## **Signal Processing for MRI**

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A Philosophical Debate Do we live in a digital world or an analog world?



#### Auguste Rodin, bronze, ca. 1880



### An Engineering Reality We live in a digitized world.

Andrew Lipson, LEGO bricks, ca. 2000

## **Fundamental fact:**

MRI data is acquired in k-space
x-space is just a derived quantity (which we happen to be interested in)

## Therefore, we need to understand:

Data



Image

## **Plan: to demonstrate that**

 The basic concepts of time/frequency signal processing can be carried over to MRI

•  $\Delta k_x$  and  $\Delta k_v$  are the relevant sampling intervals

 The imaging equation defines the transformation between conjugate variables--Fourier

 Sampling and other operations on data are performed in k-space; the convolution theorem supplies the resulting effect in the image

Both DSP and physical effects must be considered

### **Digitization of a Time-Domain Analog Signal**

### **Sampling Interval T<sub>s</sub>**



Data spaced at intervals T<sub>s</sub>

# Sampling during MRI signal acquisition



## Sampling in k-space



### For both dimensions: data is spaced at intervals $\Delta k$



Data in k-space is (usually) regularly sampled on a grid.

This sampling is entirely analogous to sampling of time-domain data: intervals are ∆k<sub>x</sub> and ∆k<sub>y</sub> instead of interval T<sub>s</sub>

**Significance of this:** 

• From a post-processing point of view, read and phase directions in MRI can be handled in an identical fashion

 Much of what you already know about signal processing of sampled time-domain signals can be immediately carried over to MRI

## The k-space sampling function is written:

$$Comb(k;\Delta k) = \sum_{n=-N/2}^{+N/2} \delta(k - n\Delta k)$$

N, number of sampled points in k<sub>x</sub> or k<sub>y</sub>



k-space data are numbers assigned to each grid point: These are the samples s<sub>samp</sub>(k<sub>x</sub>, k<sub>y</sub>)

Conceptually, we can consider  $s_{samp}(k_x, k_y)$  to be a sampled version of some continuous function,  $s(k_x, k_y)$ 

### The above dealt with signal acquisition

To proceed: consider the physics

## Relationship Between Signal and Precessing Spins During Read



## **Signal and Spins During Read**

Signal from dx dy:  $s(t; x, y) dx dy = \rho(x, y) e^{-i 2\pi v t} dx dy$ 



# Consideration of the phase encode gradient leads to the celebrated *imaging equation*

...relates k-space data,  $s(k_x, k_y)$  to the image,  $\rho(x,y)$ 

$$s(k_{x,k_y}) = \iint \rho(x,y) e^{-i\pi(k_x x + k_y y)} dx dy \equiv \Im \big[ \rho(x,y) \big]$$

$$\rho(x,y) = \iint s(k_x,k_y) e^{+i\pi(k_xx+k_yy)} dk_x dk_y \equiv \mathfrak{I}^{-1} \left[ s(k_x,k_y) \right]$$

Note: the Fourier transform arises from the physics

Combine Fourier transforms with convolution to make use of the all-powerful *Convolution Theorem* 

Convolution of x(t) and h(t)

$$y(t) = \int_{-\infty}^{\infty} x(\tau)h(t-\tau)d\tau = x(t) * h(t)$$

fold  $\rightarrow$  slide  $\rightarrow$  multiply  $\rightarrow$  integrate

### Arises naturally when considering:

the observable effects of intended or unintended actions on data

digital filters



## **The Convolution Theorem**

### Ingredients: h(t) and g(t), and their Fourier transforms H(v), G(v) S = Fourier transform S<sup>-1</sup> = Inverse Fourier transform • = multiplication \* = the convolution operator

4 ways of writing the convolution theorem:

I. ℑ {f\*g} = F • G
II. ℑ {f • g} = F \* G
III. ℑ<sup>-1</sup>{F • G} = f \* g
IV. ℑ<sup>-1</sup>{F \* G} = f • g

### Our application is based on the imaging equation:

$$\Im \{\rho(\mathbf{x}, \mathbf{y})\} = \mathbf{s}(\mathbf{k}_{\mathbf{x}}, \mathbf{k}_{\mathbf{y}})$$

