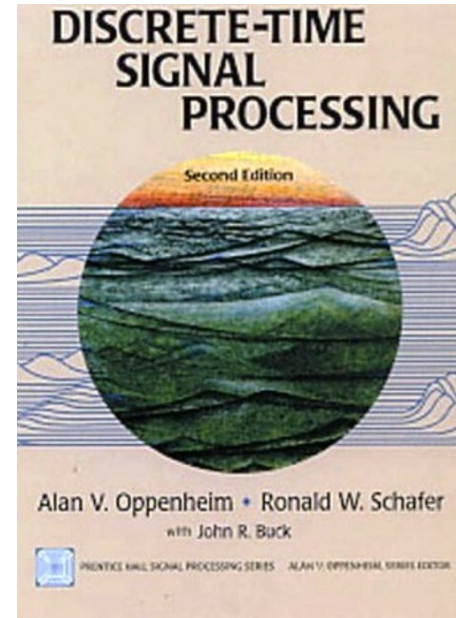


# Fast Fourier and z-transforms with almost arbitrary frequency spacing

BLonD Meeting  
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# Outline

- Discrete Fourier and z-transforms.
- Linear / circular convolution and time aliasing.
- Chirp Transform Algorithm (CTA).
- Non-Fourier: chirp-z transform (CZT).



# Setting the scene – assumptions

- Framework: discrete, finite length sequences.
- Exact calculation of the Fourier transform at almost arbitrary frequencies. (Focus: CTA.)
- Fast calculation, comparable to FFT.
- Potentially smaller size or higher resolution.

# Terminology

- Non-uniform DFT: NUDFT
  - The DFT evaluated on non-uniformly sampled data. (Type I.)
  - The DFT evaluated with non-uniform frequency spacing. (Type II.)
  - Or both. (Type III.)
- Calculated using uniformly sampled FFTs → NUFFT.
- NUDFT can be expressed as a z-transform.
- z-transform with “N log N” performance, a special NUFFT:
  - Chirp Transform Algorithm (CTA): DFT with non-uniform frequency spacing. (Type II.)
- Not a Fourier transform: Chirp-z transform (CZT).

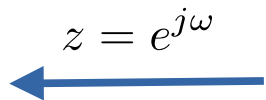
# Discrete Fourier and z-transforms

Fourier transform:

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n]e^{-j\omega n}$$

z-transform:

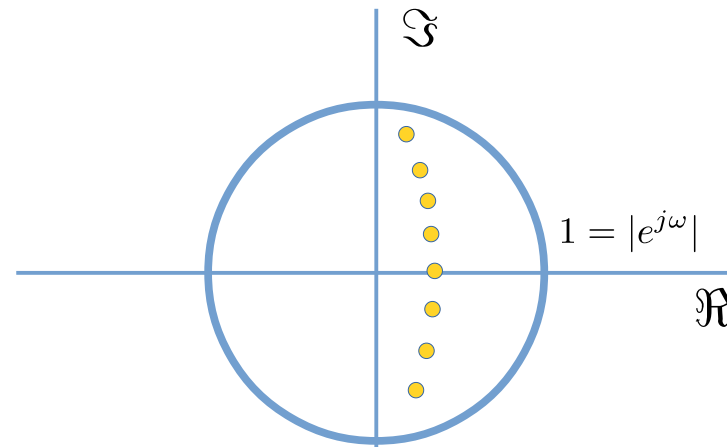
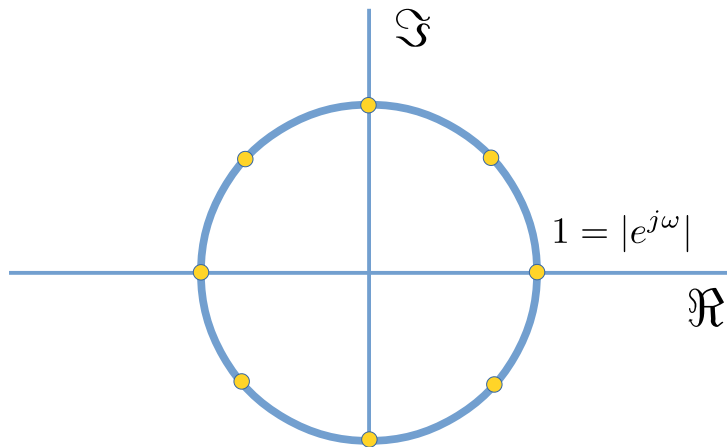
$$X(z) = \sum_{n=-\infty}^{\infty} x[n]z^{-n}$$



