Pulsed Nuclear Magnetic Resonance

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1 Background

What we call “nuclear magnetic resonance” (NMR) was developed simultaneously but independently by Edward Purcell and Felix Bloch in 1946. The experimental method and theoretical interpretation they developed is now called “continuous-wave NMR” (CWNMR). A different experimental technique, called “pulsed NMR” (PNMR), was introduced in 1950 by Erwin Hahn. Pulsed NMR is used in magnetic resonance imaging (MRI). Purcell and Bloch won the Nobel Prize in Physics in 1952 for NMR; more recently NMR was the subject of Nobel Prizes in Chemistry in 1991 and 2002. We have both NMR setups in the advanced labs: one is a variation of the CWNMR method, and the other is a pulsed NMR system.

The physics underlying NMR is the same for both the continuous and pulsed methods, but the information obtained may be different. Certainly, the words used to describe what is being done in the experiments are different: in the continuous-wave case one tunes a radio-frequency oscillator to “beat” against the resonance of a nuclear magnetic moment in a magnetic field; in the pulsed case, one applies a sequence of RF pulses called π-pulses (180 degree pulses) or π/2-pulses (90 degree pulses) and looks for “free induction decay” and “spin echoes”.

Below, we give a very simplified introduction, based on classical ideas, to the physics of NMR. More thorough discussions, focusing on the CWNMR technique, may be found in the books by Preston and Dietz [1] and Melissinos [2] (see references). In particular, the chapter in Preston and Dietz gives a nice description of the connection between quantum-mechanical and semi-classical approaches to NMR physics. For a comprehensive treatment of NMR, see the book by Slichter [3].

1.1 Semiclassical ideas

To observe NMR, one needs nuclei with a non-zero angular momentum \( \vec{I} \) and magnetic moment \( \vec{\mu} \). The relationship between these two quantities is

\[
\vec{\mu} = \gamma \vec{I},
\]

where \( \gamma \) is the gyromagnetic ratio.

A simple classical calculation would give \( \gamma = q/2M \), where \( q \) is the charge of the nucleus and \( M \) is its mass. But quantum mechanics requires this dimensionally correct result to be modified. In practice, we specify \( \gamma \) in units of the nuclear magneton \( \mu_n \equiv e\hbar/2m_p \) times a dimensionless factor \( g \) called the “spectroscopic splitting factor” (often just “\( g \) factor”):

\[
\gamma = \frac{g \mu_n}{\hbar} = \frac{g e}{2m_p},
\]
where $e$ is the elementary charge and $m_p$ is the mass of one proton. The $g$ factor is on the order of unity. It is positive for some nuclei, and negative for others. For the proton, $g = +5.586$; a table of various $g$ factors may be found in Preston and Dietz.

When a nucleus of moment $\vec{\mu}$ is placed in a magnetic field $\vec{B}_0$ it will experience a torque causing a change in the angular momentum following Newton’s second law:

$$\vec{\tau} = \vec{\mu} \times \vec{B}_0 = \frac{d\vec{I}}{dt} = \frac{1}{\gamma} \frac{d\vec{\mu}}{dt}. \quad (3)$$

Figure 1 shows some of the vectors defined with $\vec{B}_0$ lying along the $z$ axis (which is horizontal in most lab setups, including ours). The rate of change of $\vec{\mu}$ is, by Eq. (3), both perpendicular to $\vec{B}_0$ and $\vec{\mu}$ itself; hence $\vec{\mu}$ precesses about the direction of $\vec{B}_0$. From Fig. 1 it is easy to derive from geometry and Eq. (3) that the magnitudes of the vectors $\vec{\mu} \times \vec{B}$ and $d\vec{I}/dt$ obey

$$\mu B_0 \sin \theta = \frac{\mu}{\gamma} \sin \theta \left( \frac{d\phi}{dt} \right); \quad (4)$$

thus

$$\omega_0 \equiv \frac{d\phi}{dt} = \gamma B_0, \quad (5)$$

which is the Larmor angular frequency.

When we deal with nuclei of spin $I = 1/2$ (e.g., protons), quantum mechanics tells us that in a magnetic field $\vec{B}_0$ the ground state splits into two sublevels, as shown in Fig. 2. The

$$\begin{align*}
\mathcal{B}_0 &= 0 \\
\mathcal{B}_0 &\neq 0
\end{align*}$$

Figure 2: Splitting of nuclear energy levels due to applied magnetic field.
energies $U$ of the two sublevels are given by

$$U = -\vec{\mu} \cdot \vec{B}_0 = -\mu_z B_0 = -\gamma h m_I B_0, \quad (6)$$

where $m_I$ is the angular momentum projection of $I$ along the $z$ direction defined by $\vec{B}_0$, $m_I = \pm 1/2$. The difference in energy between the two sublevels is thus

$$\Delta U = U_- - U_+ = \gamma h B_0, \quad (7)$$

where $U_\pm$ is the energy of the state corresponding to $m_I = \pm 1/2$. Transitions between the two energy sublevels may be induced by photons carrying one unit of angular momentum ($\hbar$) and energy $\hbar \omega_0$ such that

$$\hbar \omega_0 = \gamma h B_0, \quad (8)$$

or $\omega_0 = \gamma B_0$, which is the same as the Larmor frequency, Eq. (5).

For free protons, the gyromagnetic ratio is

$$\gamma = 2.675 \times 10^8 \text{ radians/(seconds-tesla)},$$

so for fields in the 0.1–1 tesla range the frequencies will be in the megahertz range. These are radio frequencies (RF), and both of our setups—pulsed and continuous—work in this MHz range. (Historical note: There was an advantage to having NMR work in the RF range, as existing radio equipment and techniques were used for detection and modulation of NMR signals.) Given that magnetic fields in these experiments are frequently measured in kilogauss, a useful relationship to remember for protons is,

$$f_0 (\text{MHz}) = 4.2577 \times B_0 (\text{kilogauss}), \quad (9)$$

where $f_0 = \omega_0 / 2\pi$.

In order to excite the magnetic resonance, the applied RF must be circularly polarized, optimally with the rotation axis along the $z$ axis. This is shown in Fig. 3, where an applied field $\vec{B}_1(t)$ lies in the $x$-$y$ plane and rotates about the $z$ axis at the Larmor frequency and in the same direction as the Larmor precession. Now imagine observing the field from the reference frame that rotates along with the Larmor precession; the field $\vec{B}_1$ would be seen as a steady field relative to the moment $\vec{\mu}$, and also perpendicular to $\vec{B}_0$. The effect of $\vec{B}_1(t)$ would be to exert a torque in that rotating frame. However, if the RF were applied at a frequency different than the Larmor frequency, the torque on $\vec{\mu}$ in the rotating frame would

![Figure 3: The RF $\vec{B}_1$ field rotates with the precessing $\vec{\mu}$ in the semiclassical picture.](image-url)
vary in direction and strength over time, so the net work done by the field would be less, tending to zero as the RF angular frequency moves away from $\omega_0$. A little mathematics shows that this situation follows the standard picture of resonance phenomena—a resonance which involves the interaction between a nucleus with a magnetic field, hence the name nuclear magnetic resonance.

