TODAY'S COURSE

Chapter 1.3, 1.4 & 2.2 OAF-2 Numerical Recipes Chapter 13.3

Topics:

Aliasing & Nyquist frequency (Optimal) Filtering measuring moments of a s.p.

RECAP LECTURE 1

DETECTION OF ASTRONOMICAL SIGNALS (S.P.) plus noise is influenced by instrument transfer function and data sampling

STATISTICAL MOMENTS CHARACTERISE THE signal (plus noise)

Noise can be due to the detector, background, and/or intrinsic to the signal

Assume WSS s.p. (mean does not depend on time, or much slower than measuring process, auto-correlation depends on OFFSET ONLY)

Another math tool Power Spectral Density

(㲍 amplitude of individual sinusoids)

(will return in more depth in Chapter 6)

 $P(f) = F(f)F(f)$ * CONTINUOUS FT: $P(f) =$ \int_0^∞ $-\infty$ FOR WSS SIGNALS: $P(f) = \int R(\tau) e^{-2\pi i f \tau} d\tau$ $F(f) =$ \int_0^∞ $-\infty$ $f(t) e^{-2\pi i f t} dt$ CONTINUOUS PSD: hence:

$$
F(f)F(f) = |F(f)|^2 = \int_{-\infty}^{\infty} R(\tau)e^{-2\pi i f\tau}d\tau
$$

DATA SAMPLING

TIME DOMAIN MULTIPLY S.P. WITH SHAH FUNCTION $m_s(x) = m(x)$ 1 τ $\frac{x}{\prod(\frac{x}{x})}$ τ $m_{samp,n} = m_s(x) = m(x)\frac{1}{\tau}\Pi(\frac{x}{\tau}) = \sum m(n\tau)\delta(x - n\tau)$ *n* $DISCRETE$ $FT:$ $M_{samp,k}$ = *N* – 1 N-1
 **** $n=0$ $m_{samp,n}$ $e^{2\pi i n k/N}$

2

 $|a_j|^2$

*a*0

 D **SCRETE PSD:** $P_j = \frac{2}{a_s} |a_j|^2$ power α amplitude squared:

$$
a_0 = M_{samp,k=0} = \sum_{n=0}^{N-1} m_{samp,n} \equiv N_0
$$

$$
a_k = M_{samp,k} = \sum_{n=0}^{N-1} m_{samp,n} e^{2\pi i n k/N}
$$

Nyquist frequency Nyquist frequency = half the frequency related with the SAMPLING RATE

sharp narrow signal requires more frequencies for its description than BROAD SIGNAL

cf. the number of sin+cos necessary to describe the signal

Optical spectra: range from min to maxi frequency = bandwidth set by the width and shape of the spectral lines

Nyquist frequency Sampling at the Nyquist frequency: no loss of information if $f_{\text{nyq}} = 2f_{\text{max}}$ in signal

CONTINUOUS SIGNAL H(T) FULLY DESCRIBED by the samples

$$
h(t) = \Delta t \sum_{n = -\infty}^{\infty} h_n \frac{\sin[2\pi f_{nyq}(t - n\Delta t)]}{\pi(t - n\Delta t)}
$$

Nyquist theorem: cont'd

WINDOWING & NOISE, BRAULT & WHITE 1971, A&A

Nyquist theorem: cont'd

Sampling; Brault & White 1971, A&A

Nyquist theorem: cont'd

Sampling; Brault & White 1971, A&A

Nyquist theorem: cont'd SAMPLING CAUSES REPLICATION OF SIGNAL

Sampling; Brault & White 1971, A&A

Aliasing

Aliasing Page 496 Num Res

Convolution with shah function: replication

10

Sampling: high frequencies are filtered out Window: low frequencies are filtered out

FILTERING F REQUENCY F ILTERING $Y(f) = X(f)H(f)$ $y(t)=$ \int_0^∞ −∞ $x(t - \theta)h(\theta)d\theta$ $y(t) = x(t) * h(t)$ LEADS TO BAND LIMITED DATA

Filtering of process x with filter h

TIME FILTERING

measure a process x(t) over interval T assumed ZERO OUTSIDE T $\equiv y(t) = \Pi(\frac{t}{T})$ $\frac{c}{T}$) $x(t)$ $Y(f) = X(f) * Tsinc(Tf)$ all information about frequencies <1/T is lost!

(Optimal) Filtering Num Res Chapter 13.0-13.3 deconvolve measured signal and response function of sampled data

DISCRETE CONVOLUTION THEOREM

$$
(r * s)_j \equiv \sum_{k=-M/2+1}^{M/2} s_{j-k} r_k \Leftrightarrow S_n R_n
$$

DISCRETE DECONVOLUTION

$$
\frac{\tilde{F}(r*s)_j}{R_n} = S_n
$$

However noise and uncertainties is response can make this process unreliable

Noise removal by optimal FILTERING

 $cs(t) = s(t) + n(t)$ $s(t)$ is the smeared signal i.e. true \times response

DESIGN AN OPTIMAL FILTER ϕ (T) THAT produces a signal u(T) as close as POSSIBLE TO U(T)

> *U* $\widetilde{U(f)} =$ $C(f)\phi(f)$ *R*(*f*) Close in least square sense is minimised \int^{∞} $-\infty$ $|U|$ $U(f) - U(f)|^2 df$

Noise removal by optimal FILTERING \int_0^∞ $-\infty$ *|* $\frac{S(f) + N(f)[\phi(f)]}{R(f)} - \frac{S(f)}{R(f)}$ *|* ^{2}df

! $S(f)N(f)df$ TERMS ARE ZERO SINCE NOISE and signal are uncorrelated

$$
\int_{-\infty}^{\infty} |R(f)|^{-2} \{ |S(f)|^2 |1 - \phi(f)|^2 + |N(f)|^2 | \phi(f)|^2 \} df
$$

ϴ minimised with respect to φ

Noise removal by optimal filtering *d*θ $d\phi$ $= 0$ $-2S^2(1-\phi) + 2N^2\phi = 0$

$\phi =$ *S*2 $S^2 + N^2$ OPTIMAL FILTER $|S(f)|^2 + |N(f)|^2 = PDS(f) = |CS(f)|^2$ DOES NOT CONTAIN TRUE SIGNAL DIRECTLY!

Noise removal by optimal FILTERING

Some Applications

ON OPTIMAL DETECTION OF POINT SOURCES IN CMB MAPS Vio et al, 2002, A&A, 391, 789

An optimal filter for the detection of galaxy clusters through weak lensing Maturi, et al. 2005, A&A, 442, 851

The largest scale perturbations: A window on the physics of the beginning Wandelt, New Astronomy Review, 2006, 11, 900

C.F. X-RAY TIMING EXPERIMENTS

X-ray Timing experiments

ESTIMATING THE MOMENTS OF a stochastic process

Chapter 2.2.2

SEE ALSO APPENDIX B3.2, LENA EA.

How representative is a measurement OF A S.P.?

How representative is a measurement?

How representative is a measurement?

Different for the case of type I X-ray

BURSTS

Galloway private communication

1 Measurement of x(t) in a time T => windowing and

averaging over time ΔT

$$
\text{WINDOWNING: } y(t) = \Pi(\frac{t}{T})x(t)
$$

averaging:

$$
z(t) \equiv y_{\Delta T}(t) = \frac{1}{\Delta T} \int_{t-\Delta T/2}^{t+\Delta T/2} y(t')dt' = \frac{1}{\Delta T} \int_{-\infty}^{\infty} \Pi(\frac{t-t'}{\Delta T})y(t')dt'
$$

= low-pass filter, remember Nyquist theorem