

# ESE 531: Digital Signal Processing

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Week 1

Lecture 1: January 20, 2021

Introduction and Overview



# Lecture Outline

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- ❑ Course Topics Overview
- ❑ Learning Objectives
- ❑ Course Structure
- ❑ Course Policies
- ❑ Course Content
- ❑ What is DSP?
- ❑ DSP Examples



# Course Topics Overview

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- ❑ Discrete-Time (DT) Signals
- ❑ Time-Domain Analysis of DT Systems
- ❑ Discrete Fourier Transform (DFT)
- ❑ Fast Fourier Transform (FFT)
- ❑ Discrete-Time Fourier Transform (DTFT)
- ❑ z-Transform
- ❑ Sampling of Continuous Time Signals
- ❑ Data Converters and Modulation
- ❑ Upsampling/Downsampling
- ❑ Discrete-Time Filter Design
- ❑ Special Topics



# Learning Objectives

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- ❑ Learn the fundamentals of digital signal processing
- ❑ Provide an understanding of discrete-time signals and systems and digital filters
- ❑ Enable you to apply DSP concepts to a wide range of fields
- ❑ Gain the ability to read the technical literature on DSP
- ❑ Apply the techniques learned in a two projects focused on filter design



# Learning Objectives

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□ In other words...

□ Math, Math, Math\*

\*With MATLAB application for intuition



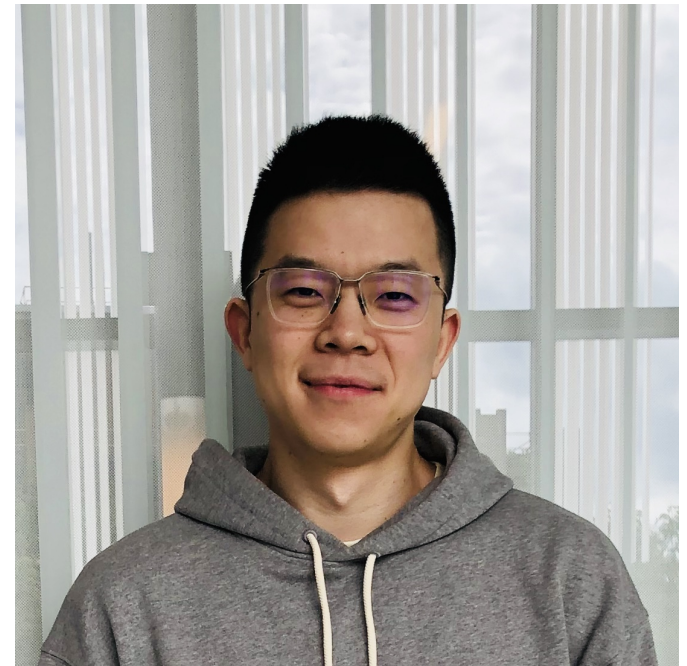
# Course Structure

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- ❑ Week's worth of material released on Sunday
  - 2-3 lecture videos in Canvas
    - Slides available on course webpage calendar
  - Assignment (HW or project) due the following Sunday
    - See course calendar for full schedule
- ❑ Website (<http://www.seas.upenn.edu/~ese531/>)
  - Course calendar is used for all handouts (lectures slides, assignments, and readings)
  - Canvas used for lecture videos, assignment submission, exams, and grades
  - Piazza used for announcements and discussions
    - Use for Zoom links for lectures and OHs

# Course Structure

- ❑ Course Staff (complete info on course website)
- ❑ Instructor: Tania Khanna
  - ESE 531 Office hours – T 9-10am, Th 3:30-4:30pm
  - Office hours – Wednesday 1-3 pm or by appointment
  - Email: [taniak@seas.upenn.edu](mailto:taniak@seas.upenn.edu)
    - Best way to reach me
- ❑ TAs:
  - Shuang Wu
    - Recitation – F TBD
      - Work out problems
      - recorded and uploaded into Canvas
    - Office hours – TBD
  - Still searching for 2<sup>nd</sup> TA





# Course Structure

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## □ Lectures

- You will do better if you keep pace with the course and not binge lectures
  - Pros: Can watch pause and rewatch
  - Cons: Can't ask questions. Use office hours!

## □ Textbook

- A. V. Oppenheim and R. W. Schaffer (with J. R. Buck), Discrete-Time Signal Processing. 3rd. Edition, Prentice-Hall, 2010
  - Homework will be from text
- Class will follow text structure... mostly





# Course Structure - Assignments/Exams

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- ❑ Homework – one week long (8 total)\* [25%]
  - Due Sundays at midnight
  - Combination of book problems and matlab problems
- ❑ Projects – two projects [35%]
  - Work individually
  - Different DSP applications of filter design
- ❑ Midterm exam [20%]
- ❑ Final exam [20%]
  - Exams administered in Canvas



# Course Policies

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See web page for full details

- ❑ Turn homework in Canvas
  - Anything handwritten/drawn must be clearly legible
  - Submit code, graphs, test results when specified
- ❑ Individual work
  - code, test simulations, analysis, writeups
  - May discuss strategies, but acknowledge help
- ❑ Late homeworks
  - 4 late days allowed
  - Can only use one max late day on projects



# I want to hear from you...

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- ❑ Accessibility Survey in Canvas
  - Submit by Sunday 1/24 for full HW credit
- ❑ Will you be in a different time zone?
- ❑ Will you have trouble seeing or hearing video lectures?
- ❑ Are there any other accessibility issues I should know about?
  
- ❑ Let me know any concerns -- I will do everything I can to ensure you achieve the learning objectives



# Course Content

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- ❑ Introduction
- ❑ Discrete Time Signals & Systems
- ❑ Discrete Time Fourier Transform
- ❑ Z-Transform
- ❑ Inverse Z-Transform
- ❑ Sampling of Continuous Time Signals
- ❑ Frequency Domain of Discrete Time Series
- ❑ Downsampling/Upsampling
- ❑ Data Converters, Sigma Delta Modulation
- ❑ Frequency Response of LTI Systems
- ❑ Signal Flow Representation
- ❑ Basic Structures for IIR and FIR Systems
- ❑ Design of IIR and FIR Filters
- ❑ Butterworth, Chebyshev, and Elliptic Filters
- ❑ Filter Banks
- ❑ Adaptive Filters
- ❑ Computation of the Discrete Fourier Transform
- ❑ Fast Fourier Transform



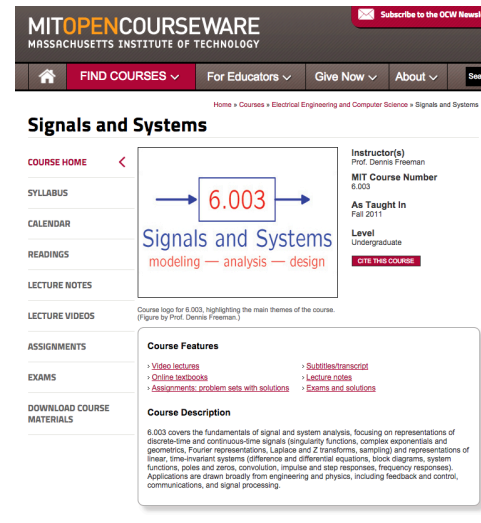
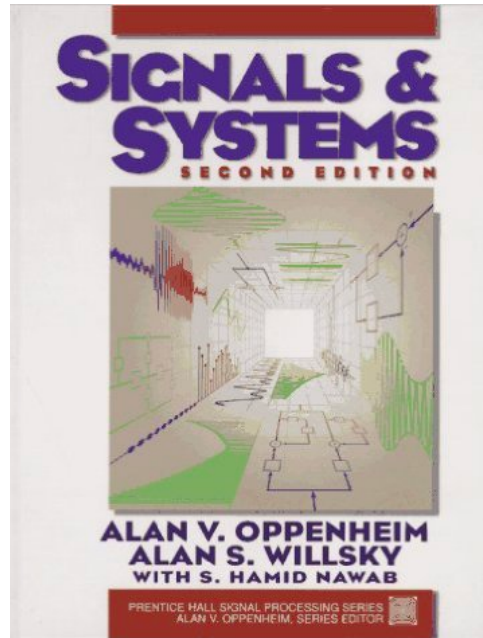
# Course Content

## ESE531 Spring 2021 Working Schedule

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Wk	Date	Lecture	Slides	Due	Reading
1	1/20 W	Week 1 Release: Discrete Time Signals & Systems			review <a href="#">course webpage</a> completely
	1/21 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	1/24 Su	Week 2 Release: Discrete Time Signals & Systems		Accessibility Survey	2.1-2.4
2	1/26 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	1/28 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	1/31 Su	Week 3 release: Discrete Time Fourier Transform, Z-Transform		HW 0, Diagnostics Quiz	2.5-2.7, 3.0-3.1
3	2/2 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	2/4 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	2/7 Su	Week 4 release: Inverse Z-Transform, Sampling and Reconstruction		MATLAB Tutorial, HW 1	3.3, 4.0-4.3
4	2/9 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	2/11 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	2/14 Su	Week 5 release: DT/CT Processing of CT/DT Signals, Re-sampling and Practical Interpolation		HW 2	4.4-4.5, 4.6-4.6.3
5	2/16 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	2/18 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	2/21 Su	Week 6 release: Non-Integer and Multi-rate Sampling, Polyphase Decomposition, and Multi-rate Filter Banks		HW 3	4.6.4-4.7
6	2/23 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	2/25 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			
	2/28 Su	Week 7 release: Data Converters, Noise Shaping, and Frequency Response of LTI Systems		HW 4	4.8-4.9, 5.0-5.4
7	3/2 T	Instructor Office Hours, 9-10am (Zoom Link in Piazza)			
	3/4 Th	Instructor Office Hours, 3:30-4:30pm (Zoom Link in Piazza)			

# Signals and Systems Review



- ❑ [https://www.seas.upenn.edu/~ese531/spring2021/knowledge\\_roundup.html](https://www.seas.upenn.edu/~ese531/spring2021/knowledge_roundup.html)
- ❑ Diagnostic Quiz in Canvas
  - Complete by 1/31 for full credit
  - Review in HW 0
    - Also Matlab Tutorial

# What is DSP

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# DSP is Everywhere

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- ❑ Sound applications
  - Compression, enhancement, special effects, synthesis, recognition, echo cancellation,...
  - Cell phones, MP3 players, movies, dictation, text-to-speech,...
- ❑ Communication
  - Modulation, coding, detection, equalization, echo cancellation,...
  - Cell Phones, dial-up modem, DSL modem, Satellite Receiver,...
- ❑ Automotive
  - ABS, GPS, Active Noise Cancellation, Cruise Control, Parking, Driverless Cars...