In the quantum-mechanical picture for a single nucleus we can say that the probability of a “spin-flip” between the $m_i = +1/2$ and $m_i = -1/2$ state is maximized when the RF photon has energy equal to $\hbar \omega_0$. One may well wonder whether a complete quantum mechanical approach would give different answers than the classical picture we have presented. It is somewhat surprising to learn that Bloch showed that the classical equations governing NMR can be derived from quantum mechanics. See the text by Slichter for a readable and thorough presentation of this result [3].

1.2 Net magnetization and relaxation

Clearly, any physical sample will consist of many atoms and the NMR signal measured by an experiment will be due to the combined effect of the magnetic field on them all. As an example, consider an assembly of $N$ protons, say, in water, glycerin, mineral oil, or animal tissue. The net magnetization $\vec{M}$ of the sample is the vector sum of all of the individual moments $\vec{\mu_i}$. For the example of spin-$\frac{1}{2}$ protons, the net magnetization along the $z$-axis, $M_z$ would be given by

$$M_z = \sum_i \gamma \hbar m_{i} = \frac{1}{2} \gamma \hbar (N_+ - N_-) ,$$

(10)

where $m_{i}$ denotes the state of the $i$th proton, and $N_+$, $N_-$ are the numbers of protons in the $+\frac{1}{2}$ and $-\frac{1}{2}$ state, respectively. When an RF field $\vec{B}_1(t)$ is applied, some protons have their spin flip to align against $\vec{B}_0$; these absorb energy from the RF field. Other protons experience a spin-flip to align with $\vec{B}_0$; these give energy back to the RF field. Thus, if the populations of the two sublevels were equal, $M_z = 0$ and no net energy would be absorbed from the RF field.

At room temperature with no RF field but with the steady $\vec{B}_0$ field there will be a small difference between the number of protons with spins aligned in the direction of $\vec{B}_0$ and the number aligned opposite to it; the magnetic sublevel with $m_i = +\frac{1}{2}$ will have a slightly larger population $N_+$ than the population $N_-$ in sublevel $m_i = -\frac{1}{2}$. The ratio of the populations is proportional to the ratio of the associated Boltzmann factors, according to the principles of statistical mechanics:

$$\frac{N_+}{N_-} = \frac{\exp(-U_+/kT)}{\exp(-U_-/kT)} = \frac{\exp(\frac{1}{2} \gamma \hbar B_0/kT)}{\exp(-\frac{1}{2} \gamma \hbar B_0/kT)} = e^{\gamma \hbar B_0/kT} .$$

(11)

If one applies the RF $\vec{B}_1(t)$ to this equilibrium population, energy goes into the system, since there is an imbalance in the number of up ($+\frac{1}{2}$) spins versus down ($-\frac{1}{2}$) spins. In a fairly short time, however, this imbalance will vanish, because the continuous transfer of energy involving each moment will cause the populations $N_+$ and $N_-$ to become equal. At this point, the net energy absorbed by the system drops to zero, $M_z = 0$ and the sample is said to be saturated.

There is still a tendency for the system to recover its equilibrium configuration, even while it is subject to the oscillating field. The collection of protons could continue to absorb energy
from an oscillating magnetic field $\vec{B}_1(t)$ if the equilibrium populations would be restored quickly following a previous absorption event. The decay of $M_z$ to its equilibrium (nonzero) value is due to energy exchanges between the proton magnetic moments and their local environments.

Let us consider the magnetization a bit more generally. The net magnetization $\vec{M}$ is a vector, and thus can be broken into components. The component along the direction of the static field $M_z$ is called the longitudinal magnetization, and the components along two orthogonal directions perpendicular to $\vec{B}_0$ are called the transverse magnetizations $M_x$ and $M_y$. Clearly, each of these components is the sum of the individual component moments $\mu_z$, $\mu_x$ and $\mu_y$, respectively, e.g., $M_x = \sum_i \mu_{xi}$. Now look again at Fig. 3 which depicts a single moment precessing about $\vec{B}_0$. It should be obvious that for a collection of magnetic moments at a particular time, each precessing independently about $\vec{B}_0$, one could have a nonzero $M_z$ but a (possibly) zero $M_x$ or $M_y$.

The equilibrium nonzero $M_z$ exists because the moments aligned parallel to $\vec{B}_0$ are at a lower energy than those aligned antiparallel. In order to return to equilibrium following absorption from the $\vec{B}_1(t)$ field, the moments would need to give energy to their surroundings, conventionally called the “lattice” (even if we are dealing with liquid or gaseous materials). The relaxation of $M_z$ to its equilibrium value is called the “longitudinal” or “spin-lattice” relaxation, and is associated with a characteristic time called $T_1$. In liquids, $T_1$ is typically very short but in solids $T_1$ may be quite long. For example, $T_1$ is a few milliseconds in water but thousands of seconds in ice. This is one reason why we use liquid samples in the continuous NMR setup, as the short $T_1$ allows for a continuous input of energy from the oscillator coil.

A nonzero $M_x$ or $M_y$ is harder to obtain because there is no reason for individual moments to prefer a particular transverse direction. At best $M_x$ and $M_y$ can oscillate between positive and negative values at the Larmor frequency, if one can first contrive to force the moments to have a net projection onto, say, the $x$-axis (and indeed one can—we’ll see how). However, the resulting oscillating $M_x$ decays because variations in the local magnetic field cause different moments to precess at different rates. This is called “spin-spin dephasing”, and the characteristic time for this process is called $T_2$, the “spin-spin” or “transverse” relaxation time.

With these two relaxation processes in mind, consider how one would detect the signal of these precessing and relaxing moments. Since they are magnetic, an obvious choice would be to use a pickup coil. The coil also needs to be oriented transverse to the static field in order to sense the precession, since in an orientation parallel to $\vec{B}_0$, the coil would only be sensitive to a slight change in flux from a non-oscillatory $M_z$; whereas the transverse component $M_x$ or $M_y$ will oscillate between positive and negative values at the Larmor frequency $\omega_0$.

This arrangement is depicted schematically in Fig. 4, which is taken from the article by Hahn [4]. The top of the figure shows that at $t = 0$, all of the moments are aligned along the $+x$ direction and the pickup coil has its axis along the $y$-axis. The static field $\vec{B}_0$ points out of the page toward the viewer, thus the moments will precess in the clockwise direction (following the sense of $\vec{M} \times \vec{B}_0$). The right side of the figure depicts the voltage measured across the pickup coil. Faraday’s law requires that this voltage be proportional to the rate of change of the flux through the coil, and it is a simple matter to prove that the maximum signal from the coil will occur at $t = 0$ in this setup. (Try it: let the field in the coil be $\mu_0 \vec{M}$ which is rotating about the $z$-axis at $\omega_0$, and calculate the rate of change of the flux, $d\Phi/dt$ from this.)
Figure 4: Induction signal measured by a pickup coil in the transverse plane following orientation of moments along the $x$-axis. Taken from the article by Hahn [4].

As the moments precess, the induced voltage oscillates at the Larmor frequency ($\nu_{\text{Larmor}}$ in the figure), and also decays. The decay is due to both the spin-spin dephasing, shown as a spreading of the moments away from the principal rotating $x$-axis—the $T_2$ process—and to the relaxation of the moments towards the $z$-axis—the $T_1$ process.

