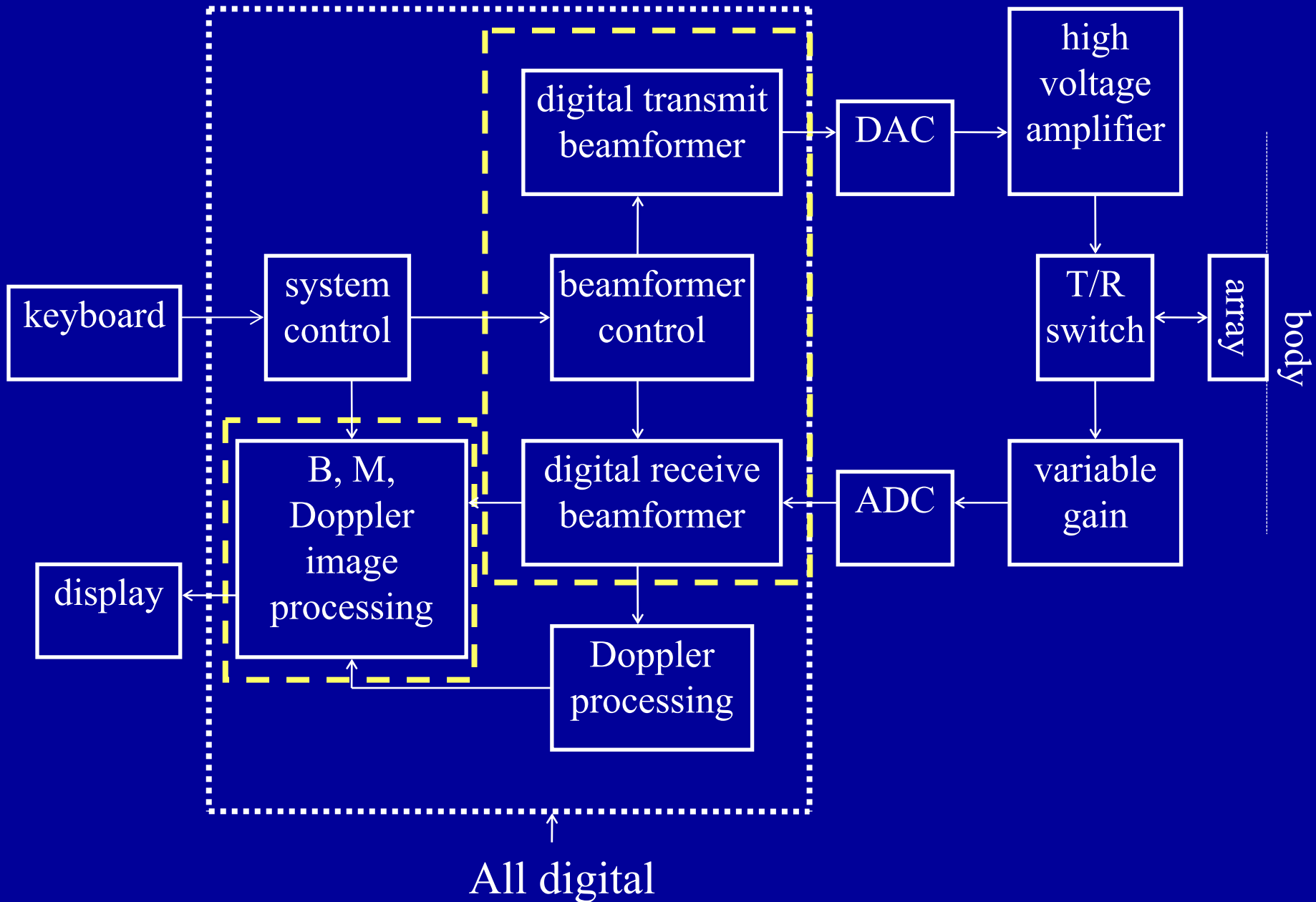
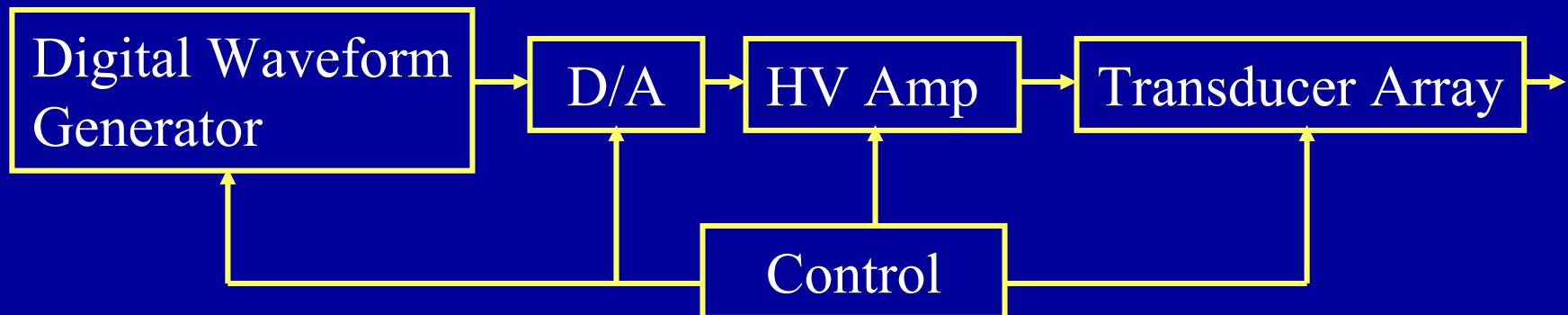


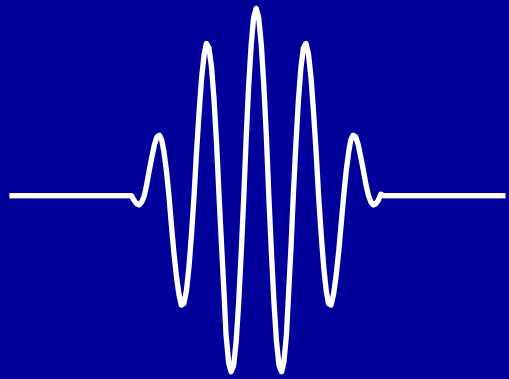
# Chapter 6: Real-Time Image Formation



# Generic Ultrasonic Imaging System

- Transmitter:
  - Arbitrary waveform.
  - Programmable transmit voltage.
  - Arbitrary firing sequence.
  - Programmable apodization, delay control and frequency control.



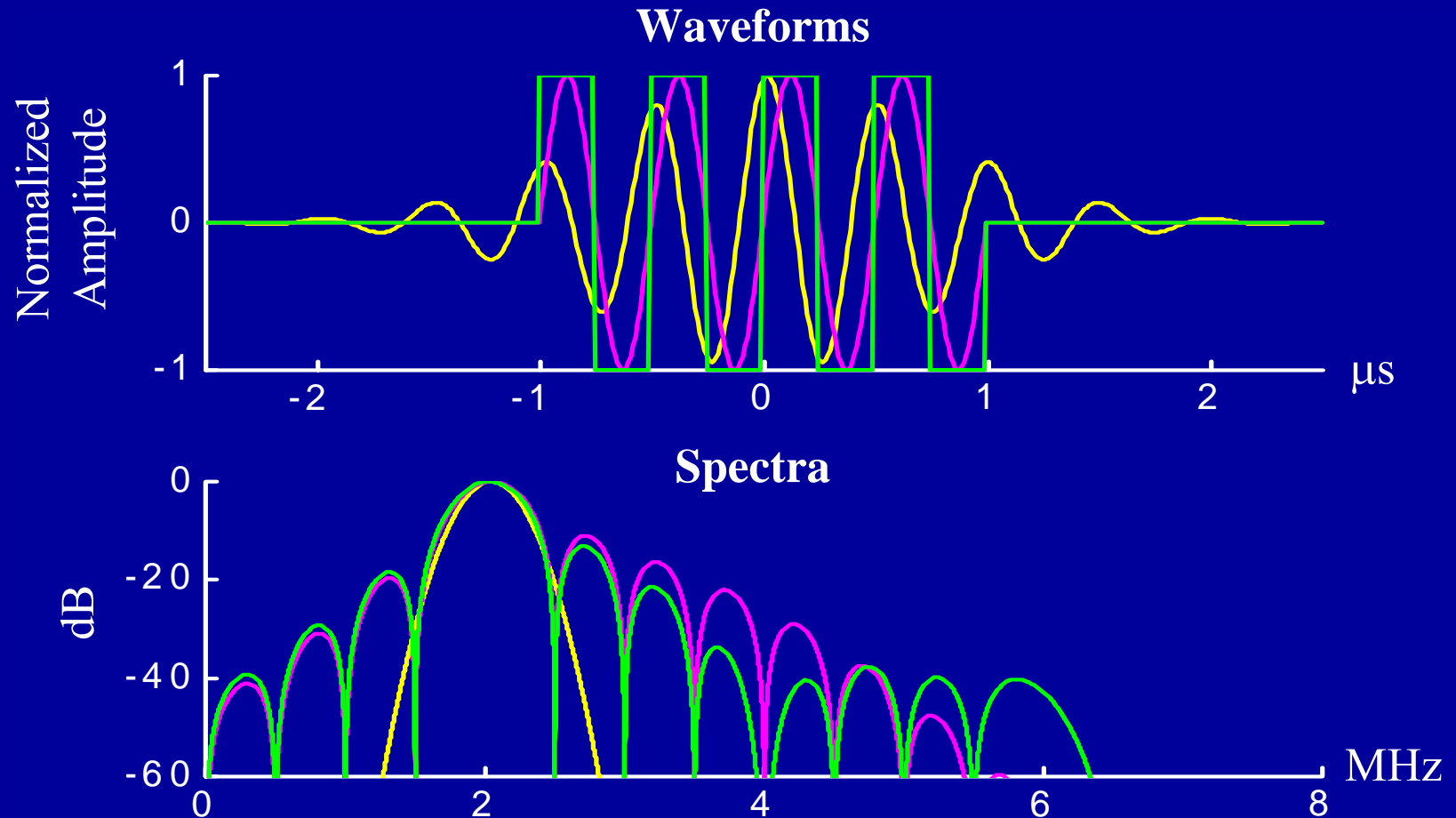


*OR*



# Transmit Waveform

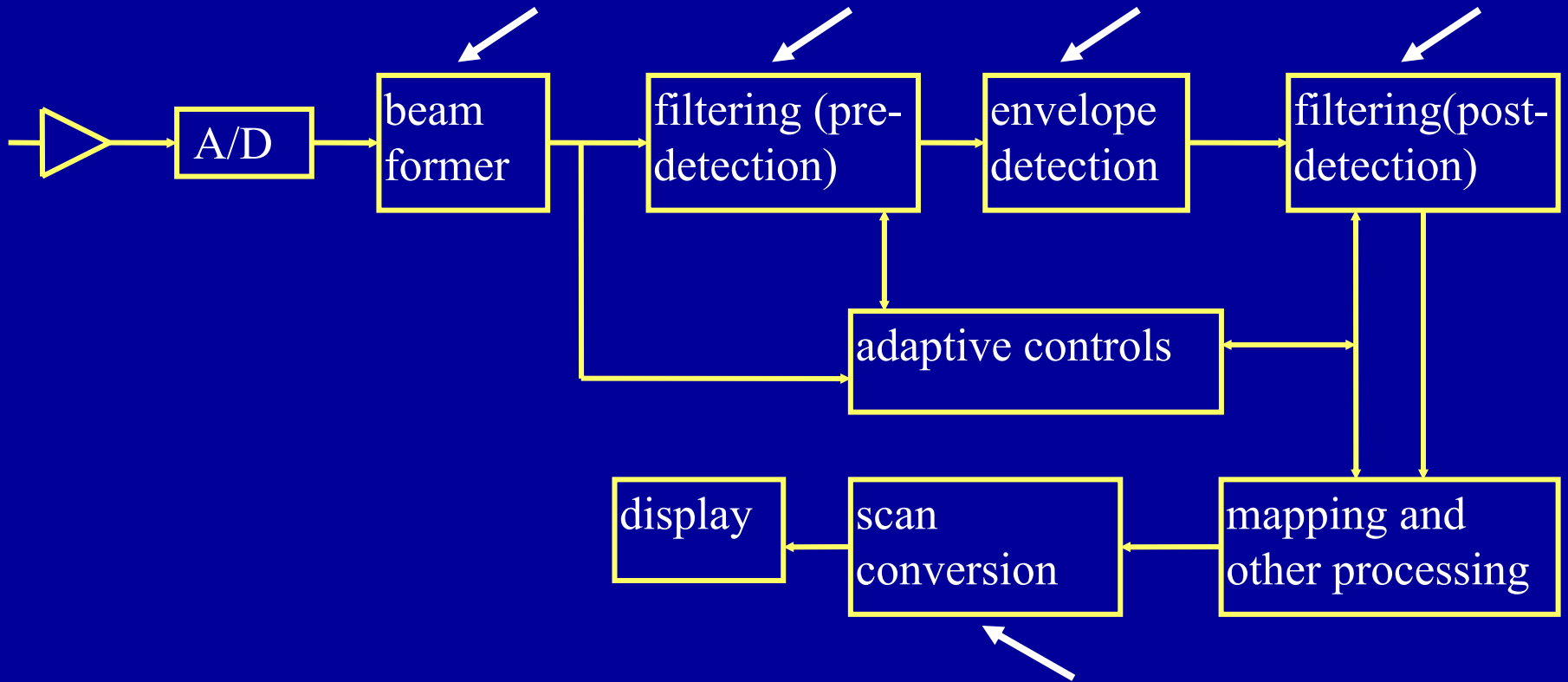
- Characteristics of transmit waveforms.