# DSP is Everywhere (con't)

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- ❑ Medical
  - Magnetic Resonance, Tomography, Electrocardiogram, Biometric Monitoring...
- ❑ Military
  - Radar, Sonar, Space photographs, remote sensing,...
- ❑ Image and Video Applications
  - DVD, JPEG, Movie special effects, video conferencing...
- ❑ Mechanical
  - Motor control, process control, oil and mineral prospecting,...



# Signal Processing

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- ❑ Humans are the most advanced signal processors
  - speech and pattern recognition, speech synthesis,...
- ❑ We encounter many types of signals in various applications
  - Electrical signals: voltage, current, magnetic and electric fields,...
  - Mechanical signals: velocity, force, displacement,...
  - Acoustic signals: sound, vibration,...
  - Other signals: pressure, temperature, biometrics...
- ❑ Most real-world signals are analog
  - They are continuous in time and amplitude
  - Convert to voltage or currents using sensors and transducers



# Signal Processing (con't)

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- ❑ Analog circuits process these signals using
  - Resistors, Capacitors, Inductors, Amplifiers,...
- ❑ Analog signal processing examples
  - Audio processing in FM radios
  - High end stereo equipment
  - Video processing in traditional TV sets



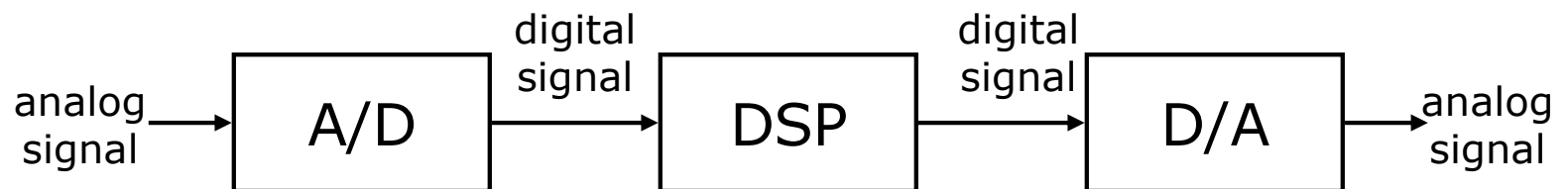
# Limitations of Analog Signal Processing

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- ❑ Accuracy limitations due to
  - Component tolerances
  - Undesired nonlinearities
- ❑ Limited repeatability due to
  - Tolerances
  - Changes in environmental conditions
    - Temperature
    - Vibration
- ❑ Sensitivity to electrical noise
- ❑ Limited dynamic range for voltage and currents
- ❑ Inflexibility to changes
- ❑ Difficulty of implementing certain operations
  - Nonlinear operations
  - Time-varying operations
- ❑ Difficulty of storing information

# Digital Signal Processing

- ❑ Represent signals by a sequence of numbers
  - Sampling and quantization (or analog-to-digital conversion)
- ❑ Perform processing on these numbers with a digital processor
  - Digital signal processing
- ❑ Reconstruct analog signal from processed numbers
  - Reconstruction or digital-to-analog conversion



- Analog input → analog output
  - Eg. Digital recording music
- Analog input → digital output
  - Eg. Touch tone phone dialing, speech to text
- Digital input → analog output
  - Eg. Text to speech
- Digital input → digital output
  - Eg. Compression of a file on computer



# Pros and Cons of Digital Signal Processing

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## □ Pros

- Accuracy can be controlled by choosing word length
- Repeatable
- Sensitivity to electrical noise is minimal
- Dynamic range can be controlled using floating point numbers
- Flexibility can be achieved with software implementations
- Non-linear and time-varying operations are easier to implement
- Digital storage is cheap
- Digital information can be encrypted for security
- Price/performance and reduced time-to-market



# Pros and Cons of Digital Signal Processing

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## ❑ Cons

- Sampling causes loss of information
- A/D and D/A requires mixed-signal hardware
- Limited speed of processors
- Quantization and round-off errors

# DSP Examples

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# Example I: Audio Compression

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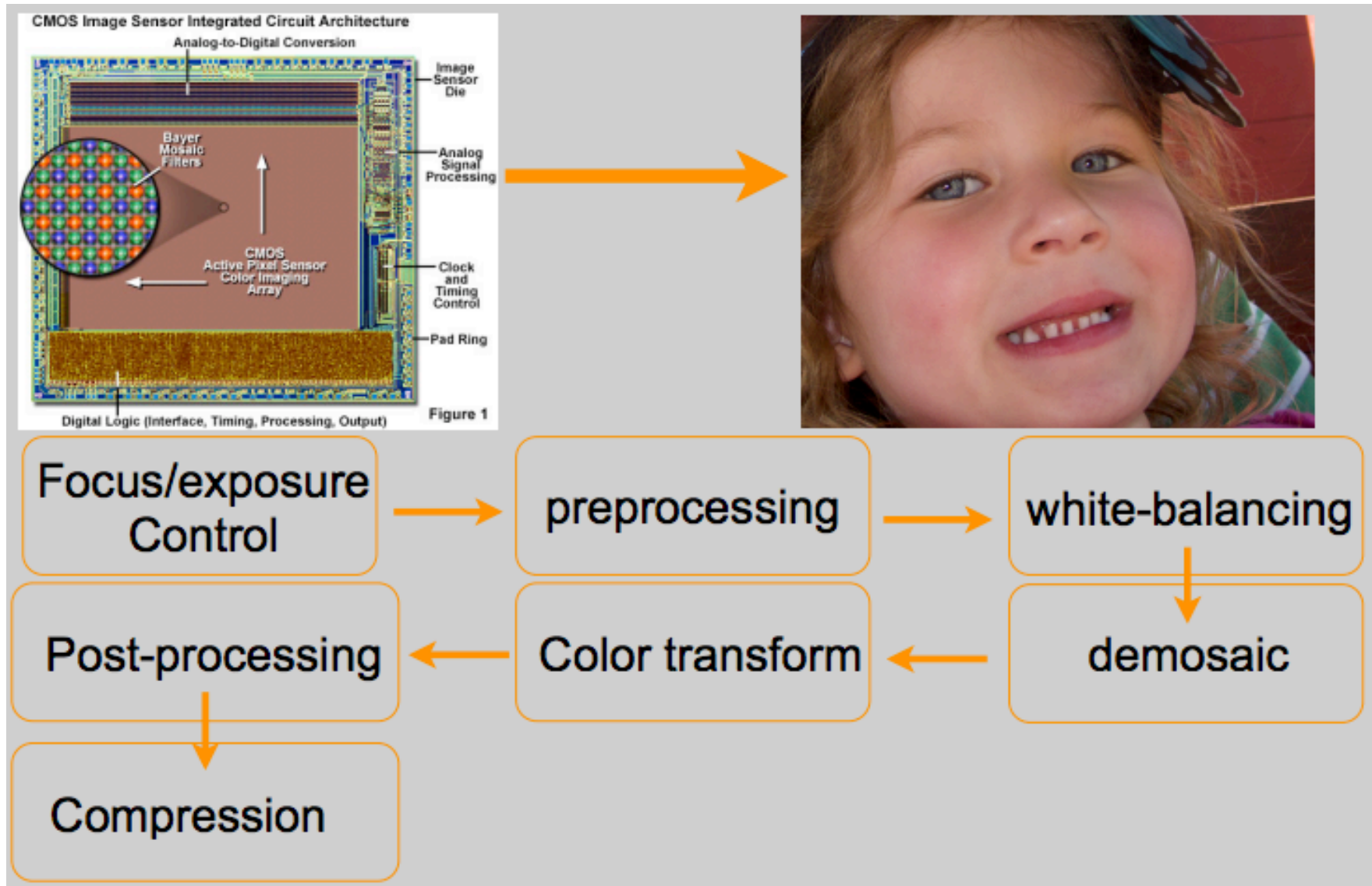
- ❑ Compress audio by 10x without perceptual loss of quality
- ❑ Sophisticated processing based on models of human perception
- ❑ 3MB files instead of 30MB
  - Entire industry changed in less than 10 years!