While the overall decay of the signal from the pickup coil comes from a combination of spin-spin and spin-lattice relaxation, experimentally, a third effect often predominates: spin-spin dephasing due to inhomogeneities in the external field $\vec{B}_0$ over the volume of the sample. Taken together, the three effects lead to a combined relaxation time $T_2^*$ defined as

$$\frac{1}{T_2^*} = \frac{1}{T_1} + \frac{1}{T_2} + \gamma \Delta B_0,$$

(12)

where $\Delta B_0$ is the variation in $B_0$. In effect, $\gamma \Delta B_0$ is equal to a spread in Larmor frequencies $\Delta \omega_0$ which, all else being equal, would cause spin-spin dephasing in a characteristic time of $1/\Delta \omega_0$. Similarly, one can assign an effective resonance width $\Delta \omega_{\text{eff}} \sim 1/T_2^*$.

In summary, two processes are involved in the relaxation of the magnetization to the equilibrium value: 1) spin-lattice relaxation, which involves energy exchange with the environment, and is characterized by the time constant $T_1$; 2) spin-spin relaxation, characterized by a time constant $T_2$, which is caused by loss of the phase relationships among the various moments and is due to the fact that each spin experiences a slightly different local magnetic field. In a practical setup, the measured signal comes from the oscillating transverse magnetization, characterized by a time constant $T_2^*$ that is related to $T_1$ and $T_2$, but is typically dominated by inhomogeneities in the static field $\vec{B}_0$. It is important to note that the recovery to equilibrium following saturation follows $T_1$, but the decay of the detected signal follows $T_2^*$. As required by Eq. (12), $T_2^* < T_1$, and in practice $T_2^* \ll T_1$. 
1.3 The two experimental methods

In the continuous NMR method, the RF excitation producing the rotating $\mathbf{B}_1$ field is applied all of the time. The resonance is created by having the $\mathbf{B}_0$ field swept slowly through the value which satisfies Eq. (5). A pickup coil surrounding the sample detects the resonance, and this signal is mixed with the fixed RF signal to create “beats” (the term for the modulation of two signals which have nearly the same frequency). The effect of the beats can be seen on an oscilloscope. Once the resonant signal is found, the relationship between the magnetic field and the resonant frequency is fixed, and depending on the givens, the information can be used to extract $B_0$ itself, $\gamma$, or related quantities. One can also crudely measure $T_2^*$ by looking at the decay of the beat signal.

In the pulsed NMR method, the RF excitation is applied to the sample in a series of short bursts, or pulses. The application of the RF field for a short time (the “pulse width”) allows the applied torque to rotate the net magnetization $\mathbf{M}$ by a specific amount. For example, one can apply a pulse of RF field to rotate all moments by $90^\circ$. If this pulse is applied to a sample initially at equilibrium with a net magnetization $\mathbf{M} = M_z \hat{k}$, then $\mathbf{M}$ will become $M_z \hat{i}$ (i.e., the same magnitude of magnetization now pointing in an orthogonal direction), which will then precess about the $z$-axis. The pickup coil will see a signal like that shown in Fig. 4 as the magnetization decays back to its equilibrium state. A pulse which accomplishes this trick is called a “$\pi/2$ pulse”, and the signal seen as a consequence is called the “free induction decay”.

A pulse of a longer duration can flip the net magnetization completely: $M_z \hat{k} \rightarrow -M_z \hat{k}$; this type of pulse is called a “$\pi$ pulse”. Interestingly, the free induction decay signal immediately following a $\pi$ pulse is zero since there is no net transverse component of the magnetization available to induce such a signal.

The real utility of the pulse method comes from using a sequence of pulses. By such sequences, one can measure accurately and independently $T_1$ and $T_2$, and also compensate for the effects of field inhomogeneity. The discussion of pulse sequences and their effects are taken up in more detail in the next sections.
The pulsed NMR experiment

This experiment is performed with the TeachSpin PS1-A pulsed nuclear magnetic resonance (PNMR) spectrometer. This instrument was designed for use in a teaching laboratory. Its modular design separates the basic functions (oscillator and RF amplifier, RF receiver/detector and pulse programmer) of a PNMR system, allowing the experimenter to investigate and understand the function of each module, and how and why the interconnections between the modules are made.

The pulsed NMR technique involves a number of subtle and somewhat complex ideas and methods. One way to learn it is to read through the theory, as laid out here or in the PS1-A manual, and then use the instrument to quickly make the measurements. Another way, perhaps more instructive, is to learn the theory and technique at the same time. In what follows, we will assume that you will proceed along the second way. If, however, you have already read the theory, and would prefer to move directly to the measurement tasks, you can turn to Section 3 (p. 25) for a summary of those.

The instruction manual for the PS1-A gives complete details on the instrument and its operation. Our instructions will focus on the particular operations needed to make the measurements. We will refer to relevant passages in the manual by means of a marginal note. Like this.

Copies of the manual are available in the lab and online at the course website.

2.1 Pulses and Free Induction Decay

We begin by stating a curious result. Recall that the magnetic moments in our system precess under the influence of the torque from a magnetic field, Eq. 3. This precession occurs with a precession angular frequency \( \omega_0 = \gamma B_0 \). If we were to transform our coordinate system to one which rotated along with this precession, i.e., we translate coordinates \( x, y \) and \( z \), as measured in the fixed lab reference frame to \( x', y' \) and \( z' \) in the rotating frame, where

\[
\begin{align*}
x' &= x \cos \omega_0 t - y \sin \omega_0 t, \\
y' &= x \sin \omega_0 t + y \cos \omega_0 t, \\
z' &= z,
\end{align*}
\]

we would find that the magnetic field \( B_0 \) vanishes! This must be true, since in this rotating frame (at exactly \( \omega_0 \)), there is no precession.\[ PS1-A \]

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Now consider what would happen if we apply a time-dependent magnetic field \( \vec{B}_1(t) \) which rotates at exactly the frequency \( \omega_0 \) and lies in the \( x-y \) plane, e.g.,

\[ \vec{B}_1(t) = (B_1 \cos \omega_0 t) \hat{i} - (B_1 \sin \omega_0 t) \hat{j}. \]

In the rotating frame, \( \vec{B}_1(t) \) would appear to be a constant. The net effect, as seen from the rotating frame, would be a precession of the moment about an axis lying in the \( x'-y' \) plane at a frequency \( \omega_1 \) equal to \( \gamma B_1 \).

Assume, for an illustration, that we have a moment \( \vec{\mu} \) initially pointing along the +z' (also the +z) axis, collinear with \( \vec{B}_0 \), and that we turn on \( \vec{B}_1(t) \) at exactly \( t = 0 \). In the rotating frame, \( \vec{B}_1(t) = B_1 \hat{i}' \), where \( \hat{i}' \) is the unit vector along the rotating +x' axis. Thus, we would see \( \vec{\mu} \) rotate about the +x' axis, initially tilting toward the +y' axis (according to \( \vec{\mu} \hat{k}' \times B_1 \hat{i}' \)).
Now imagine that we turn off $\vec{B}_1(t)$ right at the point where $\vec{\mu}$ lies along the $+y'$ axis. In the lab frame, the motion of the tip of the vector corresponding to $\vec{\mu}$ would look like a spiral ending with the moment lying in the $x$-$y$ plane precessing about the $z$ axis. This motion is shown in Fig. 5 which is taken from the article by Hahn [4].

