

# Gravitational-Wave Data Analysis: Lecture 1

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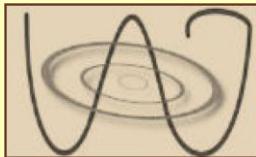


Gravitational Wave Astronomy Summer School  
May 28, 2012



# Outline for Today

- ▶ **What gravitational wave (GW) data is like**
- ▶ **Characterizing noise**
- ▶ **Frequency-domain representation of data and signals**
- ▶ **Calibration**
- ▶ **Digital filtering basics**



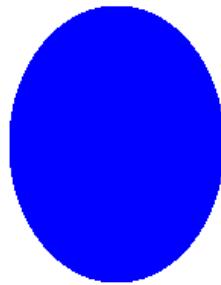
# Gravitational Waves: What They Are

GWs are perturbations of the spacetime metric

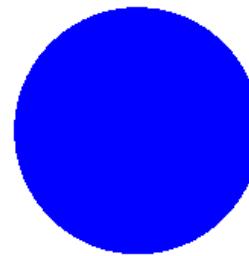
⇒ Change the effective distance between locally inertial points

According to GR, there are two polarization components

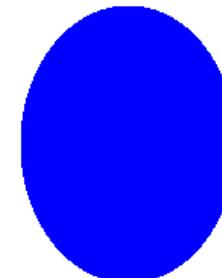
Wave can be a linear combination of polarization components



“Plus” polarization

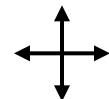


“Cross” polarization

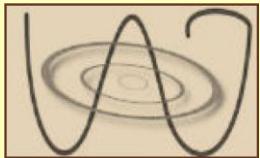


Circular polarization

...



These represent a time-dependent **strain** transverse to the direction the wave is traveling – i.e.  $\mathbf{h}_+(t)$  ,  $\mathbf{h}_\times(t)$



# The Romance of Gravitational Waves

## GWs can give us a unique view of astrophysical systems & events

GWs are powerful: can drive the dynamics of a system

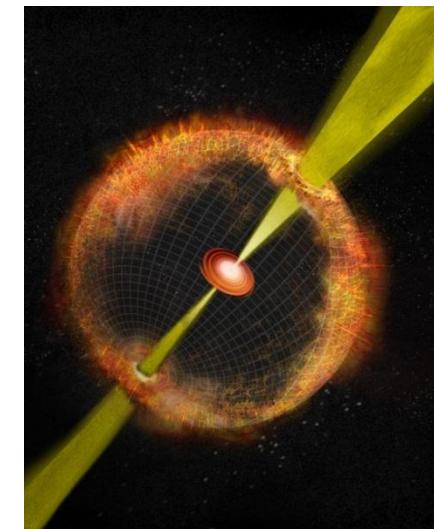
Not scattered by matter

⇒ probe the core engine of the event

Complementary to electromagnetic observations

Reveal “dark” systems such as black hole binaries

Enable tests of GR vs. other theories of gravity



## Detecting GWs is a supreme challenge

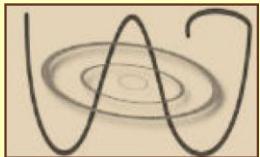
GWs are incredibly weak by the time they reach Earth –

Typical strain ~ **10<sup>-21</sup>** or even smaller !

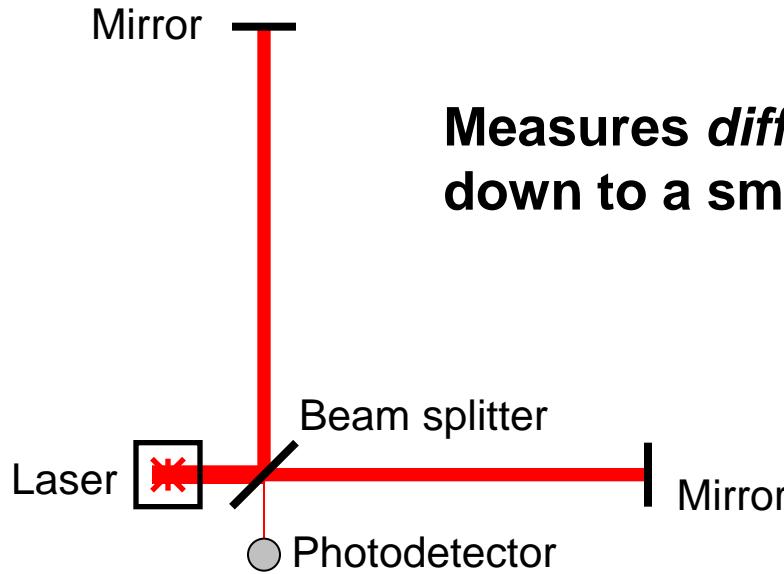
An unexplored frontier !

Can we really hope to detect them ?!?

