

Selectivity and the IF filter

Q: What about the *IF* filter?

I understand that its center frequency f_0^{IF} determines the receiver "Intermediate Frequency",

But, what should this filter bandwidth $\Delta f_{IF} = B_f$ be?

A: Remember, we want only **one** signal (the **desired** signal we down-converted) to appear at the demodulator.

So, the IF filter bandwidth should be **just wide enough** to allow for the desired **signal bandwidth** B_s —but **no wider!**

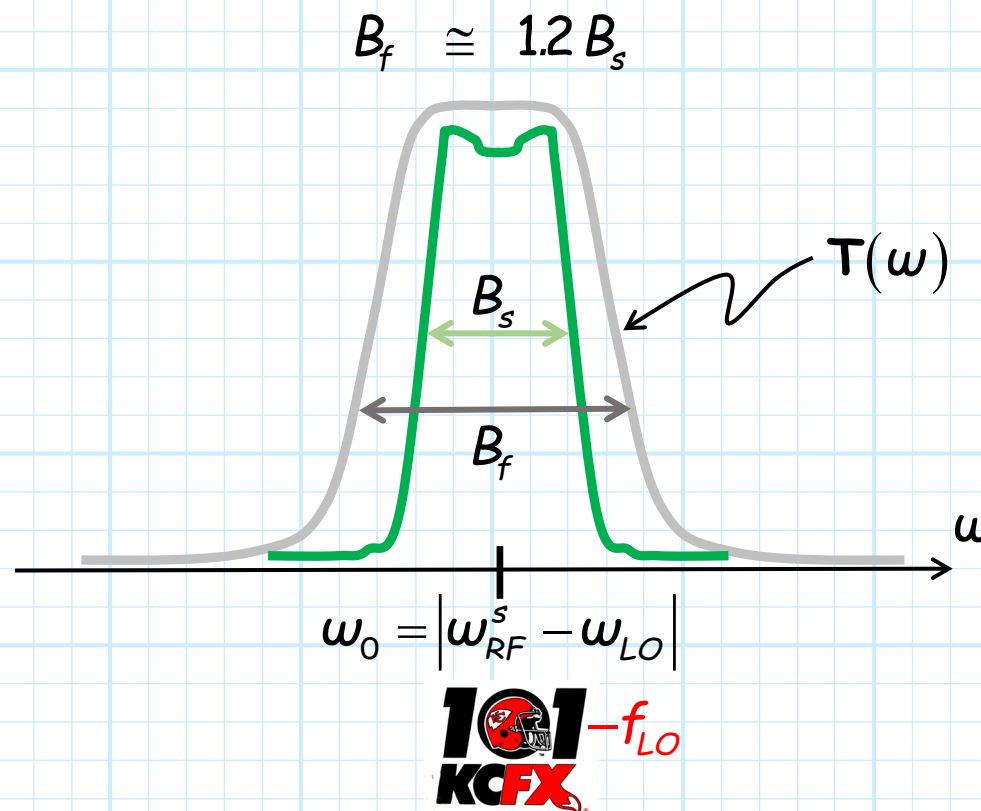
$$B_f \approx B_s$$

Make wide enough—BUT NO WIDER!

But, be **careful!**

To minimize linear distortion, you **might** need to make the IF filter 3dB bandwidth **slightly wider** (e.g. by 20%) than the signal bandwidth.

I.E., set the IF filter bandwidth to be:

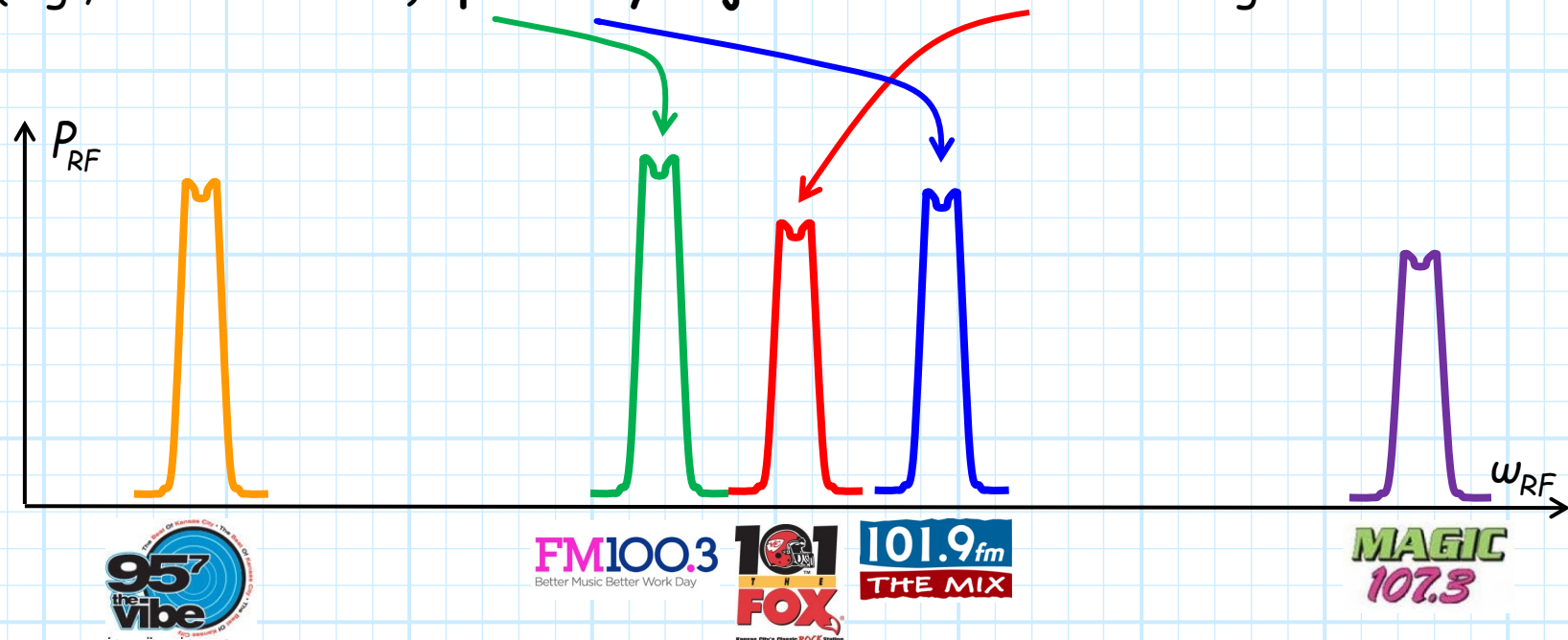


What should the filter order be?

Q: What about the IF filter "roll-off"?

How much stop-band **attenuation** is required by the IF filter?

A: The **most problematic** signals for the IF filter are the **two** signals (e.g., radio stations) **spectrally adjacent** to the **desired** signal.

