



Low-Complexity Iterative Sinusoidal Parameter Estimation

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Context

Context: Approximating a signal as a sum of sinusoids

Audio compression

Audio processing

Problem:

Estimating sinusoidal parameters is a non-linear problem

Non-linear problems are computationally expensive

Must often be done in real-time with few resources

Solution

Linearising the problem as much as possible

Using an iterative solver

Sinusoidal Parameters

A sinusoid is defined as

Amplitude }
Phase } Can be estimated linearly (e.g. FFT)
Frequency } **Non-linear**

We consider a fourth parameter

Linear amplitude modulation

$$\tilde{x}(n) = h(n) \sum_{k=1}^N \left(A_k + n A'_k \right) \cdot \cos \left((\theta_k + \Delta\theta_k) n + \phi_k \right)$$

Workaround: Linearisation

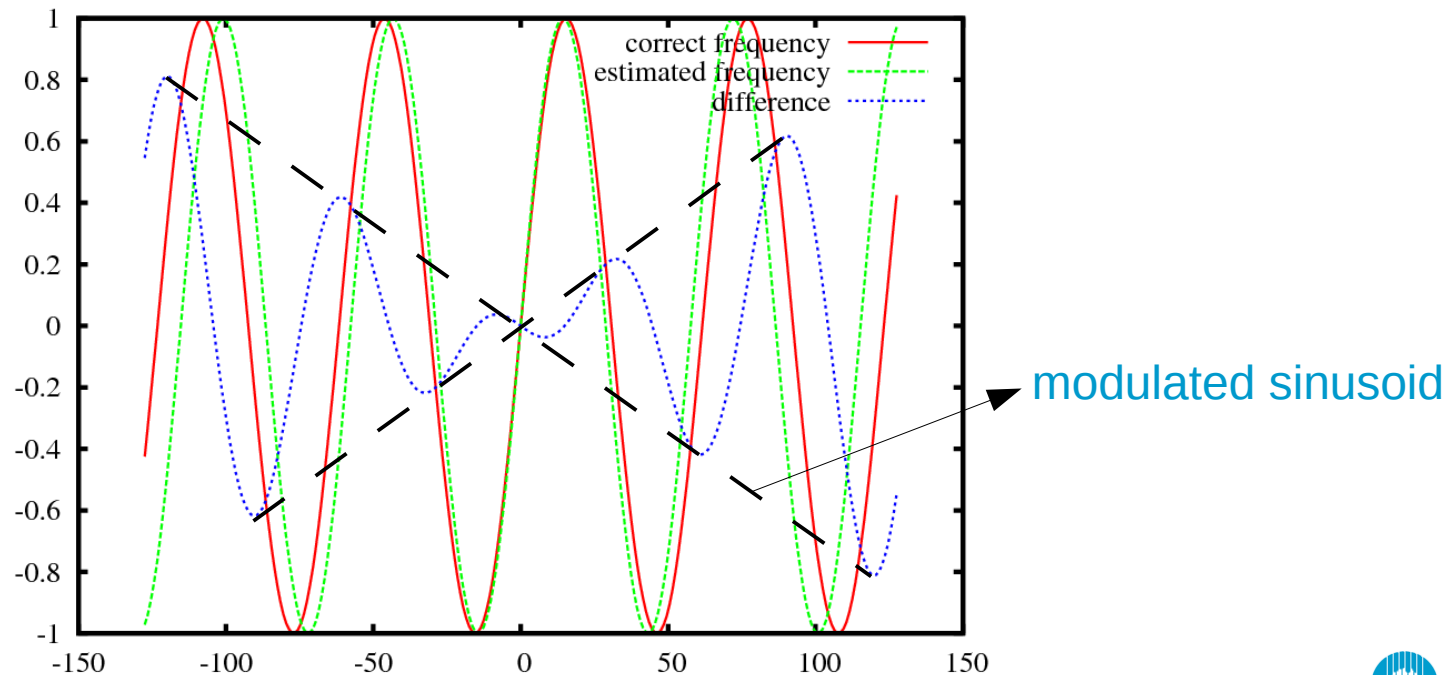
Hypothesis #1: We have an initial estimate of frequencies

Obtained though a lower resolution FFT

From previous time frame

Hypothesis #2: The error on the estimate is small

Result: Frequency behaves *almost* linearly



Linear System

Any sinusoid can be expressed as the sum of 4 basis functions

$$\tilde{x}(n) = h(n) \sum_{k=1}^N c_k \cos \theta_k n + s_k \sin \theta_k n \\ + d_k n \cos \theta_k n + t_k n \sin \theta_k n$$