Uniform sampling of the Fourier transform:

$$X(e^{j(2\pi/N)k}), \quad 0 \leq k \leq N - 1, \quad N = 8$$

Sampling the z transform:



# Convolutions in DTSP

- Linear lengths  $N, M$ :

$$x_3[n] = \sum_{k=0}^{N+M-2} x_1[k]x_2[n-k] = x_1[n] * x_2[n]$$

*Linked to cross-correlation and LTI system response*

- Circular lengths  $N > M$ :

$$x_3[n] = \sum_{m=0}^{N-1} x_1[m]x_2[n-m] = x_1[n] \circledast x_2[n]$$

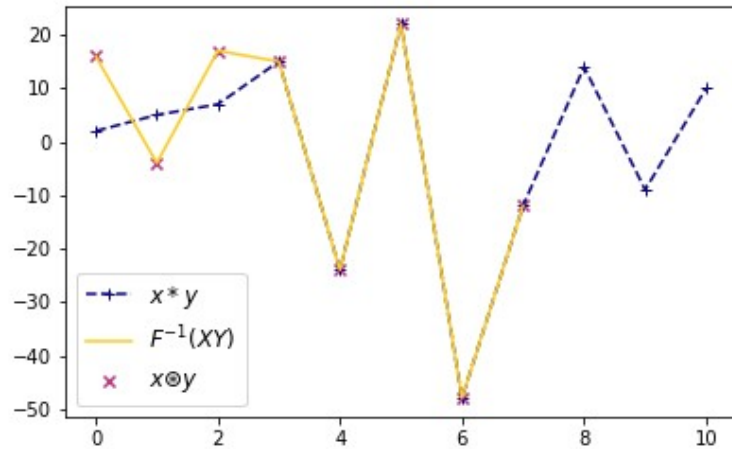
*Linked to Fourier transform*

- Relationship between linear and circular convolutions:
  - They may not be equal except for  $N-M+1$  samples.
  - Consequence: if FFT is used to compute the linear convolution, it must be at least  $N+M-1$  long to avoid time aliasing.

# Linear / circular convolution, FFT time aliasing example

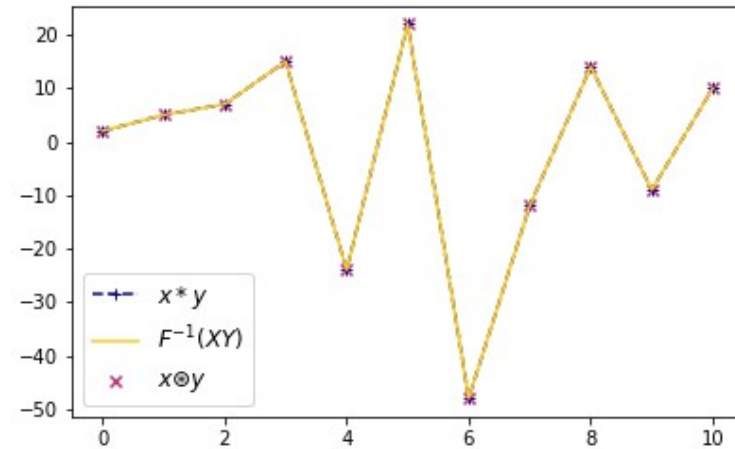
$$\text{len}(\mathbf{x}) = 4 \quad \text{len}(\mathbf{y}) = 8$$

Padding x to 8.



