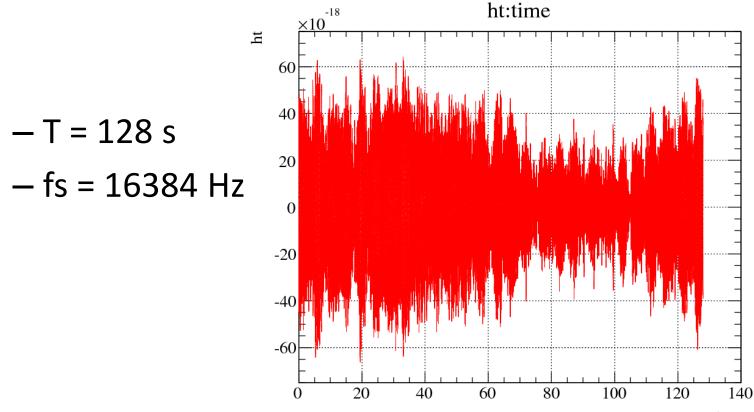
#### Line characterization

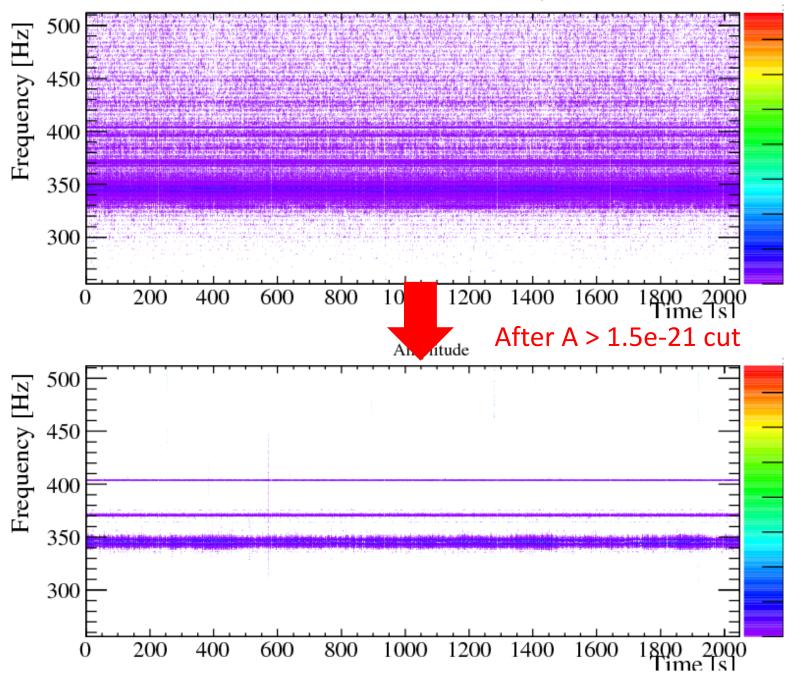
6 Oct. 2014 K. Ueno (Osaka Univ.)

## S6 data (example)

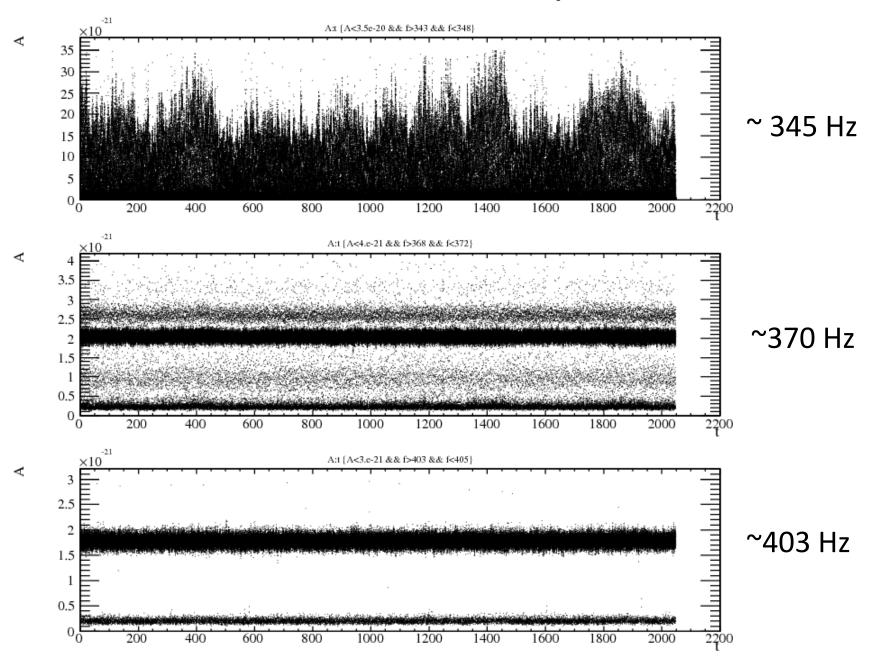
# I analyzed LIGO S6 data L-L1\_LDAS\_C02\_L2-959200000-128.gwf



