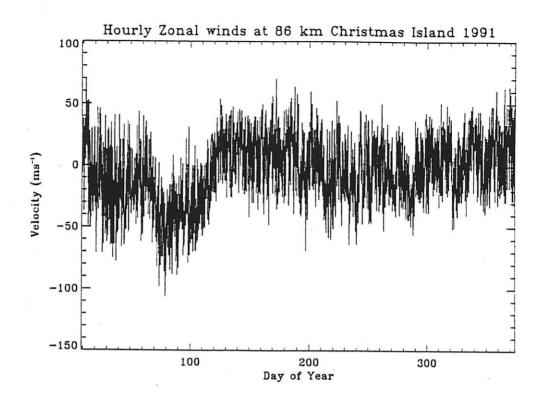
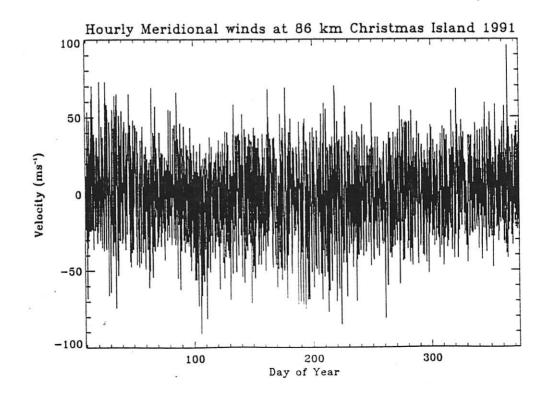
Exploring Data Using Fourier Techniques

Robert A. Vincent Department of Physics and Mathematical Physics University of Adelaide This part of the course shows how conventional Fourier techniques can be used to explore data. The methods are generally applicable to either temporal or spatial data sets.

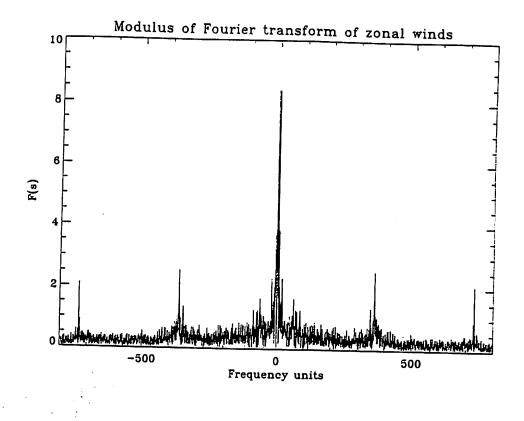
- Moving power spectra.
- Filters.
- Complex demodulation.

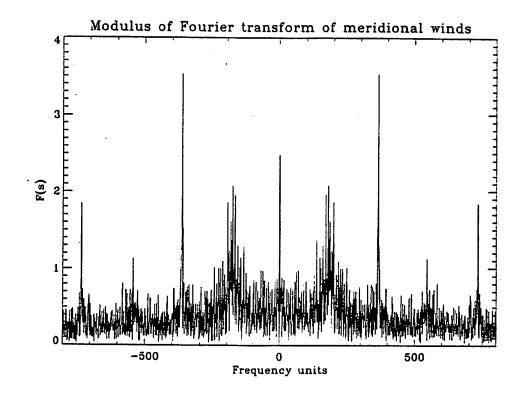
Examples are drawn from a one-year (8760 x 1 h points) time series of zonal and meridional wind components.

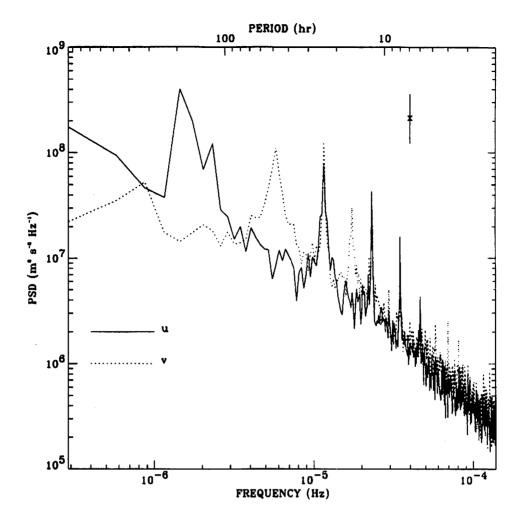




Raw Spectra: $\Delta f = 1/8760$ h-1



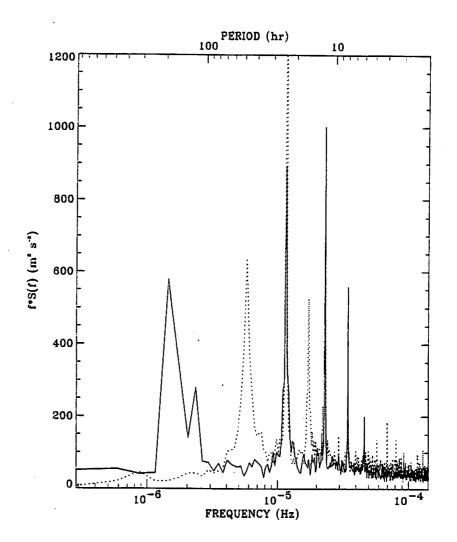




Power Spectra

(Ref.: Numerical Recipes, ch. 13; Jenkins and Watts)