 $\Im^{-1} \{ \mathbf{s}(\mathbf{k}_{\mathbf{x}}, \mathbf{k}_{\mathbf{y}}) \} = \rho(\mathbf{x}, \mathbf{y})$ 

Visible effect on the image

Version III.  $\Im^{-1}$  {s • H} =  $\rho^*$  h

Various non-idealities or filters

Ideal data in k space

 $\rho$  = the ideal image

With this, we can understand the effects that sampling, truncation, and relaxation in k-space have on the image

•Aliasing direct sampling effect

•The point spread function *truncation--signal processing relaxation--physics* 

## Aliasing aka wrap-around, aka fold-over



Thus, high frequency sinusoids, when sampled, can be mis-assigned to a lower frequency!

To avoid this, sample at a rate  $v_s = 1/T_s$  which satisfies  $v_s > 2 \cdot v$ where v is the frequency of the sinusoid



This rate, 2 • v is called the *Nyquist* rate,  $v_N$ To avoid aliasing:  $v_S > v_N \equiv 2 \cdot v$  Fourier decomposition permits extension of this theorem to a general bandlimited ( $-v_{max}$ ,  $v_{max}$ ) signal, described as:

$$g(t) = \int_{-v_{\text{max}}}^{v_{\text{max}}} G(v) e^{i2\pi v t} dv$$

Then aliasing is avoided by ensuring  $v_{\text{S}} > v_{\text{N}} \equiv 2 \bullet v_{\text{max}}$ 

#### Note: for a non-bandlimited signal, apply an anti-aliasing prefilter:



# The convolution theorem defines the effect on the image of sampling

$$Comb(k;\Delta k) = \sum_{-\infty}^{\infty} \delta(k - m\Delta k)$$

### A straightforward calculation shows:

 $\Im^{-1} \{ Comb(k; \Delta k) \} = 1/\Delta k \cdot Comb(x; 1/\Delta k) \}$ 



### We can now calculate:

 $\rho_{samp}(x) = \Im^{-1} \{s(k) \cdot Comb(k; \Delta k)\}$   $= \Im^{-1} \{s(k)\} * \Im^{-1} \{Comb(k; \Delta k)\}$   $= \rho(x) * Comb(x; 1/\Delta k)$ Obtain replicates, spaced at a distance 1/ \Delta k apart



Provided 1/  $\Delta k > L$ , there is no overlap and correct reconstruction is possible

## Using only the convolution theorem, we found that we can avoid aliasing by selecting

L/2

This is equivalent to the Nyquist sampling theorem i)  $\Delta k_x = \gamma G_x T_s$  (definition)

ii)  $\Delta k < 1/L$  (the condition derived above)

i) and ii)  $\Rightarrow$ iii)  $\gamma G_x T_s < 1/L$ 

which can be written:

iv)  $T_s < \frac{1}{\gamma G_x L}$ 

using the value of  $v_{max}$ , we obtain **v** = 0  $v_{max} = \frac{\gamma}{2} \frac{L}{2} G_x$ **v**) T<sub>s</sub> < 1/(2  $v_{max}$ ) which can also be written:  $v_s > 2 v_{max} = The Nyquist condition$  Thus, to fit the entire object into the image, one needs to sample in k-spac<mark>e such that ∆k < 1 / L is</mark> satisfied

 $\Delta k$  is called the FOV

This was derived for the read direction, but identical considerations apply in the phase encode direction



FOV = 7.5 cm Aliased in phase encode



FOV = 15 cm Non-aliased

### Point Spread Function Due to Signal Processing Actual data are samples from truncated k-space



Actual: 
$$s(k) \rightarrow \bullet s(k) = s_{trunc}(k)$$

## The convolution theorem can help define the result of this truncation

We will use: 
$$\Im^{-1}\{$$
 (k)} =  $\bigvee^{(k)} = \bigvee^{(k)} \bigvee^{(k)/x} = \bigvee^{($ 