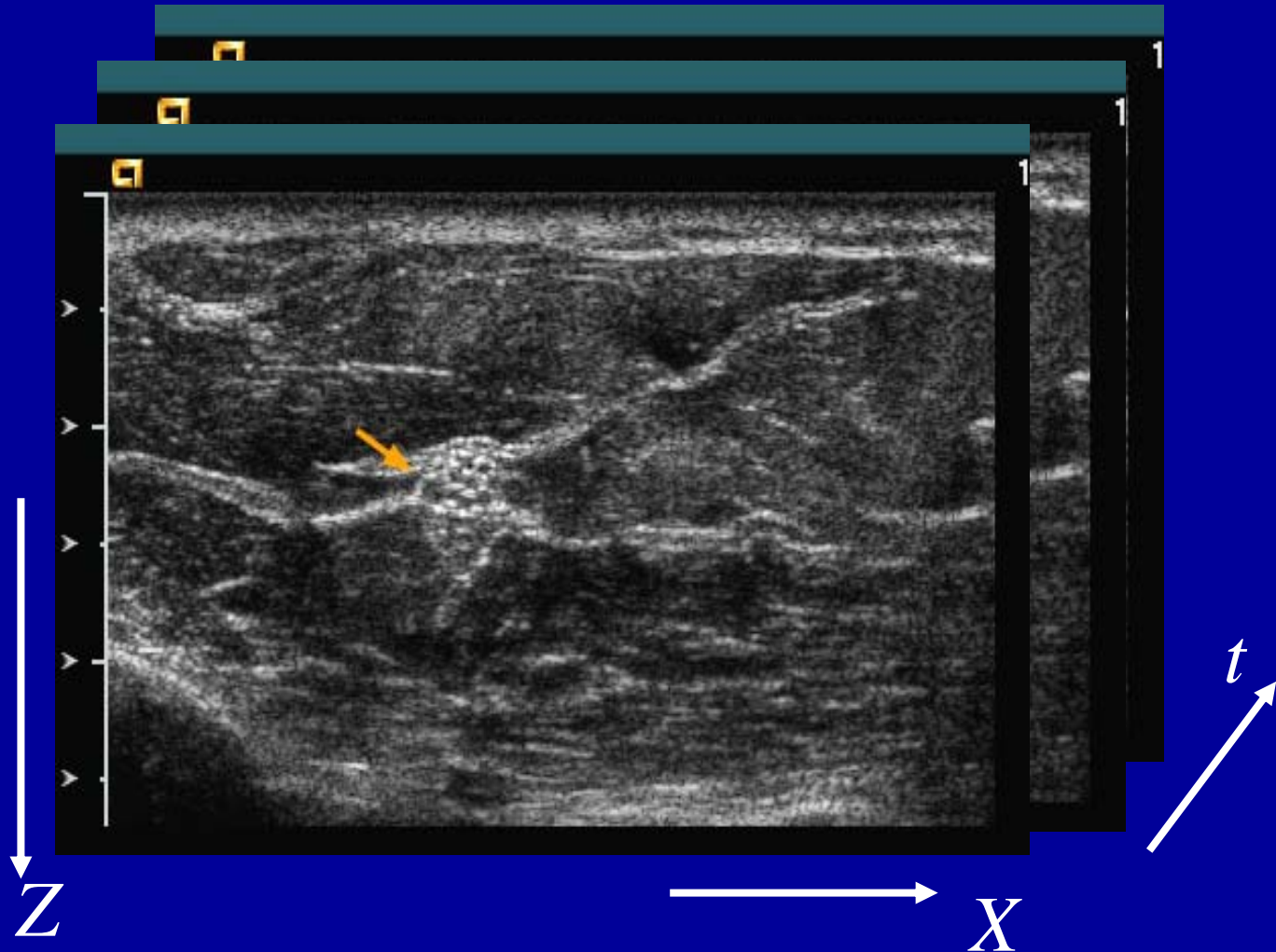
# Generic Ultrasonic Imaging System

- Receiver:
  - Programmable apodization, delay control and frequency control.
  - Arbitrary receive direction.
- Image processing:
  - Pre-detection filtering.
  - Post-detection filtering.
- Full gain correction: TGC, analog and digital.
- Scan converter: various scan format.

# Generic Receiver



# Pre-detection Filtering



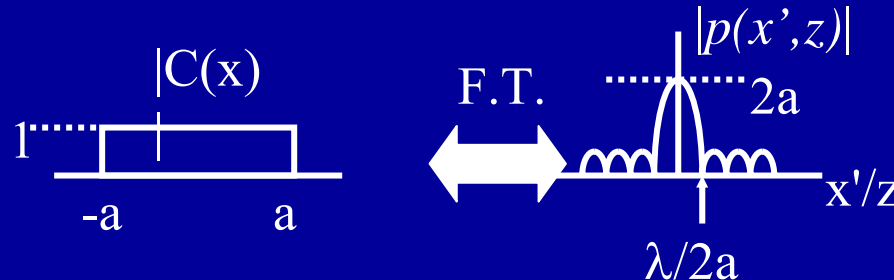


# Pre-detection Filtering

- Pulse shaping. ( $Z$ )
- Temporal filtering. ( $t$ )
- Beam shaping. ( $X$ )
  - Selection of frequency range. ( $Z \rightarrow X$ )

$$B(x', z) = \int T(x', z, \omega) R(x', z, \omega) A(\omega) d\omega$$

- Correction of focusing errors. ( $X \rightarrow X'$ )

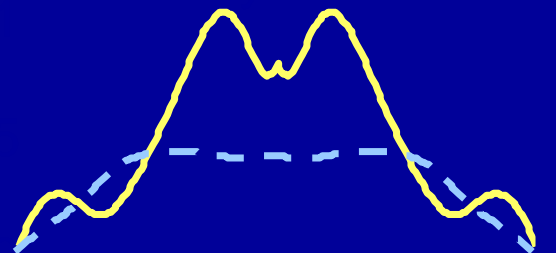
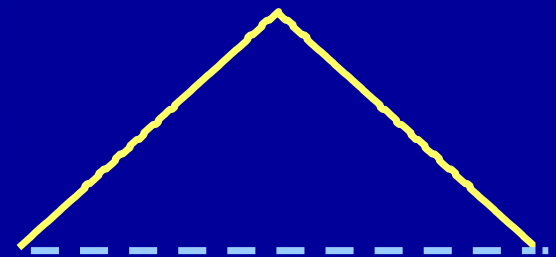
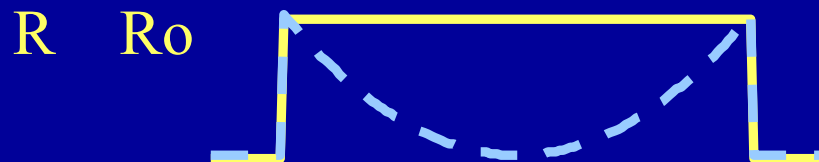
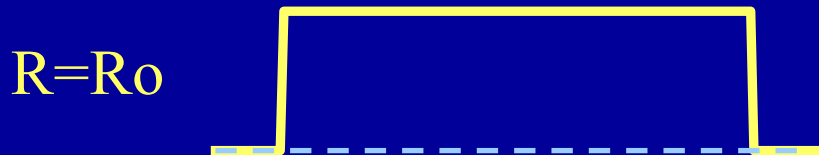


# Pulse-echo effective apertures

- The pulse-echo beam pattern is the multiplication of the transmit beam and the receive beam
- The pulse-echo effective aperture is the convolution of transmit and receive apertures

For C.W.

$$C(x) = |C(x)| e^{\frac{jkx^2}{2} \left( \frac{1}{R} - \frac{1}{R_0} \right)}$$



# Post-Detection Filtering

- Data re-sampling (Acoustic  $\rightarrow$  Display).
- Speckle reduction (incoherent averaging).
- Feature enhancement.
- Aesthetics.
- Post-processing:
  - Re-mapping (gray scale and color).
  - Digital gain.

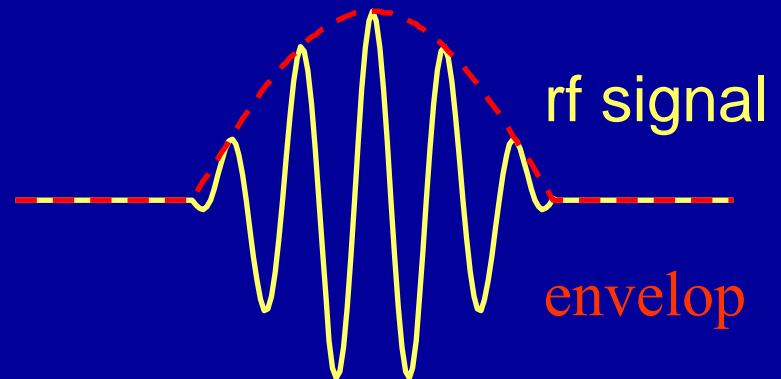
# Envelope Detection

- Demodulation based:

$$S(t) = A(t) \cos 2\pi f_0 t = \text{Re} \left\{ A(t) e^{j2\pi f_0 t} \right\}$$

$$A(t) = \text{LPF} \left\{ S(t) \cos 2\pi f_0 \right\}$$

$$D(t) = \text{abs} ( A(t) )$$

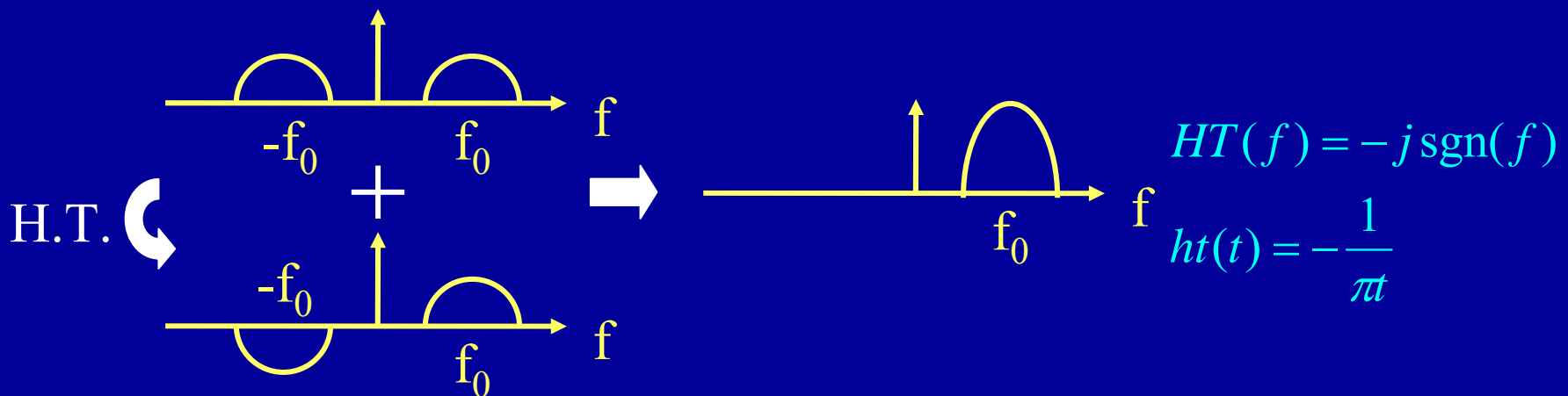


# Envelope Detection

- Hilbert Transform

$$S(t) + j \cdot H.T.\{S(t)\} = 2A(t)e^{j2\pi f_0 t}$$

$$D(t) = \text{abs}(S(t) + j \cdot H.T.\{S(t)\}) / 2$$



# Beam Former Design

# Implementaiton of Beam Formation

- Delay is simply based on geometry.
- Weighting (a.k.a. apodization) strongly depends on the specific approach.