Credit: Bill Saxton, NRAO/AUI/NSF



# Response of a GW Interferometer

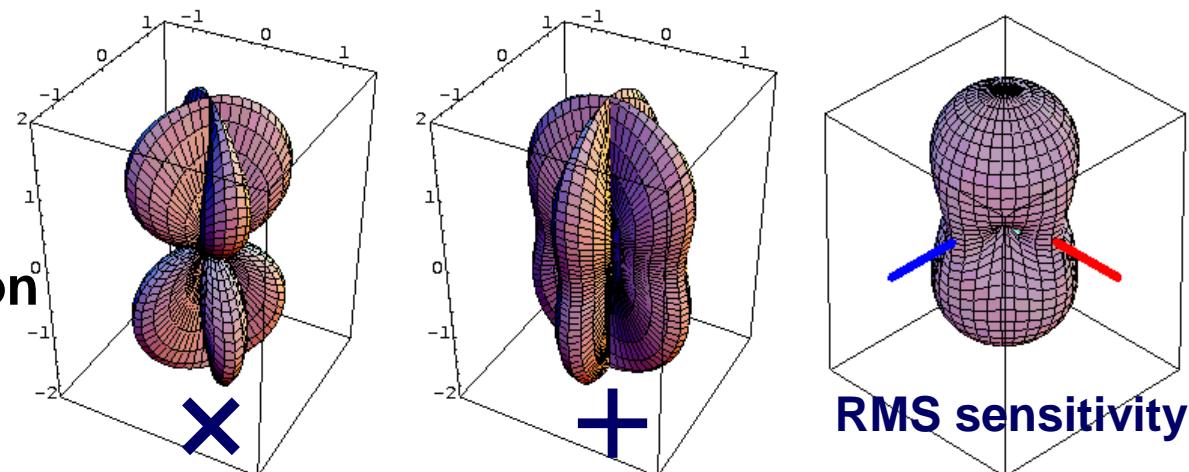


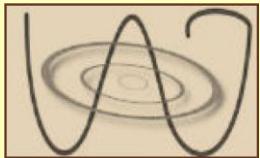
Measures *difference* in effective arm lengths down to a small fraction of a wavelength

In general, a linear combination:

$$h_{\text{det}}(t) = F_+ h_+(t) + F_x h_x(t)$$

Directional sensitivity depends on polarization in a certain (+,x) basis



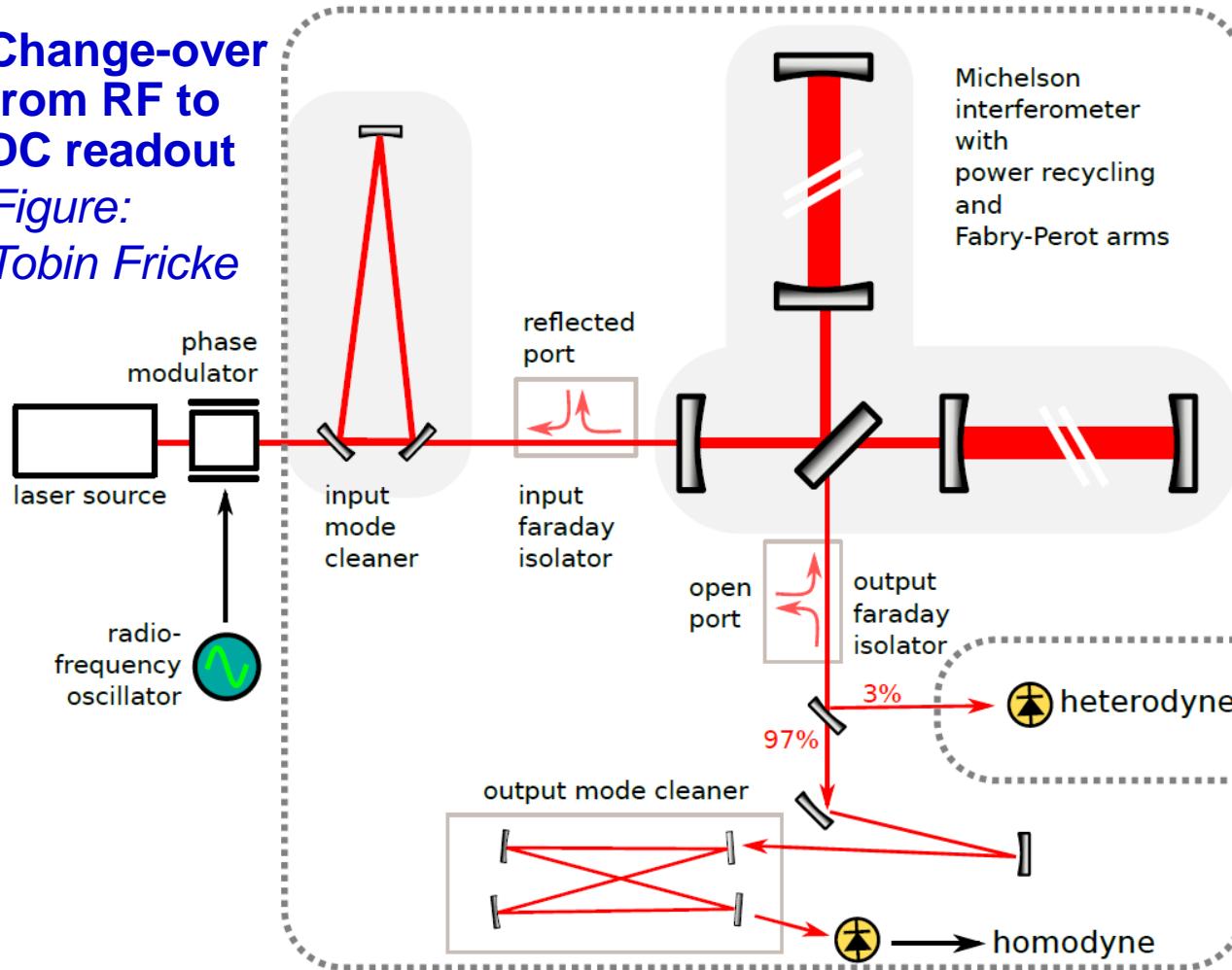


# GW Detector Readout – Overview

## The eLIGO interferometer

**Change-over  
from RF to  
DC readout**

*Figure:  
Tobin Fricke*



Heterodyne (RF)  
readout used for  
initial LIGO/Virgo

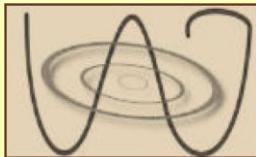
Modulate phase of  
input light (33 MHz),  
demodulate signal  
measured by  
photodiode