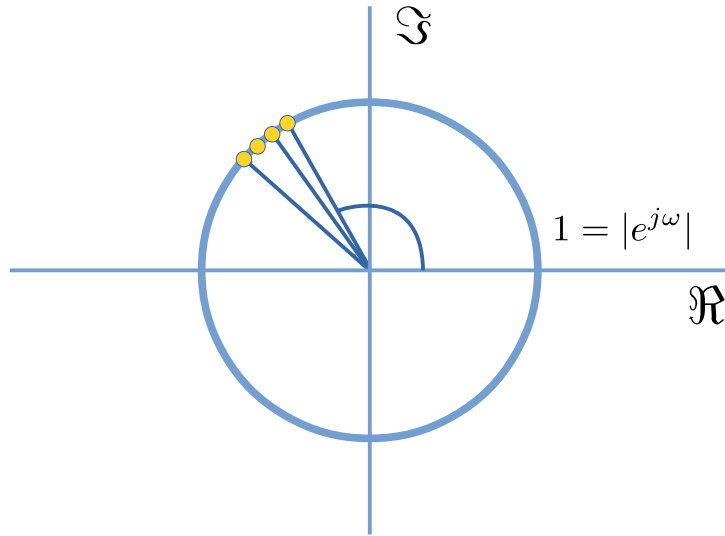
Time aliasing.

Padding x and y to 11.



No time aliasing.

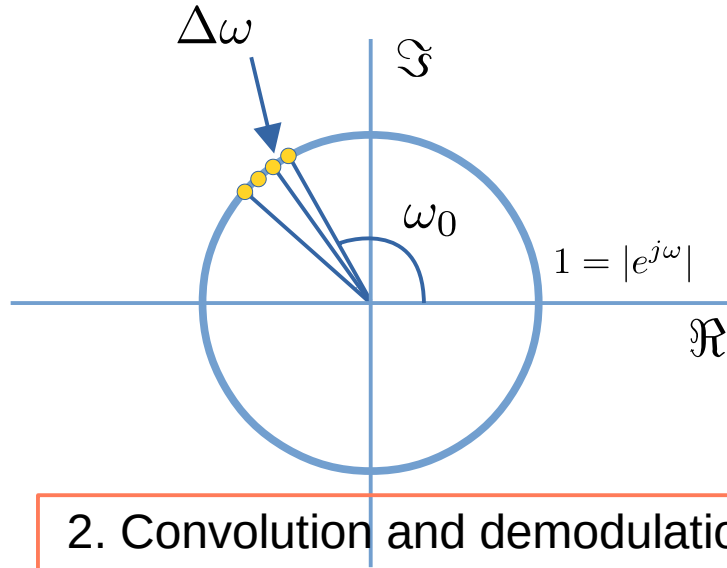
# Chirp Transform Algorithm (CTA)



- Generalisation of the DFT on the unit circle.
- Partial coverage of the unit circle.
- Spacing can be chosen.
- Special case of the CZT.
- Calculated using fast convolution FFT.



# Chirp Transform Algorithm (CTA)



$$\omega_k = \omega_0 + k\Delta\omega \quad k = 1, \dots, M - 1$$

$$X(e^{j\omega_k}) = \sum_{n=0}^{N-1} x[n]e^{-j\omega_k n} \quad W = e^{-j\Delta\omega}$$

DFT:  $M = N, \omega_0 = 0, \Delta\omega = 2\pi/N$

Substitution:  $kn = 1/2 [k^2 + n^2 - (k - n)^2]$

## 2. Convolution and demodulation

$$W^{k^2/2} \sum_{n=0}^{N-1} g[n]W^{-(k-n)^2/2},$$

Chirp  $\rightarrow$   $W^{k^2/2}$   $\leftarrow$   $W^{-(k-n)^2/2}$   $\leftarrow$  Circular conv.