# Beam Formation - Delay

- Delay is based on geometry. For simplicity, a constant sound velocity and straight line propagation are assumed. Multiple reflection is also ignored.
- In diagnostic ultrasound, we are almost always in the near field. Therefore, range focusing is necessary.

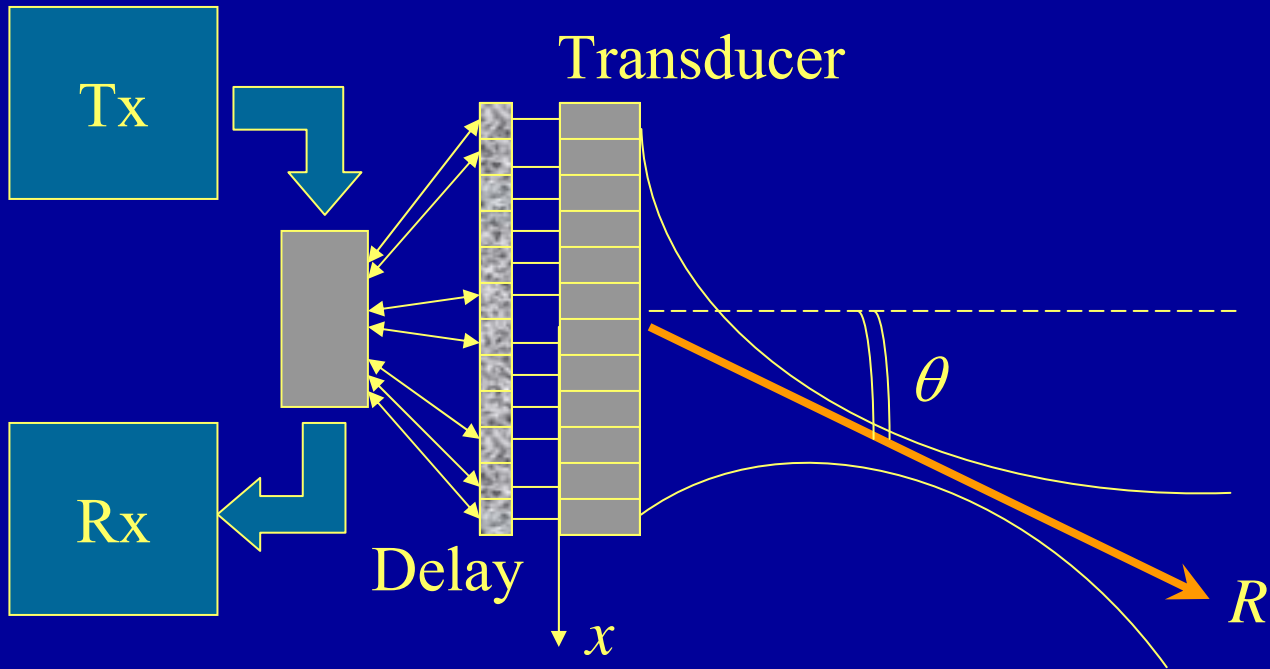


# Beam Formation - Delay

- Near field / far field crossover occurs when  $f_{\#} = \text{aperture size} / \text{wavelength}$ .
- The crossover also corresponds to the point where the phase error across the aperture becomes significant (destructive).

$$\frac{a^2}{2R} = \frac{\lambda}{8}$$

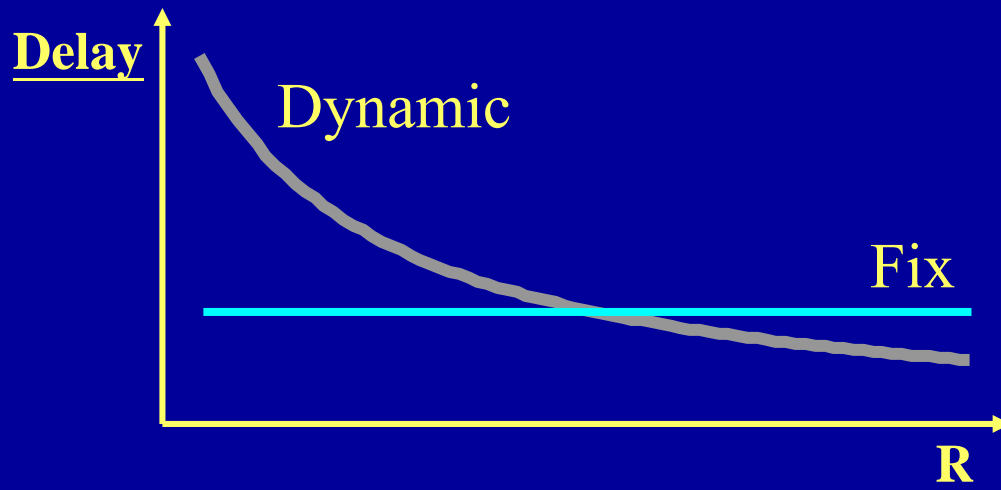
# Phased Array Imaging



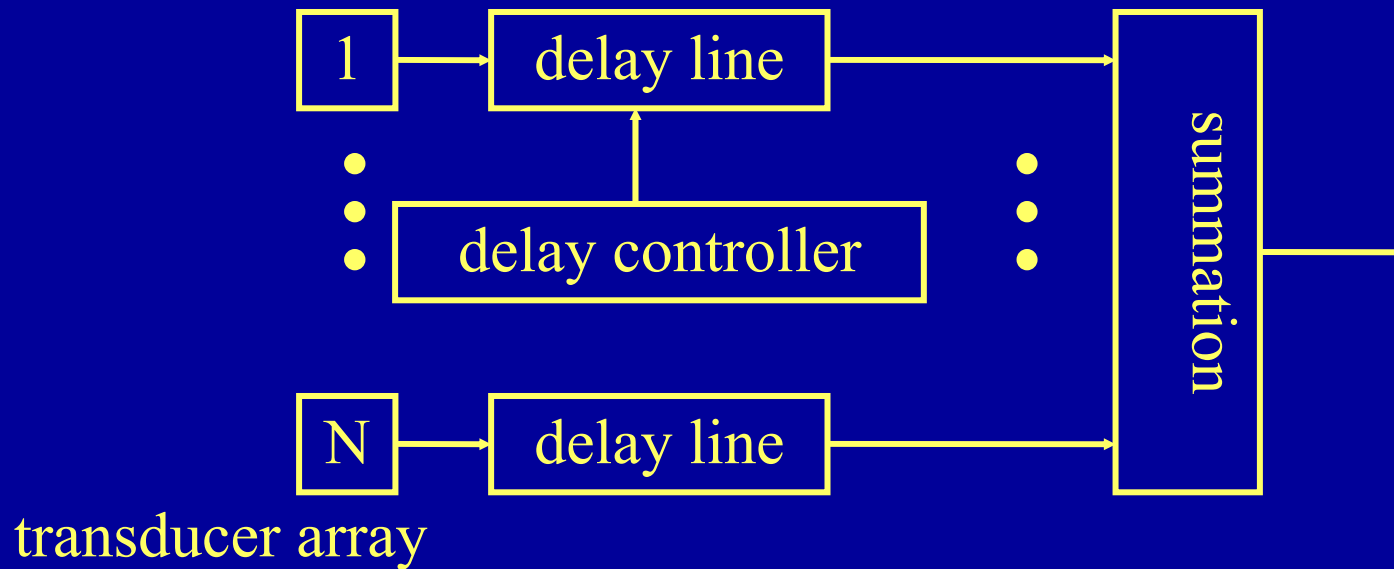
$$t_{rx}(x_i, R, \theta) = -\frac{x_i \sin \theta}{c} + \boxed{\frac{x_i^2 \cos^2 \theta}{2Rc}} \text{ Symmetry}$$

# Dynamic Focusing

- Dynamic-focusing obtains better image quality but implementation is more complicated.



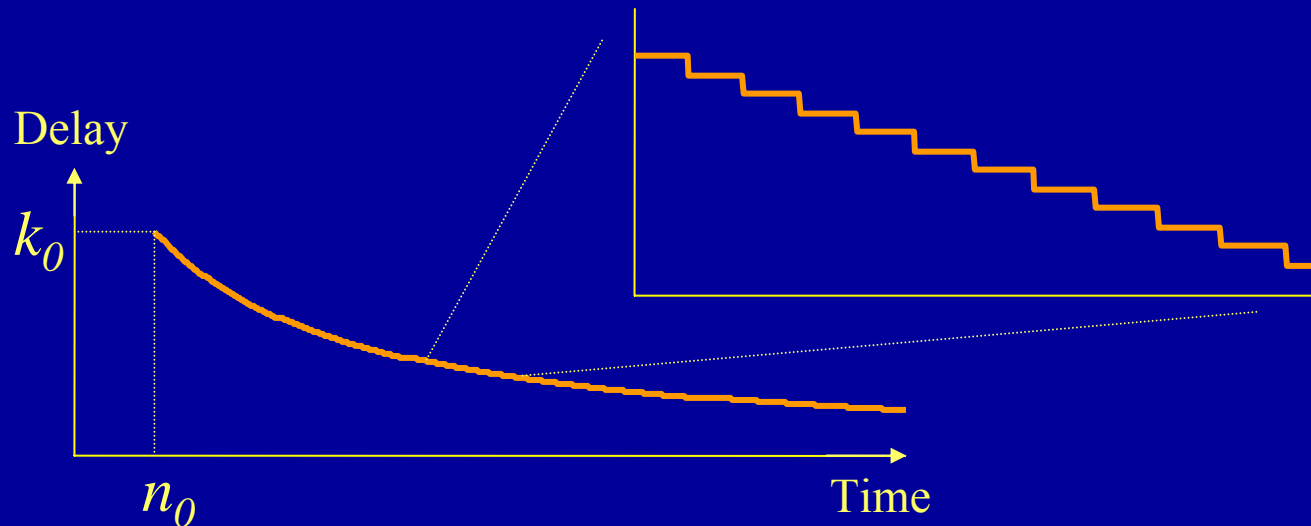
# Focusing Architecture



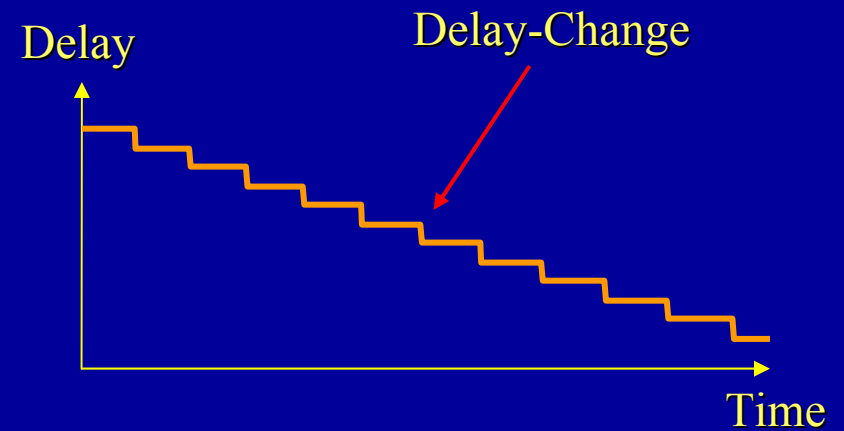
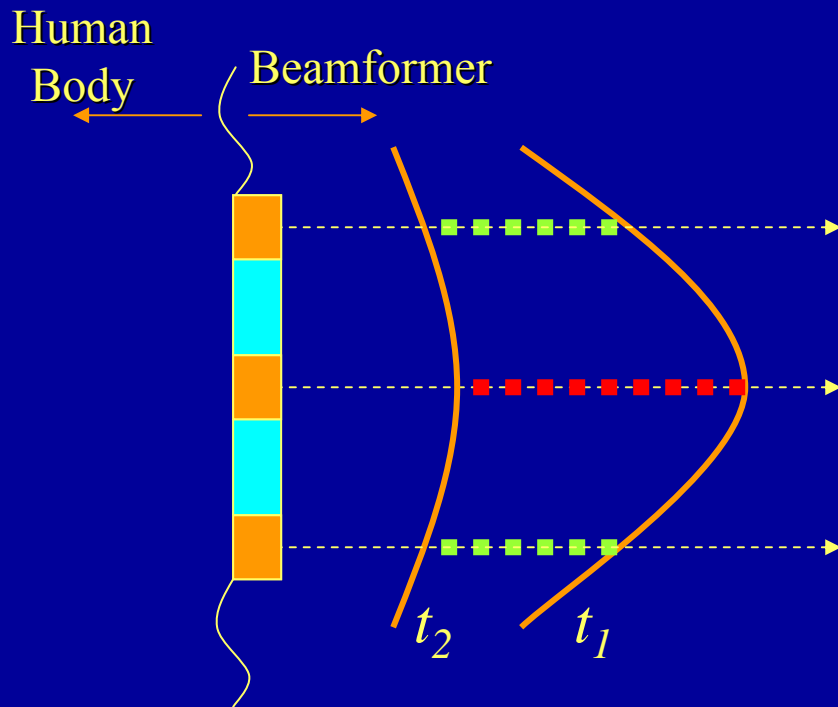
# Delay Pattern

- Delays are quantized by sampling-period  $t_s$ .

