## Data Processing and Analysis

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## **Energy and Power Spectra**

It is frequently valuable to study the power distribution of a signal in the frequency domain. For example, we may wish to have estimates for how the power in a signal is distributed with frequency, so that we can quantitatively state how much power is in a particular band of interest relative to other frequencies. Power peaks and/or troughs across specific frequency ranges may reveal important information about a physical process. Given a power spectral density function, the power across any range of frequencies can then be estimated by integrating such a function over the band of interest.

The simplest such measure of energy (or, with scaling modifications, power) in a signal as a function of frequency is the energy spectral density, which is just the square of the spectral amplitude

$$|\Phi(f)|^2 = \Phi(f)\Phi^*(f) .$$
 (1)

Applying the convolution theorem, and noting that phase conjugation in the frequency domain corresponds to reversal in the time domain, this can be recognized as the Fourier transform of the autocorrelation

$$\Phi(f)\Phi^*(f) = \mathcal{F}[\phi(t)*\phi^*(-t)] = \mathcal{F}[\phi(t) \operatorname{cor} \phi^*(t)] .$$
(2)

We can thus observe that function which has a sharp and narrow autocorrelation function will have a broad energy spectral density, while a function which has a broad autocorrelation function will have a narrow energy spectral density. This can perhaps be understood better by considering what is in fact required of a time-domain function for it to have narrow (in the limit, deltalike) autocorrelation function; the function must change rapidly, so that it does not resemble itself very much for a small shift from zero lag. For a function to change rapidly, it must have high frequency energy in its spectrum. Note that, because the units of a spectrum are  $u \cdot s = u/\text{Hz}$ , the units of (1) are  $u^2/\text{Hz}^2$ , where u denotes the physical units of  $\phi(t)$  (e.g., Volts, Amperes, meters/s, etc.).

Many interesting signals, such as those arising from an incessant excitation, are, practically speaking, unbounded in time or *continuous* (as opposed to signals that are limited in time, or *transient*). If the statistical behavior of the

signal (we will look at statistical aspects of time series much more later on in the class) doesn't change with time, so that the spectral and other properties of the signal are time-invariant, it is generally referred to as *stationary*. Examples where signals can often be considered to be stationary may include seismic, thermal, or electromagnetic noise, tides, winds, temperatures, and currents. Some signals of interest exhibit strong periodicities (tides, for example) because they are associated with astronomical or other periodic forcing. Because of their incessant nature, such signals have infinite total energy

$$E_T = \lim_{T \to \infty} \int_{-T/2}^{T/2} |\phi(t)|^2 dt = \infty,$$
(3)

so that the Fourier transform of the autocorrelation (2) won't converge. The frequency content of such signals may, however, still be examined using *power* spectral density, or simply *PSD*.

Signal power averaged over some interval T is simply the energy (3) normalized by the length of the observation

$$P_T = \frac{1}{T} \int_{-T/2}^{T/2} |\phi(t)|^2 dt = \frac{1}{T} \int_{-\infty}^{\infty} |\phi(t) \cdot \Pi(t/T)|^2 dt .$$
 (4)

As the observation interval T becomes long, this converges to the true signal power

$$P = \lim_{T \to \infty} P_T \ . \tag{5}$$

The PSD is defined as

$$PSD[\phi(t)] = \lim_{T \to \infty} \frac{1}{T} \Phi_T(f) \cdot \Phi_T^*(f)$$
(6)

where the time series has been *windowed* by multiplying the time series with a boxcar function of unit height and length T, so that

$$\Phi_T(f) = \mathcal{F}[\phi(t) \cdot \Pi(t/T)] . \tag{7}$$

Note that dimensional analysis shows that the units of the power spectral density in (6) are  $u^2/\text{Hz}$ . Further note that that PSDs will be real, symmetric functions over f for the common case where  $\phi(t)$  real (and thus has a Hermitian spectrum). For this reason, as we noted for the complex Hermitian spectrum in considering the various Fourier symmetry relationships, the power spectra of real functions are typically plotted only for positive frequencies. Because we can never do calculations on an infinite-length signal, all PSDs in practice are *estimates* of Pthat we hope approach the "true" PSD for the continuous and (conceptually) time-infinite signal that we are studying.

The simplest (but definitely not the best!) way to estimate a PSD is to simply truncate the data with a T-length rectangular time window extending across a time interval that we can define as being between -T/2 to T/2. This estimate, because it seems like the obvious thing to do, has a long history, and it is

sometimes referred to as a *periodogram*. To understand the relationship between the periodogram estimate and the true PSD (6), note that for a rectangular window of width T and a real-valued time series (which, again, has a Hermitian spectrum)

$$PSD_{periodogram} = \frac{1}{T} |\Phi_T(f)|^2 = \frac{1}{T} |\mathcal{F}[\phi(t)\Pi(t/T)]|^2$$
(8)

Using the convolution theorem, this gives

$$PSD_{periodogram} = \frac{1}{T} |\Phi(f) * \operatorname{sinc}(Tf)|^2$$
(9)

where, recall, the Fourier transform of  $\Pi(t/T)$  is

$$\operatorname{sinc}(Tf) = \frac{\sin \pi Tf}{\pi Tf} \ . \tag{10}$$

Thus, what we obtain in a periodogram estimate is the true PSD of the process convolved in the frequency domain with the sinc(Tf) function. Figure 1 shows such a periodogram estimate for a sinusoidal process. The underlying process has a delta function spectrum, with the delta function centered on the frequency of the sinusouid. However, (9) shows in a broader peak in the PSD estimate. The spearing effect of the convolution produced a limited *spectral resolution* view of the sinusoidal process.