# Historical Forms of Compression

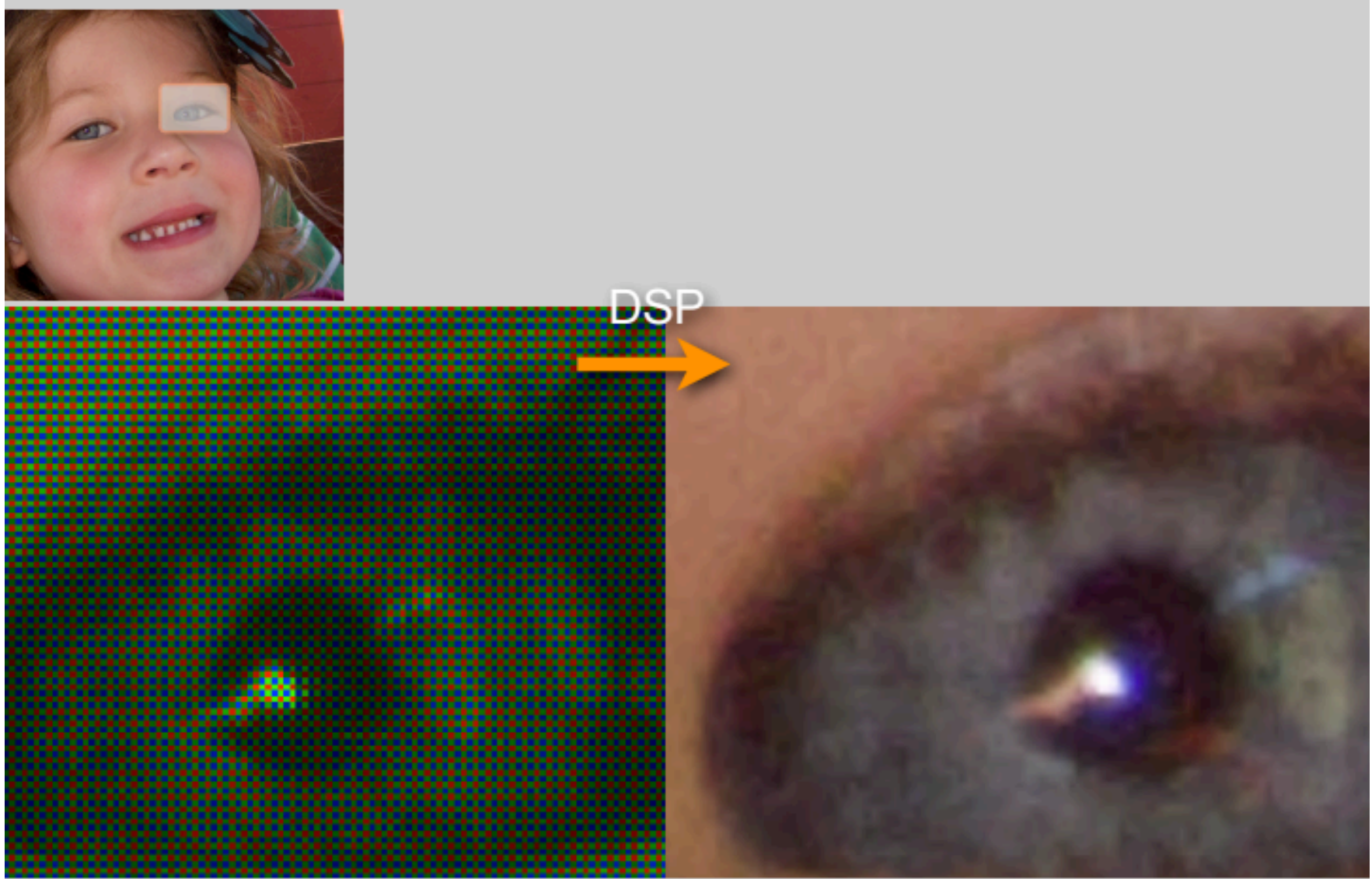
- ❑ Morse code: dots (1 unit) dashes (3 units)
  - Code Length inversely proportional to frequency of character
    - E (12.7%) = . (1 unit) Q (0.1%) = --.- (10 units)
- ❑ “92 Code”
  - Used by Western-Union in 1859 to reduce BW on telegraph lines by numerical codes for frequently used phrases
    - 1 = wait a minute
    - 73 = Best Regards
    - 88 = Loves and Kisses



# Example II: Digital Imaging Camera

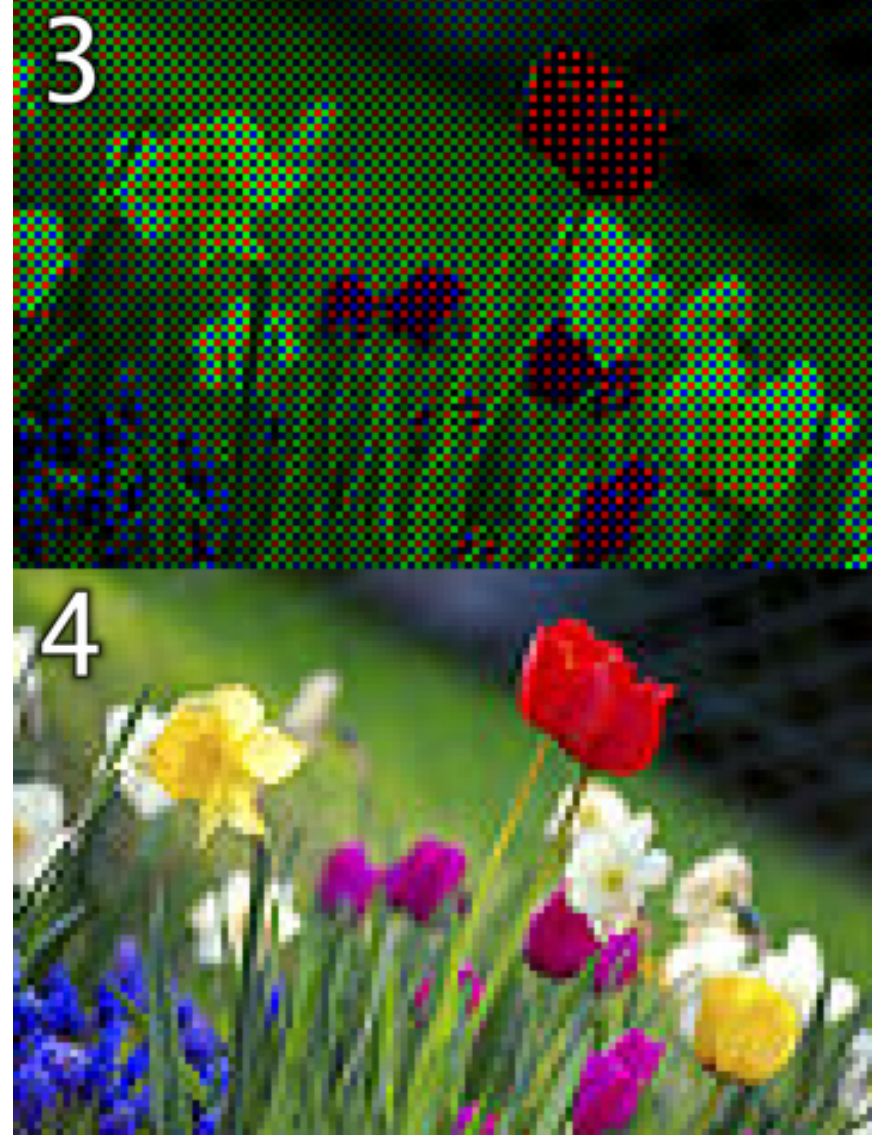
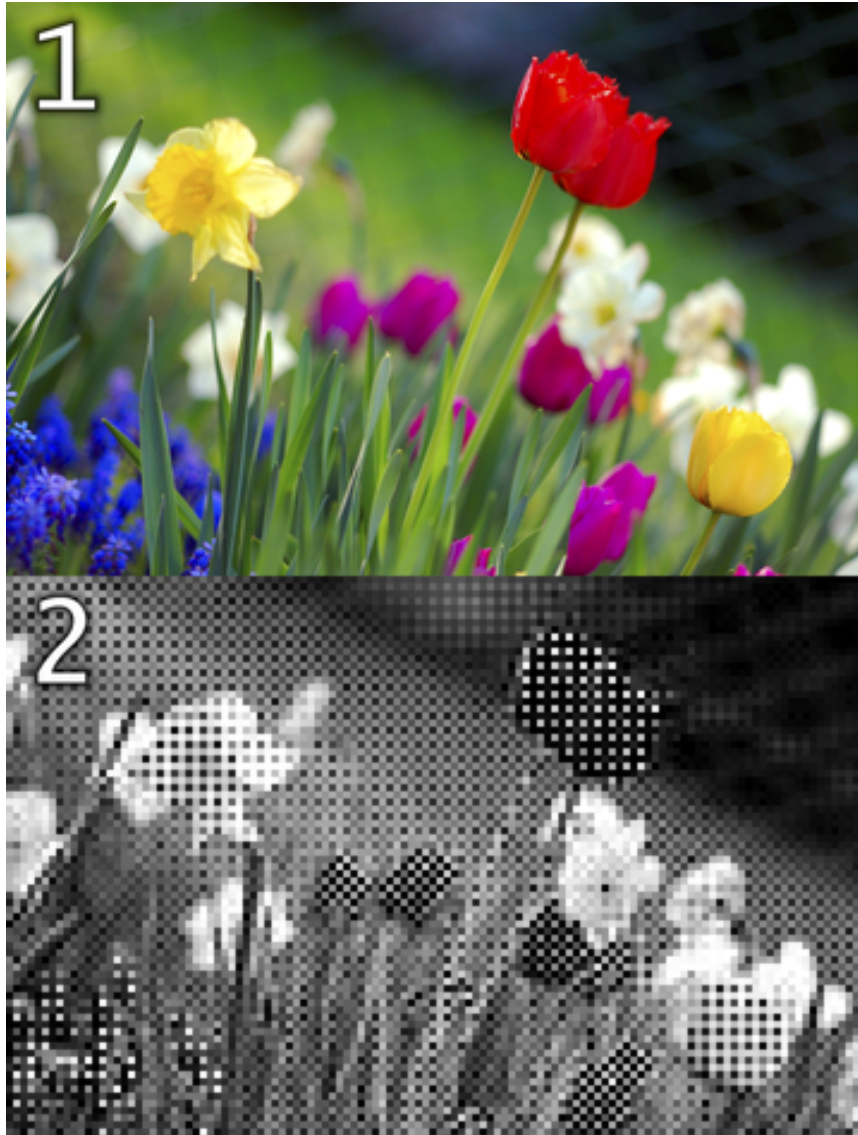


