 1) / 2$, and $I_0(\beta)$ represents the zeroth-order modified Bessel function of the first kind.

The domain supported by the function is $a > 0$; $N > 0$;

Error Conditions

The `gen_kaiser` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#),
[gen_harris](#), [gen_rectangular](#), [gen_triangle](#)

gen_rectangular

generate rectangular window

Synopsis

```

#include <window.h>
void gen_rectangular(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );

```

Description

This function generates a vector containing the Rectangular window. The length is specified by parameter *N*.

The algorithm used is

$$w[n] = 1$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_rectangular` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#),
[gen_harris](#), [gen_kaiser](#), [gen_triangle](#)

gen_triangle

generate triangle window

Synopsis

```
#include <window.h>
void gen_triangle(
float w[], /* Window vector */
int a, /* Address stride in samples for window vector */
int N /* Length of window vector */ );
```

Description

This function generates a vector containing the Triangle window. The length is specified by parameter N . See [“gen_bartlett” on page 5-43](#) for information regarding the relationship between the Bartlett window and the Triangle window.

For even n , the following equation applies:

$$w[n] = \begin{cases} \frac{2n+1}{N} & n < N/2 \\ \frac{2N-2n-1}{N} & n > N/2 \end{cases}$$

where $n = \{0, 1, 2, \dots, N-1\}$

For odd n , the following equation applies:

$$w[n] = \begin{cases} \frac{2n+2}{N+1} & n < N/2 \\ \frac{2N-2n}{N+1} & n > N/2 \end{cases}$$

where $n = \{0, 1, 2, \dots, N-1\}$

The domain supported by the function is $a > 0$; $N > 0$

Error Conditions

The `gen_triangle` function does not return an error condition.

See Also

[gen_bartlett](#), [gen_blackman](#), [gen_gaussian](#), [gen_hamming](#), [gen_hanning](#),
[gen_harris](#), [gen_kaiser](#), [gen_rectangular](#)

histogram

histogram

Synopsis

```
#include <stats.h>
int *histogram(int dm out[],
               const int dm in[],
               int out_len,
               int samples,
               int bin_size);
```

Description

The `histogram` function computes a scaled-integer histogram of its input array. The `bin_size` parameter is used to adjust the width of each individual bin in the output array. For example, a `bin_size` of 5 indicates that the first location of the output array holds the number of occurrences of a 0, 1, 2, 3, or 4.

The output array is first zeroed by the function, then each sample in the input array is multiplied by $1/\text{bin_size}$ and truncated. The appropriate bin in the output array is incremented. This function returns a pointer to the output array.

All values within the input array must be within range. In order to achieve maximum performance, out of bounds checking is not performed by this function.

Error Conditions

The `histogram` function does not return an error condition.

Example

```
#include <stats.h>
#define SAMPLES 1024

int length = 2048;
int excitation[SAMPLES], response[2048];

histogram (response, excitation, length, SAMPLES, 5);
```

See Also

[mean](#), [var](#)

idle

execute ADSP-21xxx `idle` instruction

Synopsis

```
#include <21160.h>
void idle (void);
```

Description

The `idle` function invokes the ADSP-21xxx `idle` instruction once and returns. The `idle` instruction causes the processor to stop and respond only to interrupts. For a complete description of the `idle` instruction, please refer to the *ADSP-21160 SHARC DSP Hardware Reference*.

-  In previous releases of the VisualDSP++ software (prior to release 2.1), the `idle` function repeatedly executed the `idle` instruction. This function has been changed to give you more control over the amount of time spent in the `idle` state.

Error Conditions

The `idle` function does not return an error condition.

