Part II

Gravitational-wave experiments

Data analysis techniques

In this chapter we begin our study of GW experiments. The functioning principles and the sensitivities of existing or planned detectors will be examined in great detail in the subsequent chapters. In this chapter we rather introduce a number of general concepts which characterize any GW detector, and we discuss the crucial problem of how to extract a GW signal from the (typically much larger) detector noise.

In Section 7.1 we will see how the various noise generated inside a the detector can be conveniently treated referring them to the detector input, and are characterized by a spectral strain sensitivity, which has dimensions $1/\sqrt{\text{Hz}}$. In Section 7.2 we introduce the pattern functions that encode the detector angular sensitivity. We will then discuss in Section 7.3 the optimum filtering techniques that must be applied to the detector output. The importance of this procedure stems from the fact that, with existing detectors and with reasonable estimates of the GW signal, we expect that the GW signal will be buried into a much larger noise. The fact that we try to extract a small signal from noisy detectors is certainly not a new situation in physics. Rather on the contrary, it is a typical problem in many fields, e.g. in radio engineering where it has been much studied in connections with radars, or in radio astronomy for application to pulsar searches, and standard filtering techniques have been developed. We will see how these techniques are adapted to the problem of GW detection. The proper interpretation of the results obtained with matched filtering relies on notions of probability and statistics, that we discuss in Section 7.4. Here, after an introduction to the frequentist and the Bayesian frameworks, we discuss how to reconstruct the parameters of the source and how to examine the statistical significance of the observation of an event with a given signal-to-noise ratio. Then, in Sections 7.5–7.8, we will examine the application of these concepts to various classes of GW signals, i.e. bursts, periodic signals, coalescing binaries and stochastic backgrounds.

7.1 The noise spectral density

The output of any GW detector is a time series, which describes for instance the oscillation state of a resonant mass, or the phase shift of the light recombined after traveling in the two arms of an interferometer. This output will be a combination of a true GW signal (hopefully) and of noise. To understand how signal and noise combine, it is useful to think of a GW detector as a linear system. At its input there is the GW

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signal that we want to detect. More precisely, the input and output of the detector are scalar quantities, while the GW is described by a tensor h_{ij} . So, in general, the input of the detector will have the form

$$h(t) = D^{ij}h_{ij}(t), (7.1)$$

where D^{ij} is a constant tensor which depends on the detector geometry, and is known as the detector tensor. For example, for a detector which is driven only by the (x, x) component of h_{ij} (which, as we will see, is the case for a resonant bar oriented along the x axis), $D^{ij} = 1$ if i = j = 1 and $D^{ij} = 0$ otherwise. We will later compute the explicit form of D^{ij} for interferometers and for resonant masses.

For a linear system, the output of the detector is a linear function, in frequency space, of the input h(t), that is, the output $h_{\text{out}}(t)$ of the detector (in the absence of noise) is related to the input h(t) by

$$\tilde{h}_{\text{out}}(f) = T(f)\tilde{h}(f), \qquad (7.2)$$

where T(f) is known as the transfer function of the system. However, in the output of any real detector there will also be noise, so the output $s_{\text{out}}(t)$ will be rather given by

$$s_{\text{out}}(t) = h_{\text{out}}(t) + n_{\text{out}}(t). \tag{7.3}$$

More precisely, a detector can be modeled as a linear system with many stages, labeled by $i=1,\ldots,N$, each one with its own transfer function $T_i(f)$, so the total transfer function is $T(f)=\prod_i T_i(f)$. For example, we will see in Chapter 8 that resonant-bar detectors are composed of a heavy aluminum cylinder which is set into oscillation by an incoming GW; its energy is then transferred to a lighter mechanical oscillator, coupled to the heavy bar, which works as a mechanical amplifier, then it is transformed into an electric signal by an LC circuit coupled to the light oscillator, and then this electric signal is further amplified by one or more SQUIDs, and recorded. Clearly, noise can be generated at each of these stages. Each noise will propagate to the output with a transfer function which depends on the point of the linear system at which it first appeared, see Fig. 7.1, and will contribute to total noise $n_{\text{out}}(t)$ at the output. It is convenient to refer each noise to the detector input, defining the quantity n(t) from