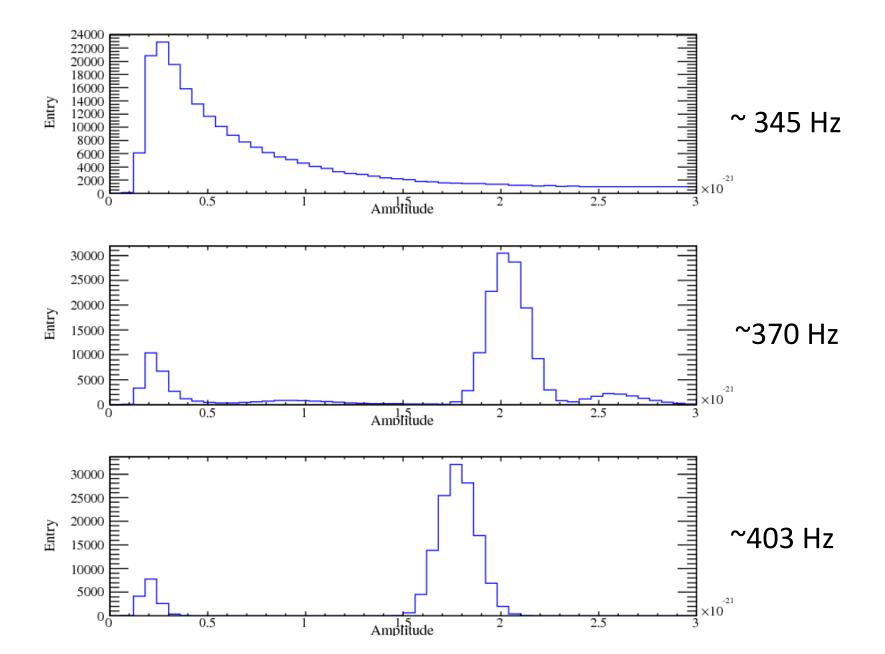
fs=1024Hz; frame=256; shift=16; #spectrum=20



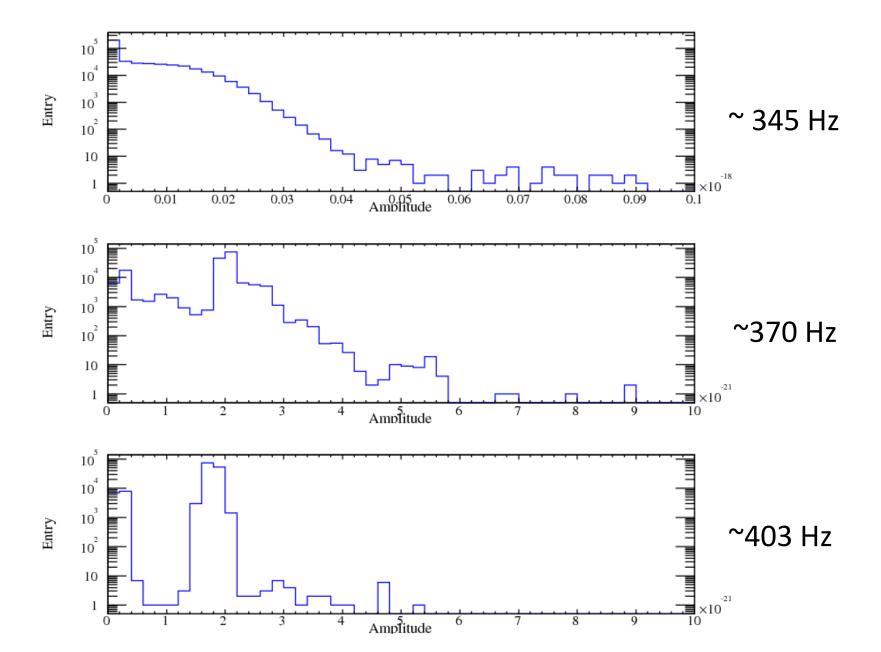
#### Time variation of amplitudes



#### Amplitude distribution



#### Amplitude distribution (logy)



#### Backups

# Non-Harmonic Analysis (NHA)

- NHA is a method to extract dominant frequency components in given time-series with high resolution, developed by a Toyama Univ. group which is led by Hirobayashi-san.
- The method itself is patent-protected and so not publicly available (even to the other KAGRA collaborators), but there are some papers which describe the outline of the method, especially,
  - <u>"Noise reduction for periodic signals using high resolution frequency</u> <u>analysis</u>" Yoshizawa et al., EURASIP Journal on Audio, Speech, and Music Processing 5, 2011

# Basic Algorithm

- 1) FFT the given time series x(n) and find the frequency which gives the largest amplitude.
- 2) You somehow minimize the following cost function about A, f, and  $\phi$ ,

$$F(A, f, \varphi) = \frac{1}{N} \sum_{n=0}^{N-1} \left\{ x(n) - A \cos\left(2\pi \frac{f}{f_s} n + \varphi\right) \right\}^2$$

starting from A and f estimated at 1). This is just a least square fit with a sinusoidal function.

- Once the best-fit values of A,f, and φ are found, the waveform of converged spectrum is subtracted from x(n).
- 4) Repeat the procedure  $1 \sim 3$  as many times as one would like.