Perfect destructive  
interference on avg

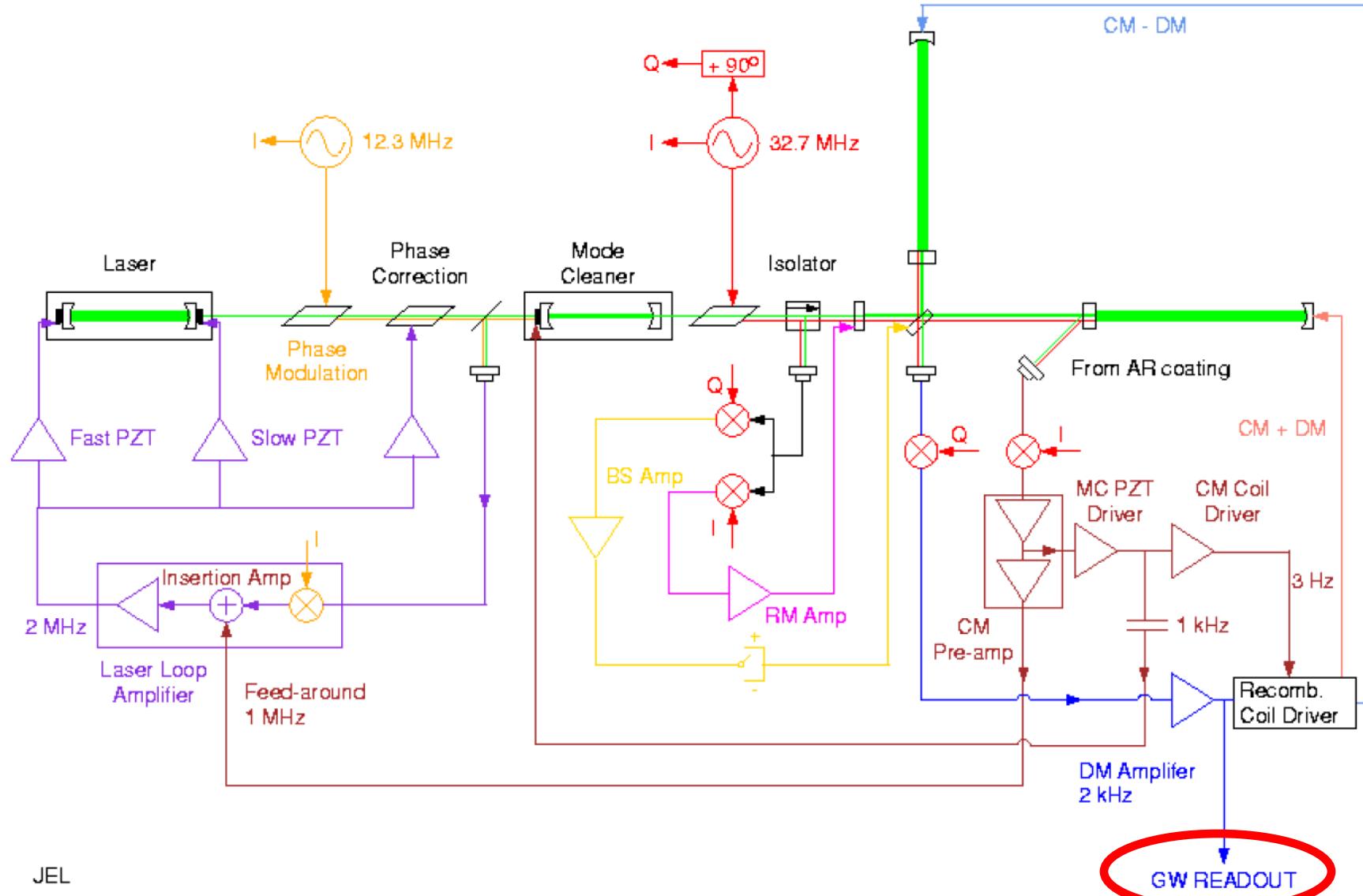
Homodyne (DC)  
readout used for  
Adv. LIGO/Virgo

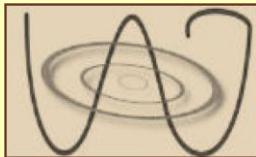
Measure intensity  
variations

Arm lengths offset



# LIGO Length Sensing and Control (RF Readout)





# Gravitational-Wave Data

**Data = Instantaneous estimate of strain for each moment in time**

i.e. demodulated channel sensitive to arm length difference

That's not the whole story – we'll come back to calibration later

**Digitized discrete **time series** recorded in computer files**

$$(t_j, x_j)$$

LIGO and GEO **sampling rate**: 16384 Hz  $\equiv f_s$



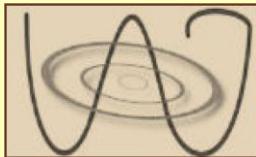
VIRGO sampling rate: 20000 Hz

Synchronized with GPS time signal

Common “frame” file format (\*.gwf)