![Diagram](image)

Figure 5: The classical motion of a moment $\vec{\mu}$ under the influence of a static $\vec{B}_0$ and rotating $\vec{B}_1(t)$ field. From Hahn [4].

Now assume that the same time-dependent field (static $\vec{B}_0 +$ rotating $\vec{B}_1(t)$ for a short time) is applied to a whole collection of moments initially at equilibrium in the static $\vec{B}_0$ field. The same motion followed by our single moment $\vec{\mu}$ will now be executed by the net magnetization $\vec{M}$, because at equilibrium, $\vec{M} = M\hat{k}$. The result? $\vec{M}$ is tilted into the $x$-$y$ plane and precesses about the $z$-axis; a pickup coil lying in this plane would detect the induced voltage, following the discussion in Section 1.2. This is the “free induction decay” (FID) signal.

The rotation of $\vec{M}$ is accomplished by turning the rotating field on and off: a pulse. Since $\vec{M}$ is rotated by 90°, we call such a pulse a “$\pi/2$ pulse”. The duration of a $\pi/2$ pulse is equal to the period of precession $t_1$ due to $\vec{B}_1$ divided by 4. From the relation $\omega_1 = \gamma B_1$, and because $\omega_1 = 2\pi f_1 = 2\pi/t_1$, we obtain the time for the $\pi/2$ pulse:

$$t_{\pi/2} = \frac{1}{4} t_1 = \frac{\pi}{2\gamma B_1}$$

(13)

To create a $\pi/2$ pulse and to see the resulting signal from a pickup coil, we need the following apparatus:

- A steady magnetic field $\vec{B}_0$, here supplied by a permanent magnet.
- A pickup coil, with its axis perpendicular to the direction of $\vec{B}_0$, along with electronics to sense the RF signal created by the coil: a receiver and detector circuit.
- An excitation coil to create the rotating $\vec{B}_1(t)$ field, with its axis also perpendicular to $\vec{B}_0$, along with electronics to create an alternating RF current of sufficient strength: an RF oscillator and amplifier. (Note: the excitation coil in our (and most) apparatus actually produces a linear oscillating magnetic field. Such a field is equivalent to two counter-rotating fields. The field that rotates in the “wrong” direction only causes a
negligibly small perturbation to the measured angular rotation frequency—the “Bloch-Siegert shift”.

- Electronics which can allow one to tune the oscillator to the resonant frequency \( f_0 \): a “mixer” circuit which produces a signal showing the difference between the oscillator and receiver signals.

- Electronics which can act as a switch to turn the RF current on and off in a carefully controlled manner: a “pulse programmer”.

- And, finally, a sample containing moments: the protons (hydrogen nuclei) in water and hydrocarbons.

This apparatus is supplied by the PS1-A system, whose block diagram is shown in Fig. 6.

2.1.1 Operation of the pulse programmer

We will start with learning to operate the pulse programmer module of the apparatus. The pulse programmer creates logic-level pulses (0-4V) which are used by the other electronics to control the application of the RF pulse and detection of the NMR signal.

The front panel of the pulse programmer is shown in Fig. 7. The programmer creates two types of pulses, called \( A \) and \( B \). To observe free induction decay, you need only use \( A \) pulses, but we will use the \( B \) pulses later on, so this exercise will look at both types.

Details of the controls are given in the PS1-A manual, here we give a brief overview of the pulse programmer function.

- Type \( A \) pulses are controlled by the A-WIDTH knob—to set the pulse duration—and the REPETITION TIME controls—to set how often the pulses occur. The REPETITION TIME controls are operational only when the MODE switch is set to INT (internal). The other settings of the MODE switch allow a pulse to be initiated by an external signal (into EXT START) or by the press of a button (MAN START).
Figure 7: Diagram of connections needed to examine output of the pulse programmer.

- Type B pulses are coupled to A pulses. They are controlled by the B-WIDTH knob—to set the pulse duration, the NUMBER OF B PULSES thumbwheel switches—to set how many B pulses occur, and the DELAY TIME thumbwheel switches—to set how long after the A pulses B pulses occur and the subsequent time between additional B pulses.

- The SYNC OUT connector and SYNC switch allows a very short (≈ 0.2 μs) pulse to be used to trigger an oscilloscope at the start of a pulse or pulse sequence. Typically, one connects SYNC OUT to the external trigger input on the scope.

- The BLANKING OUT connector is used to disable the RF detector when the A and B pulses occur in order to prevent overloading the detector amplifier. The M-G OUT connector sends a signal to the RF oscillator to enable the “Meiboom-Gill” type of multipulse sequence (see Section 2.2.2).

First, we will look at type A pulses only. Disconnect all cables from the front panel of the PS1-A except the two cables connected to the magnet and sample holder assembly. Turn on the power to the PS1-A (the switch is on the back of the case), and turn on the oscilloscope. Connect SYNC OUT to the scope’s EXT TRIG input, and connect A+B OUT to the scope’s CH 1 input. Set the oscilloscope to trigger on “EXT”; set the CH 1 sensitivity to 2V/Div; set the timebase to 20 μs/Div. First, look at A pulses only. Set the pulse programmer controls as follows:

- A-WIDTH: half-way (12 o’clock)
- MODE: INT
- REPETITION TIME: 10 ms, VARIABLE at 10% (≈ 1 ms)
- SYNC: A
- A: on
- B: off

If all is working correctly, you should see a trace on the oscilloscope similar to that in the lower trace of Fig. 8. Vary the A-WIDTH knob and note what happens. Then turn the A-WIDTH all the way clockwise, and increase the timebase on the scope to 1 ms/Div. You should see multiple pulses, spaced a couple of milliseconds apart. Note the effects when you change the REPETITION TIME controls.
Now, we will look at a pulse sequence. Reset the scope timebase to 20 μs/Div, and then make the following changes to the pulse programmer controls:

- A-WIDTH: one-quarter-way (9 o’clock)
- B-WIDTH: half-way (12 o’clock)
- REPETITION TIME: 10 ms, VARIABLE at 10%
- NUMBER OF B PULSES: 01
- A: on
- B: on
- DELAY TIME: 50 μs (0.05 × 10^0)

You should see two pulses on the scope, with the second pulse about twice as wide as the first pulse. Now, play around: vary both the A-WIDTH and B-WIDTH knobs and the DELAY TIME settings and note the results.

Next, look at the effect of setting the NUMBER OF B PULSES to a number greater than 1. In particular, pay attention to the effect of the DELAY TIME settings on multiple B pulses. It is important to see that the time between the A pulse and the first B pulse is one-half of the time between subsequent B pulses, and that the DELAY TIME setting is equal to the time between the leading edges of the A and first B pulse. See if you can reproduce the waveform shown in the top trace of Fig. 8. Also explore the REPETITION TIME settings, and expand the timebase to see multiple pictures of multiple pulse sequences.