Example

```
#include <21160.h>
idle ();
```

See Also

[interrupt](#), [interruptf](#), [interrupts](#), [interuptcb](#), [signal](#)

ifftN

N-point inverse complex input fast Fourier transform (IFFT)

Synopsis

```
#include <filter.h>
complex_float *ifft65536 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *ifft32768 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *ifft16384 (complex_float dm input[],
                          complex_float dm output[]);

complex_float *ifft8192 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *ifft4096 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *ifft2048 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *ifft1024 (complex_float dm input[],
                        complex_float dm output[]);

complex_float *ifft512 (complex_float dm input[],
                       complex_float dm output[]);

complex_float *ifft256 (complex_float dm input[],
                       complex_float dm output[]);

complex_float *ifft128 (complex_float input[],
                       complex_float output[]);

complex_float *ifft64 (complex_float dm input[],
                      complex_float dm output[]);

complex_float *ifft32 (complex_float dm input[],
                      complex_float dm output[]);
```

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```
complex_float *ifft16 (complex_float dm input[],
                      complex_float dm output[]);

complex_float *ifft8 (complex_float dm input[],
                     complex_float dm output[]);
```

Description

These functions are optimized to take advantage of the SIMD execution model of the ADSP-2116x SHARC DSP. They require complex arguments to ensure that the real and imaginary parts are interleaved in memory and are therefore accessible in a single cycle using the wider data bus of the processor.

Each of these 14 `ifftN` functions computes the N-point radix-2 inverse fast Fourier transform (IFFT) of its floating-point input (where N is 8, 16, 32, 64, 128, 256, 512, 1024, 2048, 4096, 8192, 16384, 32768 or 65536).

There are 14 distinct functions in this set. They all perform the same function with the same type and number of arguments. The only difference between them is the size of the arrays on which they operate. To call a particular function, substitute the number of points for N; for example,

```
ifft8 (input, output);
```

The input to `ifftN` is a floating-point array of N points. If there are fewer than N actual data points, you must pad the array with zeros to make N samples. Better results occur with less zero padding, however. The input data should be windowed (if necessary) before calling the function because no preprocessing is performed on the data. The functions return a pointer to the `output` array.

Error Conditions

The `ifftN` functions do not return error conditions.

Example

```
#include <filter.h>
#define N 2048
```

```
complex_float input[N], output[N];

/* input array is filled from a previous xfft2048 () or other
source */

ifft2048 (input, output);
/* Arrays are filled with FFT data */
```

See Also

[cfftN](#), [rfftN](#), [rfft2_N](#)



The `ifftN` functions use the input array as an intermediate workspace. If the input data is to be preserved it must first be copied to a safe location before calling these functions.

iir

infinite impulse response (IIR) filter

Synopsis

```
#include <filter.h>
float *iir (const float dm input[],
           float dm output[],
           const float pm coeffs[],
           float dm state[],
           int samples,
           int sections);
```

Description

The `iir` function implements an infinite impulse response (IIR) filter defined by the coefficients and delay line that are supplied in the call of `iir`. The function produces the filtered response of its input data and stores the result in the output vector. The IIR filter is implemented as a cascaded biquad. The characteristics of the filter are dependent on the coefficient values supplied by the calling program.

The floating-point inputs to the filter are contained in the array `input`, which has a size of `samples`.

The parameter `sections` specifies the number of biquad sections.

The `coeffs` array must be 4 times the number of sections in length and it also must be located in program memory. The definition is:

```
float pm coeffs[4*sections];
```

`coeffs` must be ordered in the following form:

```
coeffs[] = [a2 stage 1, a1 stage 1, b2 stage 1, b1 stage 1, a2
stage 2, ... ]
```

The `state` array holds 2 delay elements per section. It also has one extra location that holds an internal pointer. The total length must be $2 * \text{sections} + 1$. The definition is:

```
float state[2*sections + 1];
```

The user should not access the `state` array, except that it should be initialized to all zeros before the first call to `iir`. The first location of `state` is an address. Setting the address to zero tells the function that it is being called for the first time.

The algorithm used is adapted from Oppenheim and Schaffer, *Digital Signal Processing*, New Jersey: Prentice Hall, 1975.

The function returns a pointer to the `output` vector.

Error Conditions

The `iir` function does not return an error condition.

Example

```
#include <filter.h>
#define SECTIONS 4

float input[100], output[100], state[2*SECTIONS + 1];
float pm coeffs[SECTIONS * 4];
int    i;

for (i = 0; i < SECTIONS + 1; i++)
    state[i] = 0;    /* initialize state array */

iir (input, output, coeffs, state, 100, SECTIONS);
```

See Also

[fir](#)

matadd

matrix addition

Synopsis