**Many auxiliary channels recorded too**

Total data volume: a few megabytes per second per interferometer



# Leap Second Coming at End of June

<http://hpiers.obspm.fr/iers/bul/bulc/bulletinc.dat>

Paris, 5 January 2012

Bulletin C 43

To authorities responsible  
for the measurement and  
distribution of time

UTC TIME STEP  
on the 1st of July 2012

A positive leap second will be introduced at the end of June 2012.  
The sequence of dates of the UTC second markers will be:

2012 June 30,	23h 59m 59s
2012 June 30,	23h 59m 60s
2012 July 1,	0h 0m 0s

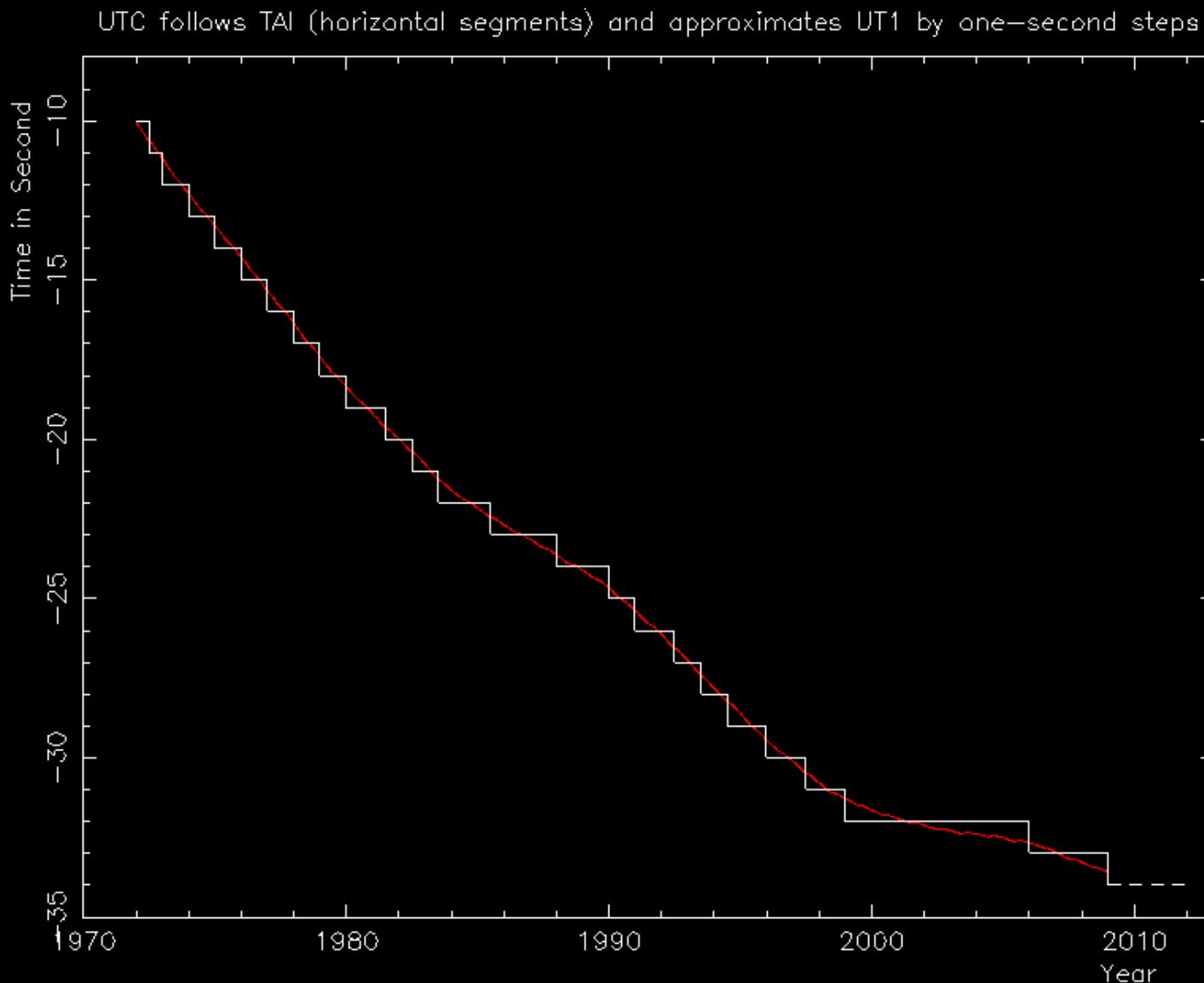
The difference between UTC and the International Atomic Time TAI is:

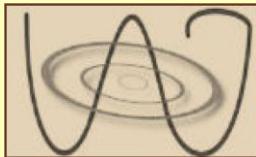
from 2009 January 1, 0h UTC, to 2012 July 1 0h UTC : UTC-TAI = - 34s  
from 2012 July 1, 0h UTC, until further notice : UTC-TAI = - 35s

Leap seconds can be introduced in UTC at the end of the months of December or June, depending on the evolution of UT1-TAI. Bulletin C is mailed every six months, either to announce a time step in UTC or to confirm that there will be no time step at the next possible date.



# Leap Seconds – Historical





# Relevance of the Sampling Rate

Is 16384 Hz a high enough sampling rate ?