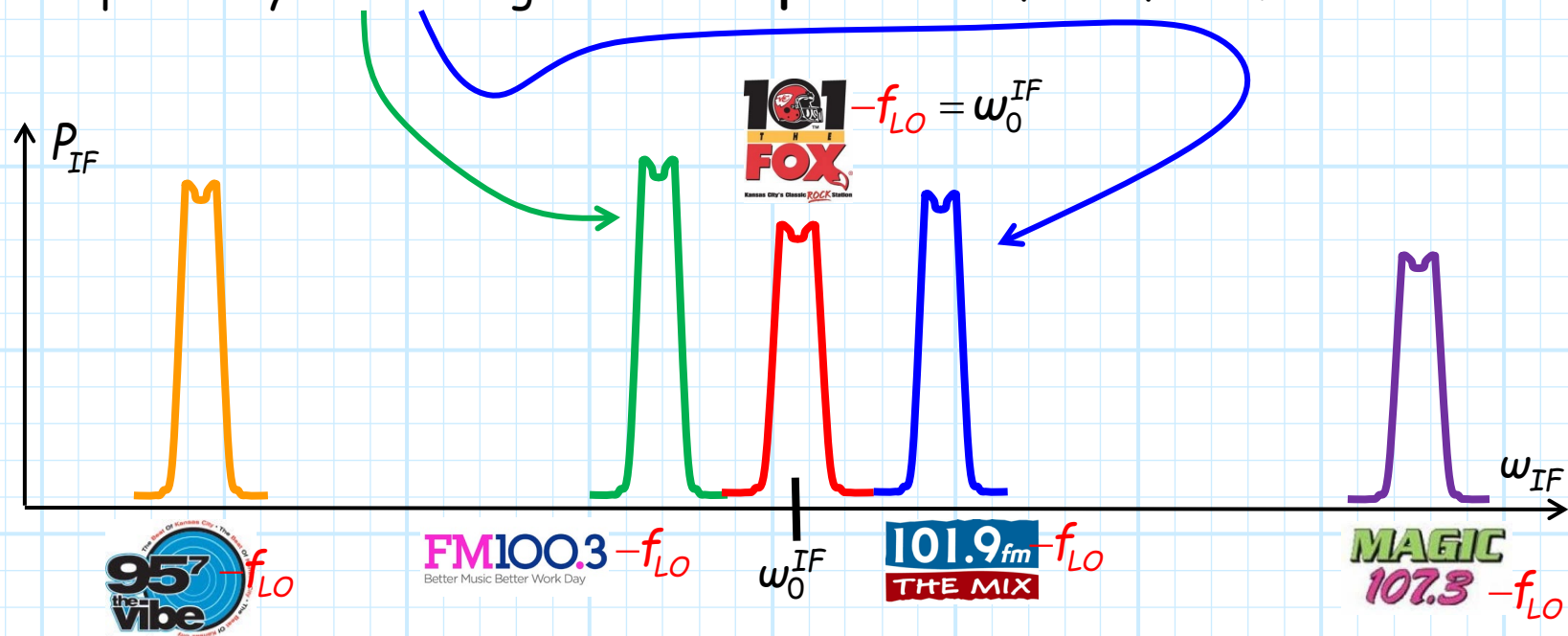


Closest to the IF passband

Q: Why are these "spectrally adjacent" signals so problematic?

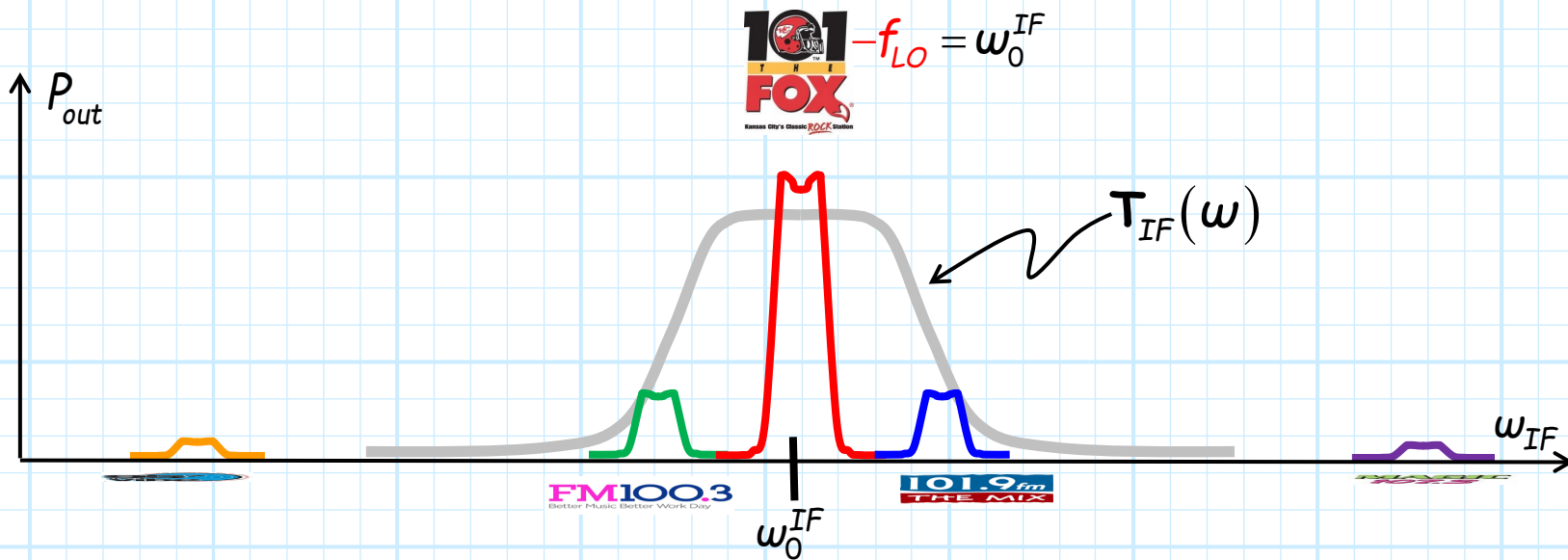
A: Recall the **entire** RF spectrum is "down-converted", such that the desired signal (with frequency ω_{RF}^s) is **centered in the passband** of the IF filter (i.e., $|\omega_{RF}^s - \omega_{LO}| = \omega_0^{IF}$).

The signals **spectrally adjacent** to this desired signal will therefore be the spectrally **closest** signals to the **passband** of the filter.