```
#include <matrix.h>
float *matadd (float output[][],
              const float x_input[][],
              const float y_input[][],
              int r,
              int s);
```

Description

The `matadd` function performs a matrix addition of the input matrices `x_input[][]` and `y_input[][]`, returning the result in `output[][]`. The `matadd` function returns a pointer to the output matrix. The dimensions of these matrices are `x_input[r][s]`, `y_input[r][s]`, and `output[r][s]`.

Error Conditions

The `matadd` function does not return an error condition.

Example

```
#include <matrix.h>
float x[10][20], y[10][20];
float z[10][20];

matadd (z, x, y, 10, 20);
```

See Also

[matsub](#)

matmul

matrix multiplication

Synopsis

```
#include <matrix.h>
float *matmul (float output[][[]],
              const float x_input[][[]],
              const float y_input[][[]],
              int r,
              int s,
              int t);
```

Description

The `matmul` function performs a matrix multiplication of the input matrices `x_input[][[]]` and `y_input[][[]]`, returning a pointer to the `output[][[]]` matrix. The dimensions of these matrices are `x_input[r][s]`, `y_input[s][t]` and `output[r][t]`.

Error Conditions

The `matmul` function does not return an error condition.

Example

```
#include <matrix.h>

float x[10][5], y[5][10];
float z[10][10];

matmul (z, x, y, 10, 5, 10);
```

See Also

[matscalmult](#)

matscalsmult

multiply matrix by scalar

Synopsis

```
#include <matrix.h>
float *matscalsmult (float output[][[]],
                    const float input[][[]],
                    float scalar,
                    int r,
                    int s);
```

Description

The `matscalsmult` function performs a scaled multiplication of the `input[][[]]` matrix, returning the result in `output[][[]]`. The `input[][[]]` matrix is multiplied by the input value `scalar`. The `matscalsmult` function returns a pointer to the output matrix. The dimensions of these matrices are `input[r][s]`, and `output[r][s]`.

Error Conditions

The `matscalsmult` function does not return an error condition.

Example

```
#include <matrix.h>
float x[10][5], z[10][5];

matscalsmult (z, x, 0.5, 10, 5);
/* multiplies the matrix x by 0.5 */
```

See Also

[matmul](#)

matsub

matrix subtraction

Synopsis

```
#include <matrix.h>
float *matsub (float output[][[]],
              const float x_input[][[]],
              const float y_input[][[]],
              int r,
              int s);
```

Description

The `matsub` function subtracts the elements of the input matrix `y_input[][[]]` from the input matrix `x_input[][[]]`, returning the result in `output[][[]]`. The `matsub` function returns a pointer to the output matrix. The dimensions of these matrices are `x_input[r][s]`, `y_input[r][s]`, and `output[r][s]`.

Error Conditions

The `matsub` function does not return an error condition.

Example

```
#include <matrix.h>
float x[10][5], y[10][5];
float z[10][5];

matsub (z, x, y, 10, 5);
```

See Also

[matadd](#)

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mean

mean of an array of floating point numbers

Synopsis

```
#include <stats.h>
float mean (const float dm input[],
           int samples);
```

Description

The `mean` function returns the mean of its floating point input array.

Error Conditions

The `mean` function does not return an error condition.

Example

```
#include <stats.h>
float result, input[256];

result = mean (input, 256);
```

See Also

[var](#)

mu_compress

μ -law compression

Synopsis

```
#include <filter.h>
int *mu_compress (const int dm input[],
                  int dm output[],
                  int samples);
```

Description

The `mu_compress` function takes an array of linear 14-bit signed speech samples and compresses them according to CCITT recommendation G.711. The output array returned contains 8-bit samples that can be sent directly to a μ -law codec.

This function uses serial port 0 to perform companding. Therefore, serial port 0 must not be in use when this routine is called.

The function returns a pointer to the compressed data.

Error Conditions

The `mu_compress` function does not return an error condition.

Example

```
#include <filter.h>
int linear[100], compressed[100];

mu_compress (linear, compressed, 100);
```

See Also

[mu_expand](#), [a_compress](#)

mu_expand

μ -law expansion

Synopsis