- Welch window used.
- Block averaged to reduce variance.
- Nine 40-day blocks used, overlapped by 50% to give 17 overlapping segments (c.f. NR §13.4).
- ~ 26 degrees of freedom give 95% confidence limits (JW) of 0.61 and 1.8
- On a log plot the 'error bars' are uniform.



Another way to display a power spectrum is in "Energypreserving" form, where f S(f) is plotted against log(f).

• Area under given segment of width ∆(log f) is proportional to variance.

$$f.S(f)\Delta(\log f) = f.S(f)\frac{\Delta f}{f} = \sigma_{\Delta}^2$$

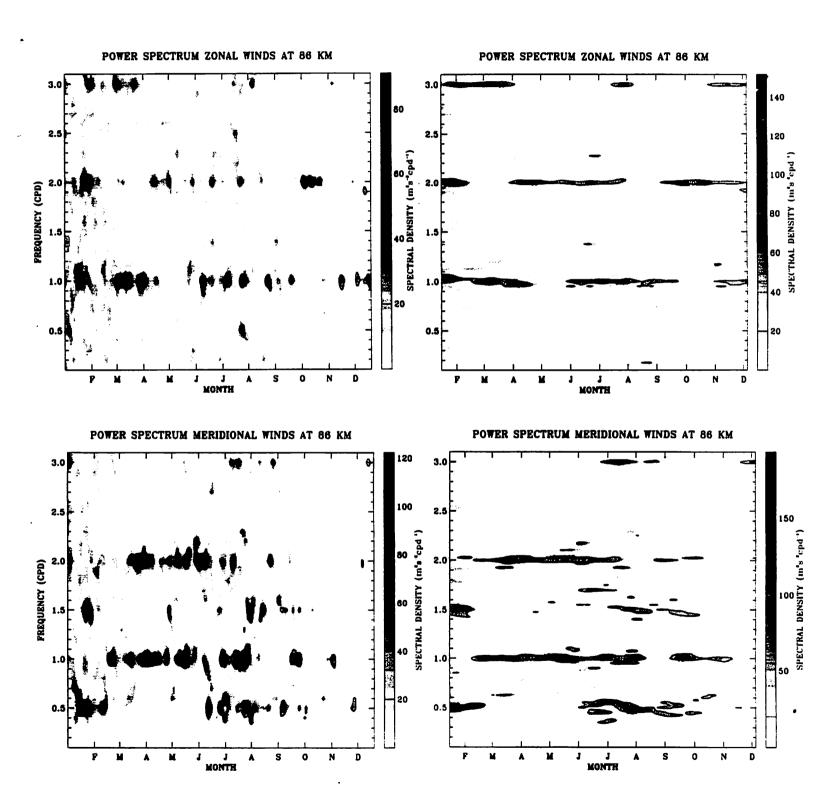
<u>Moving Power Spectra</u> (Sliding FFTs)

- 1. Select data with window of length T.
- 2. Compute spectrum, S(f).
- 3. Assign S(f) to center of window.
- 4. Step window on by amount ΔT .
- 5. Repeat 2. for data interval $\Delta T \rightarrow T + \Delta T$, etc.
- 6. Finish with 2-d array of S(f) vs t.