The spectrally adjacent signals are attenuated the least

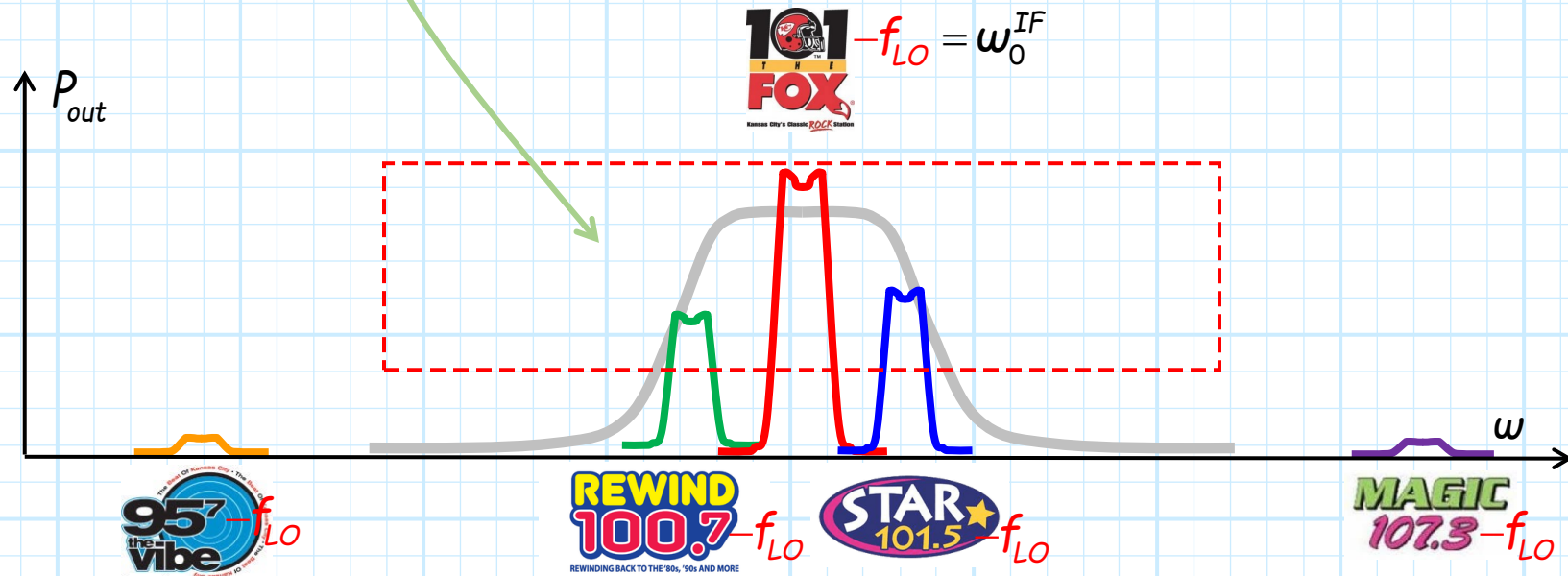
Therefore, after passing through the IF filter, **these spectrally adjacent signals are attenuated the least** of any signal outside of the passband.



Catastrophe!

Q: What happens if the **attenuation** of these adjacent signal is **insufficient**?

A: Then potentially, **more than one** signal will be presented to the **demodulator** (two small signals and one large (desired) signal).



The **result** would be a distorted and **inaccurate** demodulated signal $\hat{i}(t)$!

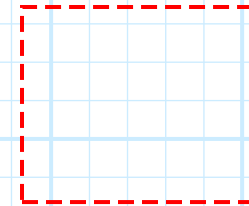


Only one signal in the "window"

Q: So "how far down" should we attenuate these adjacent signals?

A: Remember, the **desired** down-converted signal must have an output power **within** the demodulator "window":

$$P_D^{max} \geq P_{out}^s \geq P_D^{min}$$



In a **properly designed** receiver, **all** other output signals (including adjacent signals) are **attenuated** to the point that they are **less than** P_D^{min} .

$$P_{out}^{adj} < P_D^{min}$$

where P_{out}^{adj} represents the power of **spectrally adjacent** signals at the **output** of the receiver (i.e., the **input** of the demodulator).

Demodulator Dynamic Range

Now, say that the desired signal at the demodulator input is at its **largest acceptable value** (it's at the very top of the window!):

$$P_{out}^s = P_D^{\max}$$

In order for:

$$P_{out}^{adj} < P_D^{\min}$$

We find that **this** must be true:

$$\frac{P_{out}^s}{P_{out}^{adj}} > \frac{P_D^{\max}}{P_D^{\min}} \quad !!!$$

The ratio:

$$\frac{P_D^{\max}}{P_D^{\min}} \doteq DR_D$$

is known as the **Dynamic Range** of the **demodulator**.

Decibel operators!!!!

Note that dynamic range DR_D is a **unitless** value, and so is typically expressed with the **decibel operator**:

$$\begin{aligned} dB[DR_D] &= dB\left[\frac{P_D^{\max}}{P_D^{\min}}\right] \\ &= dBm[P_D^{\max}] - dB[P_D^{\min}] \end{aligned}$$

Similarly:

$$dB\left[\frac{P_{out}^s}{P_{out}^{adj}}\right] = dBm[P_{out}^s] - dBm[P_{out}^{adj}]$$

So that the power of the attenuated adjacent signal must satisfy:

$$dBm[P_{out}^{adj}] < dBm[P_{out}^s] - dB[DR_D]$$

Make this make sense

Now, **assuming** the power of the **spectral adjacent** signals P_{IF}^{adj} (before being attenuated by the filter) is **approximately** that of the **desired signal** power P_{out}^s , we can conclude alternatively that:

$$dBm[P_{out}^{adj}] < dBm[P_{IF}^{adj}] - dB[DR_D]$$

Meaning:

$$dB[DR_D] < dBm[P_{IF}^{adj}] - dBm[P_{out}^{adj}]$$

Note that the value:

$$dBm[P_{IF}^{adj}] - dBm[P_{out}^{adj}]$$

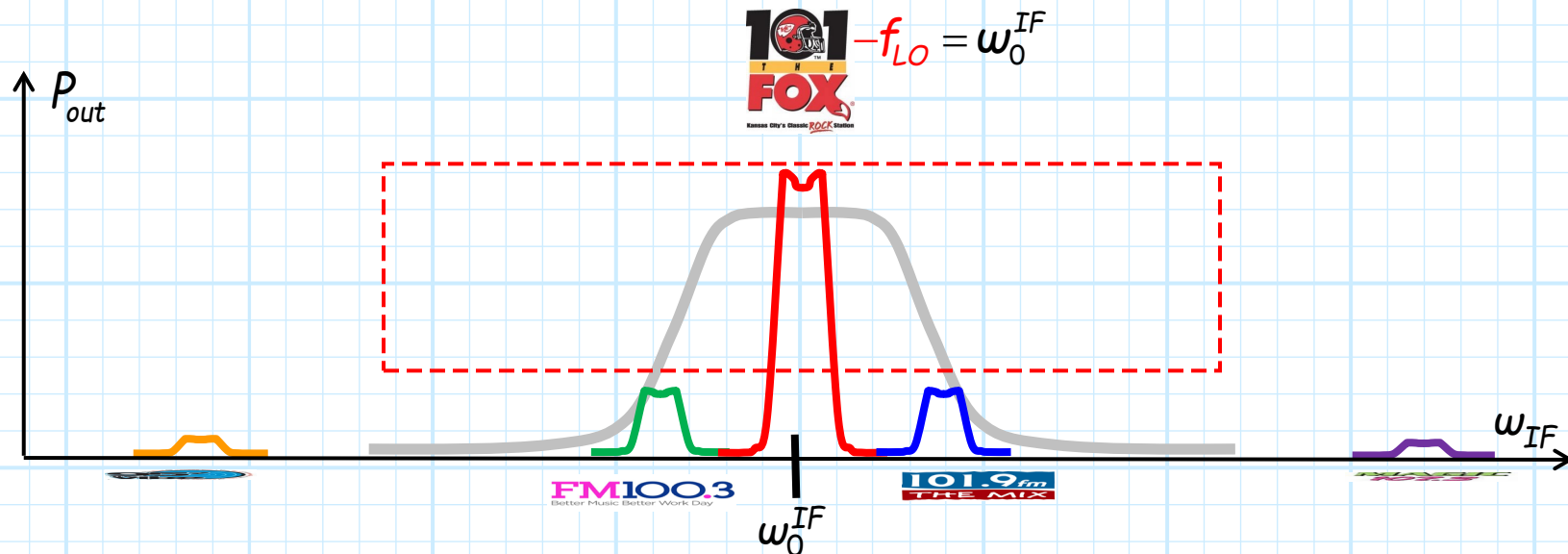
is simply the **attenuation** (in dB) applied to the spectrally adjacent signals by the IF bandpass filter.