**The Sampling Theorem:**

Discretely sampled data with sampling rate  $f_s$  can completely represent a continuous signal which only has frequency content below the **Nyquist frequency**,  $f_s / 2$

**GW signals of interest to ground-based detectors typically stay below a few kHz**

e.g. binary neutron star inspiral reaches ISCO at ~1 to 1.5 kHz



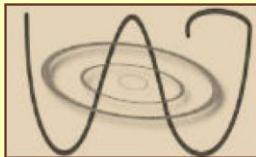
Neutron star  $f$ -modes: ~3 kHz

Black hole quasinormal modes: ~1 kHz for  $10 M_\odot$

Some core collapse supernova signals could go up to several kHz

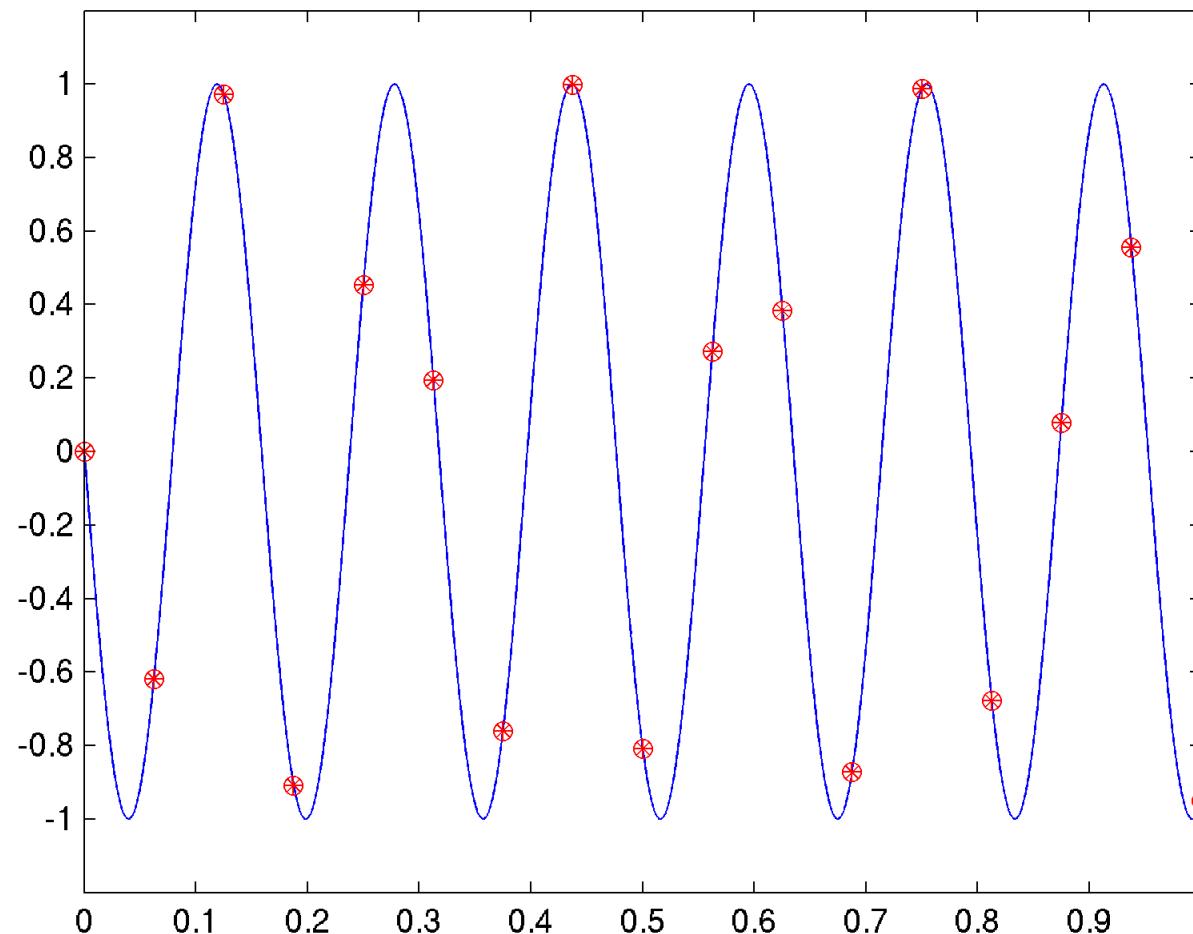
**What if the signal extends above Nyquist frequency?**

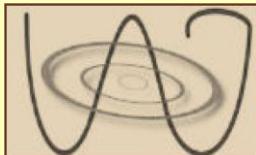
Higher frequencies are “aliased” down to lower frequencies