# Example II: Digital Imaging Camera

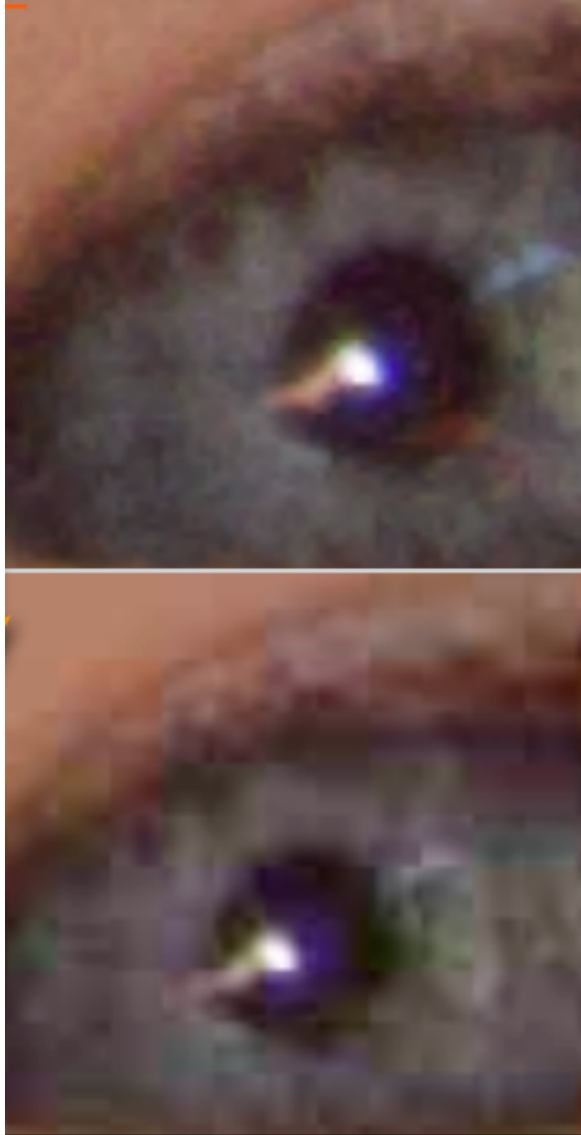




## Example II: Digital Imaging Camera

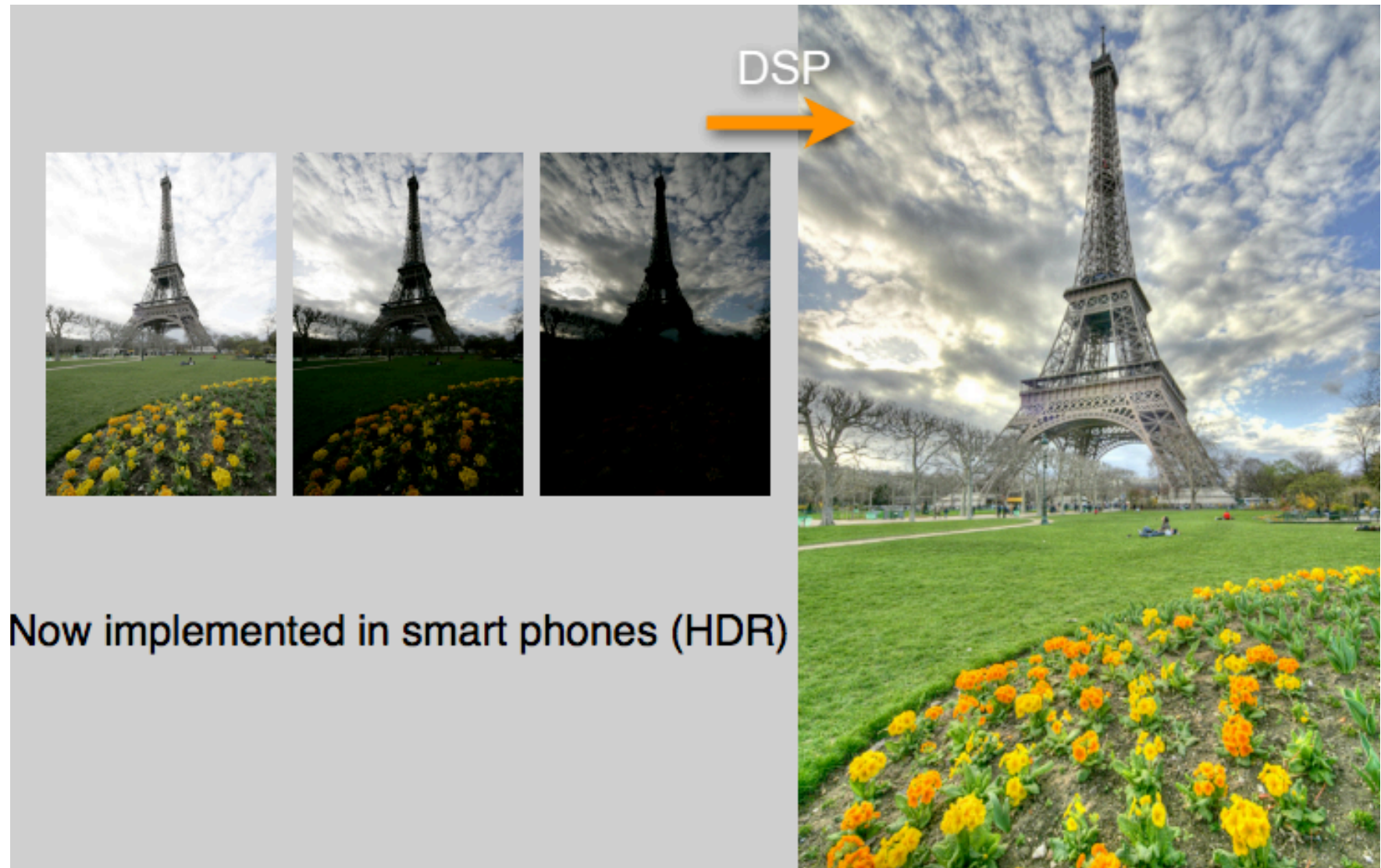


## Example II: Digital Imaging Camera



- ❑ Compression of 40x without perceptual loss of quality.
- ❑ Example of slight over compression: difference enables 60x compression!

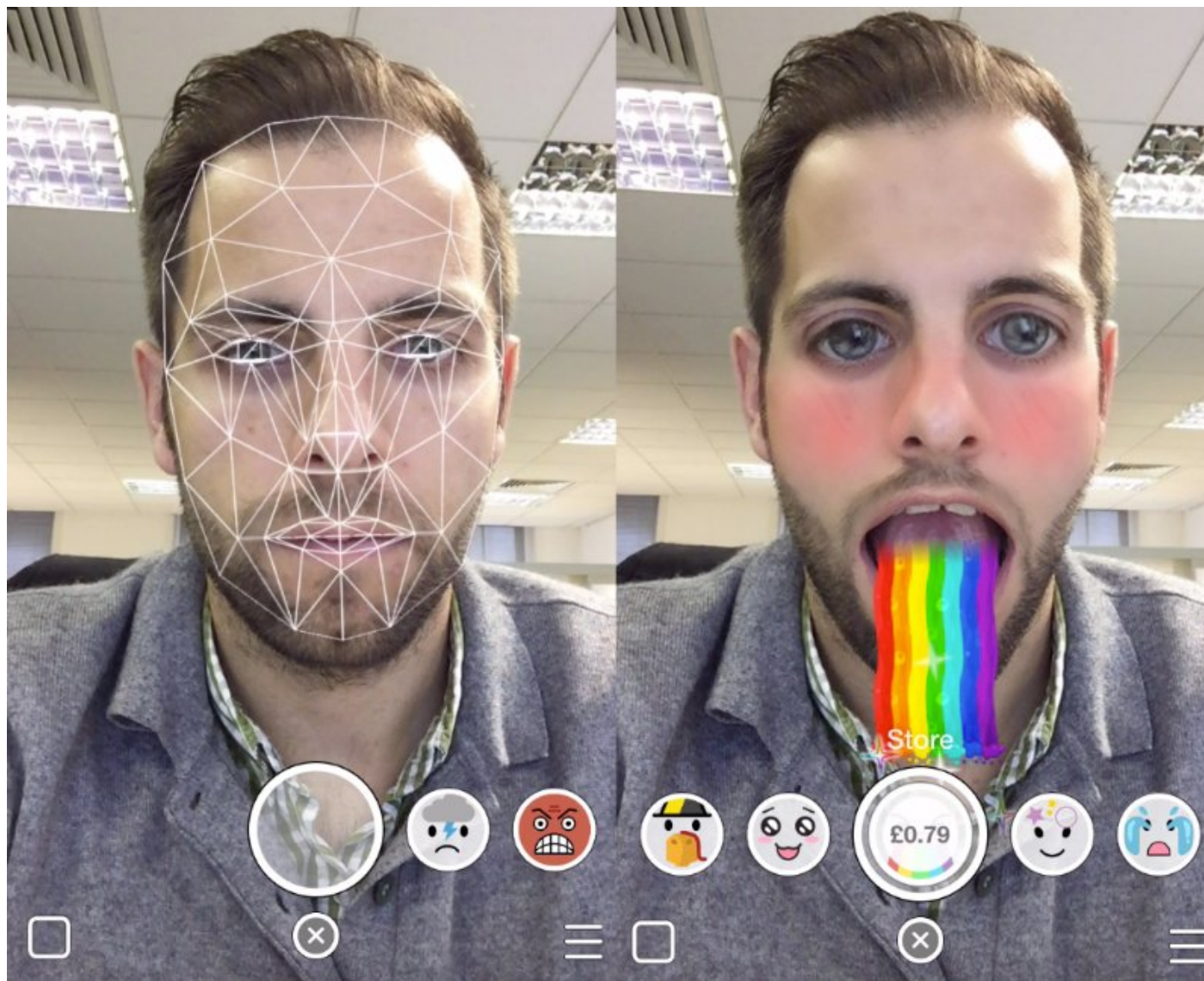
# Computational Photography







# Image Processing





# Image Processing - Saves Lives

Canadian 'swirl face'  
Thailand

jailed in

August 15, 2008

☆ Res



Images released by Interpol in 2007 show the 'unswirling' of the internet pictures that led to the capture of Christopher Paul Neil.