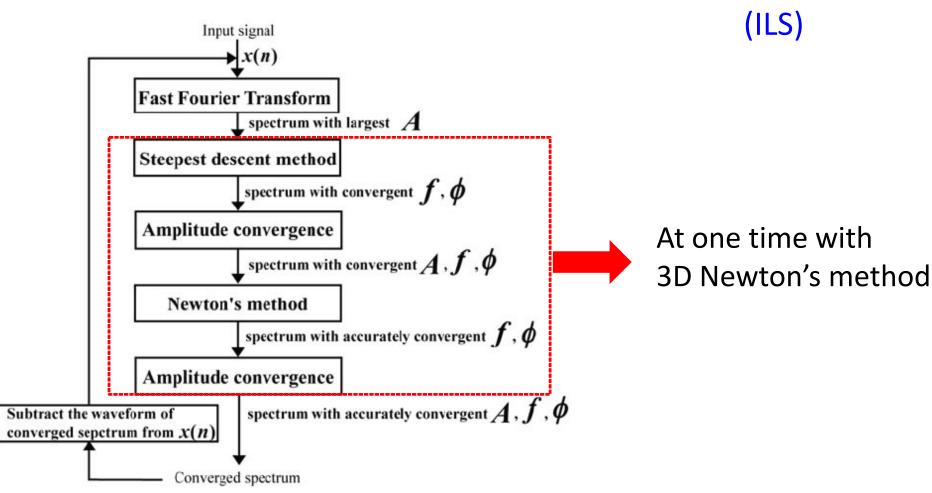
## Problems

- According to Hirobayashi-san, you need some specific environments such as MATLAB or GPGPU to execute their NHA code.
- In order to do data analysis ourselves efficiently, it's better to develop our own simple version of the code instead of using their real one.
- So I tried to implement this method from the scratch. But when I followed the paper faithfully, the solutions were found not to be stable and soon diverge.
- Instead of following the whole procedure written in the paper, I took another way to meet the same purpose. I'll refer to the new way (next slide) as "Iterative Least Square (ILS)" instead of NHA to avoid confusion.

## Difference from NHA's paper

Iterative Least Square

NHA paper

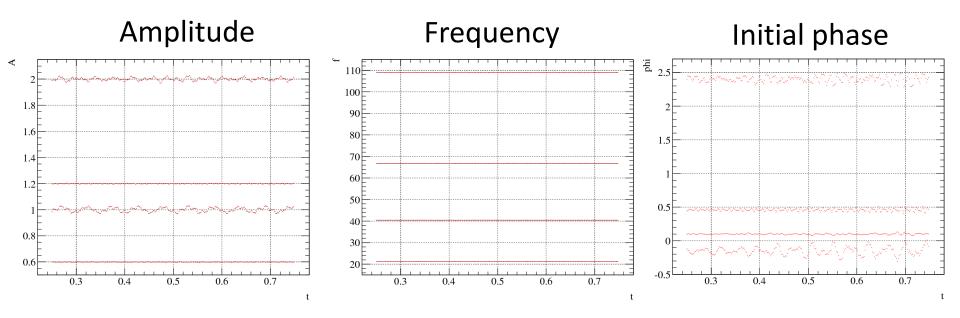


#### Test Example

- To test the performance of the ILS code, I artificially generated a signal time-series with some input parameters and tried to reconstruct them with ILS.
- The time-series is composed of four different signals, each of which has a constant amplitude and frequency.
- The sampling rate is 512 Hz, and the duration is set to 1 s.

Amplitude	Frequency [Hz]	Initial phase [rad]
2.0	66.7	-0.15
1.2	109.0	0.46
1.0	40.4	2.40
0.6	21.3	0.10

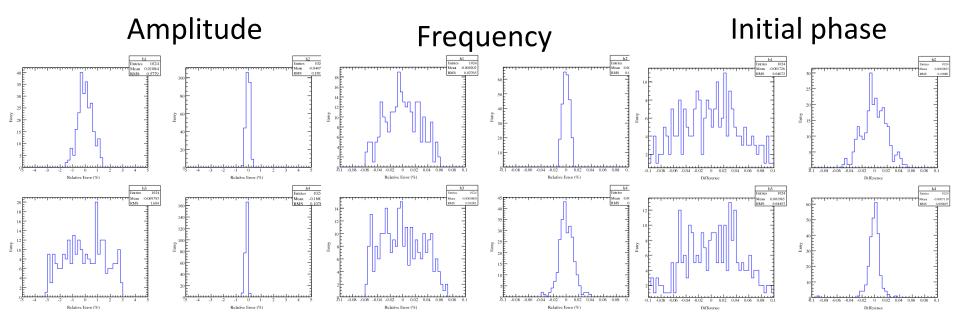
## Result (absolute)



While frequencies were well reconstructed, some of the amplitudes were not, which are caused by signal-signal interference.

What is already clear: when there are multiple signals within the same duration, the minimum of the cost function does not always correspond to the true amplitude value, and the solution shift a little bit.

### Result (resolution)



Frequencies were reconstructed within 0.1%, while amplitudes 3%.