# Aliasing

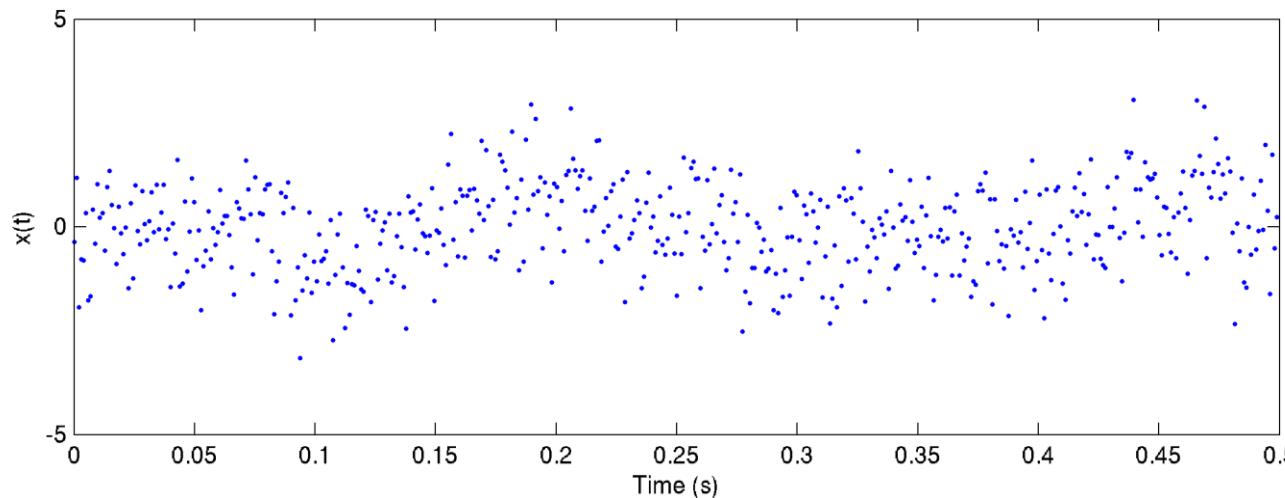
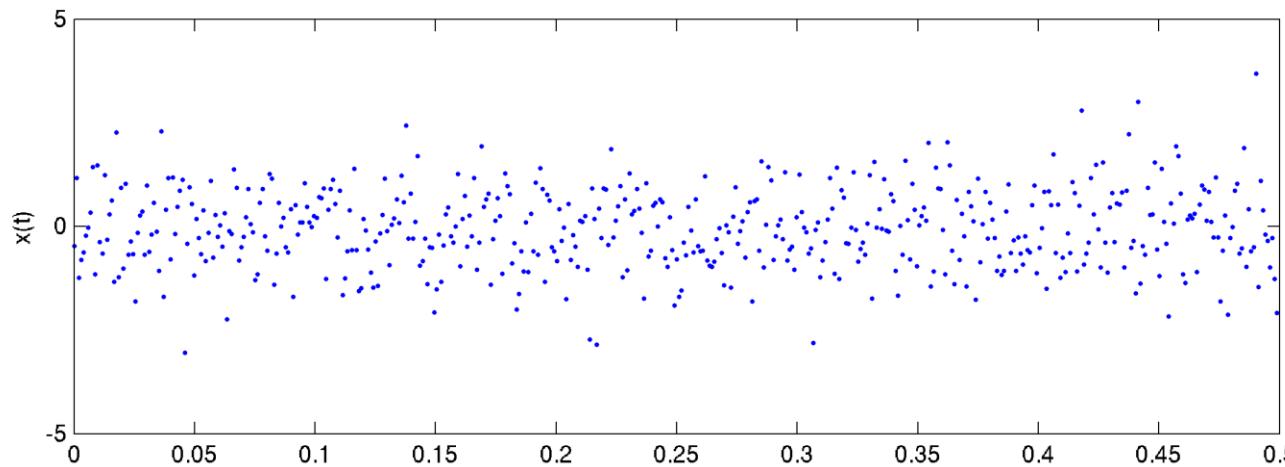
$f_s = 16 \text{ Hz}$  ; signal frequency = 9.7 Hz

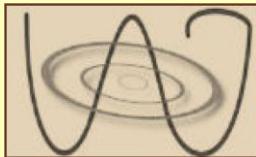




# Characterizing Noise

Noise is random, but its *properties* can be characterized





# Possible Properties of Noise

**Stationary** : statistical properties are independent of time

**Ergodic** process: time averages are equivalent to ensemble averages

**Gaussian** : A random variable follows Gaussian distribution

For a single random variable, 
$$p(x) = \frac{1}{\sqrt{2\pi\sigma_x^2}} \exp\left[-\frac{1}{2}\frac{(x - \mu_x)^2}{\sigma_x^2}\right]$$

More generally, a set of random variables (e.g. a time series) is Gaussian if the joint probability distribution is governed by a covariance matrix

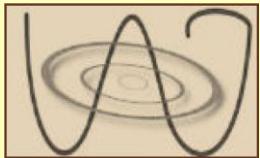
$$C_{xij} := \langle x_i x_j \rangle - \langle x_i \rangle \langle x_j \rangle$$

such that

$$p(x_1, x_2, \dots, x_N) = \frac{1}{(2\pi)^{N/2} \sqrt{\det C_x}} \exp\left[-\frac{1}{2} \sum_{i,j=0}^{N-1} C_{xij}^{-1} (x_i - \mu_{xi})(x_j - \mu_{xj})\right]$$

**White** : Signal power is uniformly distributed over frequency

⇒ Data samples are uncorrelated



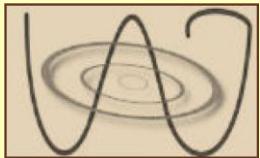
# Frequency-Domain Representation of a Time Series

## Fourier transform

$$\begin{aligned}\widetilde{x}(f) &= \int_{-\infty}^{\infty} dt \, x(t) e^{-i2\pi ft} \\ \Rightarrow \quad x(t) &= \int_{-\infty}^{\infty} df \, \widetilde{x}(f) e^{i2\pi ft}\end{aligned}$$

**A linear function, complex in general**

**Defined for all positive *and* negative frequencies**



# Frequency-Domain Representation of a *Discrete, Finite* Time Series

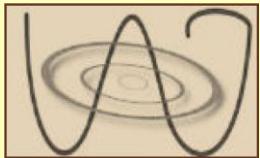
Time series  $x_j$  with  $N$  samples at times  $t_j = t_0 + j \Delta t$

Discrete Fourier transform

$$\begin{aligned}\tilde{x}_k &:= \sum_{j=0}^{N-1} x_j e^{-i2\pi jk/N} \\ \Rightarrow x_j &= \frac{1}{N} \sum_{k=-N/2}^{N/2-1} \tilde{x}_k e^{i2\pi jk/N}\end{aligned}$$