## Input series

$$g[n] = x[n]e^{-j\omega_0 n}W^{n^2/2}$$

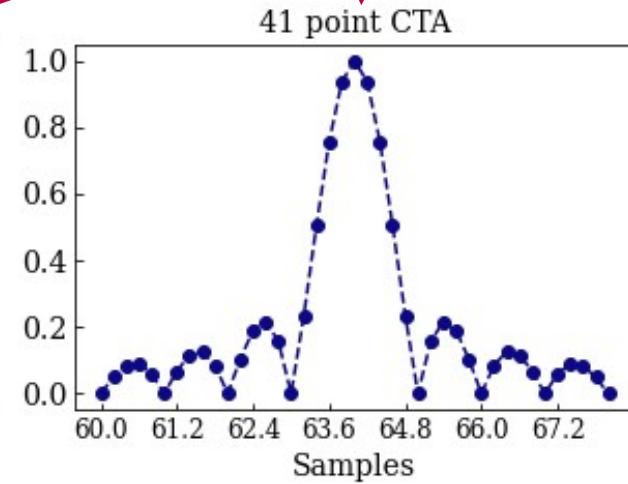
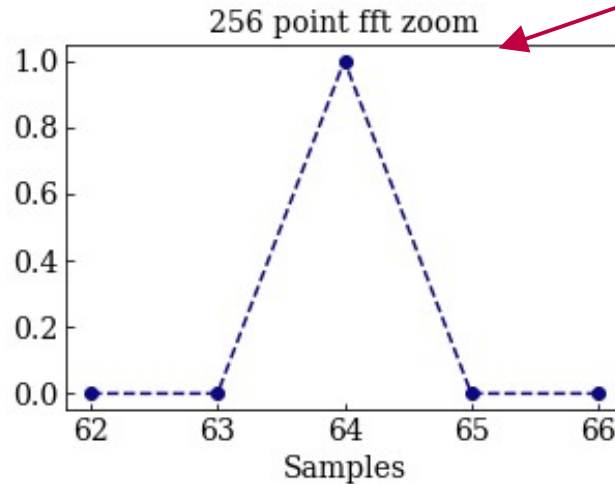
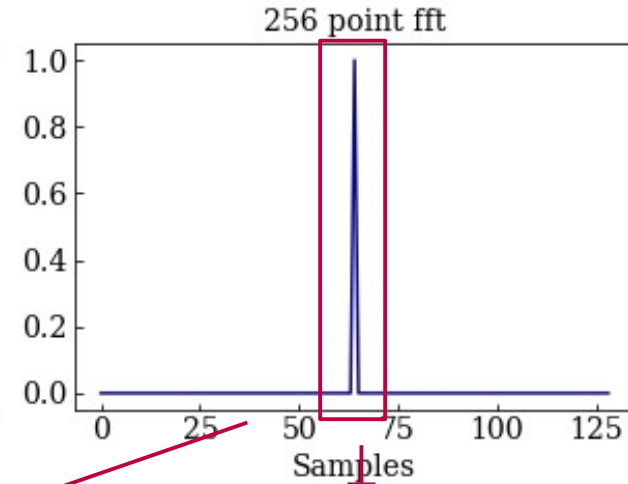
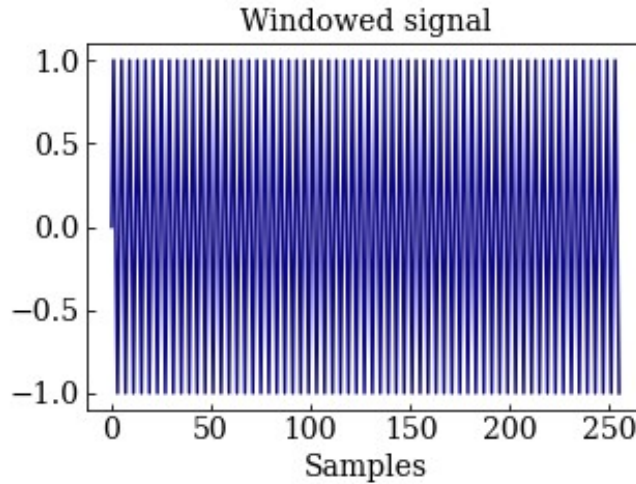
Frequency offset  $\rightarrow$   $e^{-j\omega_0 n}$