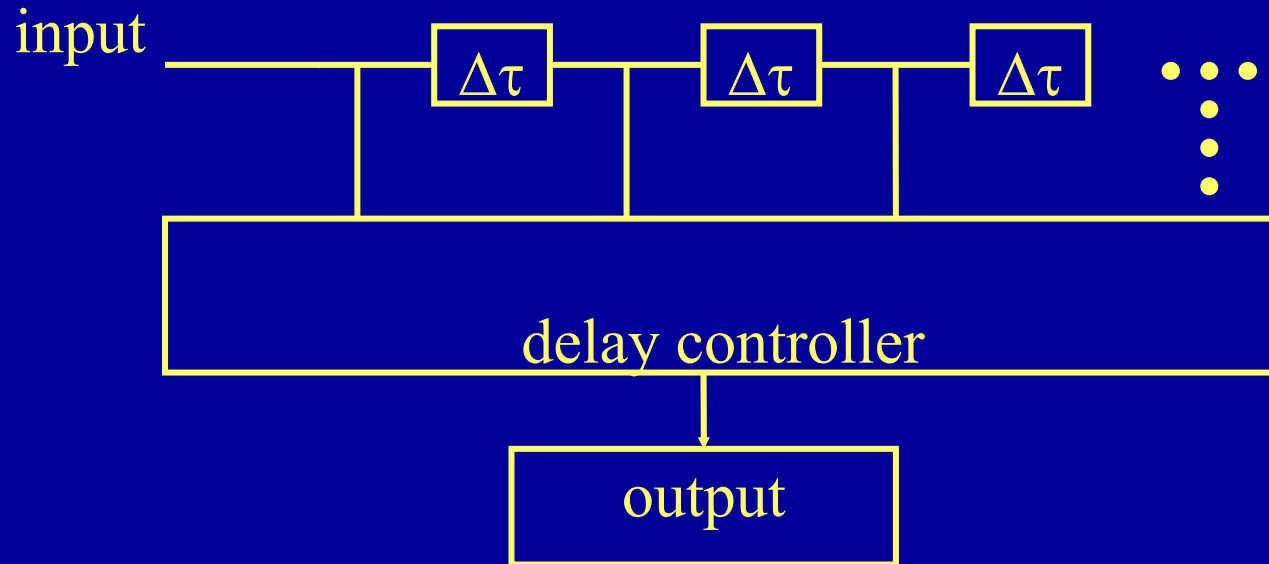
$$k_n = \text{round}\left(-\frac{x_i \sin \theta}{ct_s} + \frac{x_i^2 \cos^2 \theta}{2Rct_s}\right) = n\Delta\tau$$



# Missing Samples



# Beam Formation



$$n(t) \approx -\frac{x_i \sin \theta}{c\Delta\tau} + \frac{x_i^2 \cos^2 \theta}{c^2 t\Delta\tau}$$
$$n(t_1) - n(t_2) = 1 = \frac{x_i^2 \cos^2 \theta}{c^2 \Delta\tau} \left( \frac{1}{t_1} - \frac{1}{t_2} \right)$$

# Beam Formation - Delay

- The sampling frequency for fine focusing quality needs to be over  $32 * f_0$  ( $\gg$  Nyquist).
- Interpolation is essential in a digital system and can be done in RF, IF or BB.

$$\Delta\tau = \frac{\Delta\theta}{2\pi f_0} \leq \frac{1}{32f_0}$$

$$2\pi/32 \approx 11.25^\circ$$

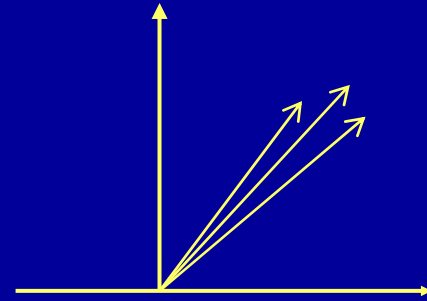


# Delay Quantization

- The delay quantization error can be viewed as the phase error of the phasors.

$$A = \sum_{n=0}^{N-1} \cos(\phi_n)$$

$$\sigma_A^2 = \sum_{n=0}^{N-1} \left( \frac{dA}{d\phi} \right)^2 \sigma_{\phi_n}^2$$

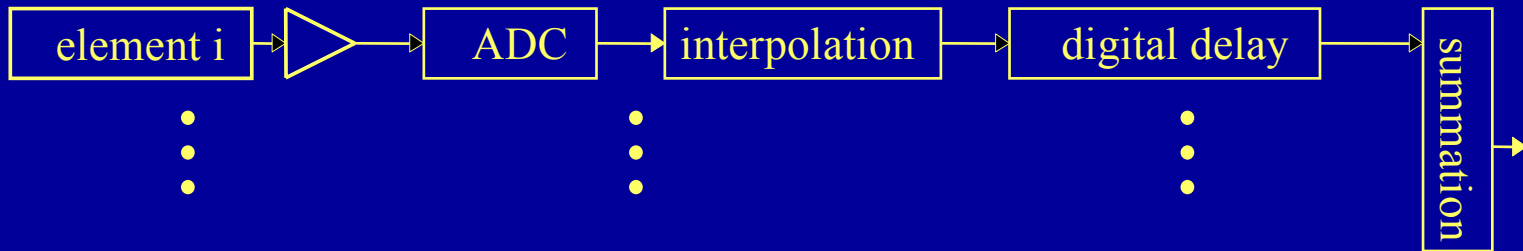


# Delay Quantization

$$\langle \sin^2 \phi \rangle = \frac{1}{2}$$
$$\sigma_{\phi_n}^2 = \sigma_{\phi}^2 = \frac{\Delta\phi^2}{12}$$
$$\sigma_A^2 = \frac{N \cdot \Delta\phi^2}{24} < 1 \Rightarrow \Delta\phi < \sqrt{\frac{24}{N}}$$

- $N=128$ , 16 quantization steps per cycles are required.
- In general, 32 and 64 times the center frequency is used.

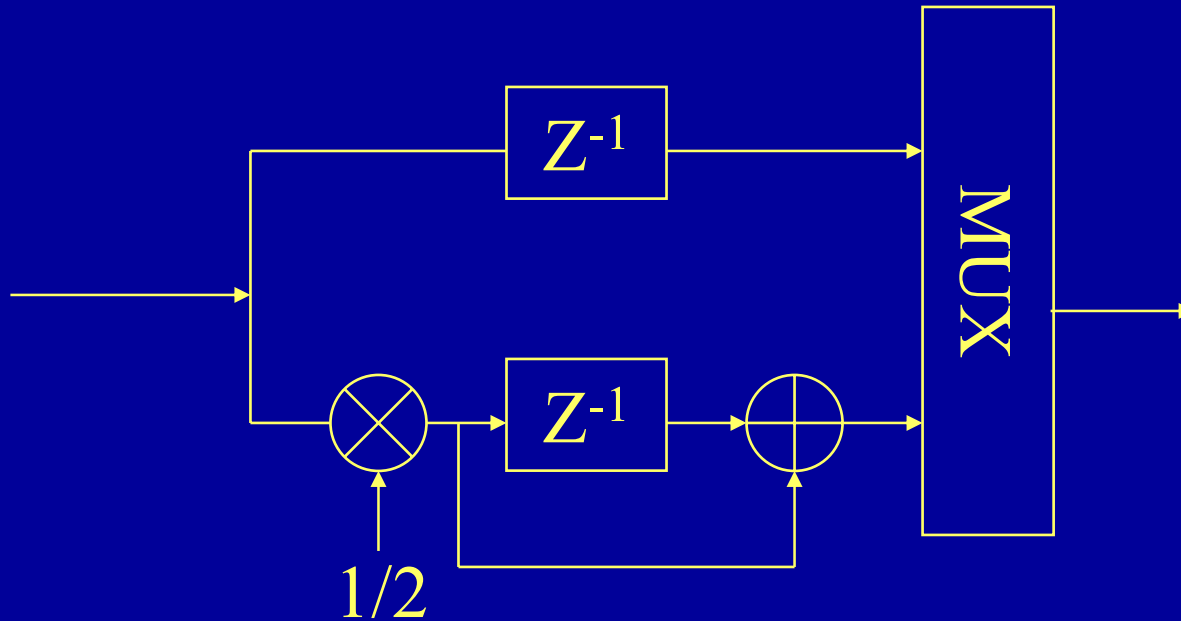
# Beam Formation - Delay



- RF beamformer requires either a clock well over 100MHz, or a large number of real-time computations.
- BB beamformer processes data at a low clock frequency at the price of complex signal processing.