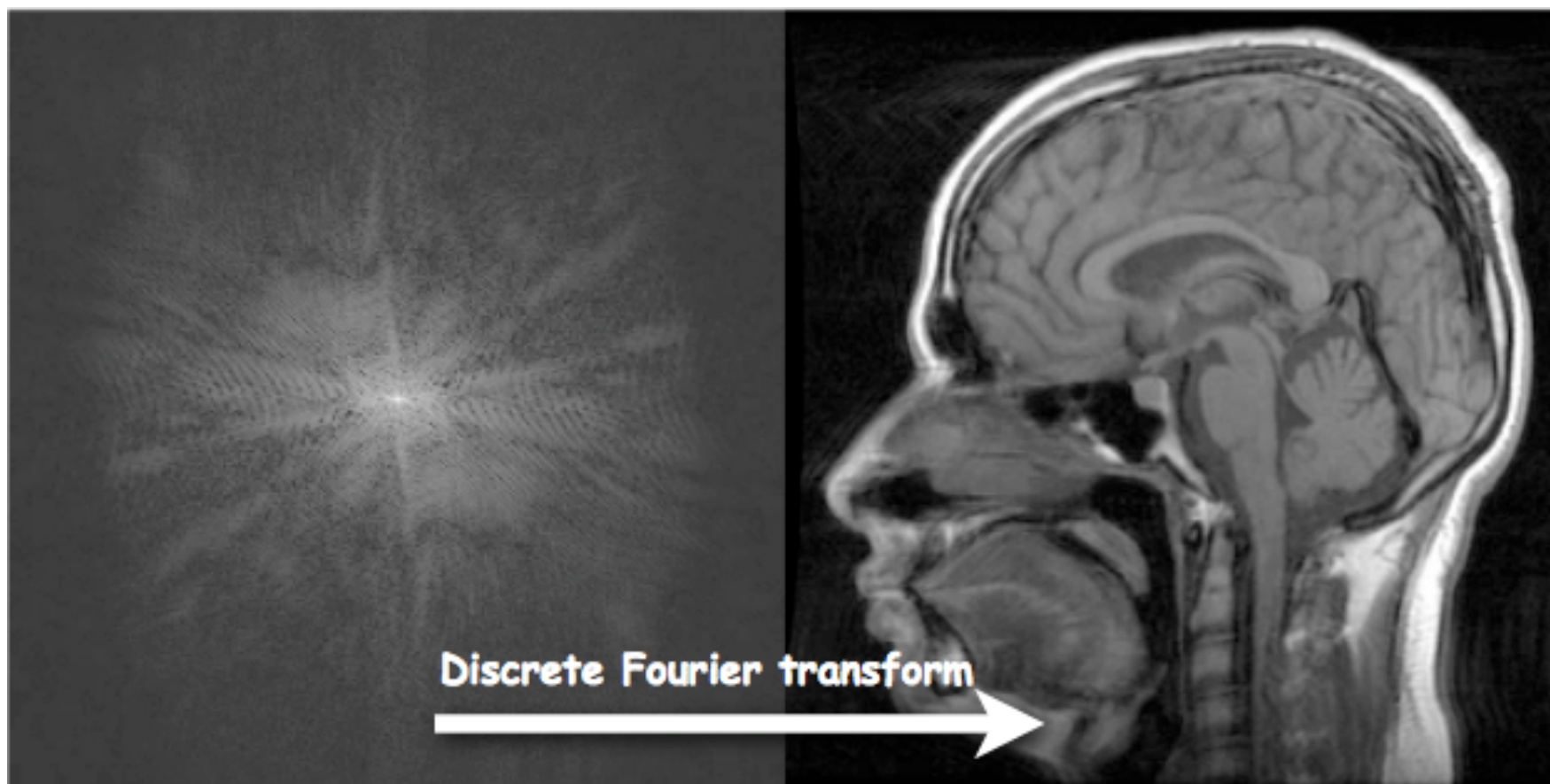


# Example III: MRI

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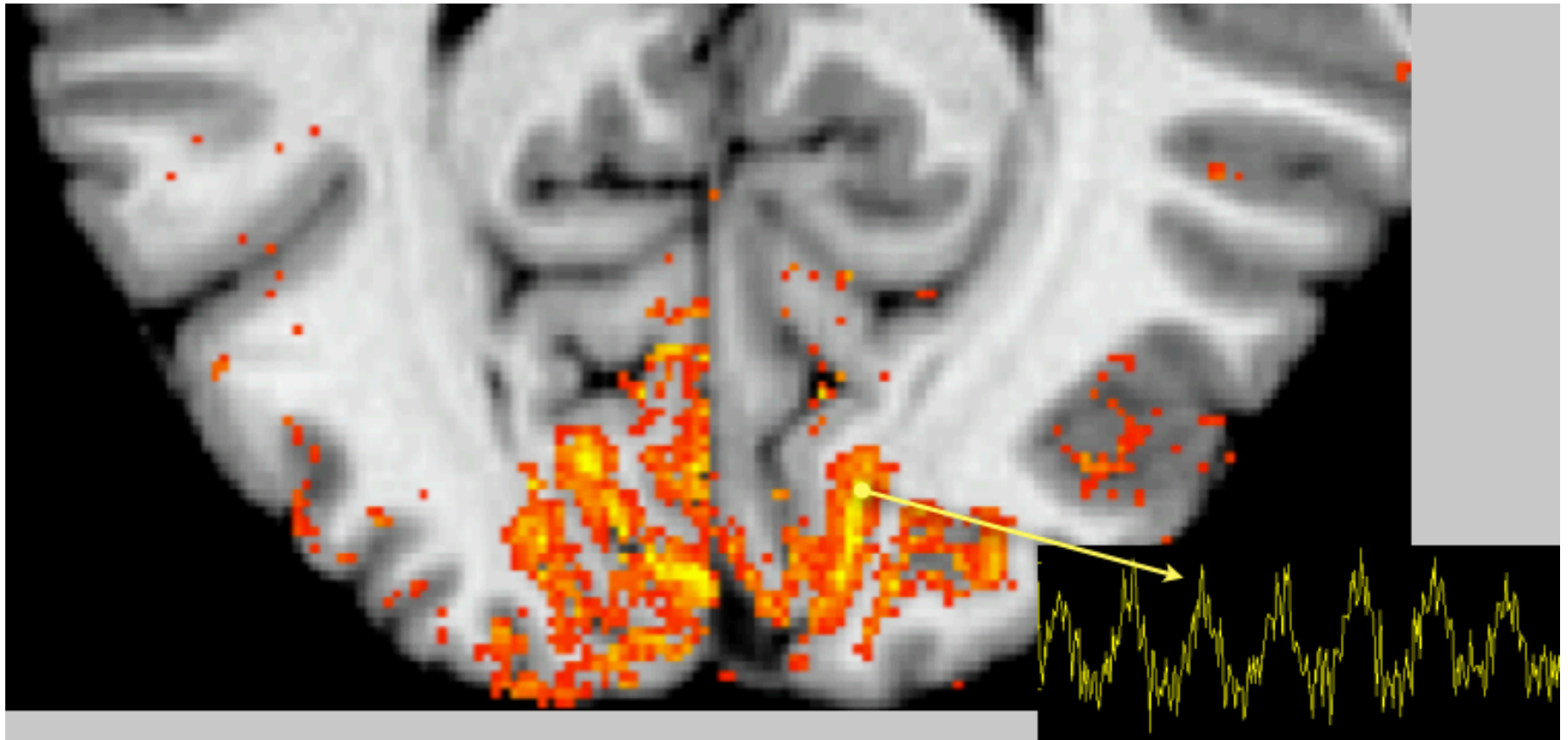
k-space (raw data)