### 2.1.2 Observation of free induction decay

Once you have thoroughly understood the operation of the pulse programmer, you are ready to observe an actual NMR signal. We will start with the simplest arrangement, a single A pulse, with the apparatus tuned to produce a π/2 pulse in order to observe the free induction decay (FID) signal.
The modules on the spectrometer are described here. Set the controls as indicated.

**15 MHz Receiver** This module senses the RF signal produced by the sample-probe coil, PS1-A, amplifies it and rectifies it. The rectified signal is sent to the oscilloscope. The unrectified RF signal is sent to the mixer in order to optimize the oscillator frequency.

- **GAIN:** about 30%
- **BLANKING:** on
- **TIME CONST:** .01 ms
- **TUNING:** 12 o’clock

**Pulse Programmer** You want to set the controls to make only A type pulses. First, calculate the time needed for a $\pi/2$ pulse from Eq. (13), given $\gamma = 2.765 \times 10^4$ rad/s-gauss and the magnetic field $B_1 \approx 12$ gauss, according to the manufacturer. Then set the programmer controls:

- **A-WIDTH:** Set to calculated $t_{\pi/2}$ (use scope)
- **B-WIDTH:** Fully CCW
- **NUMBER OF B PULSES:** 00
- **REPETITION TIME:** 100 ms, VARIABLE at 100% ($\approx 100$ ms)
- **SYNC:** A
- **A:** on
- **B:** off

**15 MHz OSC/AMP/MIXER** This module has three separate circuits in one box. The 15 MHz oscillator is controlled by the FREQUENCY ADJUST knob and the COARSE/FINE switch which changes the significant digit incremented by the knob. The oscillator signal is fed internally to the input of the power amp which is turned on (or “gated”) by the signal going into A+B IN. The output of the power amp, RF OUT is connected to the sample probe excitation coil and delivers an RF power of 150 watts peak power. The M-G IN and M-G switch are used to synchronize the phase of the RF signal with the gating in order to implement the “Meiboom-Gill” type of multipulse sequence, as described in Section 2.2.2. Set the controls as follows:

- **FREQUENCY:** 15.000000
- **CW-RF:** on
- **M-G:** on

Now connect the modules together and to the oscilloscope following the diagram shown in Fig. 9. Use the short cables to connect one module to another, and use the long cables to make connections to the oscilloscope. Connect the DETECTOR OUT to CH 1 of the digital oscilloscope, and the MIXER OUT to CH 2. Set the oscilloscope to trigger on “EXT”; set the CH 1 sensitivity to 2V/Div; set the CH 2 sensitivity to 5V/Div; set the timebase to 40 $\mu$s/Div.

Make sure that the O-ring on the sample vial is 1.5 inches from the bottom and carefully lower the mineral oil sample into the sample holder inside the magnet assembly.

After some adjustment of the oscilloscope, you should observe traces that bear some resemblance to those in Fig. 10. If you don’t get anything like Fig. 10 (or anything at all), check the connections on the PS1-A and the switch and knob settings. The CH 1 signal (lower trace) is the free induction decay (FID). This signal is half of the envelope of the RF oscillations generated by the precession of the net magnetization in the x-y plane. It is the
Figure 9: Diagram of connections on the PS1-A spectrometer to make pulsed NMR measurements.

Figure 10: Signals of free induction decay. Upper trace: mixer output showing beats indicating the frequency difference between the resonance frequency and the oscillator frequency. Lower trace: detector output showing the FID signal, which is the positive amplitude envelope of the RF signal obtained from the pick-up coil.
envelope (which represents the magnitude of the net magnetization) and not the individual oscillations that is of interest here. (A rough picture of the RF oscillations can be seen in Fig. 9.2 in the PS1-A manual.) The signal on CH 2 (upper trace) is the mixer output. The oscillations on this signal, also known as “beats”, are at the difference frequency between the detected signal (due to the precession of the net magnetization in the \(x-y\) plane) and the frequency of the RF oscillator.

Adjust the frequency on the oscillator to reduce the frequency of the beats on the mixer output to a minimum. Figure 11 shows a collection of mixer output signals for different oscillator frequencies. The goal of your adjustment is to make the number of oscillations, or zero-crossings, go to zero; the initial bump in the signal is not important. When the beats are at a minimum, you can adjust the TUNING knob to maximize the amplitude of the FID signal. Finally, pull the sample out of the holder and slide the O-ring down a few millimeters. Then slowly lower it back into the sample holder, pressing down so as to scoot the O-ring along the sample vial, while watching the FID signal on the scope. The signal will be strongest when the sample material is centered in the magnetic field. You should try to adjust the position of the O-ring to maximize the FID signal.

Figure 11: Signals from the mixer output for different oscillator frequencies near resonance. Traces from the bottom, moving up: 9 kHz low (R1), 2 kHz low (R2), on resonance (2), 2 kHz high (R3).

Now tune the \(A\) pulse width to maximize the FID signal in the following manner: turn the A-WIDTH knob completely counterclockwise (the FID signal will diminish), and then slowly turn it clockwise until the FID signal is largest. Measure the \(A\) pulse width (you will need to move some connections around) and compare it to the width you calculated earlier.

The pulse width you have just measured is the pulse width needed for a \(\pi/2\) rotation of the net magnetization \(\vec{M}\). The decay of the FID signal depends on \(T_2\), as noted in the introduction, but is also dependent on the time constant of the detector. You may wish to see what different settings of the TIME CONST switch do to the FID signal. You may also notice that the FID signal does not decay exactly following an exponential curve; this is because the
signals from the different precession frequencies mix together, and the resulting signal is more complex than a simple dying-off of the transverse magnetization. As you will see, $T^*_2$ is much shorter than either $T_2$ or $T_1$, indicating that magnetic field inhomogeneity is fairly significant in this apparatus.

Now explore what happens if you increase the A pulse width beyond $t_{\pi/2}$. You should first see the FID signal decrease, nearly to zero, and then increase again. Measure the pulse widths for the pulse that gives a minimum FID signal and the pulse that gives the following maximum. How are these pulses related to $t_{\pi/2}$? Discuss the relationship between these pulses, the rotation of $\vec{M}$, and the resulting FID signal.

Technical note: If the field of the permanent magnet changes in time (due to changes in the temperature of the magnet) the resonant frequency will change, and beats will reappear on the mixer output signal. If this happens during the course of your measurements, the oscillator frequency should be re-adjusted to minimize the beats.

2.2 Pulse sequences and measurements of $T_1$ and $T_2$

Now we will see the real power of the pulsed NMR method in its ability to measure the longitudinal (spin-lattice) relaxation time constant $T_1$ and the transverse (spin-spin) relaxation time $T_2$ even when the static magnetic field $\vec{B}_0$ is not very uniform.

2.2.1 Measuring $T_1$: 2 pulse sequence

The amplitude of the FID signal induced by a $\pi/2$ pulse is proportional to the initial longitudinal magnetization $M_z$. Immediately following a $\pi/2$ pulse, $M_z \approx 0$, which you could confirm by applying another $\pi/2$ pulse at that time and noting that the FID signal was zero. (Indeed, you will have done this already if you think of what a pulse of $2 \times t_{\pi/2}$ does!) It takes a while for $M_z$ to recover its equilibrium value, and if you hit the sample with another $\pi/2$ pulse before equilibrium has been reestablished, then this second FID signal will be weaker than the first one.