The loss of resolution caused by the convolution in (9) is undesirable, and we typically want to minimize and characterize it. As convolution is essentially a smoothing operation (recall that variances add when we convolve two functions, thus increasing their spread), our windowed estimate in (9) is a blurred image of the true spectrum. In the periodogram case, this blurring takes the specific form of convolution with a sinc function because we chose an (abrupt) boxcar data truncation on the  $\pm T/2$  interval, and the Fourier transform of the boxcar function is a sinc. The sinc function's slow  $((Tf)^{-1})$  fall-off and oscillatory side lobes are easily improved by modifying the estimation method, and the periodogram should thus never be used in practice except for quick and dirty estimates of the PSD.

The smearing of spectral resolution due to the convolution of the true spectrum with the Fourier transform of the windowing function is called *spectral leakage*, as the frequency domain convolution in (9) causes power from surrounding frequencies to "leak" into the estimate at any particular frequency. In its simplest form, spectral leakage in the periodogram will make the PSD estimate for a function that is really a sinusoid of frequency f have the appearance of sinc functions centered on the true frequencies  $(\pm f)$  of the continuous signal, rather than the true delta functions.

Spectral leakage can be reduced by increasing T, so that the Fourier transform of the windowing function becomes reciprocally (by a factor of 1/T) narrower. However, for statistical reasons involving the variance of the estimate that we will not elaborate on here, this is still a poor way to estimate the PSD.



Figure 1: A Periodogram Estimate of a Pure Sinusoidal Process

A better way to reduce spectral leakage, at the cost of eliminating the statistical contributions of data near the endpoints of the data series, is to window with a smoother time function than the boxcar that has a Fourier transform that is more delta-like by some measure. For example, consider the *Bartlett* or *Parzen* window

$$\Lambda(2t/T) = 4/T^2 \left( \Pi(2t/T) * \Pi(2t/T) \right)$$
(11)

which is a unit height triangle function spanning the interval -T/2 to T/2.  $\Lambda(2t/T)$  is easily seen by the convolution theorem to have a Fourier transform given by the sinc function squared

$$\mathcal{F}[\Lambda(2t/T)] = 4/T^2 \mathcal{F}[\Pi(2t/T) * \Pi(2t/T)] = \operatorname{sinc}^2(fT/2)$$
(12)

which falls off asymptotically as  $(Tf)^{-2}$  and is positive everywhere (although it is still oscillatory; Figure 2).

The formulation of various data windows such as the Parzen window has historically formed a rich are of research (if not a veritable cottage industry) in signal processing, and numerous function are in common usage (may of which can be readily generated using various MATLAB functions in the signal processing toolbox). The general tradeoff in window selection arises between the width of the main lobe of the leakage function and the rate of decay away from the center frequency. A few examples of commonly used windows and their corresponding spectral leakage properties when they are applied to a true sinusoidal signal (which, again, has a "true" delta function PSD), are shown in the following figures.

An interesting issue in spectral estimation that arises from the use of windows is the "throwing out" of data resulting from tapering near the data segment endpoints. The result is that we are downweighting information and thus increasing the statistical uncertainty of the PSD estimate. For long, stationary time series, one straightforward and widely-applied method of addressing this issue is to evaluate a suite of either overlapping or nonoverlapping spectral estimates for a host of window locations, and to subsequently average them and calculate statistical bounds on the mean estimate. The most commonly used technique along these lines is called *Welch's Method* (see the *pwelch* function in the MATLAB signal processing toolbox).

An elegant, more computationally-intensive, and increasingly widely utilized method of estimating spectra (see the *pmtm* function in MATLAB's signal processing toolbox) is *multitaper spectral estimation* (e.g., Thomson, proc. IEEE V. 70, No 9, September 1982). In multitaper spectral estimation, a family of statistically independent spectral estimates is obtained from a signal using an orthogonal set of windows on the estimation interval that are referred to as *prolate spheroidal tapers* (Figure 6).

In multitaper spectral estimation individual spectra obtained from the prolate spheroidal tapers are combined in a weighted sum to produce a spectral estimate with leakage that is approximately limited to some specified frequency band,  $\pm W$ . Specifically, for a specified time-bandwidth product, NW, the multitapers are the Fourier Transforms of solutions,  $U_k$ , to the frequency-domain



Figure 2: A Bartlett or Parzen window estimate of a pure sinusoidal process spectrum.