Parameters are (neglecting 2nd order terms):

$$A_k = \sqrt{c_k^2 + s_k^2}$$
$$\phi_k = \arg(c_k - js_k)$$
$$A'_k = \frac{d_k c_k + s_k t_k}{A_k}$$
$$\Delta\theta_k = \frac{d_k s_k - t_k c_k}{A_k^2}$$

Linear Solver

Direct solver is $O(LN^2)$

Iterative method: Gauss-Seidel in $O(LN)$

Basis is nearly orthogonal, guaranteed convergence

Successive projections of the error on the basis functions

First cos/sin terms, then modulated terms (faster convergence)

```
for all iteration  $i=1 \dots M$  do  
  for all sinusoid component  $k = 1 \dots 4N$  do  
     $\Delta w_k^{(i)} \leftarrow \mathbf{a}_k^T \mathbf{e}$   
     $\mathbf{e} \leftarrow \mathbf{e} - \mathbf{a}_k \Delta w_k^{(i)}$   
     $w_k^{(i)} \leftarrow w_k^{(i-1)} + \Delta w_k^{(i)}$   
  end for  
end for
```

Non-Linear Solver

Linear solution is imperfect when frequency error is too large

Non-linear solver adjusts the frequency for every iteration

- Compute one linear iteration

- Compute sinusoid parameters (including new frequency)

- Recompute the error based on the non-linear parameters

- Goto 1)

Complexity

- Only a small increase compared to the linear solution:

 - Need to re-compute the basis functions

 - Slightly longer to converge

Results

Frequency and amplitude accuracy (5 chirps with noise)

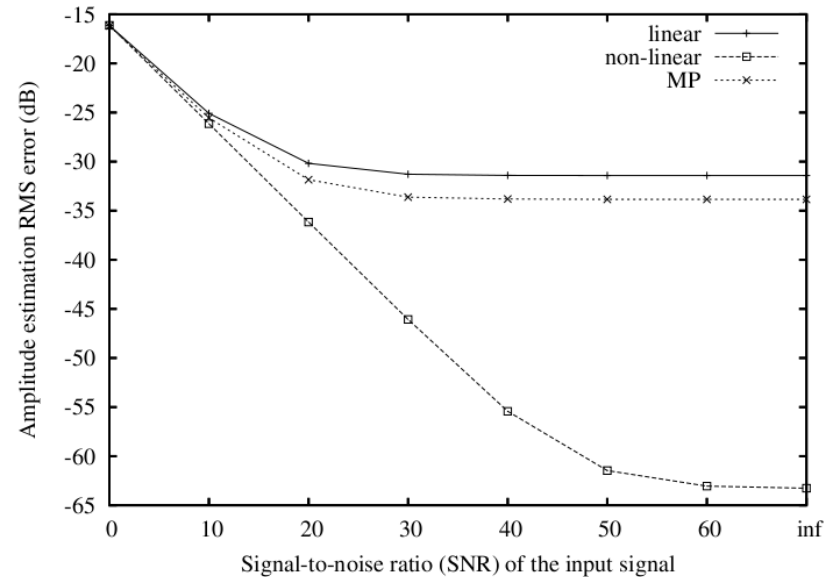
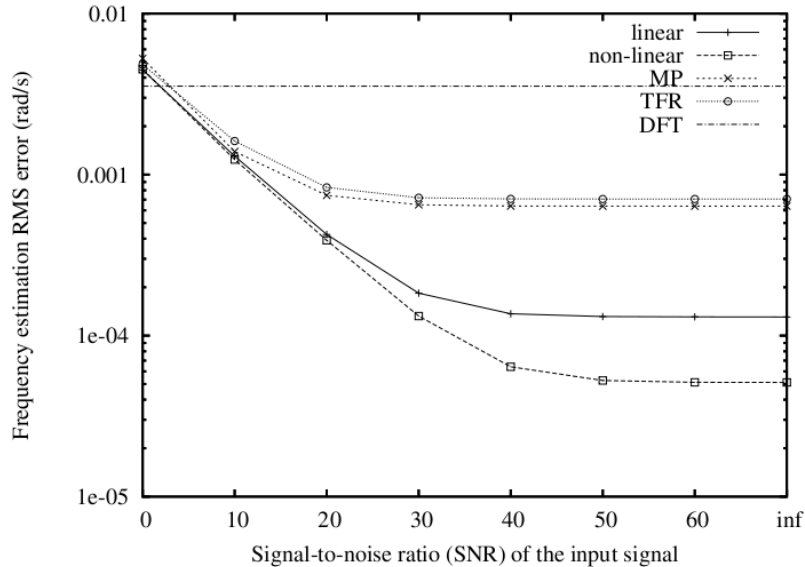
Linear solution