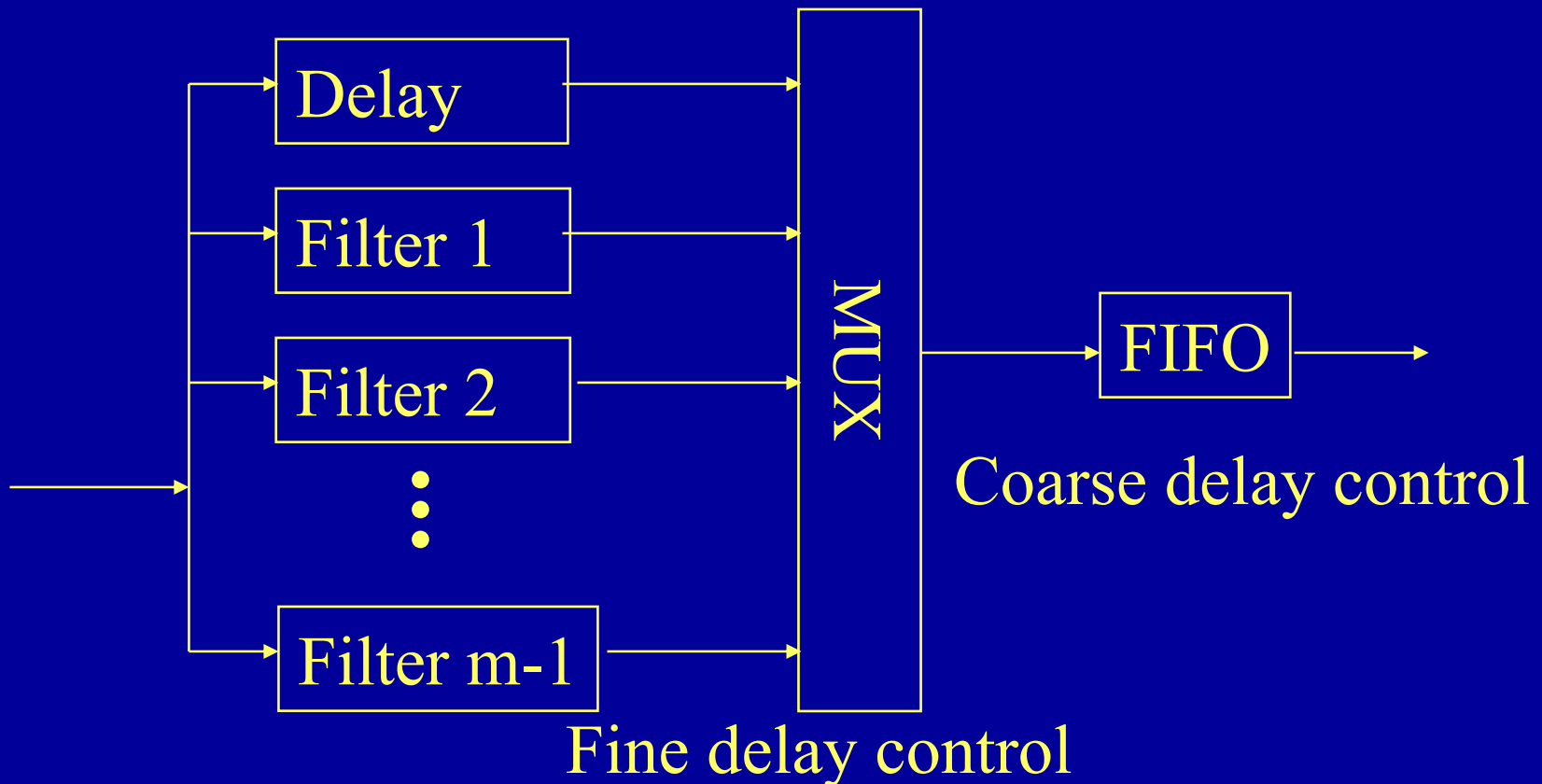
# Beam Formation - RF

- Interpolation by 2:



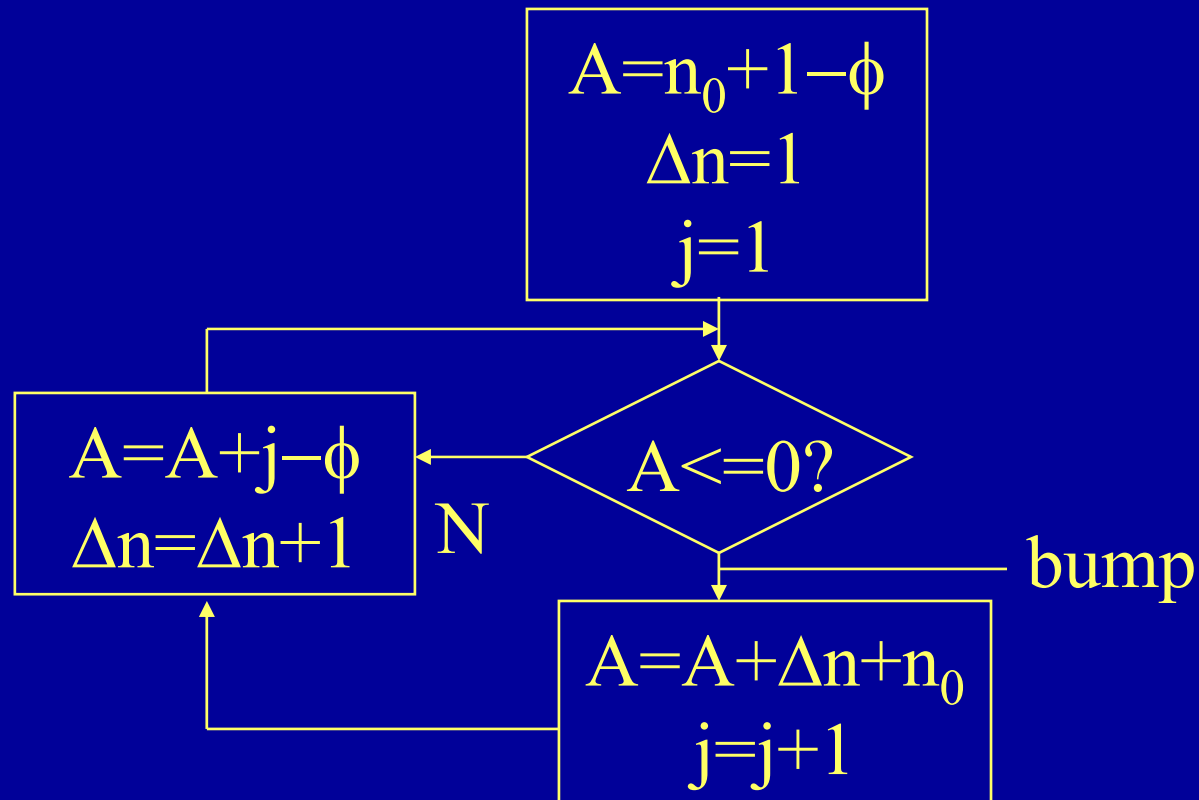
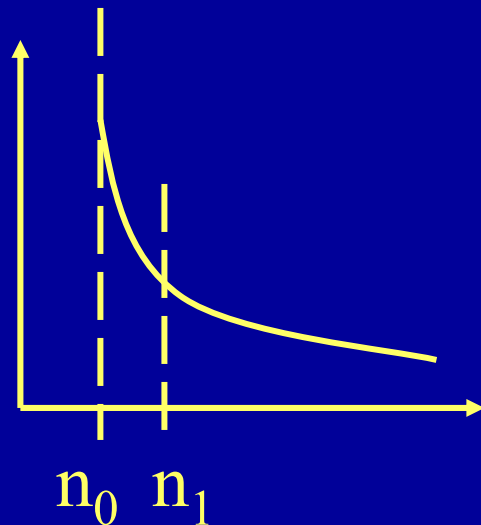
# Beam Formation - RF

- General filtering architecture (interpolation by  $m$ ):



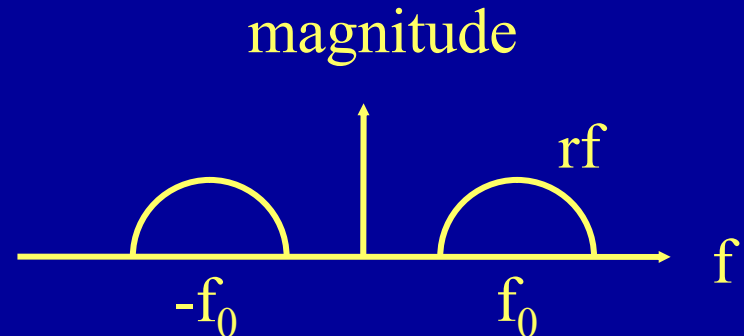
# Autonomous Delay Control

## Autonomous vs. Centralized

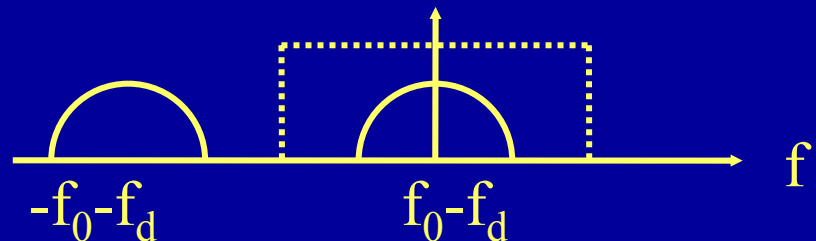


# Beam Formation - BB

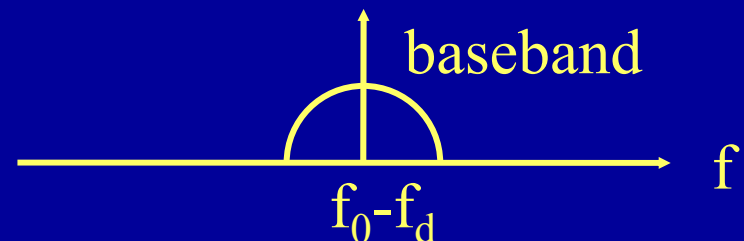
$$A(t-\tau)\cos 2\pi f_0(t-\tau)$$



$$A(t-\tau)\cos 2\pi f_0(t-\tau)e^{-j2\pi f_d t}$$



$$\text{LPF}(A(t-\tau)\cos 2\pi f_0(t-\tau)e^{-j2\pi f_d t})$$



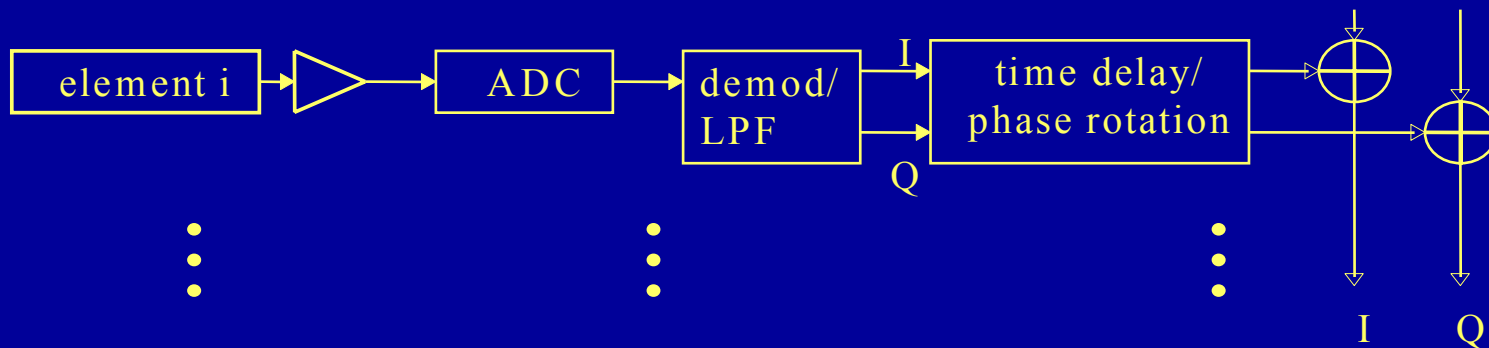
# Beam Formation - BB

$$\begin{aligned} I &= LPF\{A(t-\tau)\cos 2\pi f_0(t-\tau)\cos 2\pi f_d t\} \\ &= LPF\left\{\frac{A(t-\tau)}{2}\left(\cos 2\pi((f_0 - f_d)(t-\tau) - f_d \tau) + \cos 2\pi((f_0 + f_d)(t-\tau) + f_d \tau)\right)\right\} \\ &= \frac{A(t-\tau)}{2}\cos 2\pi((f_0 - f_d)(t-\tau) - f_d \tau) \end{aligned}$$

$$\begin{aligned} Q &= LPF\{-A(t-\tau)\cos 2\pi f_0(t-\tau)\sin 2\pi f_d t\} \\ &= LPF\left\{\frac{A(t-\tau)}{2}\left(\sin 2\pi((f_0 - f_d)(t-\tau) - f_d \tau) - \sin 2\pi((f_0 + f_d)(t-\tau) + f_d \tau)\right)\right\} \\ &= \frac{A(t-\tau)}{2}\sin 2\pi((f_0 - f_d)(t-\tau) - f_d \tau) \end{aligned}$$



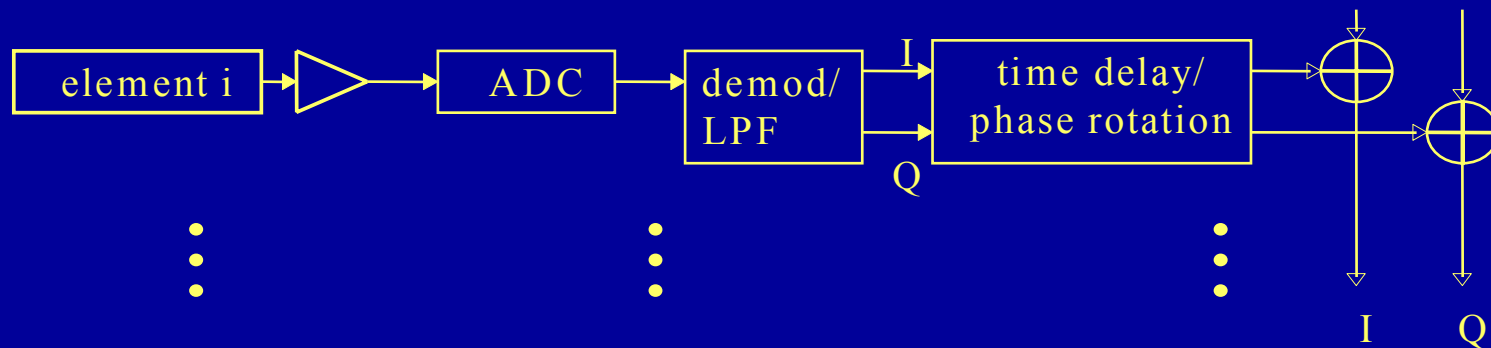
# Beam Formation - BB



$$BB(t) = \frac{A(t - \tau)}{2} e^{j2\pi\Delta f(t - \tau)} e^{-j2\pi f_d \tau}$$

$$O(t) = \sum_{i=1}^N \frac{A(t - \tau_i + \tau'_i)}{2} e^{j2\pi\Delta f(t - \tau_i + \tau'_i)} e^{-j2\pi f_d(\tau_i - \theta_i)}$$

# Beam Formation - BB



$$\Delta\tau = \frac{\Delta\theta}{2\pi\Delta f} \leq \frac{1}{32\Delta f}$$

- The coarse time delay is applied at a low clock frequency, the fine phase needs to be rotated accurately (e.g., by CORDIC).

