

cpsd

Cross power spectral density

Syntax

```
Pxy = cpsd(x,y)
Pxy = cpsd(x,y,window)
Pxy = cpsd(x,y,window,noverlap)
[Pxy,W] = cpsd(x,y,window,noverlap,nfft)
[Pxy,F] = cpsd(x,y,window,noverlap,nfft,fs)
[...] = cpsd(...,'twosided')
cpsd(...)
```

Description

$P_{xy} = \text{cpsd}(x,y)$ estimates the cross power spectral density P_{xy} of the discrete-time signals x and y using the Welch's averaged, modified periodogram method of spectral estimation. The cross power spectral density is the distribution of power per unit frequency and is defined as

$$P_{xy}(\omega) = \sum_{m=-\infty}^{\infty} R_{xy}(m) e^{-j\omega m}$$

The cross-correlation sequence is defined as

$$R_{xy}(m) = E\{x_{n+m}y_n^*\} = E\{x_n y_{n-m}^*\}$$

where x_n and y_n are jointly stationary random processes, $-\infty < n < \infty$, and $E\{\cdot\}$ is the expected value operator.

For real x and y , `cpsd` returns a one-sided CPSD and for complex x or y , it returns a two-sided CPSD.

`cpsd` uses the following default values:

Parameter	Description	Default Value
nfft	<p>FFT length which determines the frequencies at which the PSD is estimated</p> <p>For real x and y, the length of P_{xy} is $(\text{nfft}/2+1)$ if nfft is even or $(\text{nfft}+1)/2$ if nfft is odd. For complex x or y, the length of P_{xy} is nfft.</p> <p>If nfft is greater than the signal length, the data is zero-padded. If nfft is less than the signal length, the segment is wrapped using <code>datawrap</code> so that the length is equal to nfft.</p>	Maximum of 256 or the next power of 2 greater than the length of each section of x or y

fs	Sampling frequency	1
window	Windowing function and number of samples to use for each section	Periodic Hamming window of length to obtain eight equal sections of x and y
noverlap	Number of samples by which the sections overlap	Value to obtain 50% overlap

Note You can use the empty matrix `[]` to specify the default value for any input argument except x or y . For example, `Pxy = cpsd(x, y, [], [], 128)` uses a Hamming window, default `noverlap` to obtain 50% overlap, and the specified 128 `nfft`.

`Pxy = cpsd(x, y, window)` specifies a windowing function, divides x and y into overlapping sections of the specified window length, and windows each section using the specified window function. If you supply a scalar for `window`, `Pxy` uses a Hamming window of that length. The x and y vectors are divided into eight equal sections of that length. If the signal cannot be sectioned evenly with 50% overlap, it is truncated.

`Pxy = cpsd(x, y, window, noverlap)` overlaps the sections of x by `noverlap` samples. `noverlap` must be an integer smaller than the length of `window`.

`[Pxy, W] = cpsd(x, y, window, noverlap, nfft)` uses the specified FFT length `nfft` in estimating the CPSD. It also returns `W`, which is the vector of normalized frequencies (in rad/sample) at which the CPSD is estimated. For real signals, the range of `W` is $[0, \pi]$ when `nfft` is even and $[0, \pi)$ when `nfft` is odd. For complex signals, the range of `W` is $[0, 2\pi)$.

`[Pxy, F] = cpsd(x, y, window, noverlap, nfft, fs)` returns `Pxy` as a function of frequency and a vector `F` of frequencies at which the CPSD is estimated. `fs` is the sampling frequency in Hz. For real signals, the range of `F` is $[0, fs/2]$ when `nfft` is even and $[0, fs/2)$ when `nfft` is odd. For complex signals, the range of `F` is $[0, fs)$.