Attenuation must be greater than demodulator Dynamic Range!

Thus, the equation

$$dB[DR_D] < dBm[P_{IF}^{adj}] - dBm[P_{out}^{adj}]$$

Simply "says" that the **attenuation of adjacent signals** should be **greater** than the **dynamic range of the demodulator!**

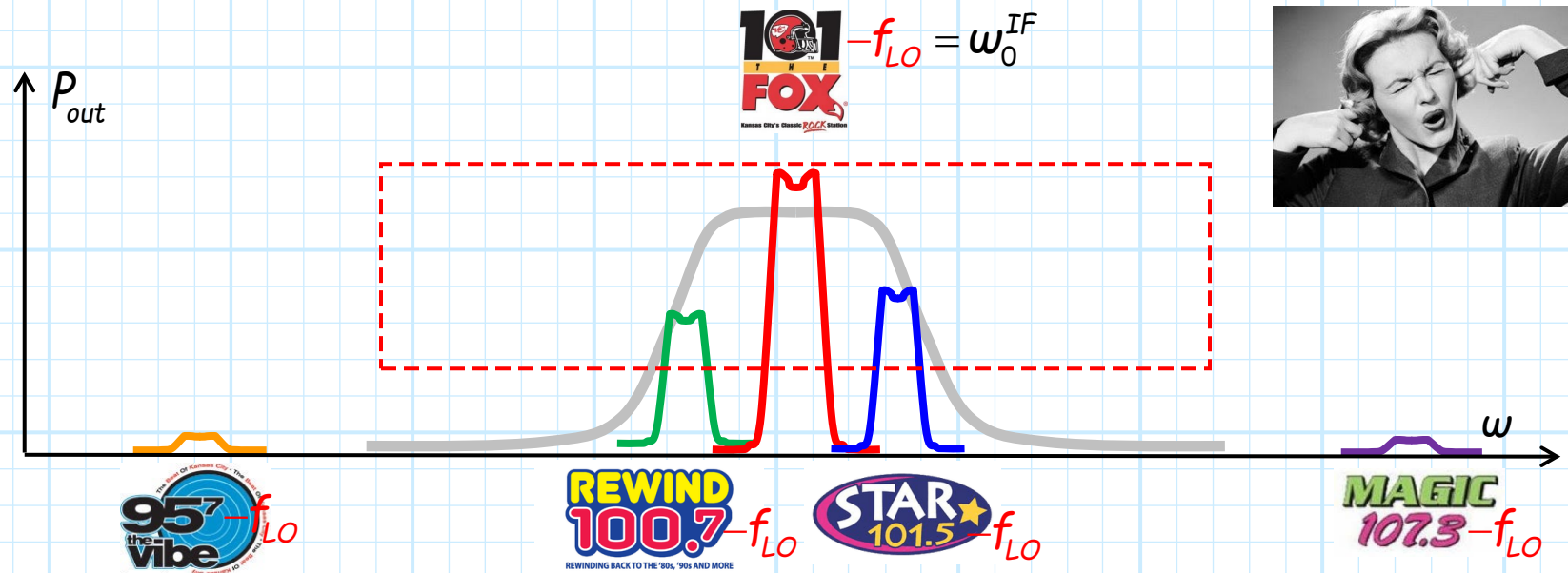


Make. It. Stop!

Q: The attenuation of these spectrally adjacent signals would seem to depend on how "close" they are to the desired signal frequency (and thus the IF filter pass-band).

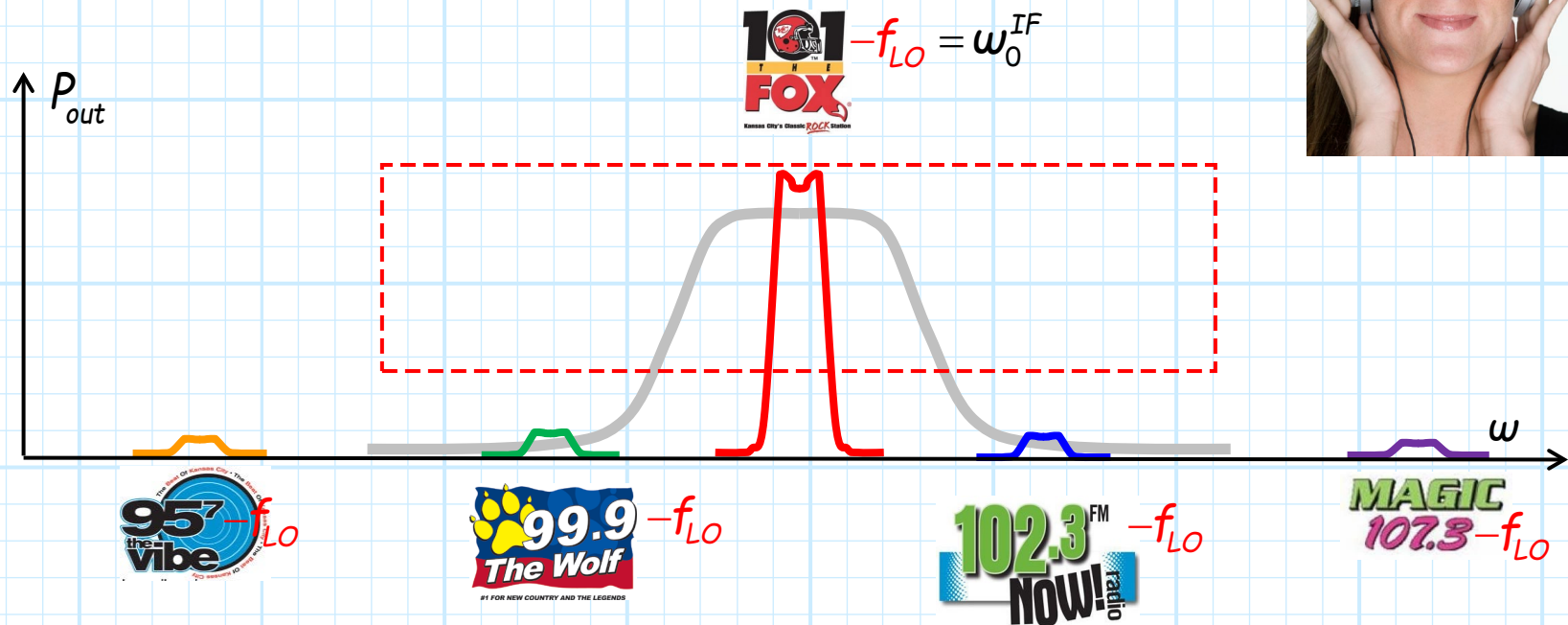
How do we **know where** (i.e., at what frequency) these spectrally adjacent signals **lie**?