Image



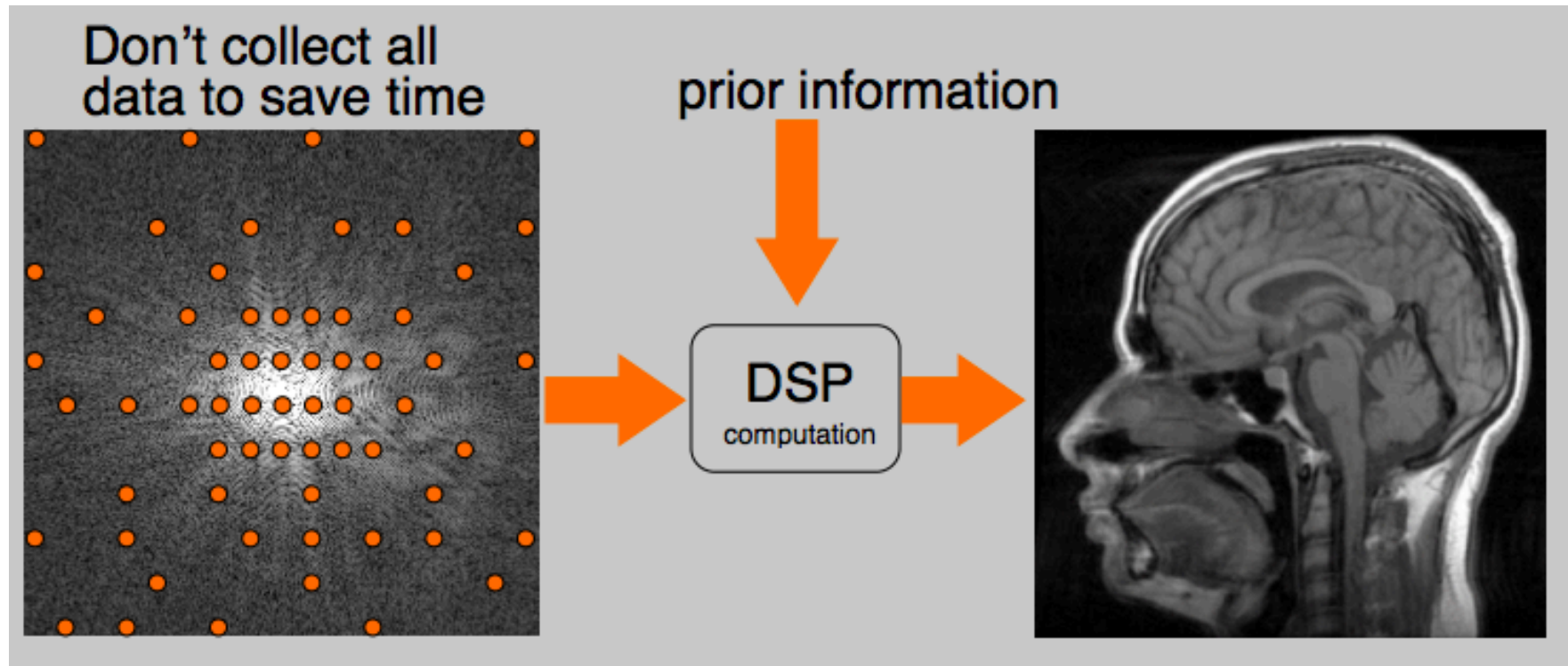
# fMRI example

- Sensitivity to blood oxygenation
  - response to brain activity Convert from one signal to another



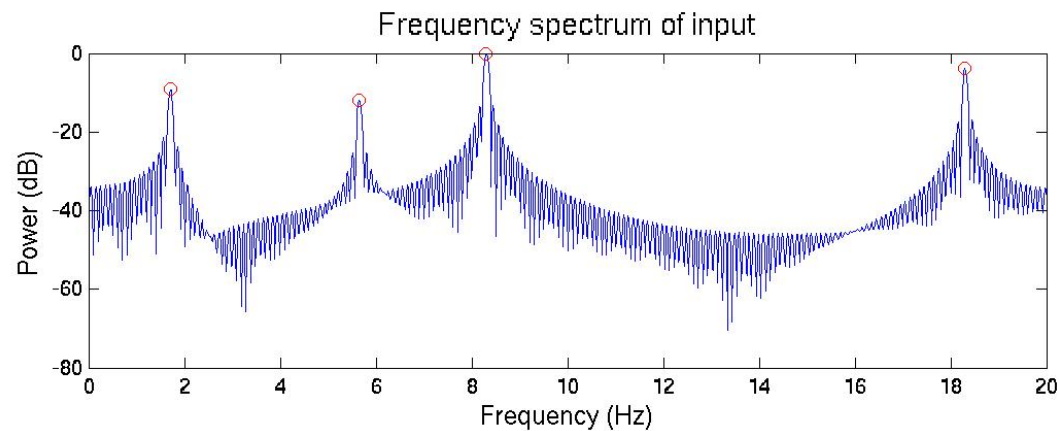
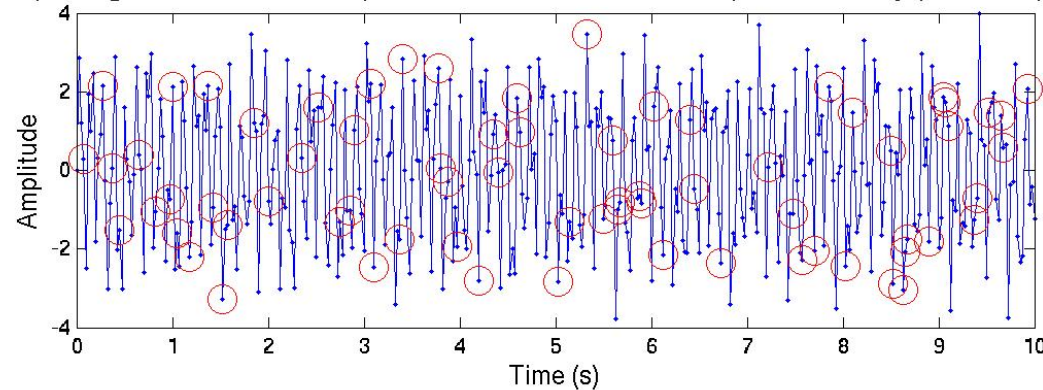
# Compressive Sampling

- ❑ Compression meets sampling



# Example: Sum of Sinusoids

Input signal with undersampled measurements circled ( $\sim 17.5\%$  of Nyquist samples)

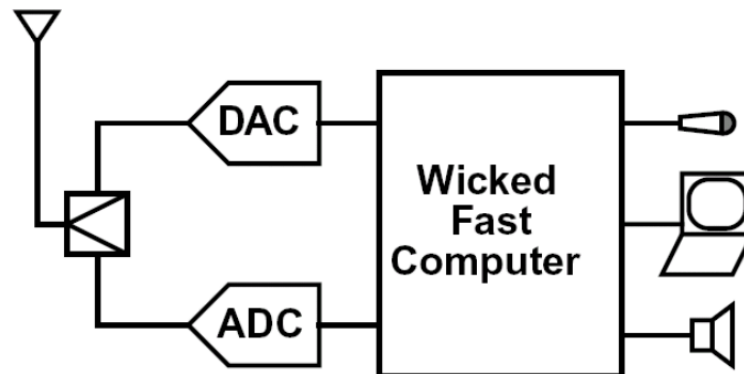


- Sense signal randomly  $M$  times
  - $M > C \cdot \mu^2(\Phi, \Psi) \cdot S \cdot \log N$
- Recover with linear program

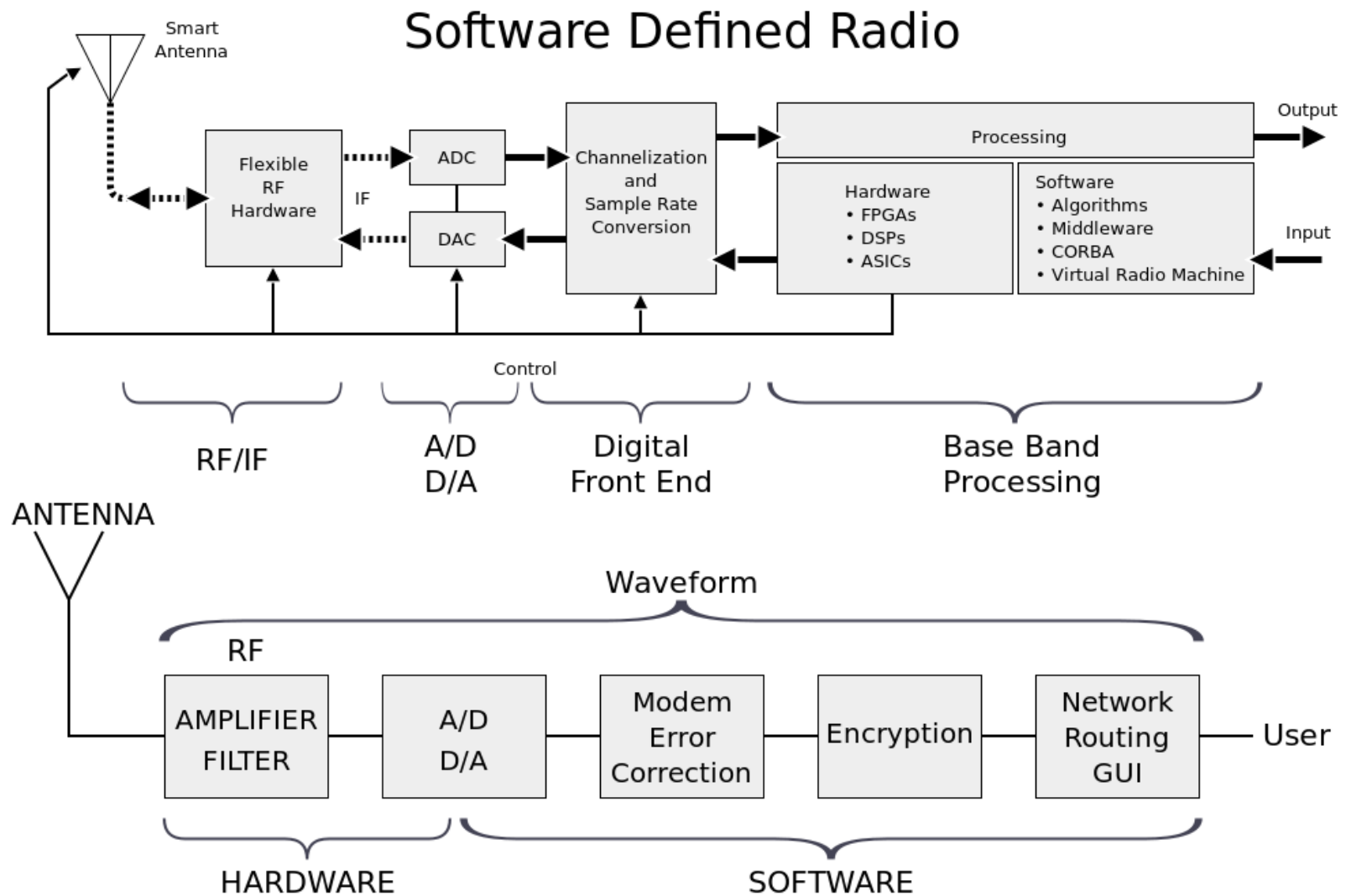


# Example IV: Software Defined Radio

- ❑ Traditional radio:
  - Hardware receiver/mixers/demodulators/filtering
  - Outputs analog signals or digital bits
- ❑ Software Defined Radio:
  - Uses RF front end for baseband signal
  - High speed ADC digitizes samples
  - All processing chain done in software