The longitudinal magnetization relaxes according to an exponential function:

$$M_z(t) = M_0 - (M_0 - M_i) e^{-t/T_1}, \quad (14)$$

where $M_0$ is the equilibrium value of $M_z$ and $M_i$ is equal to $M_z$ at $t = 0$. One way of estimating $T_1$ suggested by Eq. (14) is to decrease the time between successive $\pi/2$ pulses until you see the FID signal drop by about 1/3 (i.e., $1/e$) of its “equilibrium” value. You can see this by letting $M_i = 0$ (what you get after a $\pi/2$ pulse) in Eq. (14) and calculating $M_z(t = T_1)$. Try this: reduce the REPETITION TIME and see how the FID signal behaves. Can you make an estimate of $T_1$ by this method? (You may find that you cannot make the FID signal reduce by 1/3 if $T_1$ is too short. Don’t sweat it; this method is merely qualitative.)

A more elegant measurement of $T_1$ may be accomplished by a two pulse sequence. If one applies a longer pulse—a $\pi$ pulse—the net magnetization $\vec{M}$ can be inverted: $M_0 \rightarrow -M_0$. The decay of $M_z$ can be tracked by applying subsequent $\pi/2$ pulses at varying intervals after the $\pi$ pulse, since the FID signal is proportional to $M_z$ at any time. Of particular interest is the point at which $M_z$ momentarily vanishes, or “crosses zero” in its decay from a net negative value to the equilibrium positive value.
This pulse sequence is illustrated in Fig. 12, which is taken from reference [5]. In the figure, we see that a $\pi$ pulse initially inverts the magnetization, and that a $\pi/2$ pulse interrogates the magnetization at a time interval “TI” later.

Since the zero crossing point is easily found, we will use this method to measure $T_1$ for our samples. The spectrometer should already be configured to observe FID pulses (see Section 2.1.2 for settings and Fig. 9 for connections). The strategy is to set up the $A$ pulse so that it rotates the net magnetization by $\pi$, and then a single $B$ pulse so that it rotates the net magnetization by $\pi/2$. (As an option you can measure the $\pi/2$ and $\pi$ pulse widths to check the ratio.)

Now turn off the $A$ switch, turn on the $B$ switch and set the NUMBER OF B PULSES to 01.

Switch SYNC to $B$. Adjust the $B$ pulse width, starting from fully counterclockwise, to maximize the FID signal (mark on knob should be just below 9 o’clock position). The $B$ pulse is now rotating the net magnetization by $\pi/2$. If this $B$ pulse comes close in time (relative to $T_1$) after the $\pi$ pulse ($A$), the net magnetization will be rotated by $3\pi/2$ or $270^\circ$. Set the DELAY TIME to $30 \mu$s ($0.03 \times 10^9$ ms), turn on both $A$ and $B$ switches, set SYNC to $A$, and confirm that you get a healthy FID signal.
Now set the \textbf{DELAY TIME} between the \textit{A} and \textit{B} pulses to several milliseconds. There should now be two bumps on the scope trace, the first one being a weak transient from the $\pi$ pulse, the second one being the stronger FID signal. As the delay time is increased (millisecond intervals are handy) the amplitude of the FID signal should decrease until at some delay time, it goes away (almost) completely, as shown in Fig. 13. At this point in time, there is no net magnetization, hence no FID signal; in the language of Eq. (10), $N_+ = N_-$ for a brief moment. You will notice that if you keep increasing the delay time, the FID signal reappears, indicating that $M_z$ has gone from net negative value to a net positive value, as indicated in Fig. 12.

![Oscilloscope traces showing zero-crossing method of determining $T_1$.](image)

Figure 13: Oscilloscope traces showing zero-crossing method of determining $T_1$. The $\pi$ pulse occurs at the trigger point (T), and the FID signal from the $\pi/2$ pulse first diminishes as the delay time increases (traces R1 and R2), nearly vanishing at the delay time in the third trace from the top (R3), and then recovers at a later delay time (R4). Sample: H$_2$O$\cdot$CuSO$_4$; $T_1 \approx 2.6$ ms.

The delay time for $M_z$ to decay to 0, may be used to calculate $T_1$. Since $M_z = 0$ is halfway between $M_z = -M_0$ and $M_z = +M_0$, the zero-crossing time interval is equal to the “half-life” of the exponential decay. You can use this fact to calculate $T_1$. If you want to work it out, let $M_i = -M_0$ and $M_z(t) = 0$ in Eq. (14) and find $T_1$ in terms of $t$.

Big instrumentation hint: For samples of long $T_1$ you will find that in order to see both the \textit{A} and \textit{B} pulse responses you will need to increase the Time/Div setting on the oscilloscope so much that it becomes hard to see the pulses because they become proportionately narrower. But since you are only looking for the \textbf{DELAY TIME} setting which minimizes the FID signal from the \textit{B} pulse you can simply switch the \textsc{Sync} to \textit{B}, allowing you to view only the FID signal with a comfortable Time/Div setting.

Incidentally, you may want to see the reduction in $T_1$ when the pulse-sequence repetition time is much shorter. Why is this so?
2.2.2 Measuring $T_2$: spin echoes & multipulse sequence

You may have noticed, while adjusting the controls to set up the $\pi-\pi/2$ pulse sequence in the previous exercise, that you saw an additional bump in the scope trace at twice the delay time. This curious artifact was also noticed by Erwin Hahn in 1949 (when he was still a graduate student). Hahn realized that the signal must be due to a “rephasing” of the transverse components of the magnetic moments in the sample, leading to a nonzero transverse magnetization and subsequent FID signal. He called this process a “spin echo” [6]. In this exercise you will learn how to create a spin echo and how to use it to make a good measurement of the spin-spin relaxation time constant $T_2$.

Although a variety of pulse sequences can produce an echo signal, the strongest echo is produced by a $\pi/2-\pi$ pulse sequence. It is also easiest to understand how the transverse rephasing is caused by this particular sequence, which is illustrated in Fig. 14. The pulse

![Figure 14: Illustration of how a spin echo is created by a $\pi/2-\pi$ pulse sequence. From reference [5].](image-url)
sequence and resulting detector signal are shown at the top of the figure. A \(90^\circ (\pi/2)\) pulse tips the net magnetization into the \(x-y\) plane, creating the FID signal. This signal decays mainly due to dephasing of the moments as they precess at different rates in a nonuniform \(\vec{B}_0\) field. The lower part of the figure shows this dephasing from the point of view of the rotating reference frame. In this frame we see that some moments lag behind the group, while others lead. A \(180^\circ (\pi)\) pulse is applied at time \(\tau\) after the \(\pi/2\) pulse. This pulse effectively “flips the pancake” of the collection of dephased magnetic moments about an axis in the \(x-y\) plane. The overall precession of the group continues in the same direction, but now those moments which had lagged the group now lead the group, and vice versa. At time \(2\tau\) (TE in the figure) all of the moments come back into phase, and we see a recovery of the FID signal.