$$\tilde{n}(f) = T^{-1}(f)\tilde{n}_{\text{out}}(f), \qquad (7.4)$$

where $n_{\text{out}}(t)$ is the total noise measured at the output. That is, n(t) is a fictitious noise that, if it were injected at the detector input, and if there were no other noise inside the detector, would produce at the output the noise $n_{\text{out}}(t)$ that is actually observed. It is therefore the quantity that we can compare directly with h(t), i.e. to the effect due to the GW. We then define

$$s(t) = h(t) + n(t),$$
 (7.5)

and we can simply think of the detector as if s(t) were its output, composed of a noise n(t) and a GW signal h(t). and the detection problem is how to distinguish h(t) from n(t). In the following, when we speak of the detector output, we will always refer to s(t). If one has a theoretical model for a given source of noise $n_i(t)$, which appears at a given stage of the linear system, we can compare it with h(t) simply multiplying it by the inverse of the appropriate transfer function, in order to refer this noise to the detector input. Equivalently, of course, one could refer both the noise and the signal to the true detector output, and compare $n_{\text{out}}(t)$ to the quantity $h_{\text{out}}(t)$ whose Fourier transform is given by eq. (7.2). However, the great advantage of referring everything to the inbut is that n(t) gives a measure of the minimum value of h(t) that can be detected and h(t), apart from the geometrical factor D^{ij} which is always of order one, depends only on the incoming GW. In contrast, $h_{\text{out}}(t)$ depends on the transfer function of the system, and different detectors can have transfer functions which differ by many orders of magnitude. Thus, the use of $n_{\text{out}}(t)$ and $h_{\text{out}}(t)$ would be very unpractical when we want to compare the performances of different detectors.

So, in the above sense, we take n(t) to be the detector's noise. If the noise is stationary, as we assume for the moment, the different Fourier components are uncorrelated, and therefore the ensemble average³ of the Fourier components of the noise is of the form

$$\langle \tilde{n}^*(f)\tilde{n}(f')\rangle = \delta(f - f')\frac{1}{2}S_n(f). \tag{7.6}$$

The above equation defines the function $S_n(f)$. Since n(t) is real, $\tilde{n}(-f) = \tilde{n}^*(f)$ and therefore $S_n(-f) = S_n(f)$. If n(t) is dimensionless, as we will assume, $S_n(f)$ has dimensions Hz^{-1} . Without loss of generality, we can also assume that

$$\langle n(t)\rangle = 0. (7.7)$$

Observe that, for f = f', the right-hand side of eq. (7.6) diverges. However, in any real experiment we have a finite value of the time T used to measure $\tilde{n}(f)$, see Note 3. Restricting the time interval to -T/2 < t < T/2 we have

$$\delta(f=0) \to \left[\int_{-T/2}^{T/2} dt \, e^{i2\pi f t} \right]_{f=0} = T.$$
 (7.8)

Then, from eq. (7.6) with f = f', we get

$$\langle |\tilde{n}(f)|^2 \rangle = \frac{1}{2} S_n(f) T.$$
 (7.9)

For a function defined on the interval [-T/2, T/2], the Fourier modes have discrete frequencies $f_n = n/T$, so the resolution in frequency is given by

$$\Delta f = \frac{1}{T} \,. \tag{7.10}$$

¹One often multiplies the detector output by $T^{-1}(f)$ already at the level of data acquisition, so in this sense s(t) is really the output of the data acquisition system.