Chirp  $\rightarrow$   $W^{n^2/2}$

## 1. Chirp modulation

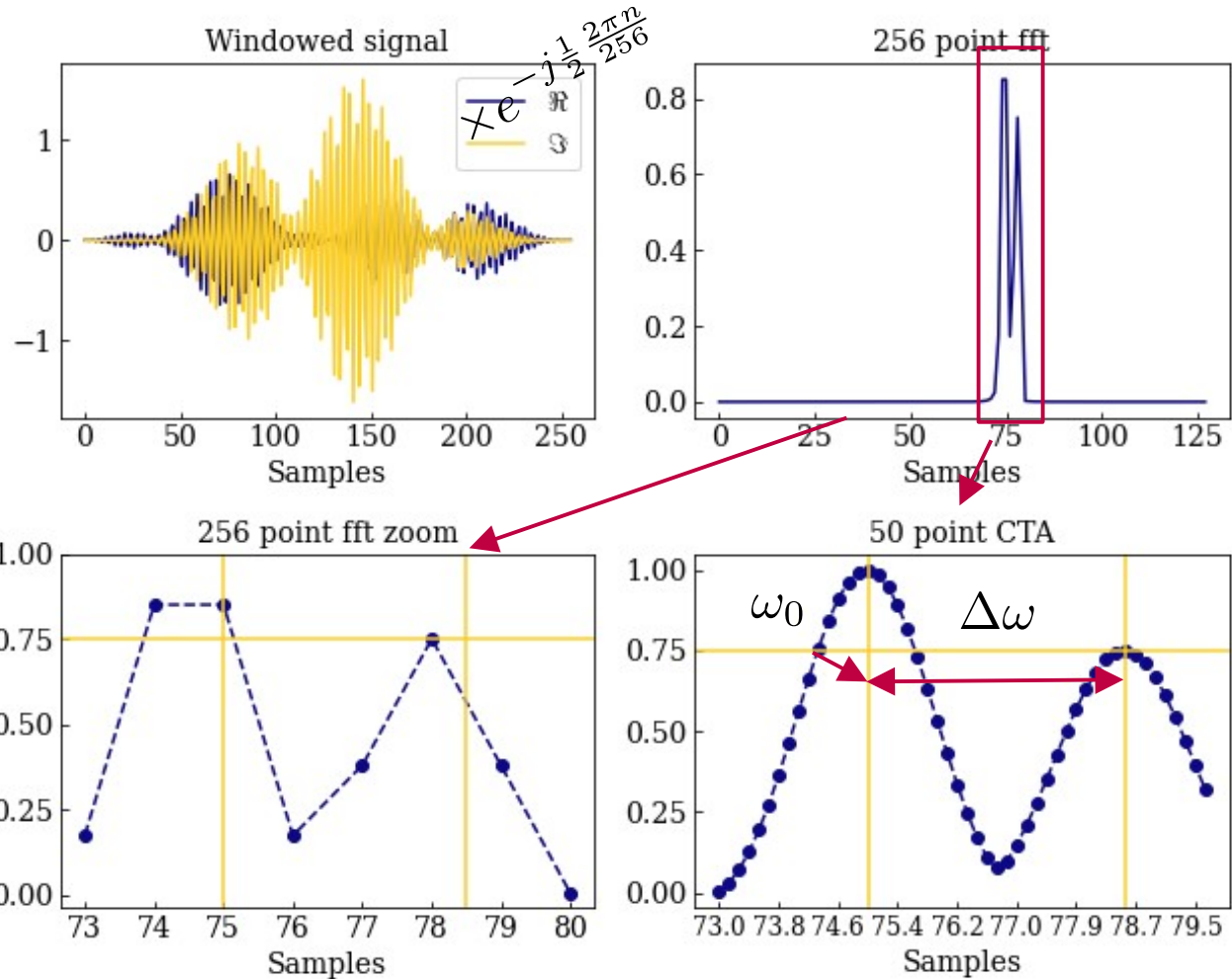
# Illustration 1

- 64/128 frequency sinusoid, amplitude 1. Signal length: 256 samples.
- Rectangular window + FFT.
- 41 point CTA around the peak (same 256 samples).
- Rectangular window transformed around 64.