Figure 3: A Welch window estimate of a pure sinusoidal process spectrum.



Figure 4: A Hann window estimate of a pure sinusoidal process spectrum.



Figure 5: A Kaiser-Bessel window estimate of a pure sinusoidal process.



Figure 6: Prolate Spheroidal Taper Functions ( $0 \le k \le 4; NW = 4$ ).



Figure 7: Fractional energy leakage outside of f = (-W, W) for the first five multitapers (NW = 4).

eigenvalue-eigenfunction equation

$$\int_{-W}^{W} \frac{\sin N\pi (f - f')}{\sin \pi (f - f')} U_k(N, W; f') \, df' = \lambda_k(N, W) \cdot U_k(N, W; f) \,. \tag{13}$$

where the  $\lambda_k$  are eigenvalues (the first 2NW of which are close to one), and N is the discrete length of the taper sequence (this is a discrete formulation for spectral estimation on sampled time series, which we shall discuss next shortly. The integral in (13) is a convolution in the frequency domain between the  $U_k$  and the Dirichlet kernel, a function that arises frequently in discrete Fourier analysis because it is the Fourier transform of the sampled counterpart of the boxcar function (more on this later). Solutions to (13) form an orthogonal family of functions which have the greatest fractional energy concentration in the frequency interval (-W, W). The eigenvalues in (13) are measures of the degree to which spectral leakage is confined to (-W, W). Spectral leakage becomes increasingly worse for higher-order tapers, with the energy leakage being given approximately as

$$1 - \lambda_k \approx \frac{\sqrt{2\pi}}{k!} (8c)^{k+1/2} e^{-2c}$$
(14)

where  $c = \pi N W$ . Figure 7 shows the fractional leakage for the first five multi-tapers.

Because of the appreciable leakage of the higher order tapers the lowest few (typically six or so, depending on the values of N and W) are typically used in



Figure 8: Prolate spheroidal taper spectral estimate (k = 0; NW = 4).

practice. Figures 8 through 12 show the five lowest order multitaper estimates for NW = 4 for the example sine wave signal used in the earlier figures. Figure 13 shows the multitaper estimate obtained by averaging them. The leakage function displayed in Figure 13 approximates a frequency boxcar of width 2W.

An example geophysical application of the PSD is to quantify the background noise characteristics of seismic stations, so as to gauge, for example, how they compare to known very quiet sites, and to assess what frequency bands good or bad for signal detection. This is of considerable importance both for earthquake and Earth structure studies and for estimating detection thresholds for clandestine events (e.g., nuclear tests). Figure 14 shows PSD estimates for a fairly quiet IRIS broadband seismic station in the Tien Shan mountains near Ala Archa, Kyrgyzstan, at periods ranging from 0.1 to  $10^3$  s (about 17 minutes). The bounding curves are empirically-based high- and low-noise extremal



Figure 9: Prolate spheroidal taper spectral estimate (k = 1; NW = 4).



Figure 10: Prolate spheroidal taper spectral estimate (k = 2; NW = 4).



Figure 11: Prolate spheroidal taper spectral estimate (k = 3; NW = 4).



Figure 12: Prolate spheroidal taper spectral estimate (k = 4; NW = 4).



Figure 13: Prolate spheroidal taper average spectral estimate  $(0 \le k \le 4; NW = 4)$ .

models for broadband stations. Noise at short periods is dominated by cultural (man-made), wind, and other rapidly varying environmental effects. The prominent noise peaks near 7 and 14 seconds are globally observed and are generated by ocean waves. The long-period power is higher on the horizontal sensors as opposed to the vertical sensors because they are sensitive to tilt caused by barometric, thermal, or other long-period noise sources. The peak near 1.6 s is unusual and may represent microseismic wave noise from the nearby Issyk Kul, one of the largest high-altitude alpine lakes in the world.

As a final indication of the great utility of the PSD, the Figure (15) shows processed PSDs from a broadband seismometer (Guralp CMG-3Tb) located in a 255-m deep borehole in the polar icecap near the South Pole. A great many of 1-hour data length, 50% overlap, PSDs using a hamming taper, were calculated from the month of May, 2003, and the resulting individual PSDs were used to assemble an empirical probability density function for the signal characteristics at he station The bifurcation of the high frequency noise is caused by intermittent periods where tractors are moving snow near the station. Pink misty areas concentrated around 1 and 20 s are PSDs that include teleseismic earthquake signals. At short periods this is among the quietest stations on Earth.



Figure 14: Earth Acceleration Power Spectral Density for background noise at the Ala Archa IRIS/IDA station as a function of period. Z, N, E refer to vertical, north, and east seismometer components. Curves labeled NM are the empirical noise model bounds of Peterson (1994) denoting to extremal PSD values from stations installed around the world. The reference (0 db) level is  $(1 \text{ m/s}^2)^2/\text{Hz}$ . PSD estimates were obtained using Welch's method.



Figure 15: Quiet South Pole (QSPA) Global Seismic Network Station, power spectral density probability density plot