A: It is certainly true that "closer" adjacent signal will be **attenuated less** if they are **closer** in frequency to the desired signal.



No issues—sounds great!

But adjacent signals are attenuated **more** if they are **further** from the desired signal:

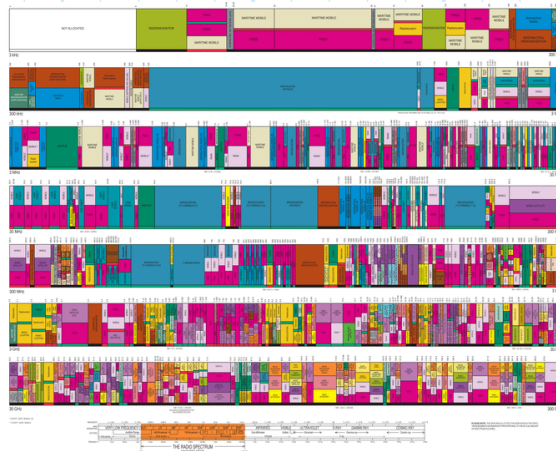


The only agency with an antenna on its seal

Fortunately, we usually know **where** these spectrally adjacent signals are, as their location is **mandated** by the **Federal Communication Commission (FCC)**!



UNITED
STATES
FREQUENCY
ALLOCATIONS
THE RADIO SPECTRUM



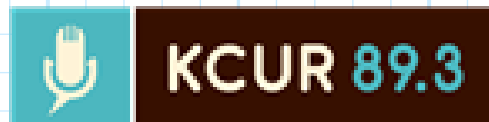
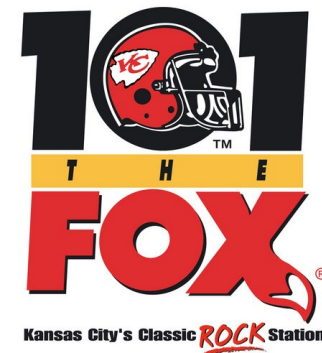
The FCC is charged with **regulating emissions** in the electromagnetic spectrum.

In addition to specifying the **precise** frequencies of transmitted signals, they specify also their **spectral separation**!

Odd multiples of 100 kHz

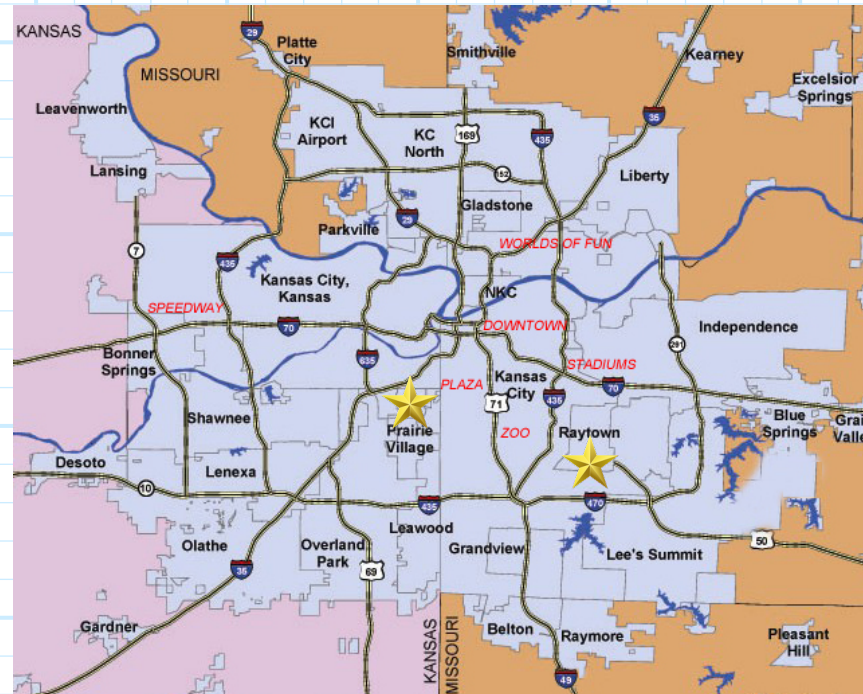
For example, the FCC long ago decided that the FM radio band would span **88 MHz to 108 MHz**.

Moreover, it specified that all FM stations transmit at frequencies that are **odd-multiples of 0.1 MHz** (e.g. 89.3 MHz, 95.7 MHz, 98.9 MHz, 101.1 MHz, 106.5 MHz).



Channel spacing

Likewise, the FCC has traditionally kept **FM stations** that are **geographically close** (e.g., in the **Kansas City metropolitan area**) separated by **at least 0.8 MHz**.



Thus, the "channel spacing" of FM radio is **0.8 MHz**.

One channel-spacing above and below

Thus, the “**worst-case** scenario” for a radio station centered at 101.1 MHz would be to have **spectrally adjacent signals** at:

$$101.1 - 0.8 = 100.3 \text{ MHz}$$



$$101.1 + 0.8 = 101.9 \text{ MHz}$$



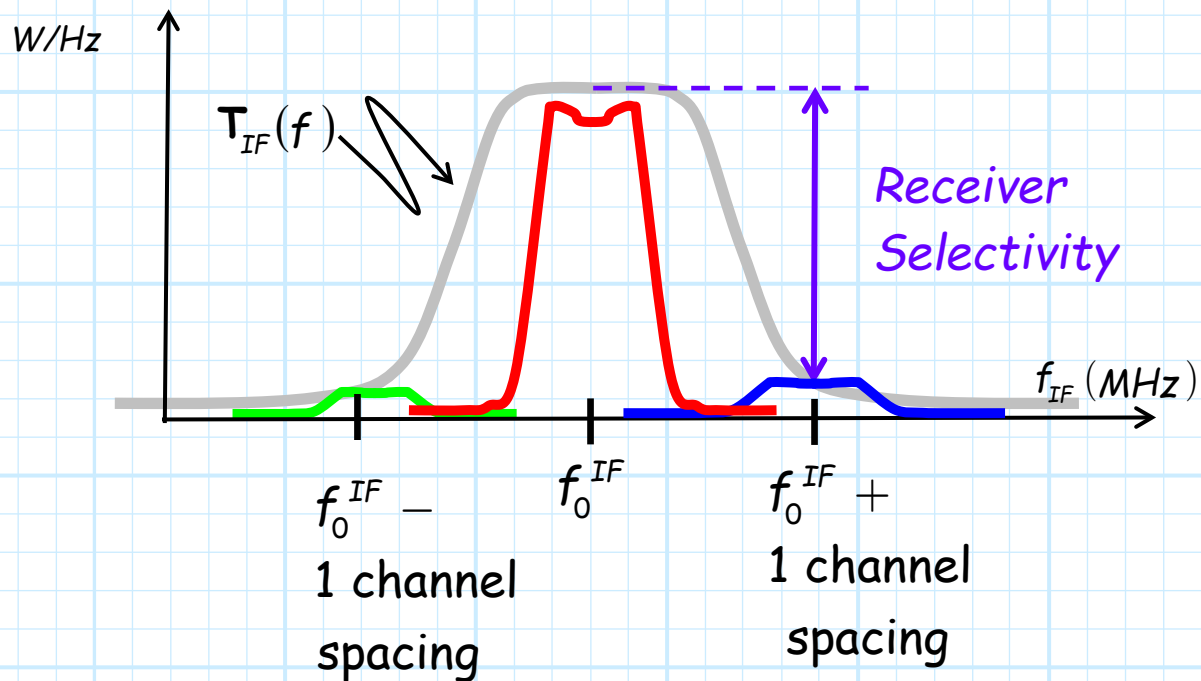
A station broadcasting at 100.3 MHz is thus one “channel spacing” **below** 101.1 MHz, while a station at 101.9 MHz is one “channel spacing” **above** 101.1 MHz.