# Illustration 2

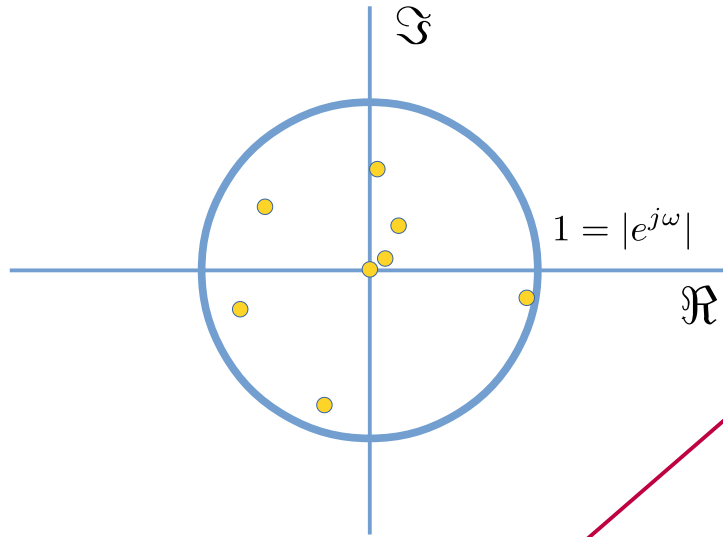
- 75/128 and 78.5/128 frequency sinusoids with 1 and  $\frac{3}{4}$  amplitude added. Signal length: 256 samples.
- Hanning window + FFT.
- CTA calculated for the 256(+50) point signal.  $\sim 6x$  resolution.
- Calculating an only two point CTA would have been sufficient in this example.



# CTA computation and use

- Cost: fast convolution needs  $2 \times \text{FFT}$  and  $1 \times \text{IFFT}$ , when implemented the usual way.
- CTA can avoid long FFTs by shortening the input or simplifying binning.
- CTA can zoom in on parts of the spectrum at relatively low computational cost.
- It is particularly suited to systems where the signal processing involves a convolution with a fixed impulse response.
- Typical use case: CCD sensors.

# Chirp z-Transform (CZT)



Same as CTA:  
slide 8.

- Going off the unit circle using a spiral.
- Getting closer to points in the z-plane.
- Amplifying buried poles.
- One, simple generalisation compared to CTA:

$$g[n] = x[n]e^{-j\omega_0 n} W^{n^2/2} \rightarrow g[n] = x[n]A^{-n}W^{n^2/2}$$

- “Radial gain” added to “tangential binning”.
- Traditional inverse via IFFT, but that is slow.
- Direct, fast inverse exists since 2019.
- CZT-ICZT became a similar pair to FFT-IFFT (N log N effort).

# Implementations in Python

- A summary: <https://gist.github.com/endolith/2783807>
- Scipy: <https://github.com/scipy/scipy/issues/4288>
- The new inverse computation is implemented:  
<https://pypi.org/project/czt/> Needs testing at the moment.
- Own code used for examples.

# Conclusions

- To overcome the limitations of uniform sampling and the unit circle, the discrete Fourier transform can be conveniently generalised using the z-transform and z-plane.
- CTA adds flexibility to the DFT by allowing a high resolution subsampling of an arbitrary arch on the unit circle. It still maintains  $N \log N$  performance.
- CZT, the generalised CTA, evaluates the z-transform along a spiral contour of the z-plane. Performance is  $N \log N$ .
- The recently invented  $N \log N$  performance ICZT completes the CZT-ICZT pair similar to the FFT-IFFT pair of transforms.

# Outlook for BLonD

- CTA / CZT are efficient tools for enhanced spectral analysis. → Mode analysis.
- Because the convolution is calculated via FFT, frequency domain multiplication can be implemented for some relevant samples only that can be spaced flexibly. → Potential gain for induced voltage.
- Next step: check performance with the [example from Markus](#) for fast induced voltage.



# References

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