```
#include <filter.h>
int *mu_expand (const int dm input[],
               int dm output[],
               int samples);
```

Description

The `mu_expand` function takes an array of 8-bit compressed speech samples and expands them according to CCITT recommendation G.711 (μ -law definition). The output returned is an array of linear 14-bit signed samples. The function returns a pointer to the output array.

This function uses serial port 0 to perform companding. Therefore, serial port 0 must not be in use when this routine is called.

The function returns a pointer to the expanded data.

Error Conditions

The `mu_expand` function does not return an error condition.

Example

```
#include <filter.h>
int compressed_data[100], expanded_data[100];

mu_expand (compressed_data, expanded_data, 100);
```

See Also

[mu_compress](#), [a_expand](#)

poll_flag_in

test input flag

Synopsis

```
#include <21160.h>
int poll_flag_in (int flag, int mode);
```

Description

The `poll_flag_in` function tests the specified flag (0, 1, 2, 3) for the specified transition (0=low to high, 1=high to low, 2=flag high, 3=flag low, 4=any transition, 5=read flag). The function returns a zero *after* the specified transition has occurred in modes 0-3. In mode 4 it returns the state of the flag after the transition. In mode 5 it returns the value of the flag without waiting.

Table 5-4. `poll_flag_in` Macros and Values

Flag Macro	Value	Mode Macro	Value
READ_FLAG0	0	FLAG_IN_LO_TO_HI	0
READ_FLAG1	1	FLAG_IN_HI_TO_LOW	1
READ_FLAG2	2	FLAG_IN_HI	2
READ_FLAG3	3	FLAG_IN_LOW	3
READ_FLAG3	3	FLAG_IN_TRANSITION	4
READ_FLAG3	3	RETURN_FLAG_STATE	5

This function assumes that the flag direction in the `MODE2` register is already set as an input (the default state at reset).

Error Conditions

The `poll_flag_in` function returns a negative value for an invalid flag or transition mode.

Example

```
#include <21160.h>

poll_flag_in (0, 3);
    /* return zero after transition has occurred */
```

See Also

[interrupt](#), [interruptf](#), [interrupts](#), [interuptcb](#), [set_flag](#)

rfftN

N-point real input fast Fourier transform

Synopsis

```
#include <filter.h>
complex_float *rfft65536 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft32768 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft16384 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft8192 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft4096 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft2048 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft1024 (float dm input[],
                          complex_float dm output[]);

complex_float *rfft256 (float dm input[],
                        complex_float dm output[]);

complex_float *rfft128 (float dm input[],
                        complex_float dm output[]);

complex_float *rfft64 (float dm input[],
                       complex_float dm output[]);

complex_float *rfft32 (float dm input[],
                       complex_float dm output[]);

complex_float *rfft16 (float dm input[],
                       complex_float dm output[]);
```

Description

These functions are optimized to take advantage of the SIMD execution model of the ADSP-2116x SHARC DSP, and requires complex output results to ensure that the real and imaginary parts are interleaved in memory and are therefore accessible in a single cycle using the wider data bus of the processor.

Each of these `rfftN` functions are similar to the `cfftN` functions except that they only take real inputs. They compute the N-point radix-2 fast Fourier transform (RFFT) of their floating point input (where N is 16, 32, 64, 128, 256, 512, 1024, 2048, 4096, 8192, 16384, 32768 or 65536).

There are thirteen distinct functions in this set. They all perform the same function with the same type and number of arguments. The only difference between them is the size of the arrays on which they operate.

Call a particular function by substituting the number of points for N, as in the following example:

```
rfft16 (input, output);
```

The input to `rfftN` is a floating-point array of N points. If there are fewer than N actual data points, you must pad the array with zeros to make N samples. However, better results occur with less zero padding. The input data should be windowed (if necessary) before calling the function because no preprocessing is performed on the data.

The `rfftN` functions return a pointer to the output array.

Error Conditions

The `rfftN` functions do not return any error conditions.

Example

```
#include <filter.h>
#define N 2048

float input[N];
complex_float output[N];