# $\Delta\Sigma$ -Based Beamformers

# Why $\Delta\Sigma$ ?

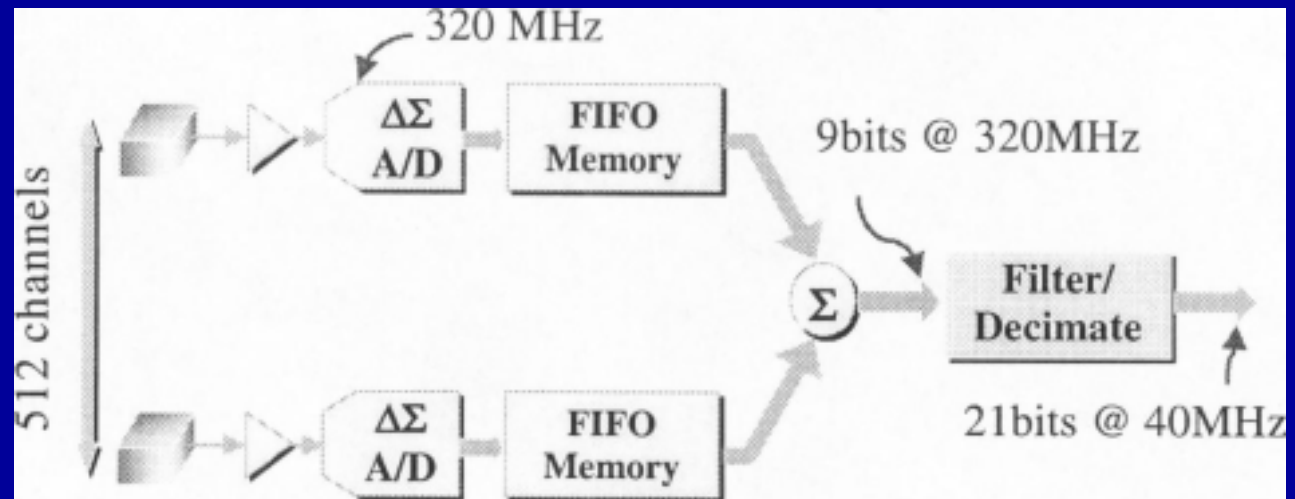
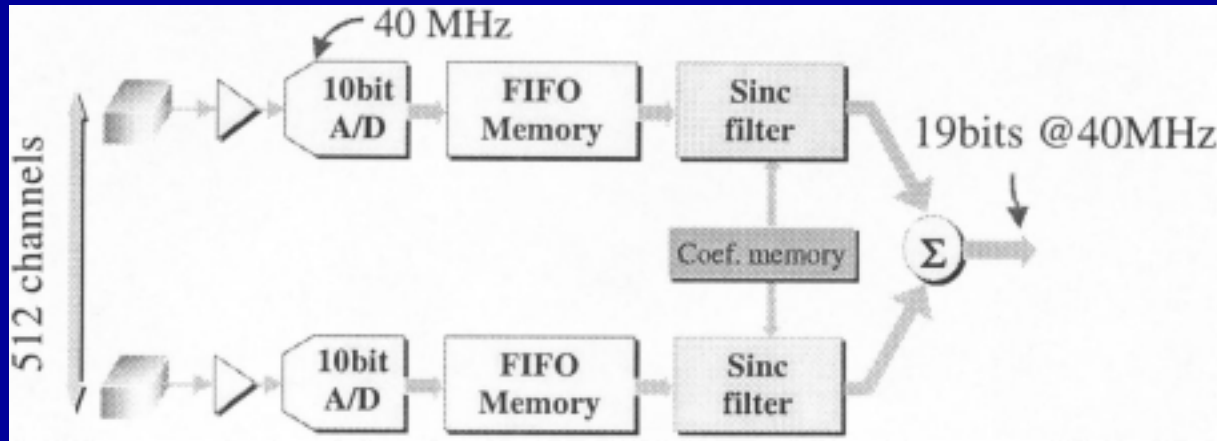
## Current Problems

- High Delay Resolution --  $32 f_0$  (requires interpolation)
- Multi-Bit Bus

## $\Delta\Sigma$ Advantages

- High Sampling Rate -- No Interpolation Required
- Single-Bit Bus -- Suitable for Beamformers with Large Channel-Count

# Conventional vs. $\Delta\Sigma$



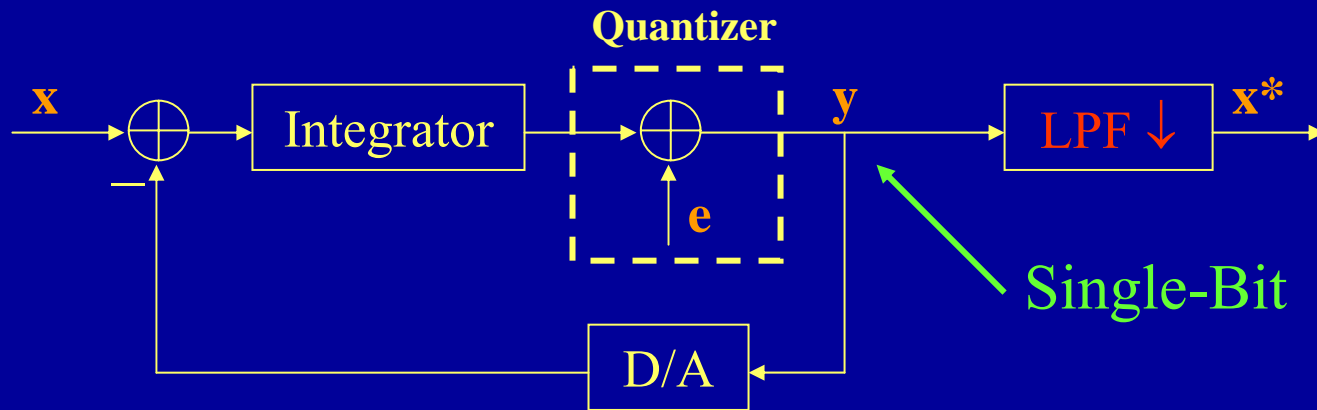
# Advantages of Over-Sampling

- Noise averaging.
- For every doubling of the sampling rate, it is equivalent to an additional 0.5 bit quantization.
- Less requirements for delay interpolation.
- Conventional A/D not ideal for single-bit applications.

# Advantages of $\Delta\Sigma$ Beamformers

- Noise shaping.
- Single-bit vs. multi-bits.
- Simple delay circuitry.
- Integration with A/D and signal processing.
- For hand-held or large channel count devices.

# Block-Diagram of the $\Delta\Sigma$ Modulator



- Over-Sampling
- Noise-Shaping
- Reconstruction

- 
- The SNR of a  $32 f_0$ , 2nd-order, low-passed  $\Delta\Sigma$  modulator is about 40dB.

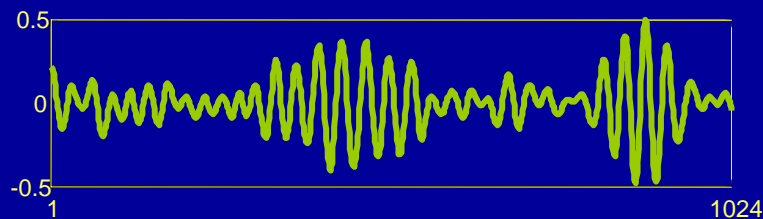


# Property of a $\Delta\Sigma$ Modulator

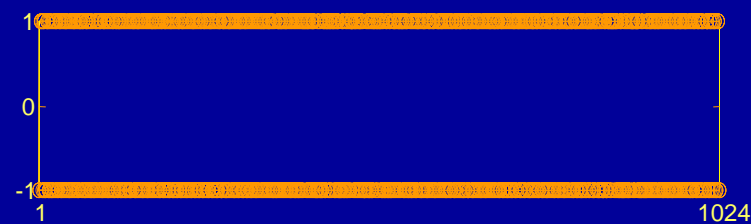
## Waveform

## Spectrum

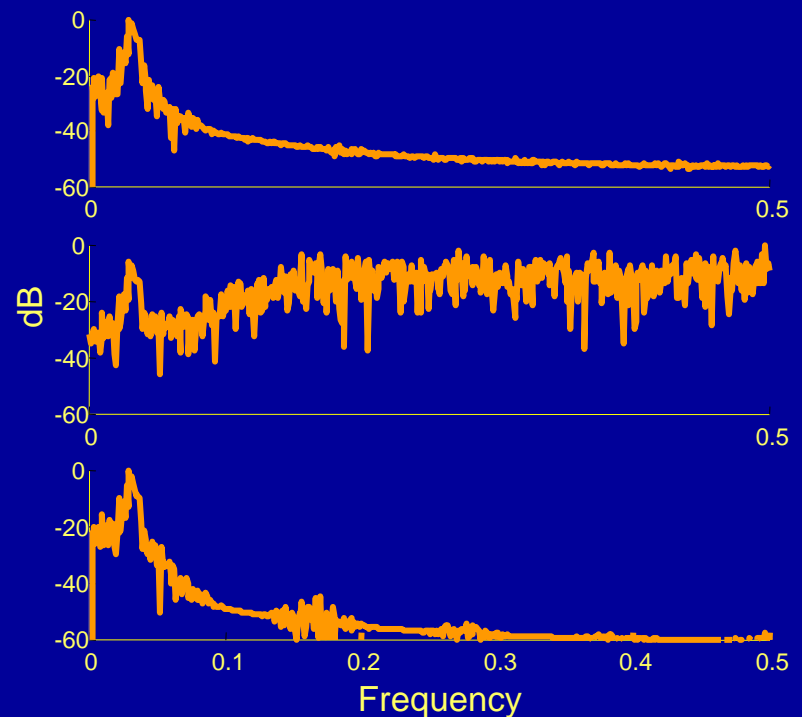
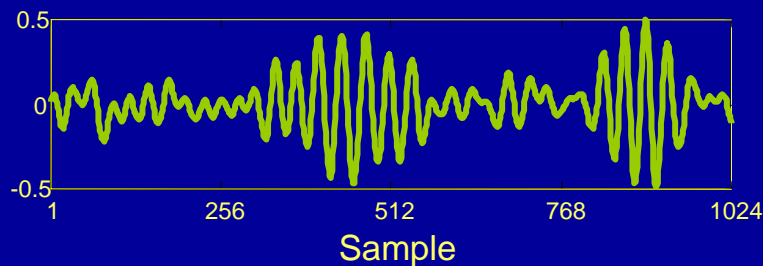
**x**