Now, we can provide a **formal definition** for receiver selectivity!

Receiver selectivity

The amount by which the IF filter attenuates (in dB) adjacent channels is defined as the selectivity of the receiver.

$$\text{Selectivity} = -10 \log_{10} \left[T_{IF}(f_0^{IF} \pm 1 \text{ channel spacing}) \right]$$



Typical values of receiver selectivity range between 30 dB and 60 dB.

Also selectivity

Using our **earlier notation**, we express **this** transmission as:

$$\mathbf{T}_{IF}(f_0^{IF} \pm 1 \text{ channel spacing}) = \frac{p_{out}^{adj}}{p_{IF}^{adj}}$$

So that selectivity can **also** be expresses as:

$$\begin{aligned} \text{Selectivity} &= -10 \log_{10} \left[\mathbf{T}_{IF}(f_0^{IF} \pm 1 \text{ channel spacing}) \right] \\ &= -10 \log_{10} \left[\frac{p_{out}^{adj}}{p_{IF}^{adj}} \right] \\ &= \text{dBm} \left[p_{IF}^{adj} \right] - \text{dBm} \left[p_{out}^{adj} \right] \end{aligned}$$

An important receiver design rule

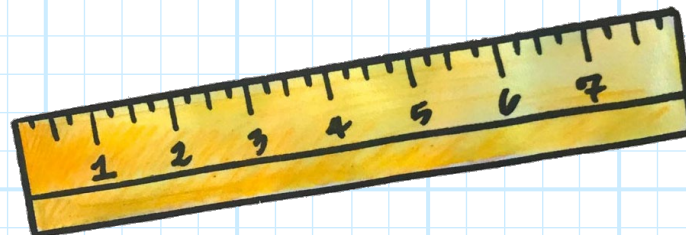
Then our **earlier conclusion**:

$$dB[DR_d] < dBm[P_{IF}^{adj}] - dBm[P_{out}^{adj}]$$

Can **also** be expressed as:

$$\text{Selectivity} > dB[DR_d]$$

The **receiver selectivity** should exceed the **dynamic range** of the demodulator!



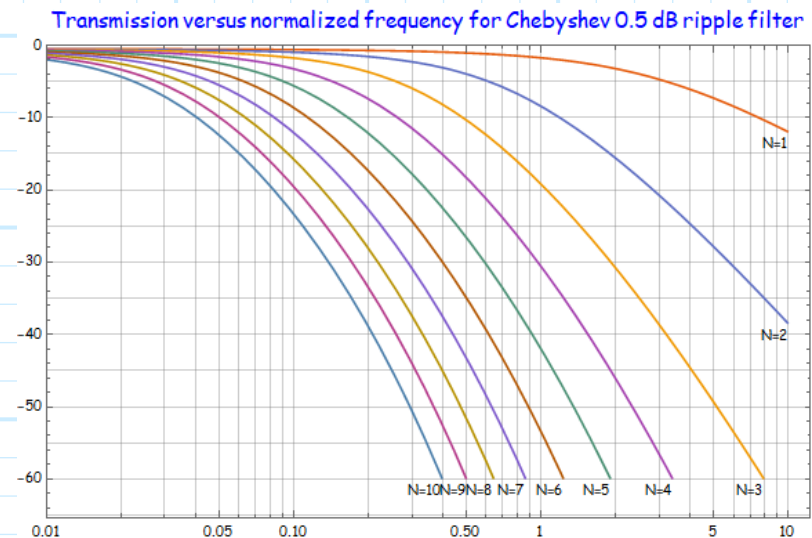
Increase the filter order!

Q: But *what if the selectivity doesn't exceed the dynamic range of the demodulator?*

Since the bandwidth B_f of the filter *cannot* be decreased (i.e., $B_f \cong 1.2 B_s$), how can we *increase Selectivity*?

A: You *increase* receiver selectivity by increasing the *order N* of the IF filter!

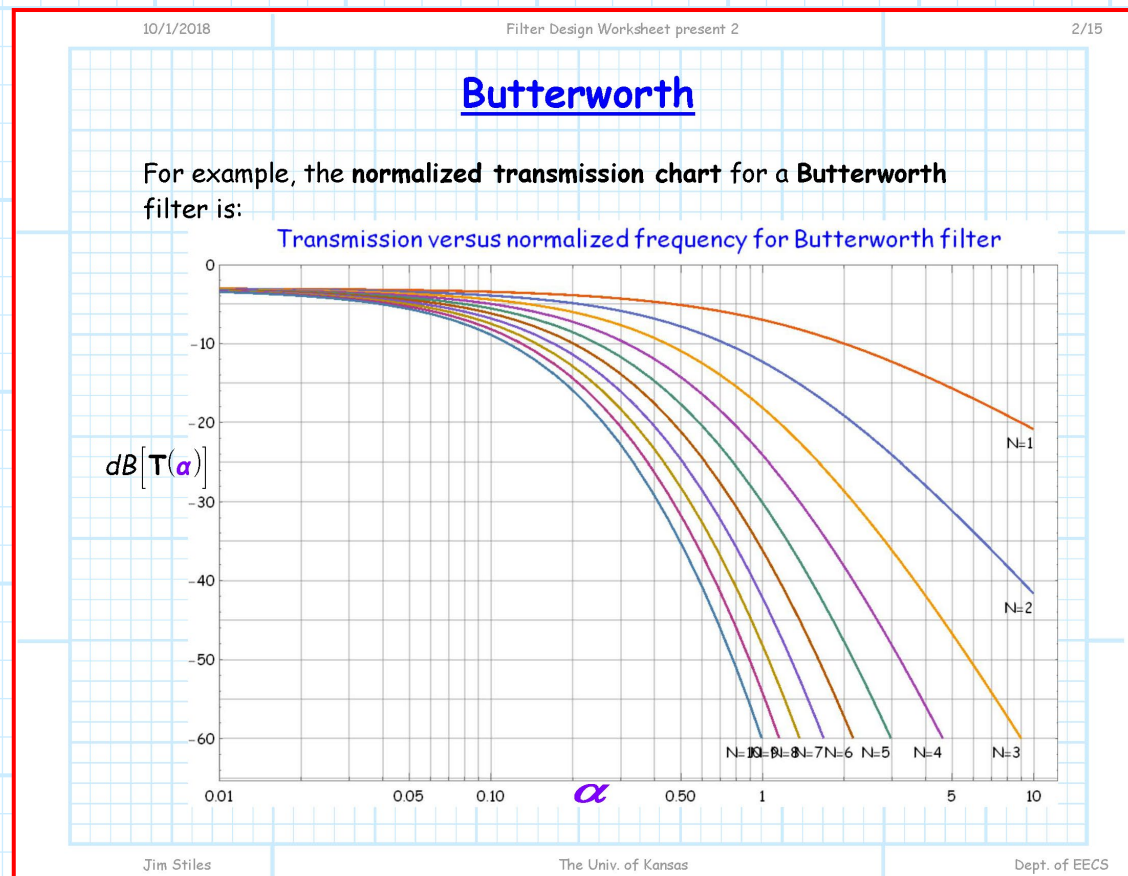
Remember, the larger the order N , the **steeper** the "roll-off"; and thus the **greater** the attenuation and Selectivity.