<u>Time Resolution vs Frequency Resolution</u> <u>Some Examples</u>

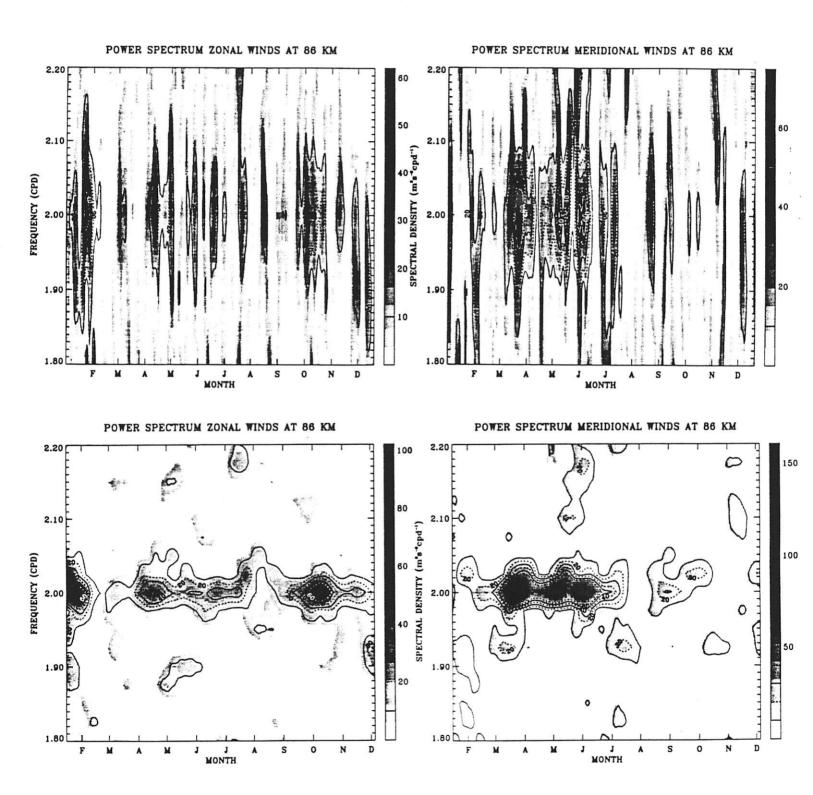
 $\Delta T = 10 \text{ day}$

 $\Delta T = 30 \text{ day}$



• Resolution depends on window. $\Delta T = 10 \text{ day} \Rightarrow \Delta f = \sim \pm 0.15 \text{ cpd}$ (Welch window) $\Delta T = 40 \text{ day} \Rightarrow \Delta f = \sim \pm 0.04 \text{ cpd}$ (Welch window)

• Watch out for side-lobes.



Filters

Time-Domain Filters

A simple filter is *n*-point weighted moving average:

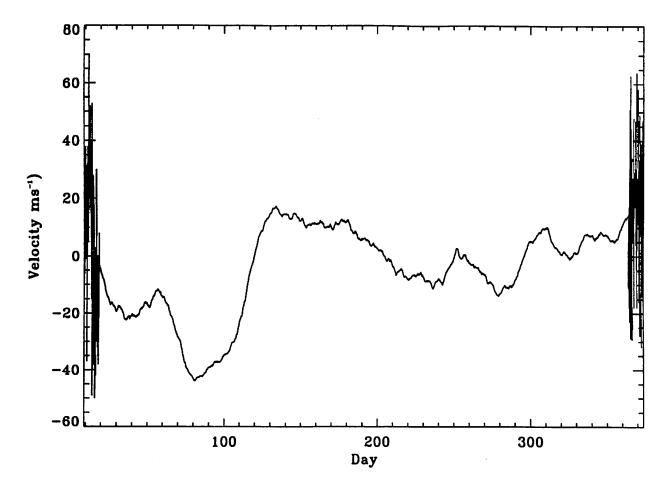
$$g_i(t) = \frac{1}{W} \sum_{j=0}^{n-1} w_j f_{i+j-n/2}$$
 where $W = \sum_j w_j$

for i = n/2....N-n/2

 w_i are weights.

(e.g 'Triangular-average' is often used to lightly smooth data: $w_i = \frac{1}{4}, \frac{1}{2}, \frac{1}{4}$)

Simplest example is <u>Rectangular</u> or <u>Box-car</u> average: $w_j = 1$ and W = n



Zonal winds smoothed with a 481-point (20-day) moving average.

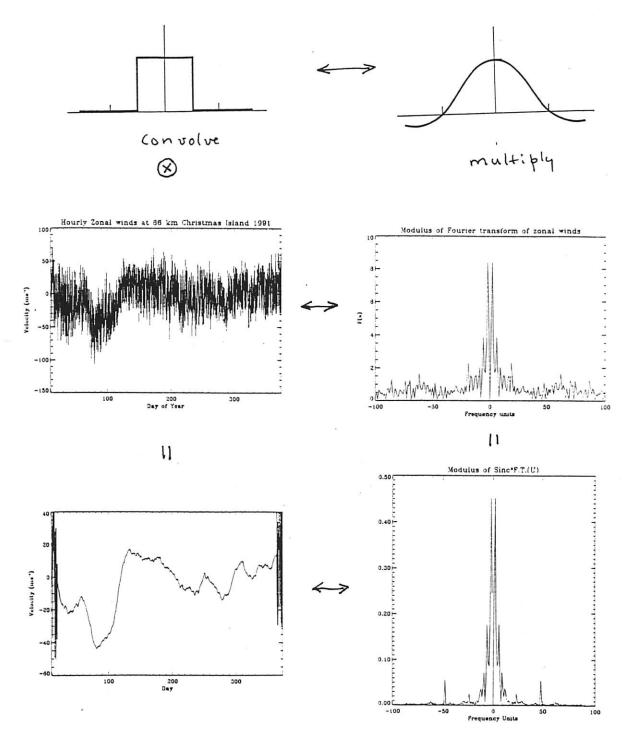
Note loss of 240 points (10 days) of data at each end. Also note structure with time scales of less than 20 d.