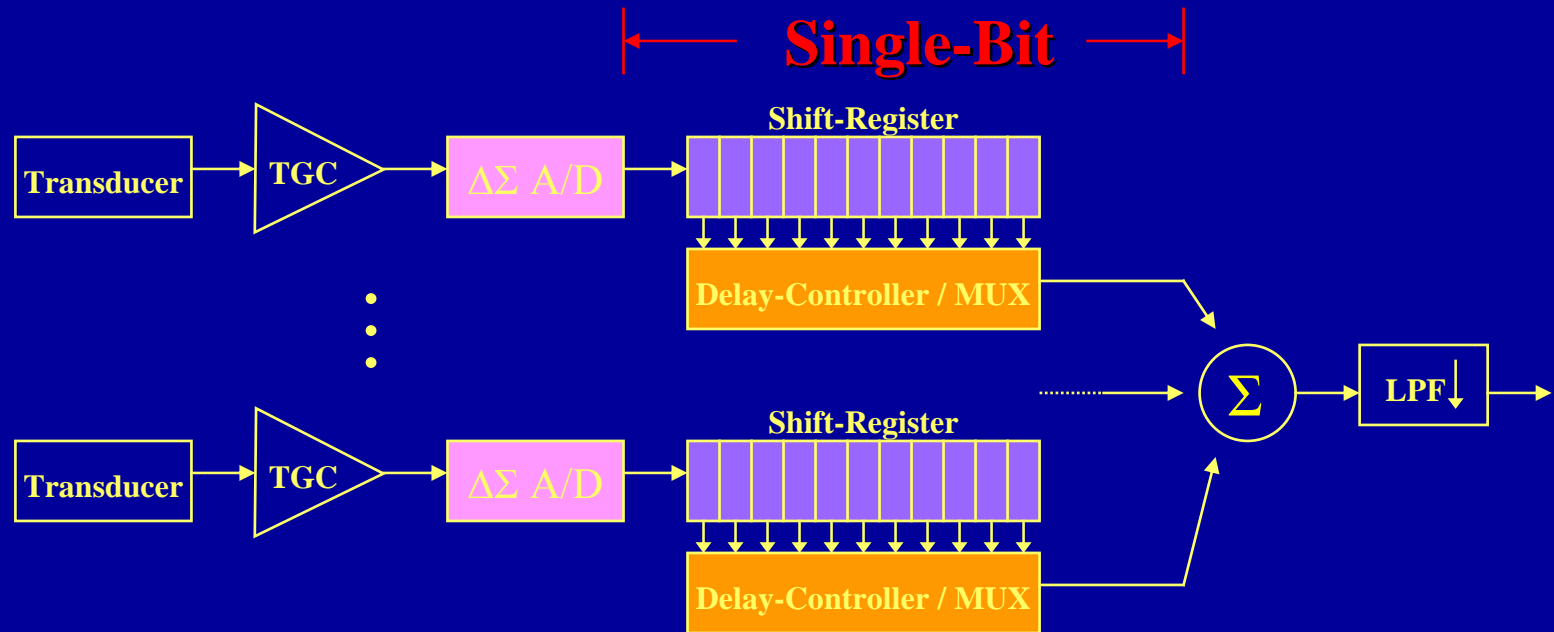
**y**



**x\***

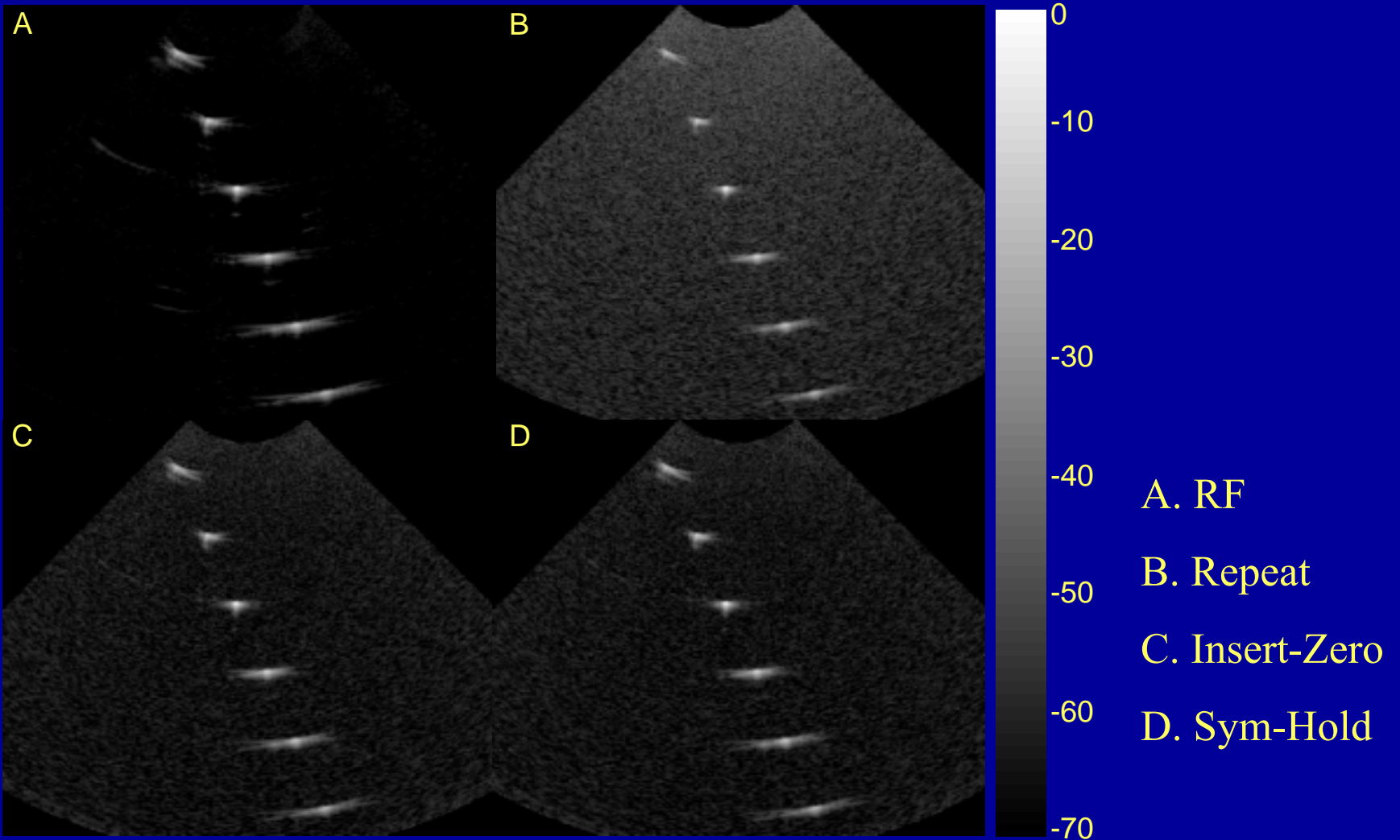


# A Delta-Sigma Beamformer

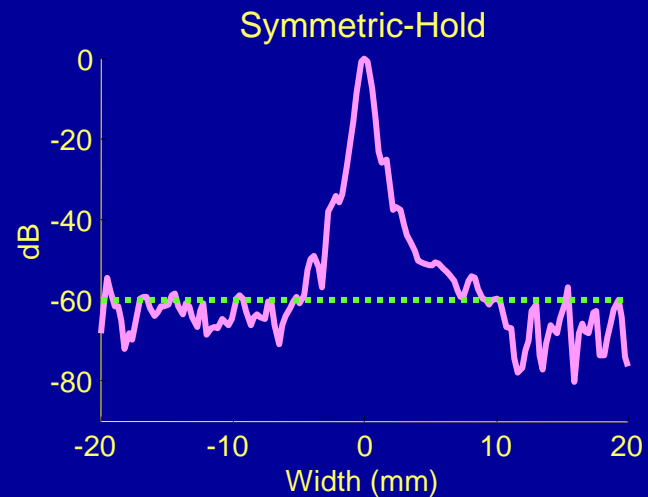
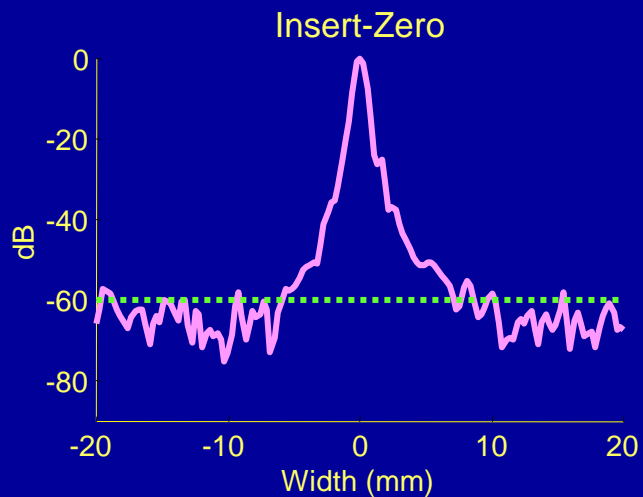
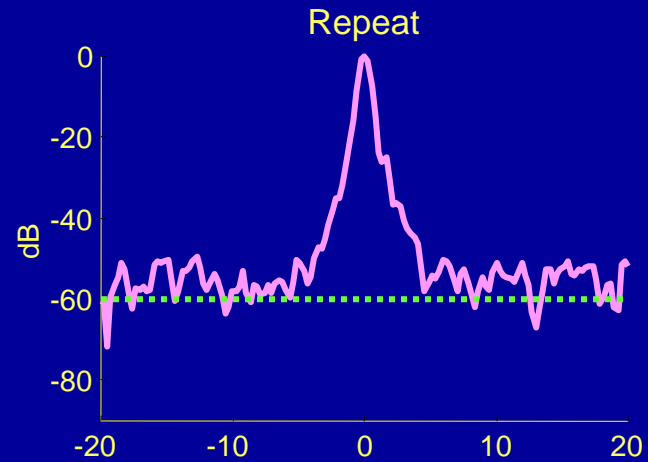
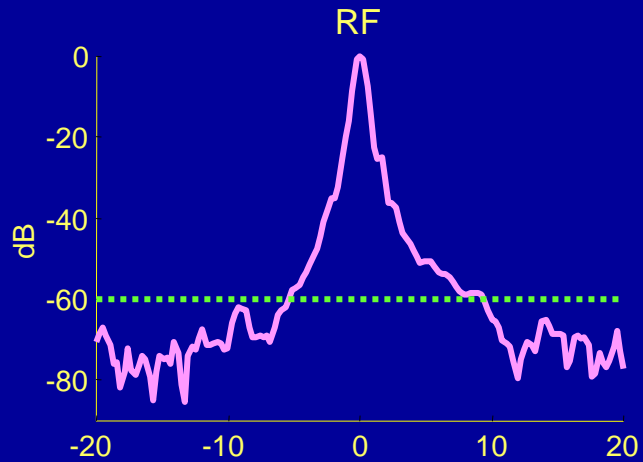