Remember?

Q: But how would I determine the necessary order value N of the IF filter?

A: Remember your "Filter Design Worksheet" !



You can do this!



For **example**, let's say **you** are designing an **FM radio receiver**—one that will be connected to an FM demodulator with **40 dB** of dynamic range.

Say then your wish for your receiver to have a Selectivity of at least **50 dB** (10 dB **more** than the demodulator dynamic range!).

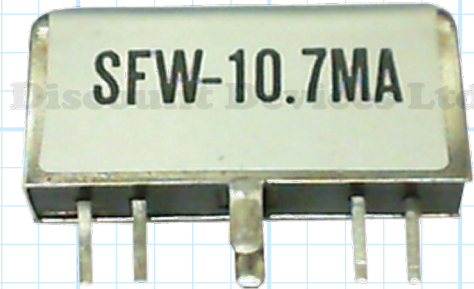
The **bandwidth** of an FM radio **signal** is about $B_s = 0.17 \text{ MHz}$, so **you** make the bandwidth of the IF filter about 200 kHz:

$$B_f \cong 1.2 B_s = 0.2 \text{ MHz}$$

10.7 MHz IF is the standard for FM

A **standard** IF center frequency for FM radio stations is:

$$f_0^{IF} = 10.7 \text{ MHz}$$



Using **this** IF center frequency, the frequency of signals at **one** channel-spacing away from the desired signal would thus arrive at the **IF filter** with frequencies:

$$f_{IF}^{adj} = 10.7 \pm 0.8 \text{ MHz} = 10.1 \text{ MHz}, 11.5 \text{ MHz}$$

Now, the **normalized bandwidth** of the IF filter is:

$$\% \Delta = \frac{f_H - f_L}{f_0} = \frac{B_f}{f_0^{IF}} = \frac{0.2}{10.7} = 0.0187$$

Normalized frequencies

And so for **this** filter, the **normalized** frequency of the **adjacent signal** one channel-spacing **above** is:



$$\begin{aligned}\alpha &= \left| \frac{1}{\% \Delta} \left(\frac{f}{f_0} - \frac{f_0}{f} \right) \right| - 1 \\ &= \left| \frac{1}{0.0187} \left(\frac{11.5}{10.7} - \frac{10.7}{11.5} \right) \right| - 1 \\ &= 6.72\end{aligned}$$

While the **normalized** frequency of the **adjacent signal** one channel-spacing **below** is:

$$\begin{aligned}\alpha &= \left| \frac{1}{\% \Delta} \left(\frac{f}{f_0} - \frac{f_0}{f} \right) \right| - 1 \\ &= \left| \frac{1}{0.0187} \left(\frac{10.1}{10.7} - \frac{10.7}{10.1} \right) \right| - 1 \\ &= 5.18\end{aligned}$$

The lower frequency is the worst case

Notice the **smaller** of these two results:

$$\begin{aligned}\alpha &= \left| \frac{1}{\% \Delta} \left(\frac{f}{f_0} - \frac{f_0}{f} \right) \right| - 1 \\ &= \left| \frac{1}{0.0187} \left(\frac{101}{10.7} - \frac{10.7}{101} \right) \right| - 1 \\ &= 5.18\end{aligned}$$

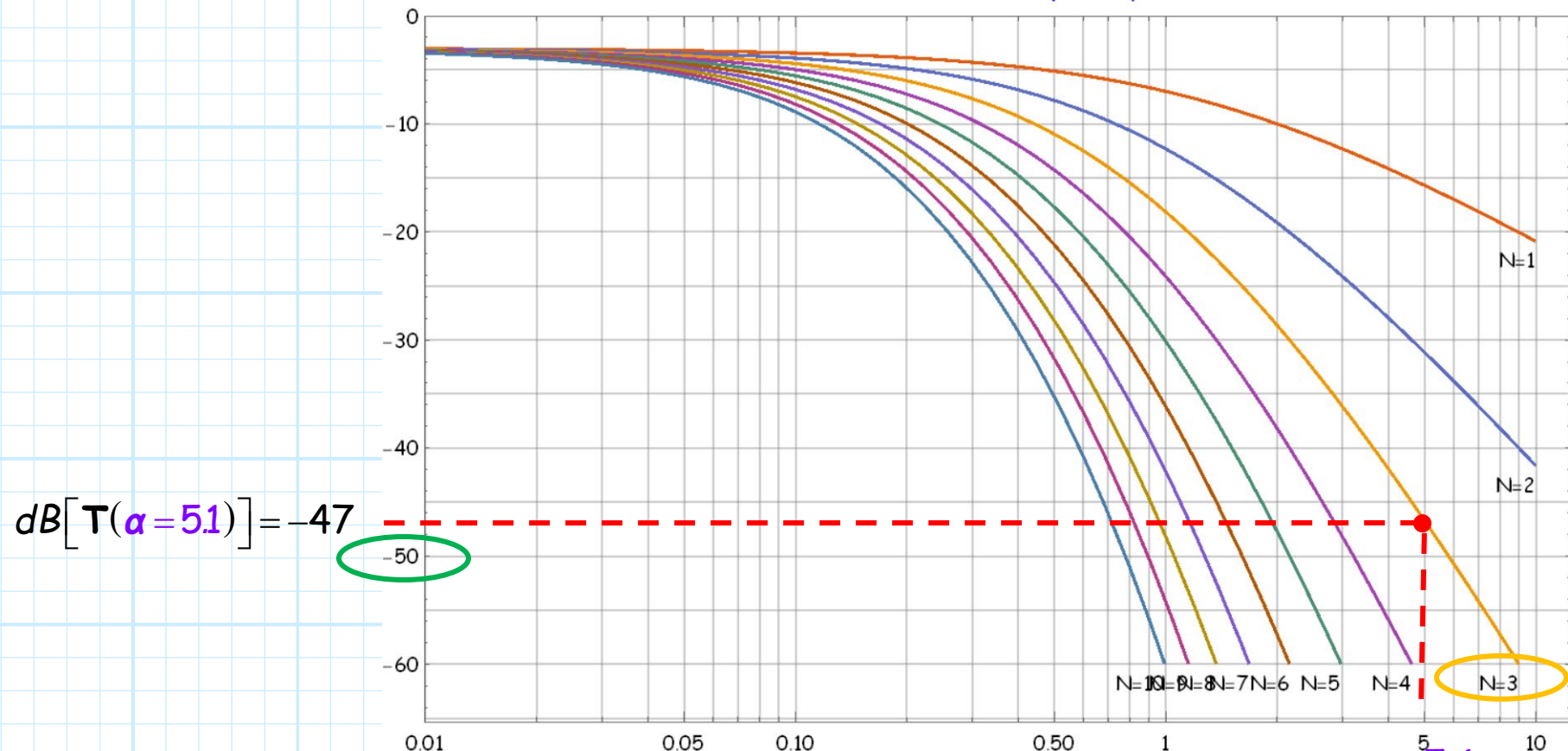
is the **normalized** frequency of the adjacent signal **one channel-spacing below** (this is **always** the case!).