/* Real input array fills from a converter or other source */
rfft2048 (input, output);
/* Arrays are filled with FFT data */
```

See Also

[cfftN](#), [rfftN](#), [rfft2_N](#)



The `rfftN` functions use the input array as an intermediate workspace. If the input data is to be preserved it must first be copied to a safe location before calling these functions.

rfft2_N

Dual N-point real input fast Fourier transform

Synopsis

```
#include <filter.h>
complex_float *rfft2_32768 (float dm input[],
                           complex_float dm x_output[],
                           complex_float dm y_output[]);

complex_float *rfft2_16384 (float dm input[],
                           complex_float dm x_output[],
                           complex_float dm y_output[]);

complex_float *rfft2_8192 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_4096 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_2048 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_1024 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_512 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_256 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);

complex_float *rfft2_128 (float dm input[],
                          complex_float dm x_output[],
                          complex_float dm y_output[]);
```

```
complex_float *rfft2_64 (float dm input[],
                        complex_float dm x_output[],
                        complex_float dm y_output[]);

complex_float *rfft2_32 (float dm input[],
                        complex_float dm x_output[],
                        complex_float dm y_output[]);

complex_float *rfft2_16 (float dm input[],
                        complex_float dm x_output[],
                        complex_float dm y_output[]);
```

Description

These functions are optimized to take advantage of the SIMD execution model of the ADSP-2116x SHARC DSP. They require complex arguments to ensure that the real and imaginary parts are interleaved in memory and are therefore accessible in a single cycle using the wider data bus of the processor.

Each of these `rfft2_N` functions are similar to the `rfftN` functions except that they take an interleaved real input and return two complex FFT results. The `x_output` vector represents the odd input values, and the `y_output` vector represents the even input values. Therefore the input vector must be twice the length of N.

The `rfft2_N` functions compute the N-point radix-2 fast Fourier transform (RFFT) of their floating point input (where N is 16, 32, 64, 128, 256, 512, 1024, 2048, 4096, 8192, 16384, or 32768).

There are twelve distinct functions in this set. They all perform the same function with the same type and number of arguments. The only difference between them is the size of the arrays on which they operate.

Call a particular function by substituting the number of points for N, as in the following example:

```
rfft2_16 (input, x_output, y_output);
```

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The input to `rfft2_N` is a floating-point array of $2*N$ points. This data can be either two interleaved input data sets or one contiguous set, depending on the processing strategy to be employed. If there are fewer than $2*N$ actual data points, you must pad the array with zeros to make $2*N$ samples. However, better results occur with less zero padding.

The input data should be windowed (if necessary) before calling the function because no preprocessing is performed on the data. Zero padding and windowing must be applied with the interleaved characteristic of the data set in mind.

The `rfft2_N` functions return a pointer to the array `x_output`.

Error Conditions

The `rfft2_N` functions do not return any error conditions.

Example

```
#include <filter.h>
#define N 2048

float input[2 * N];
complex_float x_output[N];
complex_float y_output[N];

/* Real input array fills from a converter or other source */

rfft2_2048 (input, x_output, y_output);
/* Arrays are filled with FFT data */
```

See Also

[cfftN](#), [ifftN](#), [rfftN](#)



The `rfft2_N` functions use the input array as an intermediate workspace. If the input data is to be preserved it must first be copied to a safe location before calling these functions.

rms

root mean square

Synopsis

```
#include <stats.h>
float rms (const float dm input[],
          int length);
```

Description

The `rms` function returns the square root of the mean of the square of its floating-point input array.

Error Conditions

The `rms` function does not return an error condition.

Example

```
#include <stats.h>
float input[256], results;

results = rms (input, 256);
```

See Also

[mean](#), [var](#)

rsqrt, rsqrtf

reciprocal square root

Synopsis

```
#include <math.h>
double rsqrt (double x);
float rsqrtf (float x);
```

Description

The `rsqrt` and `rsqrtf` functions return the reciprocal positive square root of their argument.

The `rsqrt` and `rsqrtf` functions return a value that is accurate to 20 bits of the mantissa. This accuracy corresponds to a maximum relative error of 2^{-20} over its input range.

Error Conditions

The `rsqrt` and `rsqrtf` functions return zero for a negative input.

Example