- No Interpolation
- Single-Bit Bus

# Results

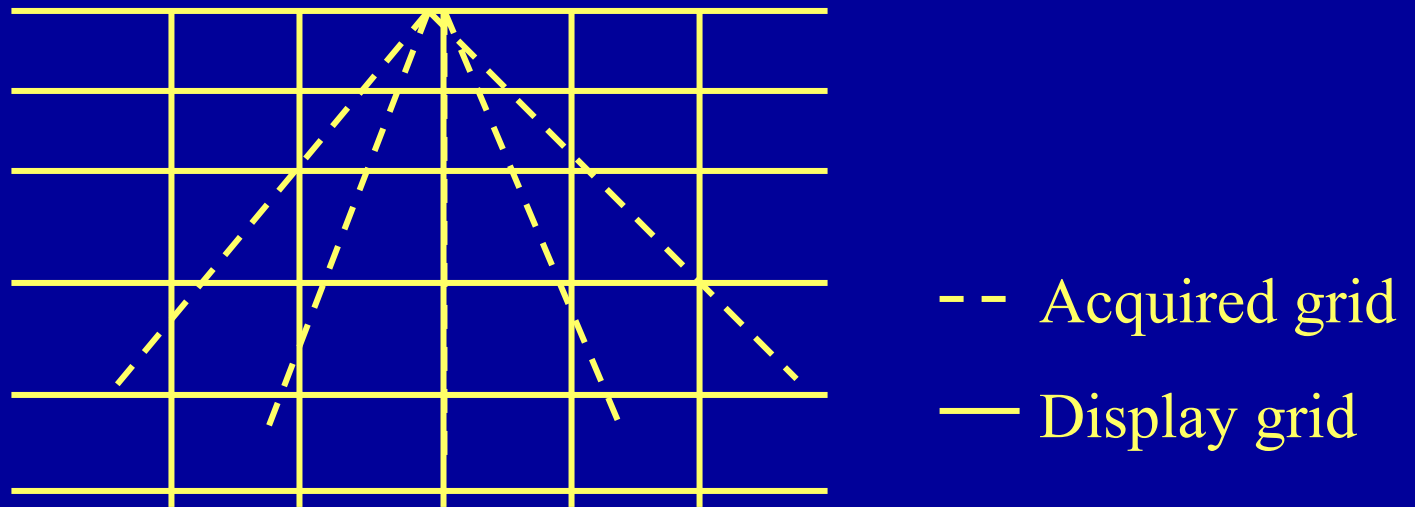


# Cross-Section-Views of Peak 3

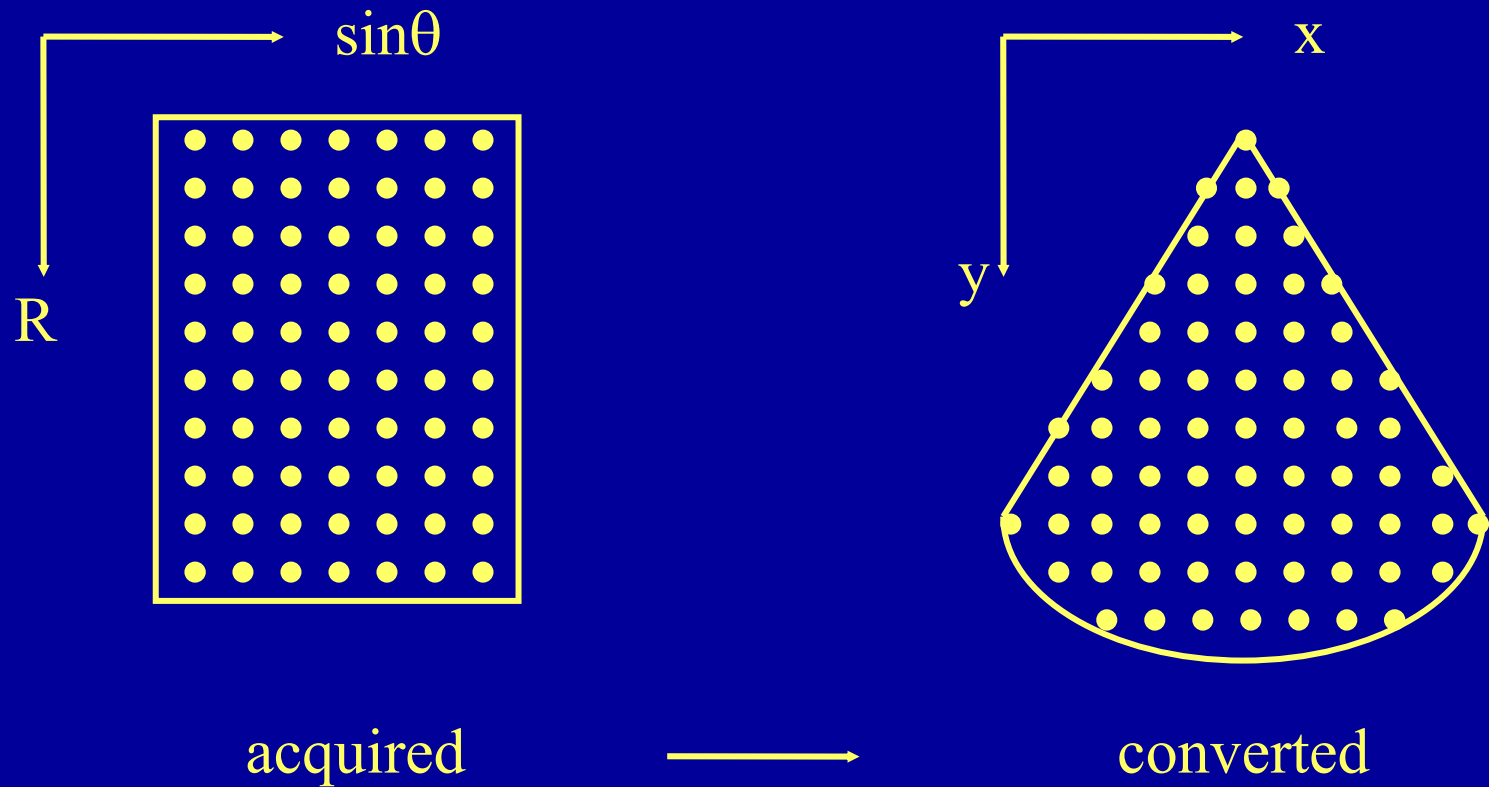


# Scan Conversion

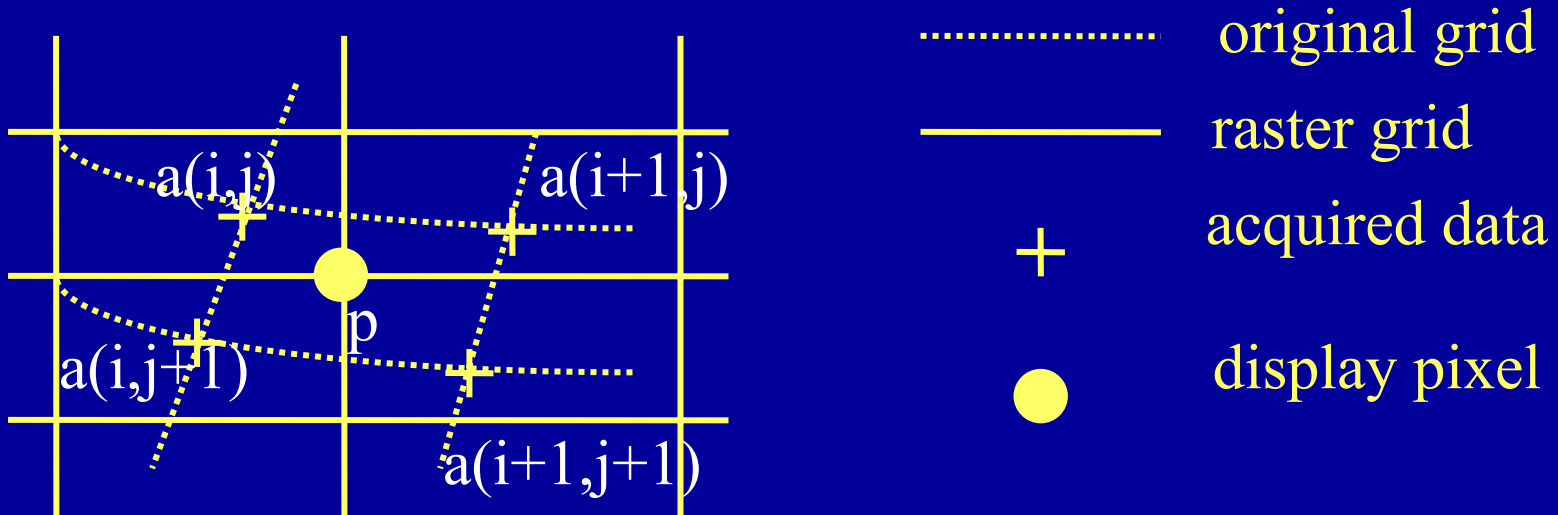
- Acquired data may not be on the display grid.



# Scan Conversion



# Scan Conversion



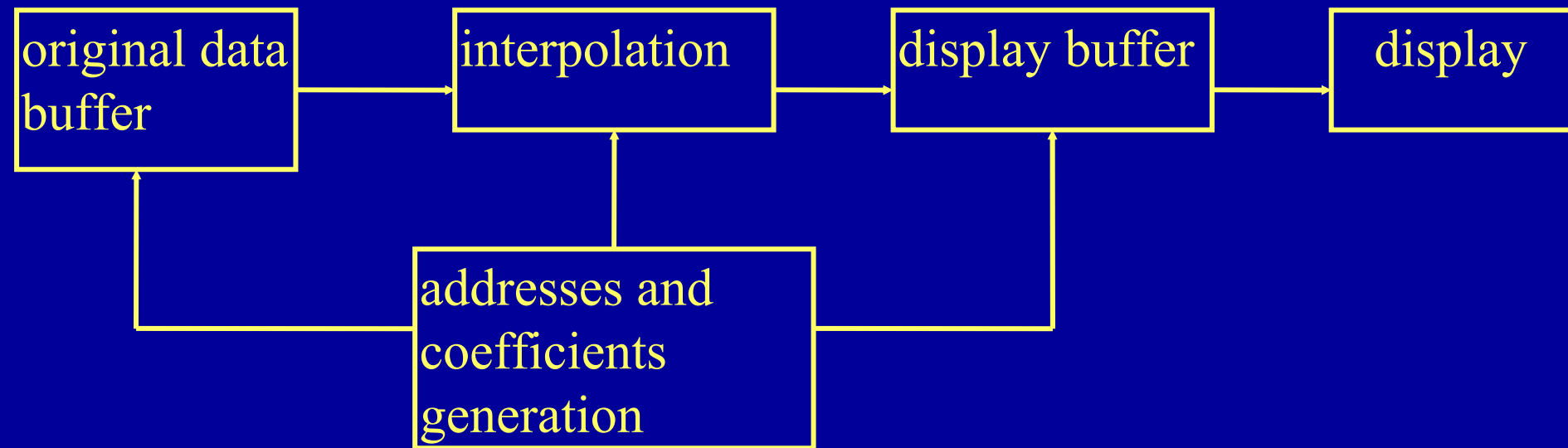
$$p(m,n) = c_{m,n,i,j} a(i,j) + c_{m,n,i+1,j} a(i+1,j) + c_{m,n,i,j+1} a(i,j+1) + c_{m,n,i+1,j+1} a(i+1,j+1)$$

# Moiré Pattern





# Scan Conversion



# Temporal Resolution (Frame Rate)

- Frame rate =  $1/\text{Frame time}$ .
- Frame time = number of lines \* line time.
- Line time =  $(2 * \text{maximum depth}) / \text{sound velocity}$ .
- Sound velocity is around 1540 m/s.
- High frame rate is required for real-time imaging.

# Temporal Resolution

- Display standard: NTSC: 30 Hz. PAL: 25 Hz (2:1 interlace). 24 Hz for movie.
- The actual acoustic frame rate may be higher or lower. But should be high enough to have minimal flickering.
- Essence of real-time imaging: direct interaction.

# Temporal Resolution

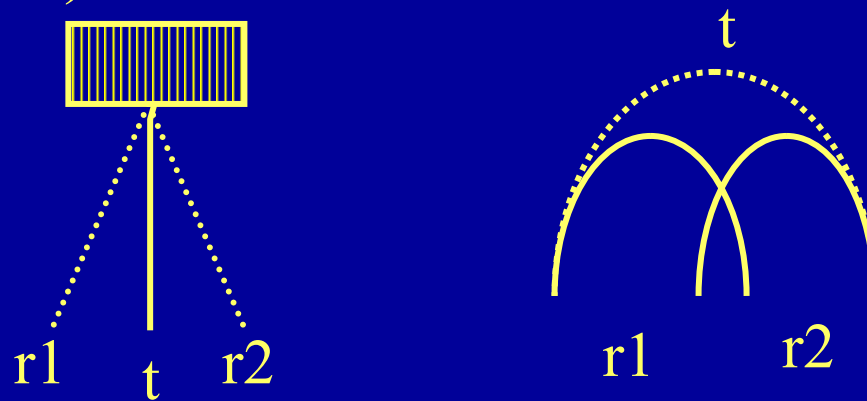
- For an actual frame rate lower than 30 Hz, interpolation is used.
- For an actual frame rate higher than 30 Hz, information can be displayed during playback.
- Even at 30 Hz, it is still possibly undersampling.

# Temporal Resolution

- B-mode vs. Doppler.
- Acoustic power: peak vs. average.
- Increasing frame rate:
  - Smaller depth and width.
  - Less flow samples.
  - Wider beam width.
  - Parallel beam formation.

# Parallel Beamformation

- Simultaneously receive multiple beams.
- Correlation between beams, spatial ambiguity.
- Require duplicate hardware (higher cost) or time sharing (reduced processing time and axial resolution).



# Parallel Beamformation

- Simultaneously transmit multiple beams.
- Interference between beams, spatial ambiguity.