²Some more nomenclature: we will always use the word "event" to indicate that in the detector happened something, which deserves further scrutiny. At this stage, it could be due to a GW or (much more likely) to noise. An event which is already assumed to have been generated by a GW will be called a "GW signal". The letter s conventionally used to denote the detector output s(t) = h(t) + n(t) does not stand for "signal" (the signal in this nomenclature is h(t)). It can rather be taken to denote the "strain amplitude" of the detector.

³The ensemble average is the average over many possible "realizations" of the system. In practice we have only one physical system, our detector, but we can follow it in time, so the ensemble average is replaced by a time average (this implicitly assumes that the system is ergodic). Then the ensemble average is computed measuring the noise n(t) over a given time interval T, and considering this as a "realization" of the system. From this we obtain $\tilde{n}(f)$ (with a resolution in frequency $\Delta f = 1/T$). We then repeat the procedures over a subsequent time stretch, again of duration T and separated by a sufficient time shift from the first realization, so that the correlation between the noise n(t) in the two stretches can be neglected, and we define this as a second independent realization of the system. Finally, we average $\tilde{n}(f)$ over many independent realizations. It is useful to keep in mind that a time-scale T is implicit in this procedure, and will indeed appear in the equations below.

Fig. 7.1 A schematic representation of a detector as a linear system. The full transfer function T(f) is the product of the separate transfer function. Here $T(f) = T_1(f)T_2(f)$, and $\tilde{n}_{\text{out}}(f) = T_1(f)T_2(f)n_a(f) + T_2(f)n_b(f)$.

We can then write eq. (7.9) also in the form

$$\frac{1}{2}S_n(f) = \langle |\tilde{n}(f)|^2 \rangle \Delta f. \qquad (7.11)$$

The factor 1/2 is conventionally inserted in the definition (7.6) of $S_n(f)$, so that $\langle n^2(t) \rangle$ is obtained integrating $S_n(f)$ over the physical range $0 \leq f < \infty$, rather than from $-\infty$ to ∞ ,

$$\langle n^{2}(t)\rangle = \langle n^{2}(t=0)\rangle$$

$$= \int_{-\infty}^{\infty} df df' \langle n^{*}(f)n(f')\rangle$$

$$= \frac{1}{2} \int_{-\infty}^{\infty} df S_{n}(f)$$

$$= \int_{0}^{\infty} df S_{n}(f). \qquad (7.12)$$

The function $S_n(f)$ is known as the noise spectral density (or the noise spectral sensitivity, or the noise power spectrum). More precisely, it is called a *single-sided spectral density*, to emphasize that $\langle n^2(t) \rangle$ is obtained from it integrating only over the physical range of frequencies f > 0. Alternatively, we can write

$$\langle n^2(t) \rangle = \int_{-\infty}^{\infty} df \, S_n^{\text{double sided}}(f) \,,$$
 (7.13)

with $S_n^{\text{double sided}}(f) = (1/2)S_n(f)$. Throughout this book, when we will use the term spectral density or power spectrum, we will always refer to the single-sided quantity.

Equivalently, the noise of a detector can be characterized by $\sqrt{S_n(f)}$, which is called the *spectral strain sensitivity*, or *spectral amplitude*, and has dimensions $Hz^{-1/2}$. Note that, if the noise increases by a factor λ , $n(t) \to \lambda n(t)$, then $S_n(f) \to \lambda^2 S_n(f)$ while the strain sensitivity scales linearly.

Actually the definition (7.6), even if rather intuitive, is not mathematically rigorous, because the function n(t) in general does not satisfy the conditions necessary for having a well-defined Fourier transform; for instance, on the interval $-\infty < t < \infty$, n(t) does not necessarily go to zero at $t \to \pm \infty$, so $\tilde{n}(f)$ in general does not exist. A more precise definition of the spectral density is obtained considering the auto-correlation function of the noise,

$$R(\tau) \equiv \langle n(t+\tau)n(t)\rangle \tag{7.14}$$

A Gaussian stochastic process n(t) is characterized uniquely by its average value $\langle n(t) \rangle$, that for a stationary noise is a constant and can be set to zero with a constant shift of n(t), and by its auto-correlation function. Typically, the knowledge of the noise at time t gives us very little information on the value of the noise at a subsequent time $t + \tau$ with τ sufficiently large, that is, for $|\tau| \to \infty$, $R(\tau)$ goes to zero quite fast, e.g.

