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Can someone provide me a Python program to calculate fundamental frequency and other frequencies of an unknown signal with 0.1 or 0.01 Hz accuracy?

Question

Asked March 10, 2017 in the project [Numerical Relay](#)

I'm using BeagleBone Black to detect frequencies present in an unknown signal (whose fundamental frequency can range from 45 Hz to 55 Hz). I've used the following Python program to obtain a value for fundamental frequency of the unknown signal.

```
import numpy as np
```

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All Answers (3)

**Joerg Fricke** added an answer

March 13, 2017

Sorry, I can't help you with Python, but I think your problem lies on a more general level: If you apply FFT to a number of samples taken in the time interval T then the output represents the frequencies $0, f, 2f, \dots$ with $f = 1/T$. If you need a resolution of 0.1 Hz you have to feed into the FFT samples taken within 10 s; a resolution of 0.01 s would be based on samples taken in 100 s. If this are too much data to process in real time, the next question is: What is the maximum frequency you are interested in? With 16 kHz sampling frequency your upper limit would be 8 kHz.

If your signal has a spectrum containing virtually only *one* frequency with minor noise (as on the power lines, as long as there are no control signals added) then a much simpler algorithm would be (as used in frequency counters if the frequency is low):

Determine the time T between each second zero voltage crossing, and compute $f = 1 / T$. By computing the average value of several results you could eliminate noise.

To determine the presence of other frequencies above the line frequency, you could first send your samples through a (digital) high pass filter, and then decide if the amplitude is above a threshold value.

But if you really need a spectrum, and your hardware is too slow, I'm afraid you have to use a Digital Signal Processor.

I hope this helps.

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2 Recommendations



Fernando Soares Schlindwein added an answer

March 13, 2017

A few comments:

When you sample a signal $x(t)$ to be $x(nT)$ the sampled series is sometimes written as ... $x(n)$ as a 'lazy notation' for $x(nT)$. Similarly in frequency we have $X(k/(NT))$, and this is sometimes written as $X(k)$ as the 'lazy notation'. The frequency resolution is

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2 Recommendations



U. Dreher added an answer

March 13, 2017

Joerg Fricke is absolutely correct regarding the necessity to process samples from 10 s resp. 100 s long intervals. In your special case a high sampling frequency might be counter-productive: if your interest is on the grid frequency, a comparably low sampling frequency might help to solve the problem (reduce the number of samples to process) - provided you've got an appropriate anti-aliasing filter. Otherwise aliasing eff

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2 Recommendations



Sandipan Dey

4.69 · University of Maryland, Baltimore County

Answer

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