### The resulting 1-D image is given by:

 $\rho_{trunc}(\mathbf{x}) = \mathfrak{I}^{-1}\{\mathbf{s}_{trunc}(\mathbf{k})\}$ 

=  $\mathfrak{I}^{-1}$ { Rect(k) • s(k<sub>x</sub>)}

= ℑ<sup>-1</sup>{ Rect(k)} \* ℑ<sup>-1</sup>{s(k)}

$$= \frac{\sin(2\pi k_{\max}x)}{2\pi k_{\max}x} * \rho(x)$$

### Therefore, a delta function density distribution in one dimension becomes:



Ideal point object

**Smearing from truncation** 

Actual image: blurred

### More truncation gives more blurring





### In two dimensions: two dimensional truncation!

$$s(k_x, k_y) \rightarrow s(k_x, k_y) = s_{trunc}(k_x, k_y)$$

The image is given by:

$$\rho_{trunc}(x, y) = \mathfrak{I}^{-1}\left\{s_{trunc}(k_x, k_y)\right\}$$

Conv Thm 
$$\rightarrow = \mathfrak{I}^{-1}\left\{rect(k_x, k_y)\right\} * \mathfrak{I}^{-1}s(k_x, k_y)$$

$$=\frac{\sin(2\pi k_{x,\max}x)}{2\pi k_{x,\max}x} \cdot \frac{\sin(2\pi k_{y,\max}y)}{2\pi k_{y,\max}y} * \rho(x,y)$$

### What does this point spread function look like?



**Truncation pattern in 2D k-space** 

**PSF in 2D x-space** 

# As in 1 dimension, width of point spread function is inverse to width of truncation



### **Truncation pattern in 2D k-space**





#### **PSF in 2D x-space**



## Next example: Point Spread Function Due to *Physics* The effect of $T_2^*$ decay



In the gradient echo experiment, both  $T_2$  and  $T_2$  decay start from the beginning of each k-space line, at  $-k_{max}$ 



## **Gradient echo sequence**



$$s(k) \bullet rect(k) \rightarrow s(k) \bullet rect(k) \bullet e^{-t/T_2^*}$$

**Rewrite in terms of k:** 



**Therefore:** 

$$e^{-t/T_2^*} = e^{-TE/T_2^*}e^{-k/\varphi GT_2^*}$$

$$s(k) \bullet rect(k) \rightarrow s(k) \bullet rect(k) \bullet e^{-TE/T_2^*} e^{-k/\varphi GT_2^*}$$

$$PSF(x) = \mathfrak{I}^{-1}\left(rect(k) \bullet e^{-TE/T_2^*} e^{-k/\mathscr{V} GT_2^*}\right)$$

$$= e^{-TE/T_{2}^{*}} \int_{-k_{m}}^{k_{m}} e^{2\pi i k x} e^{-k/\varphi GT_{2}^{*}} dk$$
Note:  $T_{s} = \frac{2k_{m}}{\varphi G_{read}}$ 
Therefore:  $T_{s} = \frac{-k_{m}}{T_{s}} \int_{-k_{m}}^{G_{read}} \int$ 

## $\left| PSF \right| = \left| PSF(x;T_s/T_2^*) \right|$



PSF as for truncation

### **Point Spread Function Due to** *T*<sub>2</sub> *decay*

 $1/T_2^* = 1/T_2 + 1/T_2^{\prime}$ 

In the spin echo experiment • T<sub>2</sub> decay starts from the beginning of each k-space line, at -k<sub>max</sub>

• T<sub>2</sub><sup>'</sup> effectively "starts" at k = 0, in the middle of acquisition



# This PSF can be described by its full width at half-maximum (FWHM)



## **Conclusions:**

• The basic concepts of time/frequency signal processing can be carried over to x-space/k-space in MRI

• The imaging equation defines the relevant Fourier conjugate variables

•  $\Delta k_x$  and  $\Delta k_y$  are the sampling intervals, analogous to  $T_s$ 

 Sampling and other operations on data are performed in k-space; the convolution theorem supplies the resulting effects on the image