Non-linear solution

Matching pursuit

Time-frequency reassignment

DFT

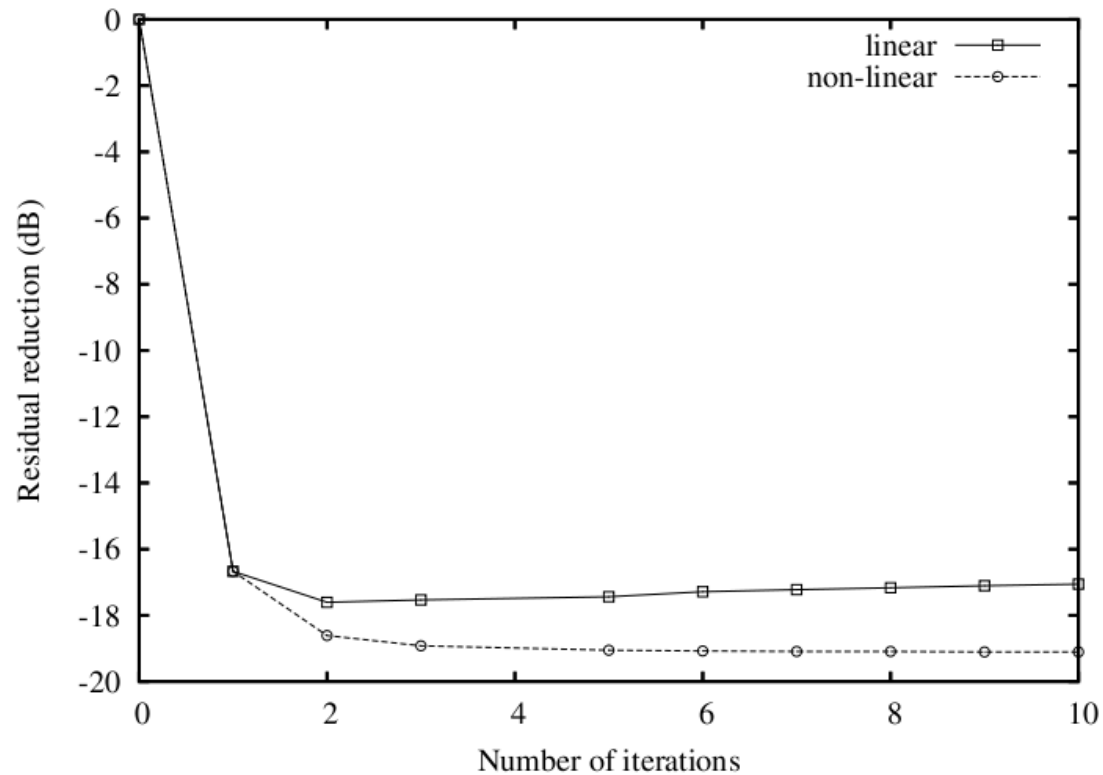


Convergence

Convergence on a music signal

Linear solution requires 2 iterations

Non-linear solution requires 3 iterations



Complexity

L: Length of the input data (256)

M: Number of iterations (2 for linear, 3 for non-linear)

N: Number of sinusoids (20)

P: Matching pursuit oversampling (32)

Algorithm	Complexity	Typical (Mflops)
MP (slow)	$4LN^2P$	3,300
MP (FFT)	$\frac{5}{2}LNP \log_2 LP$	1,300
linear (18)	$64N^3 + 32LN^2$	900
non-linear ([5])	$O(N^4 + LN^2)$	>500*
linear (proposed)	$(8M + 5)LN$	27
non-linear (prop.)	$(17M - 4)LN$	60

Conclusion

A low-complexity method for estimating sinusoid parameters

- Linearisation of the estimation problem

- Iterative solution (Gauss-Seidel)

- Optional non-linear solution

Reduces complexity by 1-2 orders of magnitude compared to other algorithms

Future work

- Improve initial frequency estimates

- Extend to the estimation of frequency modulation

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