Remember:

"Convolution (\otimes) in one domain is equivalent to multiplication in the other domain"

$$f \otimes g \leftrightarrow F.G$$

or
$$f.g \leftrightarrow F \otimes G$$



The action of a filter on an input signal, y(t), can be summarized as:

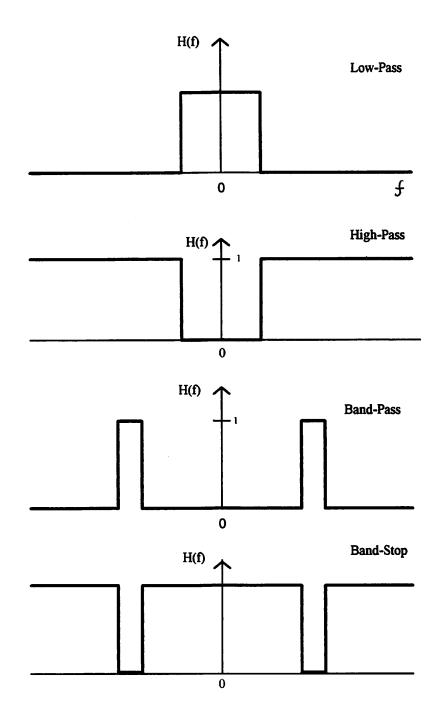
<u>Time Domain</u>	Frequency Domain	
<i>y(t)</i>	\leftrightarrow	Y(f)
8		multiply
h(t)	\leftrightarrow	H(f)
↓ [.]		\downarrow
g(t)	\leftrightarrow	G(t)

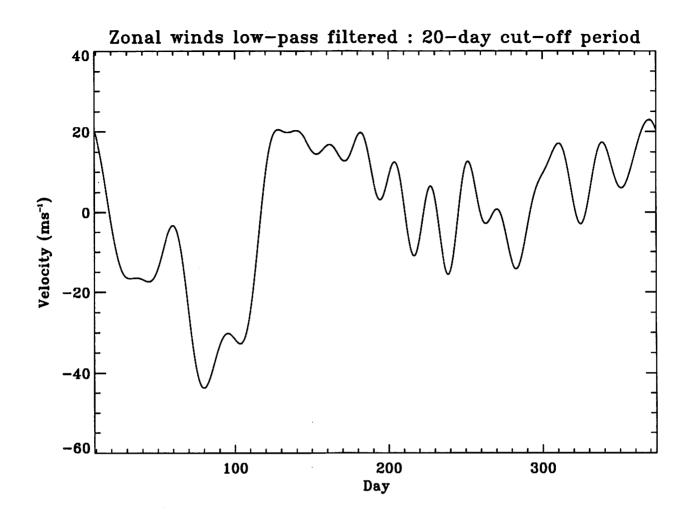
H(f) is the *Transfer Function*. Easily found if $y(t) = \delta(t) \leftrightarrow Y(f) = 1$

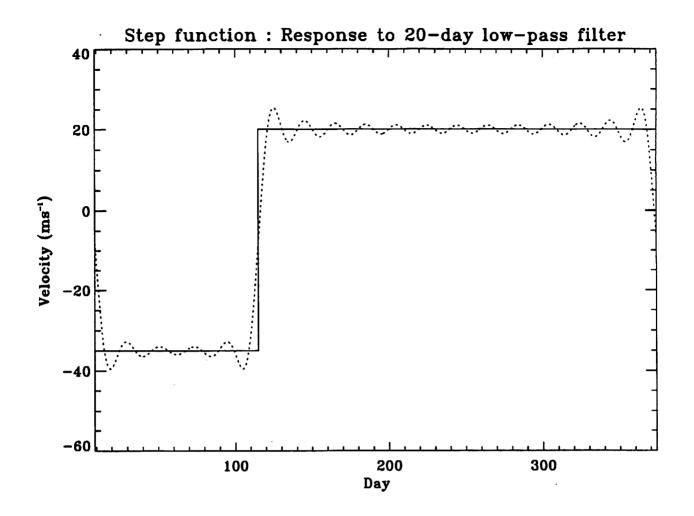
h(t) is Impulse response.

or specify filter by Step response, S(f): $H(f) = 2\pi i S(f)$

Ideal Filters



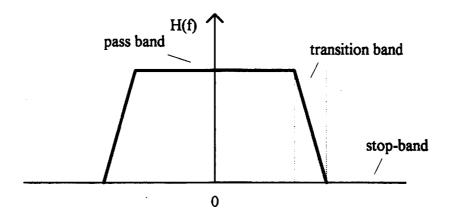




- Note overshoot and ringing Gibbs phenomenon.
- End effects due to windowing of data.