Hahn nicely illustrates the formation of the echo by means of an analogy [4, p. 8]:

Let a team of runners with different but constant running speeds start off at a time \(t = 0\) as they would do at a track meet. At some time \(T\) these runners would be distributed around the race track in apparently random positions. The referee fires his gun at a time \(t = \tau > T\), and by previous arrangement the racers quickly turn about-face and run in the opposite direction with their original speeds. Obviously, at a time \(t = 2\tau\), the runners will return together precisely at the starting line.

Before worrying about \(T_2\) and its relationship to spin echoes, first set up the spectrometer so that you can see an echo. The spectrometer should already be configured to observe FID pulses (see Section 2.1.2 for settings and Fig. 9 for connections).

Now you want to set the \(A\) pulse so that it rotates the net magnetization by \(\pi/2\) and then make a single \(B\) pulse so that it rotates the net magnetization by \(\pi\). Start with just the \(A\) pulse (\(B\) pulses turned off) and adjust the electronics (\(A\)-WIDTH, TUNING and FREQUENCY) to maximize the FID signal. Set the repetition time to 1 sec, 30\% (again, assuming mineral oil as a sample; with other samples you may need a longer time for full recovery of \(M_z\)) and make sure that the M-G switch is on. Next, turn on the \(B\) pulses and set the \(\text{NUMBER OF B PULSES}\) to 01 and the \(\text{DELAY TIME}\) to 2.5 ms. Finally, adjust the \(B\)-WIDTH (starting from fully counterclockwise) to maximize spin echo which occurs at twice the delay time. You should get a scope display similar to that in Fig. 15.

Now note what happens when you increase the delay time. The reduction in the magnitude of the echo signal is due to a loss of “phase memory” among the magnetic moments. This in turn is due to variations in the local magnetic field of each moment over time; it does not depend on the static variation in the applied \(\vec{B}_0\) field that is responsible for the comparatively short FID decay time. In other words, the reduction in the spin echo amplitude follows \(T_2\), not \(T_2^*\). Hahn explains [4, p. 9]:

The decay of the echo may be understood in terms of the race track analogy if it is assumed now that the runners become fatigued after the start of the race. For this reason they may change their speeds erratically or even drop out of the race completely. Consequently, following the second gun shot (the second pulse) some of the racers may return together at the starting line, but not all of them.

In terms of the analogy, \(T_2^*\) depends on the inherent differences in base speeds among the runners; in terms of our collection of moments this is like the different precession speeds of the
Figure 15: Oscilloscope trace showing one spin echo. The small transient between the initial FID signal and the echo is from the $\pi$ pulse. Note that the initial FID signal is larger in magnitude than the echo, indicating loss of phase memory among the spins. Sample: H$_2$O+CuSO$_4$.

spins in different parts of the sample. However, $T_2$ depends on the changing of each runner’s speed over time; similarly, each moment will change its precession speed as it feels the fluctuating effects of neighboring moments, diffusing molecules, etc. So, a way to measure $T_2$ is to track the peak height of the spin echo as a function of the delay time. For example, you could look for the change in delay time which reduces the echo by 1/2, and thus obtain the “half life” of the spin echo.

Unfortunately, this method is still beset by the effects of a non-homogeneous $\vec{B}_0$ field. If the molecules in the sample (typically a liquid) can diffuse easily to other parts, then the precession speeds of the moments in those molecules will change as they move into regions of different static field. The dephasing of the moments will occur more quickly than it would if the field were uniform.

Hahn was aware of the effects of diffusion in a nonuniform field, and developed a theoretical expression for the amplitude of a spin echo which depended on $T_2$ as well as the diffusion constant and magnetic field gradient. Later, Carr and Purcell discovered that the effects of diffusion could be minimized if one used a multipulse sequence, wherein many successive $\pi$ pulses followed after the initial $\pi/2$ pulse [7]. The result of such a sequence is a single scope trace showing many spin echoes. In effect, each echo is like a “start pulse” for the later $\pi$ pulse that creates a subsequent echo. The decay of the echo amplitudes follows an exponential curve, of time constant $T_2$ as long as the echo spacing is sufficiently close together relative to the diffusion time. A diagram showing the method of the multipulse sequence is given in Fig. 16 and an example of the scope trace with multiple echoes is shown in Fig. 17.

Use the multipulse sequence method to measure $T_2$ for the samples. The instrument should already be set up to look at a single spin echo. All you need to do is increase the NUMBER OF B PULSES. Note the effect; you should get a picture like Fig. 17. With more $B$ pulses it may now be easier to optimize the echo signals by fine tuning the $A$ and $B$ pulse widths; do this,
Figure 16: Multipulse spin echo sequence. Later 180° (π) pulses refocus the spins to produce additional spin echoes following on the first echo. The envelope of the decay follows an exponential in terms of $T_2$. From reference [5].

Figure 17: Oscilloscope trace showing many spin echoes. The small transients between the peaks are from the π pulses. $T_2$ can be measured from the decay of the envelope of the peaks; here we see a half-life time of $\approx 1.7$ ms, so $T_2 \approx 2.5$ ms. Sample: H$_2$O+CuSO$_4$. 

Sample H$_2$O+CuSO$_4$. 

$T_2$
and strive for maximum echo amplitudes and minimum between-echo “glitches” (short pulses). As the NUMBER OF B PULSES is increased, the sweep speed of the scope (time per division) must be reduced to display all the echoes. But as the sweep speed is reduced, so is the sampling rate (indicated at the top of the scope screen). If the sampling rate is too low, the echo signals will not be captured accurately. Check this out on the scope with 3 or 4 echoes and the DELAY TIME set to several milliseconds. Reduce the sweep speed of the scope to 25 ms/Div or slower and observe how the amplitude of the echoes is no longer accurately captured. You can work around this limitation by using the scope’s “Peak Detect” feature. With this feature, only the most extreme voltage at each pixel time is recorded. The Peak Detect mode is available under the scope’s “Acquire” menu.

Also, you may need to increase the repetition time to allow the system to come to equilibrium between pulse sequences. As pointed out in the PS1-A manual, “…each pulse sequence must wait at least 3 $T_1$ (preferably 6–10 $T_1$’s) before repeating the pulse train.” If the repetition time is too short, the FID and echo signals will be reduced. Once you have selected the number of $B$ pulses for your $T_2$ measurement, adjust the repetition time so that a slight variation in this time does not affect the amplitude of the echoes. For samples with a very long $T_1$, you may get better results by using manual triggering of the pulse sequence. You can do this by setting the MODE switch to MAN, and then firing the sequence with the MAN START button. The scope will hold the most recent waveform indefinitely.

Once a satisfactory spin echo sequence has been acquired, time and voltage measurements need to be taken of the maxima of the echoes to map out the envelope. This is most easily accomplished using the paired cursors on the scope (activate the “Cursor” menu for this option) which allow for both time and voltage measurements. It is useful to record the $\Delta V$ and $\Delta t$ measurements as well as the absolute values at each echo maximum.