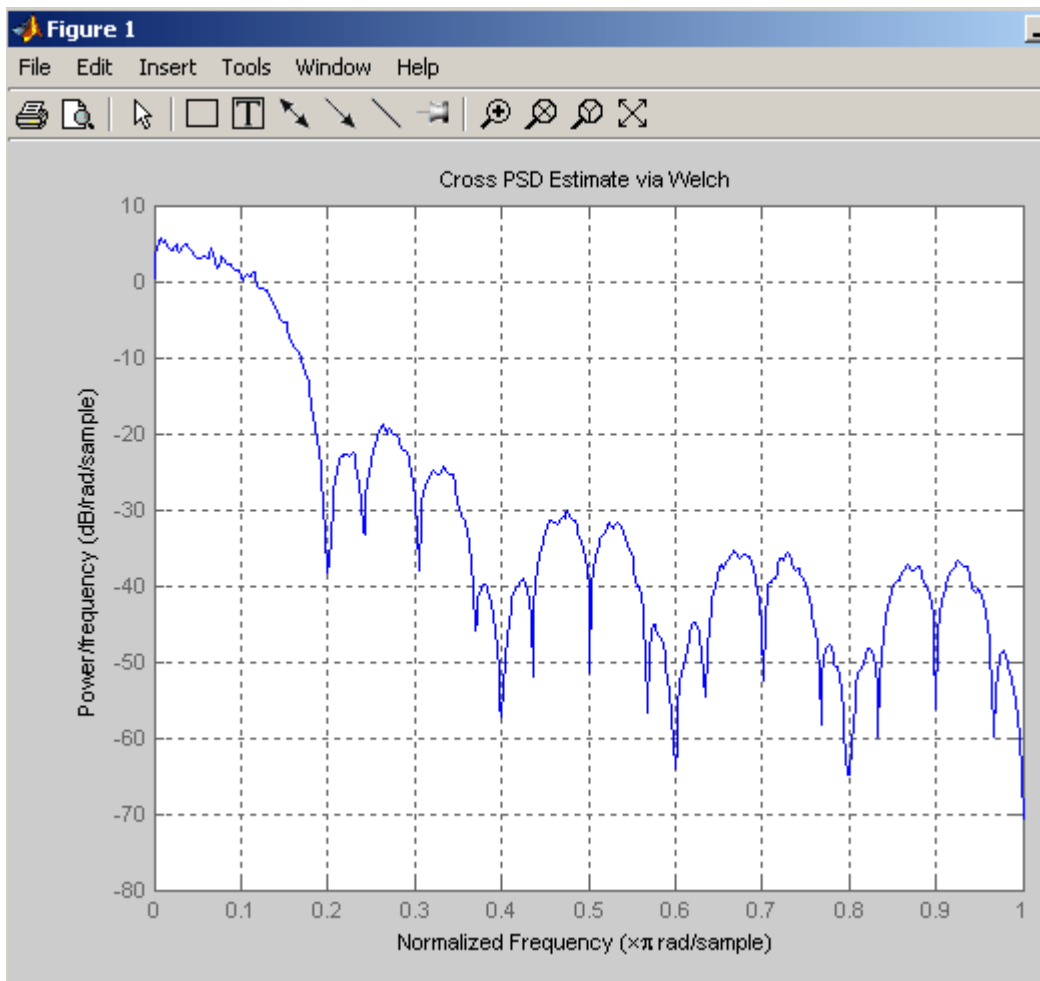
`[...] = cpsd(..., 'twosided')` returns the two-sided CPSD of real signals x and y . The length of the resulting `Pxy` is `nfft` and its range is $[0, 2\pi)$ if you do not specify `fs`. If you specify `fs`, the range is $[0, fs)$. Entering 'onesided' for a real signal produces the default. You can place the 'onesided' or 'twosided' string in any position after the `noverlap` parameter.

`cpsd(...)` plots the CPSD versus frequency in the current figure window.

Examples

Generate two colored noise signals and plot their CPSD with a confidence interval of 95%. Specify a length 1024 FFT, a 500 point triangular window with no overlap, and a sampling frequency of 10 Hz:

```
randn('state', 0);
h = fir1(30, 0.2, rectwin(31));
h1 = ones(1, 10)/sqrt(10);
r = randn(16384, 1);
x = filter(h1, 1, r);
y = filter(h, 1, x);
cpsd(x, y, triang(500), 250, 1024)
```



Algorithm

`cpsd` uses Welch's averaged periodogram method. See the references listed below.

References

- [1] Rabiner, L.R., and B. Gold. *Theory and Application of Digital Signal Processing*, Englewood Cliffs, NJ: Prentice-Hall, 1975. Pgs.414-419.
- [2] Welch, P.D. "The Use of Fast Fourier Transform for the Estimation of Power Spectra: A Method Based on Time Averaging Over Short, Modified Periodograms." *IEEE® Trans. Audio Electroacoust*, Vol. AU-15 (June 1967). Pgs.70-73.
- [3] Oppenheim, A.V., and R.W. Schaffer. *Discrete-Time Signal Processing*, Upper Saddle River, NJ: Prentice-Hall, 1999, pp. 737.

See Also

[dspdata](#), [mscohere](#), [pburg](#), [pcov](#), [peig](#), [periodogram](#), [pmcov](#), [pmtm](#), [pmusic](#), [pwelch](#), [pyulear](#), [spectrum](#), [tfestimate](#)

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