exponentially, $R(\tau) \sim \exp\{-|\tau|/\tau_c\}$. The limiting case is white noise, in which the noise at time t and at any subsequent time $t + \tau$ are totally uncorrelated, so for $\tau \neq 0$ we have $\langle n(t+\tau)n(t)\rangle = \langle n(t+\tau)\rangle\langle n(t)\rangle = 0$, and $R(\tau) \sim \delta(\tau)$.

The auto-correlation function therefore goes to zero very fast as $\tau \to \pm \infty$, and it satisfies the requirements for performing the Fourier transform. We can then define the (one-sided) noise spectral density $S_n(f)$ by

$$\frac{1}{2}S_n(f) \equiv \int_{-\infty}^{\infty} d\tau \, R(\tau) \, e^{i2\pi f \tau} \,. \tag{7.15}$$

The reality of $R(\tau)$ implies $S_n(-f) = S_n^*(f)$, while invariance under time translations gives $R(-\tau) = \langle n(t-\tau)n(t) \rangle = \langle n(t)n(t+\tau) \rangle = R(\tau)$, which implies $S_n(-f) = S_n(f)$. Inverting eq. (7.15),

$$R(\tau) \equiv \langle n(t+\tau)n(t)\rangle$$

$$= \frac{1}{2} \int_{-\infty}^{\infty} df \, S_n(f) e^{-i2\pi f \tau} \,, \tag{7.16}$$

and in particular

$$R(0) = \langle n^2(t) \rangle$$

$$= \frac{1}{2} \int_{-\infty}^{\infty} df \, S_n(f)$$

$$= \int_{0}^{\infty} df \, S_n(f) . \tag{7.17}$$

Comparing this result with eq. (7.12) we see that, when $\tilde{n}(f)$ exists, eqs. (7.6) and (7.15) are equivalent definitions of S_n . Otherwise, only eq. (7.15) applies. Equation (7.15) is known as the Wiener–Khintchin relation.

If $R(\tau) \sim \delta(\tau)$, we see from eq. (7.15) that $S_n(f)$ is independent of frequency and therefore we have white noise. If instead $S_n(f)$ depends on f, one speaks generically of colored noise. A typical example is 1/f noise, which is a generic denomination for a noise where $S_n(f)$ has a power-law behavior, $S_n(f) \sim 1/f^{\gamma}$, over many decades in frequency.

7.2 Pattern functions and angular sensitivity

From eq. (1.58), we know that a GW with a given propagation direction $\hat{\mathbf{n}}$ can be written as

$$h_{ij}(t, \mathbf{x}) = \sum_{A=+,\times} e_{ij}^{A}(\hat{\mathbf{n}}) \int_{-\infty}^{\infty} df \, \tilde{h}_{A}(f) \, e^{-2\pi i f(t - \hat{\mathbf{n}} \cdot \mathbf{x}/c)}, \qquad (7.18)$$

where e_{ij}^A are the polarization tensors given in eq. (1.54). We take $\mathbf{x} = 0$ as the location of the detector. For a detector which is sensitive only to GWs with a reduced wavelength much larger than its size, such as

resonant masses and ground-based interferometers, we have $2\pi f \hat{\mathbf{n}} \cdot \mathbf{x} = \hat{\mathbf{n}} \cdot \mathbf{x}/\lambda \ll 1$ over the whole detector, and we can neglect the spatial dependence of $h_{ab}(t, \mathbf{x})$. So, to study the interaction of GWs with such detectors we can write simply

$$h_{ij}(t) = \sum_{A=+,\times} e_{ij}^{A}(\hat{\mathbf{n}}) \int_{-\infty}^{\infty} df \, \tilde{h}_{A}(f) \, e^{-2\pi i f t}$$
$$= \sum_{A=+,\times} e_{ij}^{A}(\hat{\mathbf{n}}) h_{A}(t) \,. \tag{7.19}$$