# Software Defined Radio





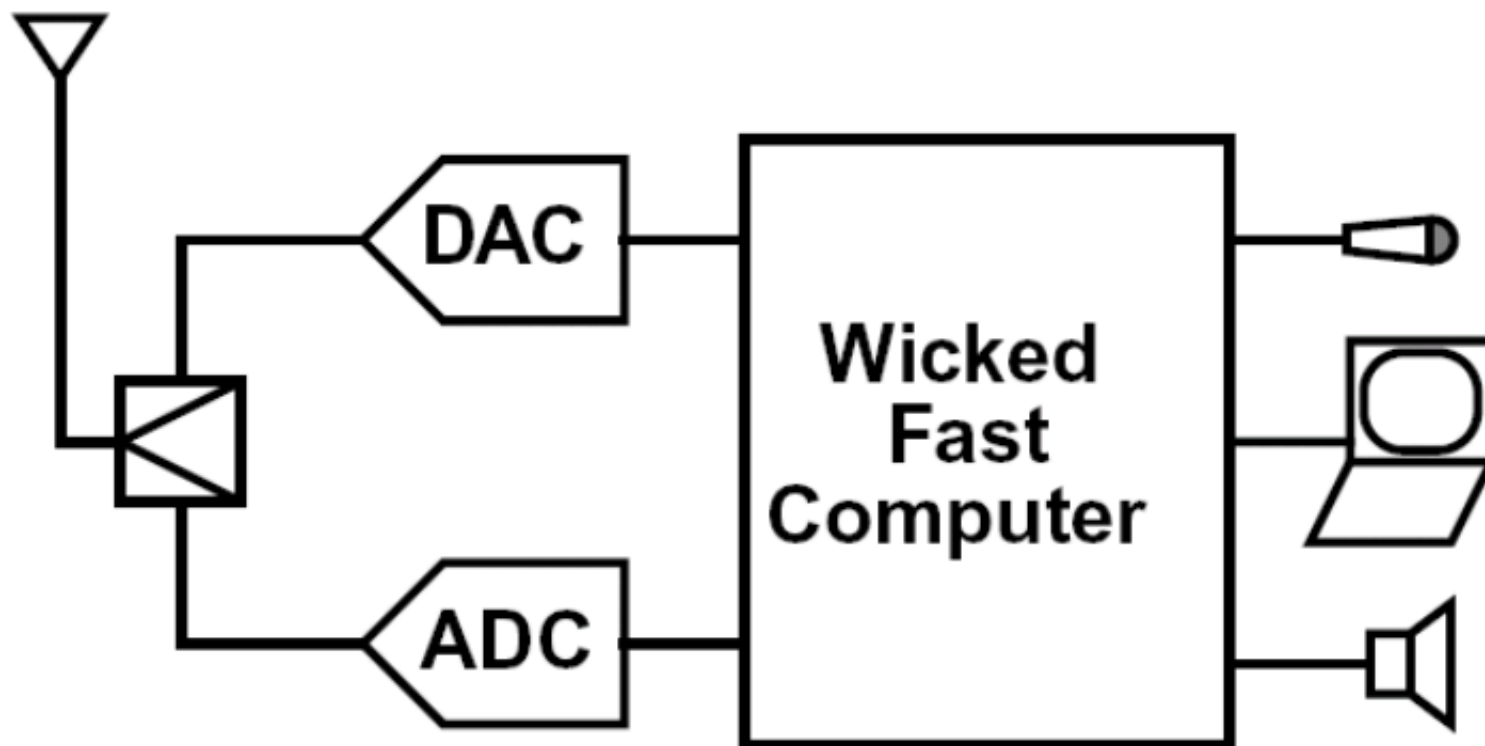
# Software Defined Radio

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- ❑ Advantages:
  - Flexibility
  - Upgradable
  - Sophisticated processing
  - Ideal Processing chain
  - not approximate like in analog hardware
- ❑ Already used in consumer electronics
  - Cellphone baseband processors
  - Wifi, GPS, etc....

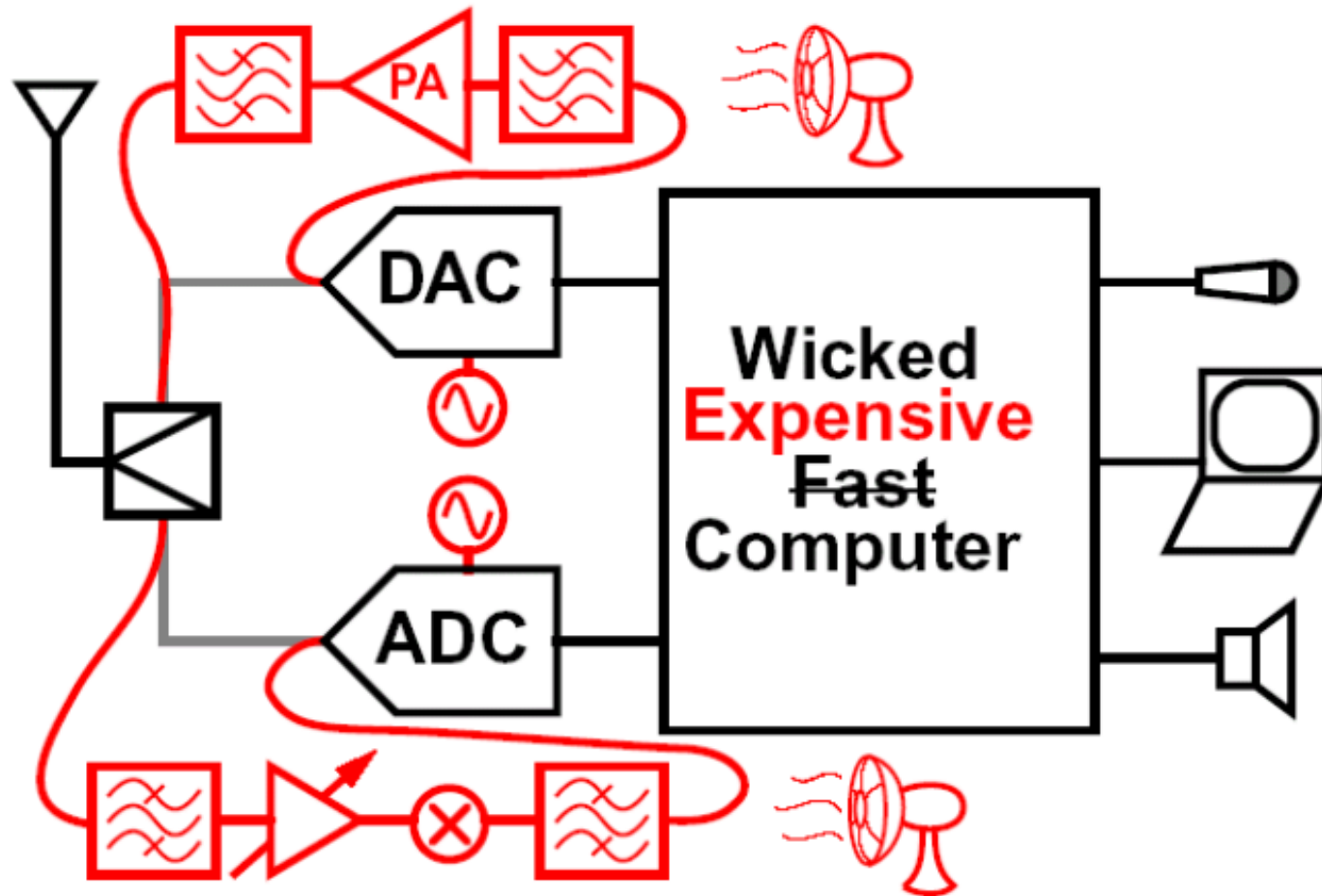


# Software Radio Vision



[Schreier, "ADCs and DACs: Marching Towards the Antenna," GIRAFE workshop, ISSCC 2003]

# Software Radio Reality



[Schreier, "ADCs and DACs: Marching Towards the Antenna," GIRAFE workshop, ISSCC 2003]



# Shameless Plug

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- ❑ If you are interested in how Analog to digital converters work and how to make them
- ❑ Take ESE 568!
- ❑ Good to know both sides of the system



# Future of ADC design

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- ❑ Today's ADCs are extremely well optimized
- ❑ For non-incremental improvements, we must explore new ideas in signal processing that tackle ADC inefficiency at the system level
  - Compressed sensing
  - Finite innovation rate sampling
  - Other ideas?