Frequency spacing is **inversely proportional to  $N$**

Efficient way to calculate complete discrete Fourier Transform:  
**Fast Fourier Transform (FFT)**



# Power Spectral Density

**Parseval's theorem:**

$$\int_{-\infty}^{\infty} dt |x(t)|^2 = \int_{-\infty}^{\infty} df |\tilde{x}(f)|^2$$

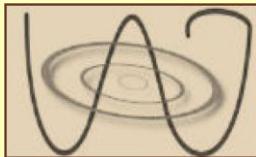
⇒ Total energy in the data can be calculated in either time domain or frequency domain

$|\tilde{x}(f)|^2$  can be interpreted as energy spectral density

**When noise (or signal) has infinite extent in time domain, can still define the power spectral density (PSD)**

$$\lim_{T \rightarrow \infty} \frac{1}{T} |\tilde{x}_T(f)|^2$$

**Watch out for one-sided vs. two-sided PSDs**



# Estimating the PSD

**Generally we need to determine the PSD empirically, using a finite amount of data**

**Simplest approach: FFT the data, calculate square of magnitude of each frequency component – this is a **periodogram****

For stationary noise, one can show that the frequency components are statistically independent

**This estimate is unbiased (has the correct mean), but has a large variance – so average several periodograms**

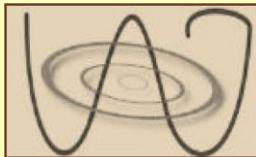
Alternately, smooth periodogram; give up frequency resolution either way

**Generally apply a “window” to the data to avoid **spectral leakage****

Leakage arises from the assumption that the data is periodic!

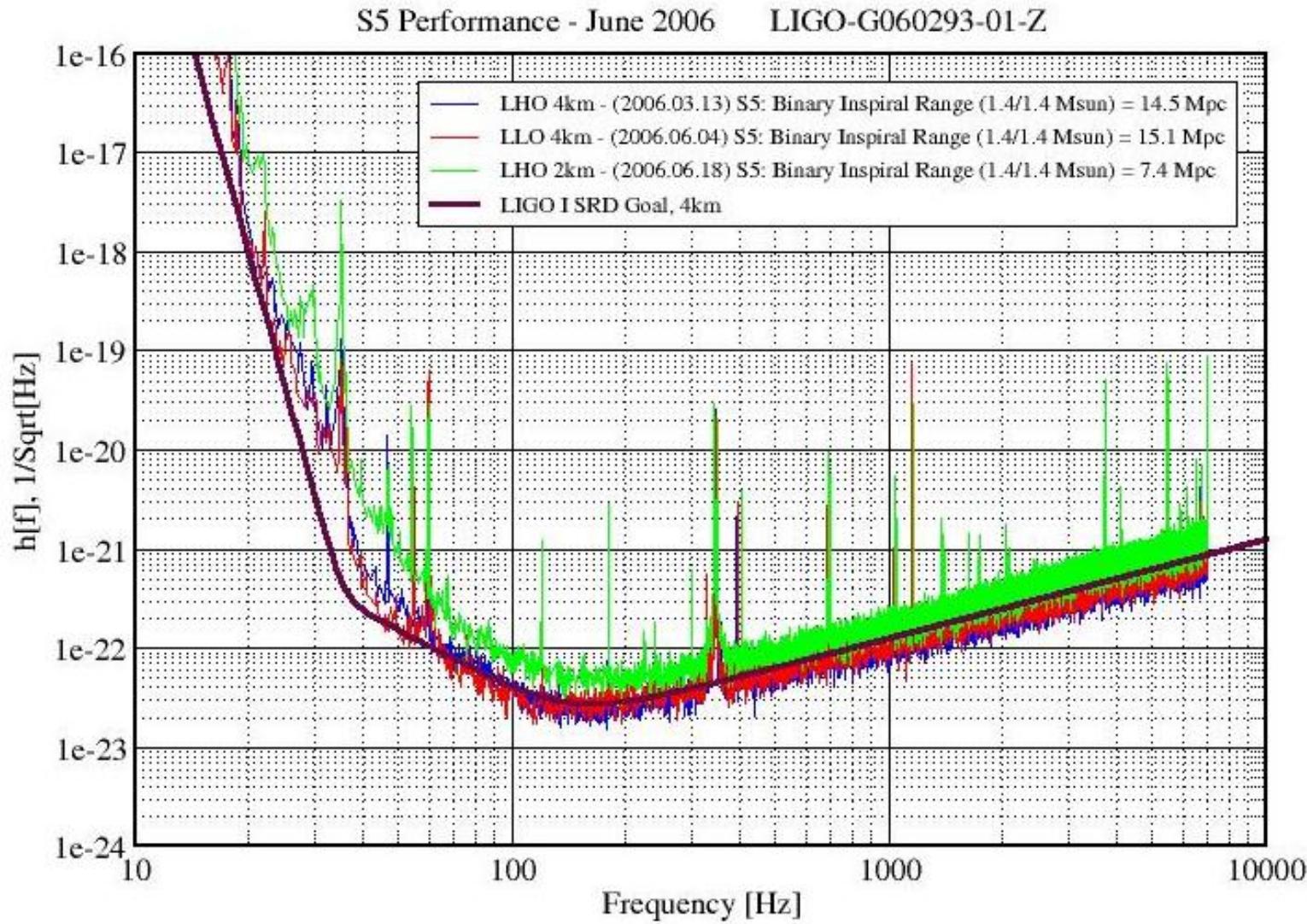
Tapered window forces data to go to zero at ends of time interval

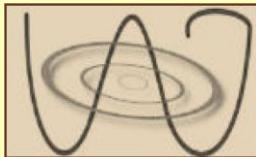
**Welch’s method** of estimating a PSD averages periodograms calculated from windowed data



# Amplitude Spectral Density of LIGO Noise

Strain Sensitivity for the LIGO 4km Interferometers





# Interpretation of Time Series Data

**Recorded data values are *not* simply proportional to GW strain**

A linear system, but that does not guarantee proportionality !