Thus, the **lower** adjacent signal at 10.1 MHz (i.e., $\alpha = 5.18$) will be **attenuated less** than the other adjacent signal—we should use **this** value determine the proper filter order to achieve **50dB** of Selectivity!

Close, but not close enough

From the normalized **Butterworth** filter transmission chart, we see that a **3rd-order** filter will achieve about **47 dB** of attenuation at $\alpha = 5.18$.

Transmission versus normalized frequency for Butterworth filter



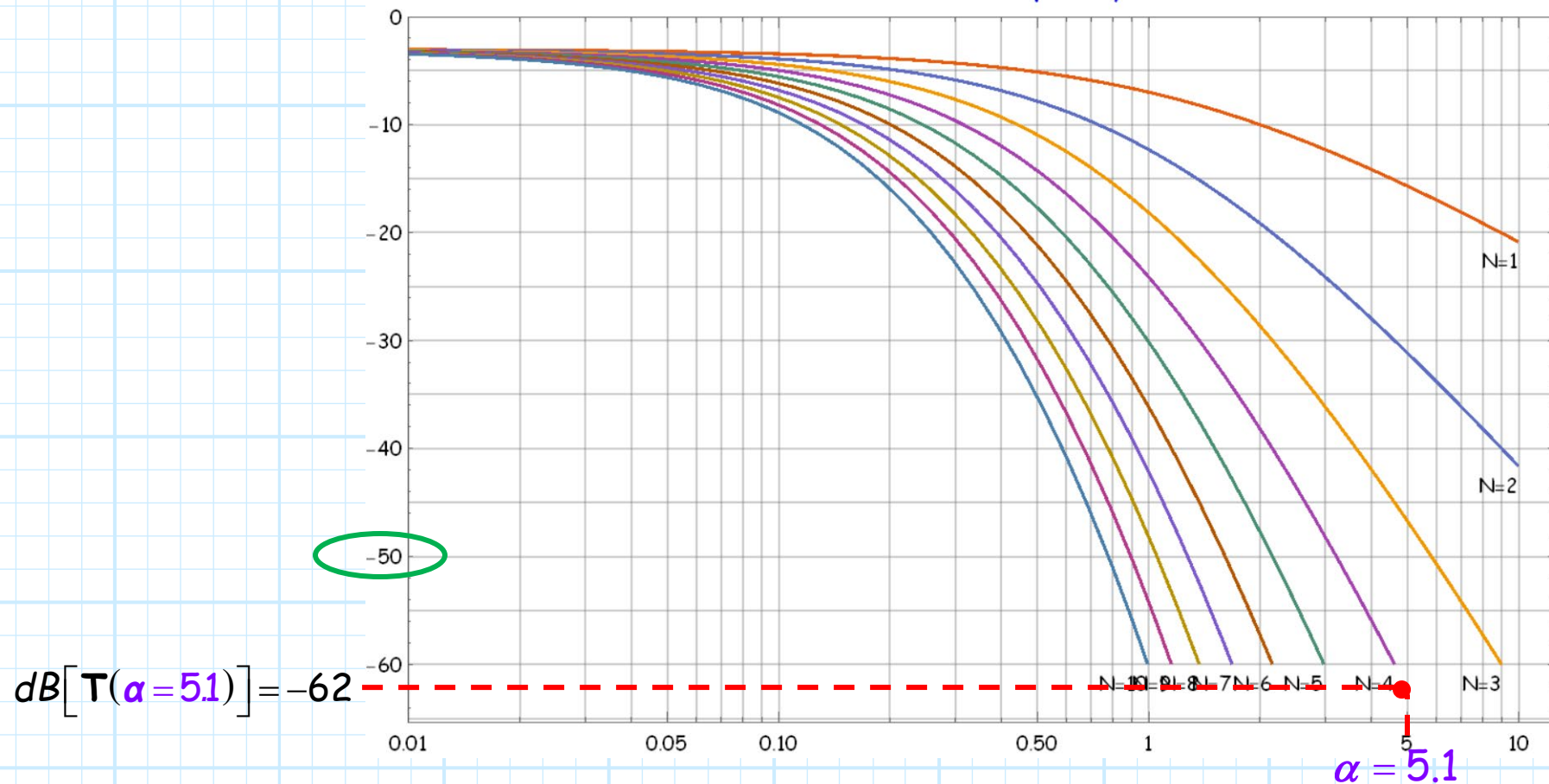
Although 47dB is **close** to 50dB, it is does **not** exceed 50dB!

Thus, a 3rd-order Butterworth filter is **entirely insufficient**.

The other hand holds 62 dB

On the **other** hand, a **4th-order** filter will provide about **62dB** of attenuation at $\alpha = 5.18$ —this **does** exceed 50dB (by a whopping 12dB).

Transmission versus normalized frequency for Butterworth filter



A **4th-order** Butterworth filter is necessary to achieve the required Selectivity for this FM radio design!

It is what it is!!!!



$> 50\text{dB!}$

Q: *So, do I proudly state that the Selectivity of this receiver design is **more than 50dB?***

A: NO!!!

Since your 4th-order IF filter attenuates the lower channel by 62 dB, we say that **the selectivity of the receiver is 62 dB!**

$= 62\text{dB!}$



I believe it has something to do with phase delay



Q: *Couldn't we further increase the receiver selectivity by using a **Chebyshev** filter (instead of a Butterworth)?*

A: Because of the **faster roll-off** of Chebyshev filter, this would **indeed** increase the receiver selectivity (or, we could get by with **just** a 3rd-order filter).

But remember, an **IF filter** is a **narrowband filter** (i.e., the **signal bandwidth** and the **filter bandwidth** are approximately **equal**).

Thus, using a Chebyshev **IF** filter is a **very bad** idea!

Do you remember why?