```
#include <math.h>
double y;

y = rsqrt (2.0);    /* y = 0.707 */
```

See Also

[sqrt](#), [sqrtf](#)

set_flag

set ADSP-21xxx flags

Synopsis

```
#include <21160.h>
int set_flag (int flag, int mode);
```

Description

This function is used to set the ADSP-21xxx flags to the desired output value.

The function accepts as input a flag number [0-3] and a mode. The mode can be specified as a macro (defined in 21160.h) or a value [0-3].

Table 5-5. Flag Function Macros and Values

Flag Macro	Value	Mode Macro	Value
SET_FLAG0	0	SET_FLAG	0
SET_FLAG1	1	CLR_FLAG	1
SET_FLAG2	2	TGL_FLAG	2
SET_FLAG3	3	TST_FLAG	3

In addition to setting the flag to the specified value, the function also sets the MODE2 register to specify that the flag is used for output, not input.

If the TST_FLAG macro (or a 3) is specified as the mode, the current value (0 or 1) of the flag is returned as the result of the function.

The set_flag function returns a zero upon success (except as noted in the previous paragraph).

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Error Conditions

The `set_flag` function returns a non-zero for an error.

Example

```
#include <21160.h>

set_flag (SET_FLAG0, CLR_FLAG);
set_flag (SET_FLAG0, SET_FLAG);
```

See Also

[poll_flag_in](#)

set_semaphore

set bus lock semaphore

Synopsis

```
#include <21160.h>
int set_semaphore (void dm *semaphore,
                  int set_value,
                  int timeout);
```

Description

The `set_semaphore` function is used to control bus lock in multiprocessor ADSP-21xxx systems.

A -1 is returned if the bus is locked and the bus lock timeout is exceeded.

A 0 is returned if the bus is not locked and a semaphore is set.

Error Conditions

The `set_semaphore` function does not return an error condition.

See Also

No references to this function.

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timer_off

disable ADSP-21xxx timer

Synopsis

```
#include <21160.h>
unsigned int timer_off (void);
```

Description

The `timer_off` function disables the ADSP-21xxx timer and returns the current value of the `TCOUNT` register.

Error Conditions

The `timer_off` function does not return an error condition.

Example

```
#include <21160.h>
unsigned int hold_tcount;

hold_tcount = timer_off ();
/* hold_tcount contains value of TCOUNT */
/* register AFTER timer has stopped */
```

See Also

[timer_on](#), [timer_set](#)



The function is supplied only as an in-lined procedure; that is, the compiler substitutes the appropriate statements for any reference to the procedure. Therefore, any source that references `timer_off` must include the `21160.h` header file.

timer_on

enable ADSP-21xxx timer

Synopsis

```
#include <21160.h>
unsigned int timer_on (void);
```

Description

The `timer_on` function enables the ADSP-21xxx timer and returns the current value of the TCOUNT register.

Error Conditions

The `timer_on` function does not return an error condition.

Example

```
#include <21160.h>
unsigned int hold_tcount;

hold_tcount = timer_on ();
/* hold_tcount contains value of TCOUNT */
/* register when timer starts */
```

See Also

[timer_off](#), [timer_set](#)



The function is supplied only as an in-lined procedure; that is, the compiler substitutes the appropriate statements for any reference to the procedure. Therefore, any source that references `timer_on` must include the `21160.h` header file.

timer_set

initialize ADSP-21xxx timer

Synopsis

```
#include <21160.h>
int timer_set (unsigned int tperiod,
               unsigned int tcount);
```

Description

The `timer_set` function sets the ADSP-21xxx timer registers `TPERIOD` and `TCOUNT`. The function returns a 1 if the timer is enabled, or a zero if the timer is disabled.

Note: Each interrupt call takes approximately 50 cycles on entrance and 50 cycles on return. If `TPERIOD` and `TCOUNT` are set too low, you may incur an initializing overhead that could create an infinite loop.

Error Conditions

The `timer_set` function does not return an error condition.

Example