Tapered or Transition Filters

Tapered Low-Pass



See Forbes, Elgar, Rabiner and Gold, and Kuc, for further discussion.

Simple taper is raised-cosine function (see Kuc, ch 9)

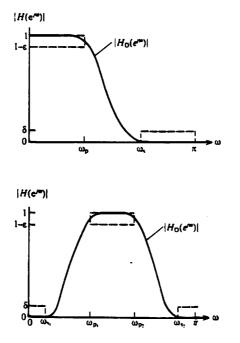
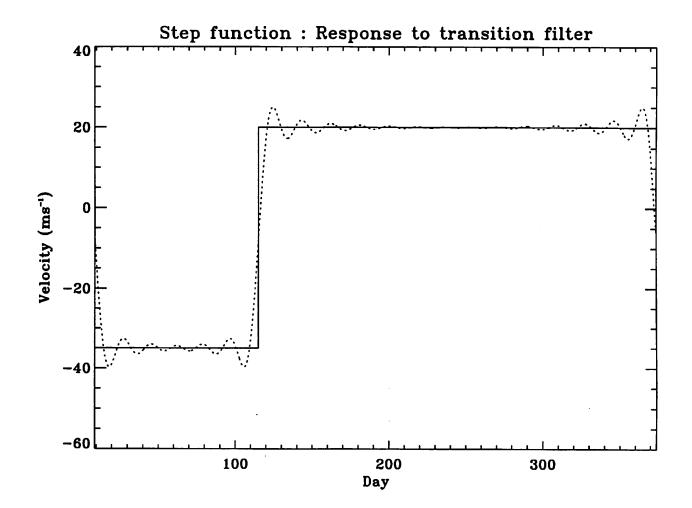
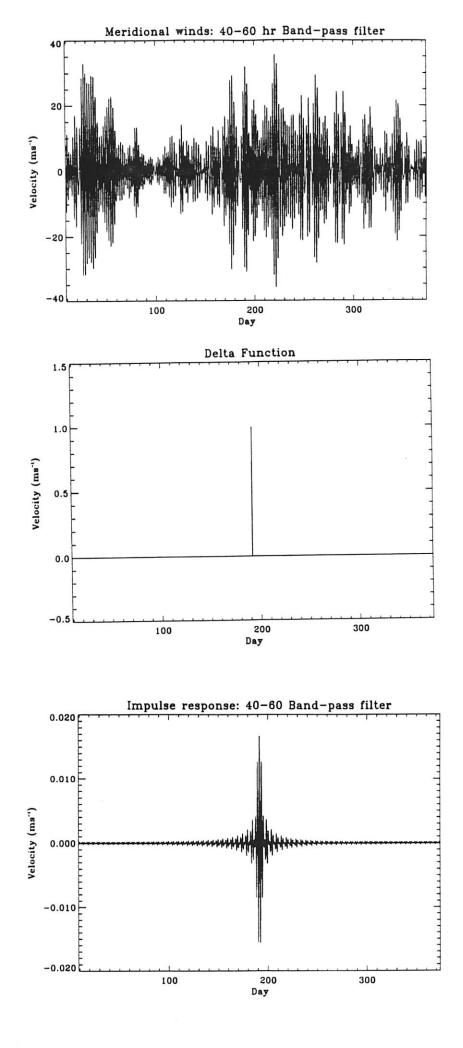


FIGURE 9-33 The desired magnitude response for lowpass and bandpass filters is obtained by connecting the passband and stopband with a raised-cosine function.





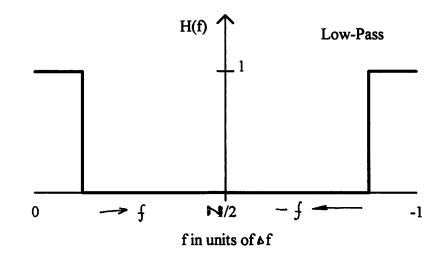
Filtering in the Frequency Domain

(See also *NR*, §13.5)

- Define filter for both positive and negative frequencies.
- For real data chose H(f) that is real and even in frequency i. e. $H(-f) = H(f)^*$.
- Sharp edges in *H(f)* will produce ringing (Gibb's phenomenon) for impulsive input. Test filter before using. May need to taper edges.
- Forward transform time series, y(t), to give Y(f).
- Form product of Y(f) and H(f).
- Inverse transform G(f) to give g(t).