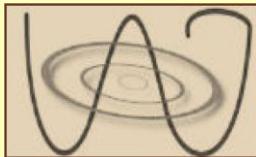
Frequency-dependent amplitude and phase relation (i.e. transfer function)

Instrumental and practical reasons

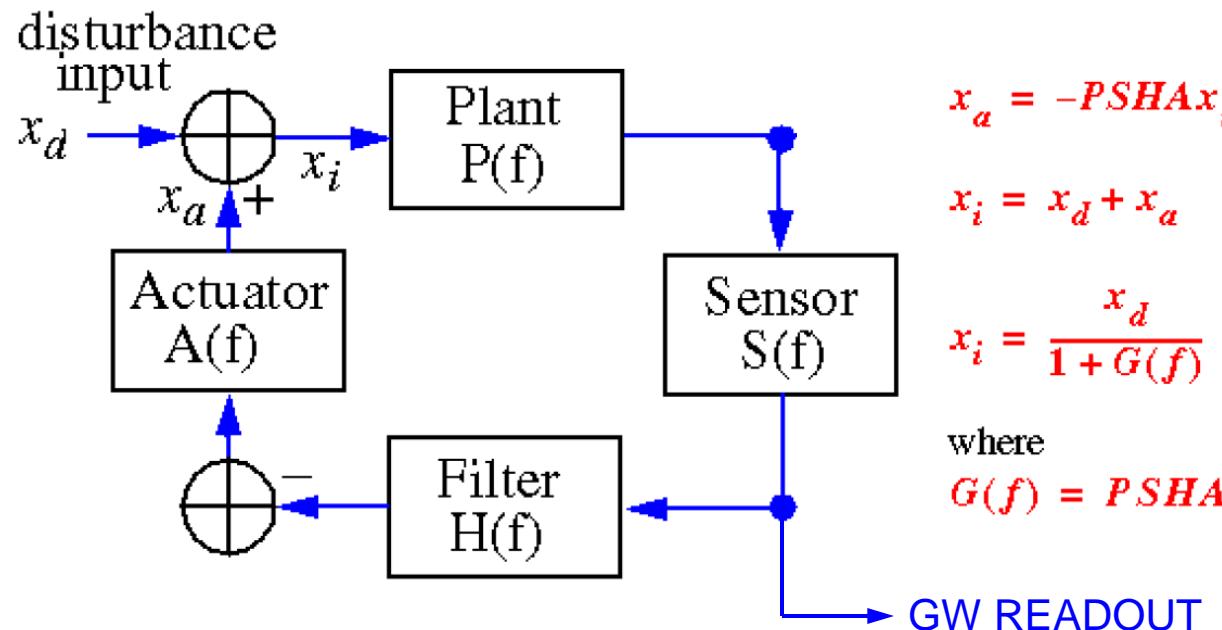
⇒ **Raw time series is a distorted version of GW strain signal**

e.g. a delta-function GW signal produces an output with a characteristic shape and duration (“**impulse response**”)

**Want to recover actual GW strain for analysis**



# Calibration

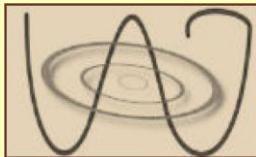


**Monitor  $P(f)$  continuously with “calibration lines”**

Sinusoidal arm length variations with known absolute amplitude

**Apply frequency-dependent correction factor to get GW strain**

$$h = (\text{GW READOUT}) \times \frac{1 + G(f)}{P(f) S(f)}$$



# Basics of Digital Filtering

**A filter calculates an output time series from a linear combination of the elements of an input time series**

## Finite Impulse Response (FIR) filter

Calculated *only* from the input time series

Typical form:  $y_i = b_0x_i + b_1x_{i-1} + b_2x_{i-2} + \dots + b_{N-1}x_{i-N}$

## Infinite Impulse Response (IIR) filter

Also uses prior elements of the output time series

e.g.  $y_i = b_0x_i + b_1x_{i-1} + b_2x_{i-2} + \dots + b_{N-1}x_{i-N} + a_1y_{i-1} + a_2y_{i-2} + \dots$

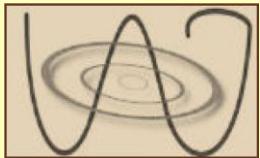
## Choice of coefficients determines transfer function

Many filter design methods, depending on goals

## Causality and phase lag

Linear-phase and zero-phase filters

**Watch for transient in filter output at beginning of data stream!**



# Applications of filtering

**High-pass, low-pass, band-pass, band-stop, etc.**

## **Anti-aliasing for down-sampling**

Low-pass filter to cut away signal content above new Nyquist frequency

## **Whitening / Dewhitening**



# Time for some exercises ...

**Based on Matlab** – but the UTB laptops have Octave

**Work by yourself or with a partner**

**How to get help:**

- Ask me or a neighbor

- Use Matlab's/Octave's built-in help

- Consult a book – I have one here

**The items in the handout are intended as a guide**

- Feel free to explore !