# Filter Design Example

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# Optimal Filter Design

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- ❑ Window method
  - Design Filters heuristically using windowed sinc functions
- ❑ Optimal design
  - Design a filter  $h[n]$  with  $H(e^{j\omega})$
  - Approximate  $H_d(e^{j\omega})$  with some optimality criteria - or satisfies specs.

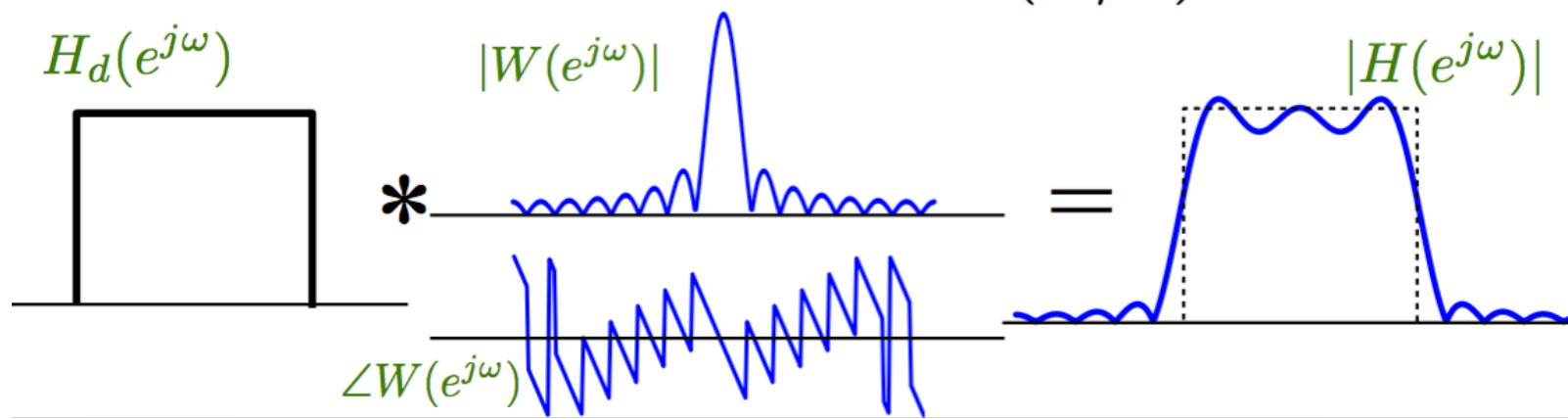
# FIR Design by Windowing

- Desired filter,

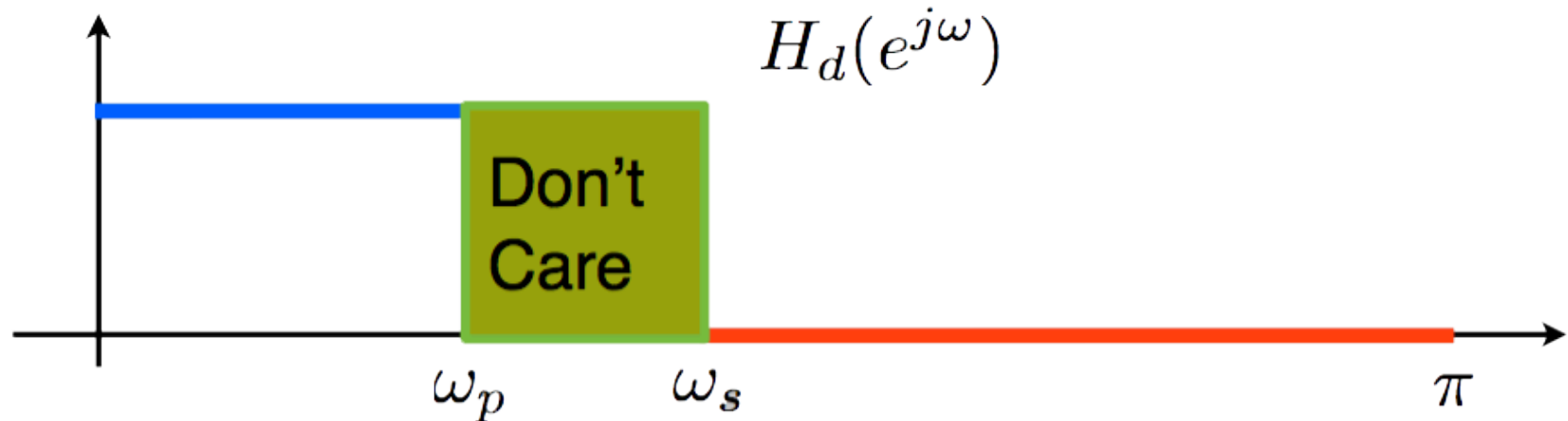
$$H(e^{j\omega}) = H_d(e^{j\omega}) * W(e^{j\omega})$$

- For Boxcar (rectangular) window

$$W(e^{j\omega}) = e^{-j\omega \frac{M}{2}} \frac{\sin(\omega(M+1)/2)}{\sin(\omega/2)}$$



# FIR Design by Optimality



□ Least Squares:

$$\text{minimize} \int_{\omega \in \text{care}} |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$

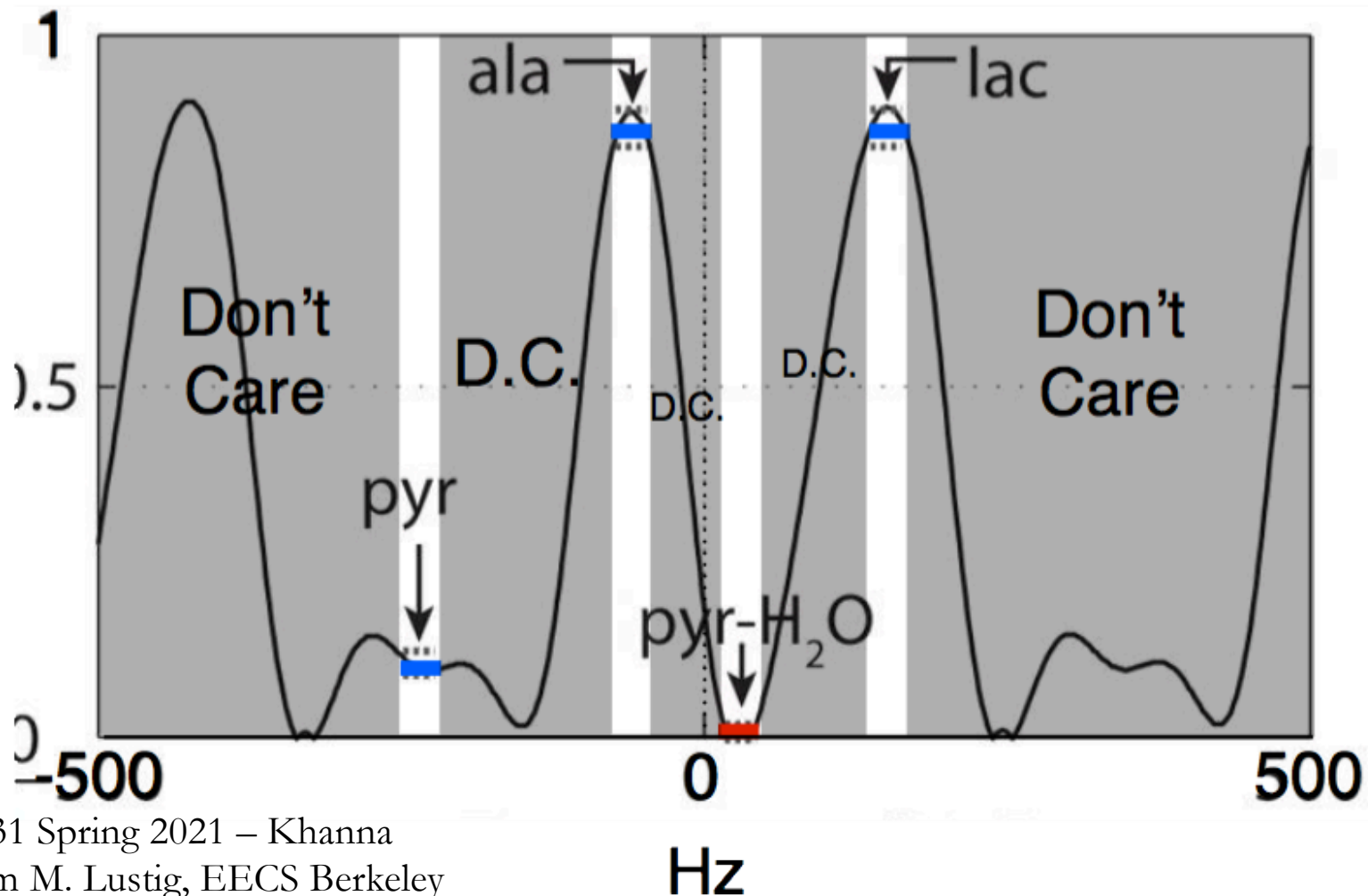
□ Variation: Weighted Least Squares:

$$\text{minimize} \int_{-\pi}^{\pi} W(\omega) |H(e^{j\omega}) - H_d(e^{j\omega})|^2 d\omega$$



# Example of Complex Filter

- ❑ Larson et. al, “Multiband Excitation Pulses for Hyperpolarized  $^{13}\text{C}$  Dynamic Chemical Shift Imaging” JMR 2008;194(1):121-127
- ❑ Need to design 11 taps filter with following frequency response:





# Admin

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- ❑ Find web, get text, start HW 0 and assigned reading...
  - <http://www.seas.upenn.edu/~ese531>
  - <https://piazza.com/upenn/spring2021/ese531/>
  - <https://canvas.upenn.edu/>
- ❑ Accessibility Survey due 1/24
- ❑ Diagnostic quiz due 1/31
  - Use for review of material