• **REMEMBER** that FFT orders transform in frequency space as:

0, Δf , $2\Delta f$,..... $\pm N\Delta f/2$,.... $-2\Delta f$, $-\Delta f$ where $\Delta f = 1/N\Delta t$



COMPLEX DEMODULATION

(Ref: Bloomfield, ch. 6)

A convenient way to investigate variations of amplitude and phase (frequency).

Consider the time series:

 $f(t) = A(t)\cos(\omega t + \phi(t))$ or $f(t) = A(t)\cos(\omega t)$

where A and ϕ (or ω) are slowly varying functions of time.

Complex demodulation allows A and ϕ to be described as a function of t.

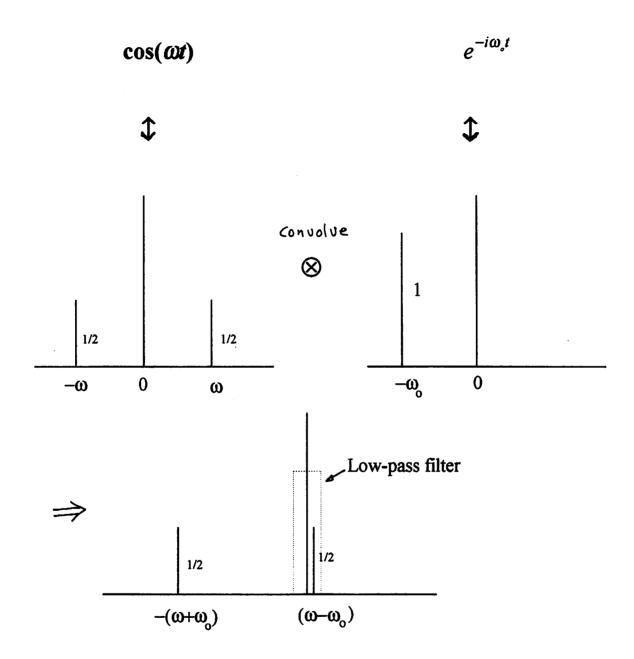
First consider the simple function

 $f(t) = \cos(\omega t)$

Multiply by $e^{-i\omega_o t}$, where $f_0 = \omega_0/2\pi$ is the frequency we wish to demodulate about, to get:

 $g(t) = \cos(\omega t) e^{-i\omega_s t}$

In the frequency domain this is equivalent to:



After low-pass filtering we get:

$$g'(t) = \frac{1}{2}\cos\{(\omega - \omega_o)t\}$$

(note, have to double g' to recover amplitude).

To apply:

- 1. Choose center frequency f_o .
- 2. Multiply time series by $e^{-i\omega_o t}$
- 3. Low-pass filter to give g'.
 Width of filter determines rate at which temporal variations can be studied.
- 4. To recover amplitude and phase:

$$A(t) = 2.|g'(t)|$$

$$\phi(t) = \tan^{-1} \left(\frac{imag(g')}{real(g')} \right)$$

Example: Time series with 8760 points ('hrs'). $f = 1/48 h^{-1}$ $f_o = 1/50 h^{-1}$. Filter bandwidth of the filter is 240 points (i.e. '10 days').

Note:

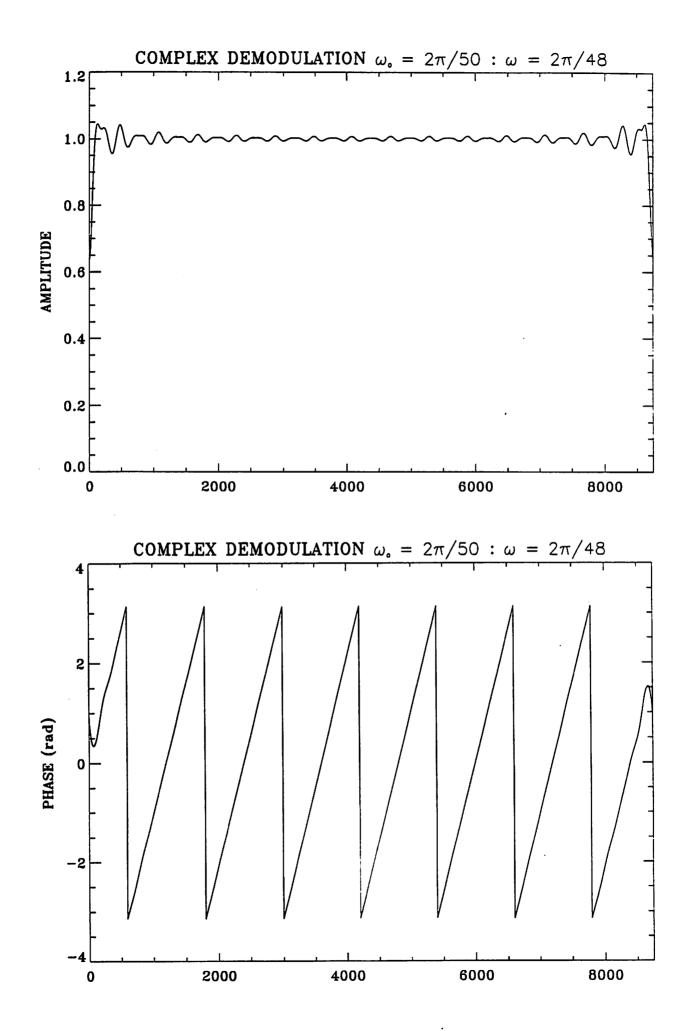
End effects due to filter.