2.3 Student Exercises

Now that you know how to operate the spectrometer and measure $T_1$ and $T_2$ accurately, you need to carry out some investigations. Here are some possibilities. You are required to do at least two of these, but you may do as many as you like.

**Exercise 1: Characteristics of different materials**

Measure $T_1$ and $T_2$ for a few of the following materials: mineral oil, Vaseline, glycerine, water, ethyl alcohol. Note: some have very long time constants, so you will need to keep a watch on the pulse repetition time and, in the case of $T_2$, the number of pulses and associated delay time. If your multipulse envelope does not follow an exponential curve, you have work to do. (Reduce the delay time and increase the number of $B$ pulses.)

**Exercise 2: Effect of phase transition on NMR signals**

Attempt to get a signal from the paraffin (wax) sample. It is weak, isn’t it? Now warm the sample tube with a candle flame until the wax melts (no hotter!!!) and quickly insert the tube into the magnet assembly so you can see the much stronger signal from the liquid paraffin. If you work quickly, you may be able to track $T_2$ as it cools. (Following $T_1$ is harder.)
Exercise 3: Effect of doping with paramagnetic ions

If you have done the continuous NMR experiment, you will be familiar with the idea behind this experiment. There is a jar with a saturated CuSO$_4$ solution available, as well as sample vials and deionized water. Use a syringe to put one drop of the CuSO$_4$ solution into a clean vial, and then add 9 drops of pure water. Measure $T_1$ and $T_2$ for the sample. Then make another sample with 1 drop of your mixture, and 9 drops of pure water, and repeat the measurement. Continue in this manner until you have covered several decades of dilution, and then plot your results. (Log-log scales are handy.) If you look up solubility data in a handbook, you can calculate the absolute concentration of the Cu$^{++}$ ions. Compare your results to those given by Hahn [6, p. 586].

Exercise 4: Effect of diffusion on spin echo signal

As discussed above, diffusion of spins in a magnetic field gradient will affect the spin-echo amplitude. Hahn [6], and later, Carr and Purcell [7], derived an expression for the decay envelope which included the diffusion effects. It has this form:

$$M_{\text{peak}}(t) = M_i \exp(-t/T_2) \exp(-Kt^3/n^2),$$

(15)

where $M_{\text{peak}}(t)$ is the amplitude of the echo peak at time $t$, $M_i$ is the amplitude of the initial $\pi/2$-pulse response, $K$ is a function of the diffusion constant and field gradient, and $n$ is the number of intervening $\pi$ pulses between the time of the initial $\pi/2$ pulse and the echo at time $t$. As shown by Carr and Purcell

$$K = \gamma^2 \left( \frac{\partial B_0}{\partial z} \right)^2 D,$$

(16)

where $\gamma$ is as defined in Eq. 1, $\partial B_0/\partial z$ is the gradient of the static field along the field direction, and $D$ is the diffusion constant of the sample material. As is evident from Eq. (15), if $T_2^3 \gg K$ and $n = 1$, the decay of the echo envelope should follow a form proportional to $e^{-Kt^3}$. But if we increase the number of $\pi$ pulses while making the time between them shorter, that is increase $n$, we can reduce the effect of the $e^{-Kt^3}$ factor.

In this exercise you want to test Eq. (15). Choose a sample with a long $T_2$—pure water is good. Set the instrument to make spin echoes; you will want a long repeat time, since $T_1$ is also long. Then study how the spin echo envelope evolves as you increase the NUMBER OF B PULSES while decreasing the DELAY TIME.

If you use the newer TDS 3012 (color screen) oscilloscope (not the older TDS 320, 340 or 360 monochrome scopes), you can obtain a picture of the pulse-height envelope that shows the $e^{-Kt^3}$ form to best effect. Use the manual trigger mode (MODE set to MAN, use MAN START to trigger), and the infinite persistence feature of the scope (available under the “Display” menu button)—this feature holds all previous waveforms indefinitely. Set the NUMBER OF B PULSES to 1, and then make a succession of traces by advancing the DELAY TIME and firing the sequence with the MAN START switch. The scope will simply add all of the signal traces to the display, and you can print out the final picture. If you do not have access to the TDS 3012, you can still do the experiment by measuring the pulse height of the echo for each value of the DELAY TIME, and then plotting the results.
Exercise 5: The Meiboom-Gill multipulse sequence

The Carr-Purcell multipulse method of obtaining $T_2$ suffers from a systematic experimental error: if the width and field strength of the $\pi$ pulses are not exactly right, then the phase error from each pulse will accumulate, leading to a decay of the echo envelope that is too rapid. To wit: if the $\pi$ pulses produce, say, a rotation of $186^\circ$ per pulse (rather than the expected $180^\circ$), then after, say, ten such pulses, the effective rotation would accumulate to a tilting of the moments by $60^\circ$ out of the $x$-$y$ plane, and the FID signal would be about half what it should be ($\propto \cos 60^\circ$).

Meiboom and Gill discovered a way to correct for this accumulated error by a simple trick. They found that by shifting the phase of the RF signal by $90^\circ$ for the first $\pi/2$ pulse relative to the phase of the RF signal for the subsequent $\pi$ pulses, that the phase error of two successive $\pi$ pulses could be made to cancel, and thus, over many pulses, there would be no accumulated error. In effect, the $90^\circ$ phase shift causes the first $\pi/2$ rotation of $\vec{M}$ to be about the $y'$ axis (in the rotating frame of reference), but the following $\pi$ rotations will be about the $x'$ axis [8].

As noted, the PS1-A has a provision for implementing (or not) the Meiboom-Gill phase shift on the first pulse. You can study the effect of this by setting up a long multipulse sequence and then noting how the spin echo envelope depends on the setting of the M-G switch and the B-WIDTH position. In particular, you can see a very interesting modulation of the echo envelope with the M-G switch off and the B-WIDTH set notably low. In your notebook discussion, see if you can explain the modulation in terms of the effect of accumulated phase errors. Mineral oil works well as a sample.

3 A Summary of Tasks for the Prepared Student

If you have read through the write-up and feel that you know the theoretical underpinnings well enough, you may want to follow the list below as you do the lab.

1. Learn how to use the pulse programmer. You should know how to set up single pulse sequences, double pulse sequences, and multipulse sequences, with various repetition times, delay times, and $A$ and $B$ pulse widths.

2. Set up the instrument to observe free induction decay (FID) by means of a $\pi/2$ pulse applied to a mineral oil sample. Study the effect of varying the $A$ pulse width.

3. Study the effect of shortening the pulse repetition time on the FID signal. Use this effect to estimate the longitudinal relaxation time $T_1$.

4. Set up a $\pi$–$\pi/2$ pulse sequence to make an accurate measurement of $T_1$ by means of the zero-crossing method. Measure $T_1$ for mineral oil.

5. Set up the instrument to observe spin echoes using the $\pi/2$–$\pi$ pulse sequence. Then use the multipulse sequence with many more $\pi$ pulses to display the envelope of the echo amplitudes. Use this to obtain an accurate measurement of $T_2$ for mineral oil.

6. Carry out at least two of the exercises in Section 2.3. Make sure you obtain representative printouts of the oscilloscope displays along with your hand-written data.
References


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pulsed_NMR_rev.tex -- Updated 21 February 2011