```
#include <21160.h>
if (timer_set (1000, 1000) != 1)
    timer_on ();    /* enable timer */
```

See Also

[timer_on](#), [timer_off](#)



The function is supplied only as an in-lined procedure; that is, the compiler substitutes the appropriate statements for any reference to the procedure. Therefore, any source that references `timer_set` must include the `21160.h` header file.

var

variance

Synopsis

```
#include <stats.h>
float var (const float dm input[],
          int length);
```

Description

The `var` function returns the variance of its floating-point input array.

Error Conditions

The `var` function does not return an error condition.

Example

```
#include <stats.h>
float input[256], result;

result = var (input, 256);
```

See Also

[mean](#)

vecdot

vector dot product

Synopsis

```
#include <vector.h>
float vecdot (const float dm x_input[],
              const float dm y_input[],
              int samples);
```

Description

The `vecdot` function computes the dot product of the vectors, `x_input` and `y_input`, which are `samples` in size. The scalar result is returned by the function.

Error Conditions

The `vecdot` function does not return an error condition.

Example

```
#include <vector.h>
float answer, x[100], y[100];

answer = vecdot (x, y, 100);
```

See Also

[cvecdot](#)

vecsadd

vector scalar addition

Synopsis

```
#include <vector.h>
float *vecsadd (const float dm input[],
               float scalar,
               float dm output[],
               int samples);
```

Description

The `vecsadd` function computes the sum of each element of the vector, `input`, added to the scalar. Both the input and output vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecsadd` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y;

vecsadd (x, y, answer, 50);
```

See Also

[cvecsadd](#)

vecsm1t

vector scalar multiply

Synopsis

```
#include <vector.h>
float *vecsm1t (const float dm input[],
               float scalar,
               float dm output[],
               int samples);
```

Description

The `vecsm1t` function computes the product of each element of the vector, `input`, multiplied by the scalar. Both the input and output vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecsm1t` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y;

vecsm1t (x, y, answer, 50);
```

See Also

[cvecsm1t](#)

vecssub

vector scalar subtraction

Synopsis

```
#include <vector.h>
float *vecssub (const float dm input[],
               float scalar,
               float dm output[],
               int samples);
```

Description

The `vecssub` function computes the difference of each element of the vector, `input`, minus the scalar. Both the input and output vector are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecssub` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y;

vecssub (x, y, answer, 50);
```

See Also

[cvecssub](#)

vecvadd

vector addition

Synopsis

```
#include <vector.h>
float *vecvadd (const float dm x_input[],
               const float dm y_input[],
               float dm output[],
               int samples);
```

Description

The `vecvadd` function computes the sum of each of the elements of the vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecvadd` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y[50];

vecvadd (x, y, answer, 50);
```

See Also

[cvecvadd](#)

vecvmlt

vector multiply

Synopsis

```
#include <vector.h>
float *vecvmlt (const float dm x_input[],
               const float dm y_input[],
               float dm output[],
               int samples);
```

Description

The `vecvmlt` function computes the product of each of the elements of the vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecvmlt` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y[50];

vecvmlt (x, y, answer, 50);
```

See Also

[cvecvmlt](#)

vecvsub

vector subtraction

Synopsis

```
#include <vector.h>
float *vecvsub (const float dm x_input[],
               const float dm y_input[],
               float dm output[],
               int samples);
```

Description

The `vecvsub` function computes the difference of each of the elements of the vectors, `x_input` and `y_input`, and places the results in `output`. All three vectors are `samples` in length. The function returns a pointer to the output vector.

Error Conditions

The `vecvsub` function does not return an error condition.

Example

```
#include <vector.h>
float answer[50], x[50], y[50];

vecvsub (x, y, answer, 50);
```

See Also

[cvecvsub](#)

zero_cross

count zero crossings

Synopsis

```
#include <stats.h>
int zero_cross (const float dm in[],
               int length);
```

Description

The `zero_cross` function returns the number of times that a signal represented in the input array crosses over the zero line. If all input values are zero, the function returns a zero.

Error Conditions

The `zero_cross` function does not return an error condition.

Example

```
#include <stats.h>
float input[256];
int results;

results = zero_cross (input, 256);
```

See Also

No references to this function.

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