Positive phase shifts with slope of 1 cycle per 1200 points - equivalent to frequency difference of $0.02 h^{-1}$.

The phase gradient is a measure of the 'local' frequency difference from demodulation frequency, f_{σ}

$$f - f_o = \frac{1}{2\pi} \frac{d\phi}{dt}$$

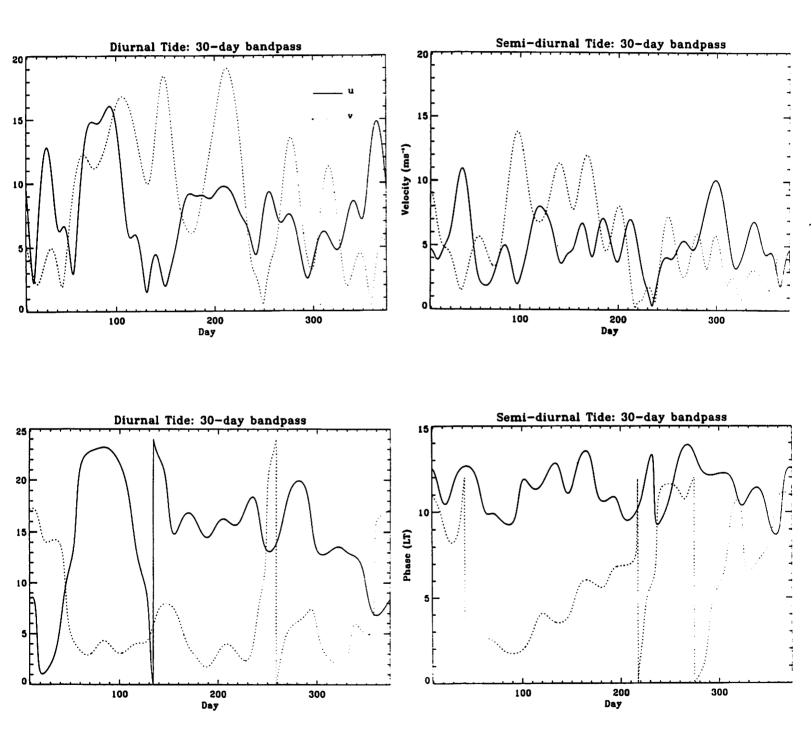
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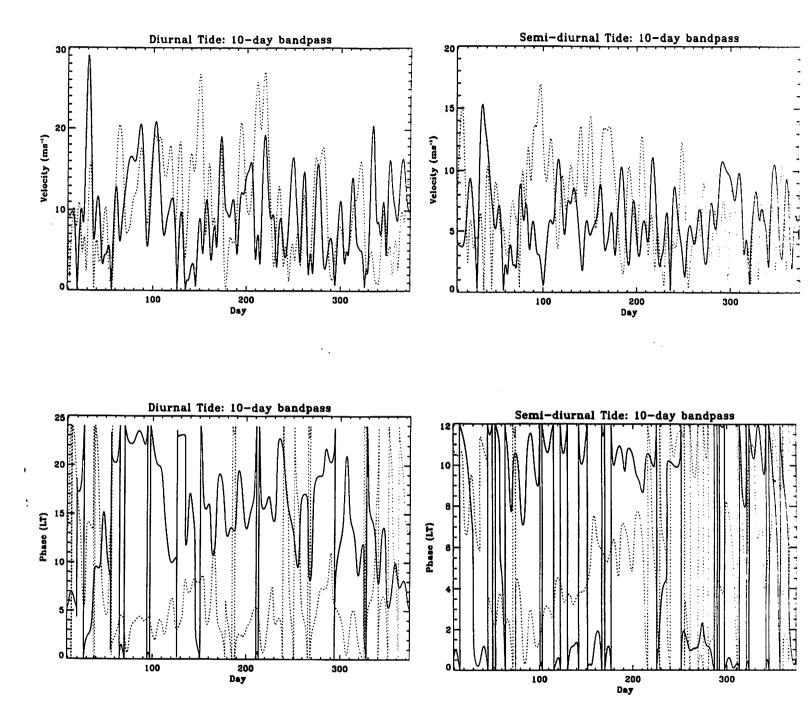


Complex Demodulation can be thought of as a "local" harmonic analysis i.e., a moving fit of

$$f(t) = A(t) \cos\{\omega t + \phi(t)\},\$$

but carried out on all the data simultaneously.



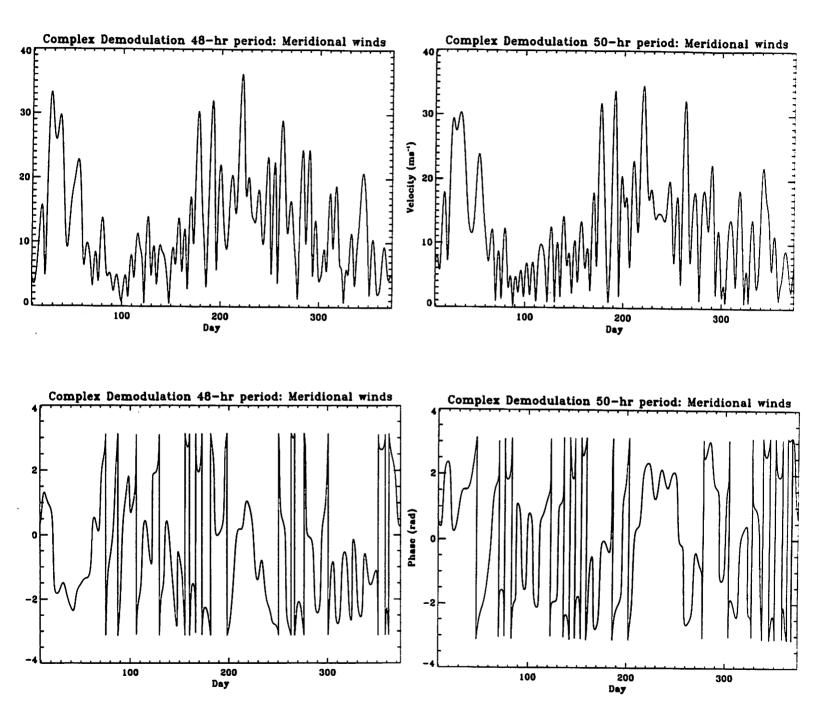


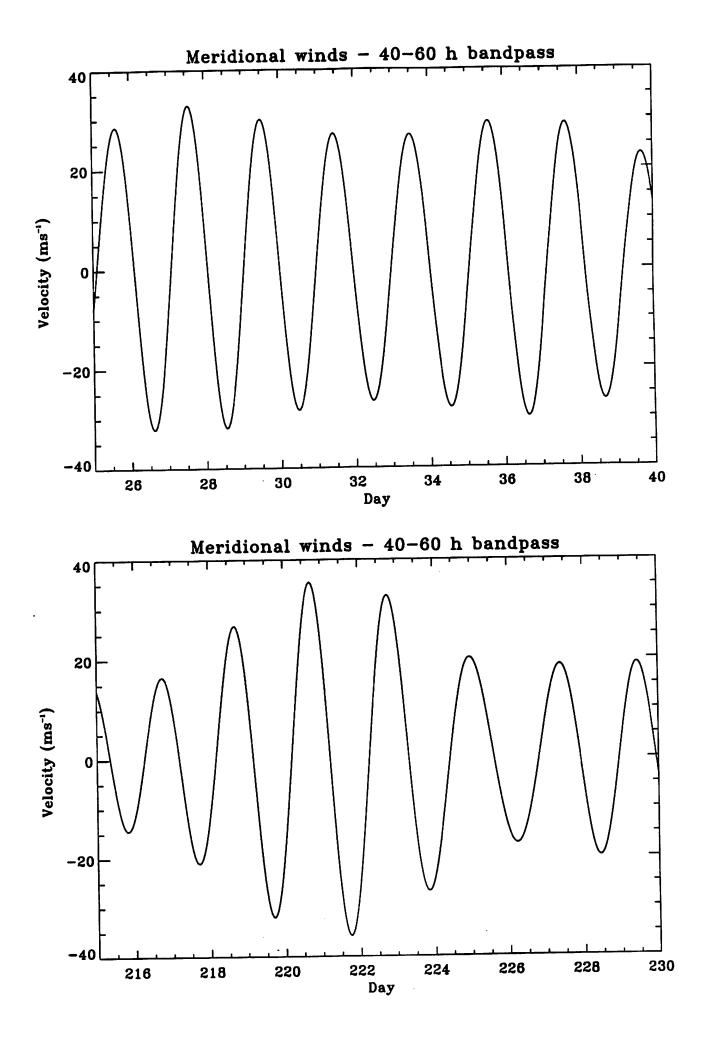
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