Combining this with eq. (7.1) we see that the contribution of GWs to the scalar output of the detector can be written as

$$h(t) = \sum_{A=+\times} D^{ij} e_{ij}^{A}(\hat{\mathbf{n}}) h_{A}(t).$$
 (7.20)

It is then convenient to define the detector pattern functions $F_A(\hat{\mathbf{n}})$,

$$F_A(\hat{\mathbf{n}}) = D^{ij} e_{ij}^A(\hat{\mathbf{n}}). \tag{7.21}$$

The pattern functions depend on the direction $\hat{\mathbf{n}} = (\theta, \phi)$ of propagation of the wave, and in terms of them eq. (7.20) becomes

$$h(t) = h_{+}(t)F_{+}(\theta,\phi) + h_{\times}(t)F_{\times}(\theta,\phi)$$
. (7.22)

The above equations assume that we have chosen a system of axes $(\hat{\mathbf{u}}, \hat{\mathbf{v}})$, in the plane orthogonal to the propagation direction $\hat{\mathbf{n}}$ of the wave, with respect to which the polarizations h_+ and h_\times are defined. It is interesting to see what happens if we change this system of axes, performing a rotation by an angle ψ in the transverse plane. Then the axes $(\hat{\mathbf{u}}, \hat{\mathbf{v}})$ are rotated to new axes $(\hat{\mathbf{u}}', \hat{\mathbf{v}}')$ given by

$$\hat{\mathbf{u}}' = \hat{\mathbf{u}}\cos\psi - \hat{\mathbf{v}}\sin\psi,
\hat{\mathbf{v}}' = \hat{\mathbf{u}}\sin\psi + \hat{\mathbf{v}}\cos\psi,$$
(7.23)

where we used the same conventions on the sign of ψ as in eqs. (2.188) and (2.194). With respect to the $(\hat{\mathbf{u}}, \hat{\mathbf{v}})$ axes, the amplitudes of the plus and cross polarizations have values h_+ and h_\times , while with respect to the $(\hat{\mathbf{u}}', \hat{\mathbf{v}}')$ axes, they have the values h'_+ and h'_\times . Equations (1.49) and (1.50) show that h'_+ and h'_\times are related to h_+ and h_\times by

$$h'_{+} = h_{+} \cos 2\psi - h_{\times} \sin 2\psi$$
, (7.24)

$$h_{\times}' = h_{+} \sin 2\psi + h_{\times} \cos 2\psi . \tag{7.25}$$

In the new frame, the definition (1.54) states that the polarization tensors are given by

$$(e_{ij}^{+})'(\hat{\mathbf{n}}) = \hat{\mathbf{u}}_{i}'\hat{\mathbf{u}}_{j}' - \hat{\mathbf{v}}_{i}'\hat{\mathbf{v}}_{j}', \qquad (e_{ij}^{\times})'(\hat{\mathbf{n}}) = \hat{\mathbf{u}}_{i}'\hat{\mathbf{v}}_{j}' + \hat{\mathbf{v}}_{i}'\hat{\mathbf{u}}_{j}'.$$
 (7.26)

Then, using eq. (7.23), we find

$$(e_{ij}^+)'(\hat{\mathbf{n}}) = e_{ij}^+(\hat{\mathbf{n}})\cos 2\psi - e_{ij}^\times \sin 2\psi,$$
 (7.27)

$$(e_{ij}^{\times})'(\hat{\mathbf{n}}) = e_{ij}^{+}(\hat{\mathbf{n}})\sin 2\psi + e_{ij}^{\times}\cos 2\psi. \tag{7.28}$$

The pattern functions F_A depends on the polarization tensors e_{ij}^A through eq. (7.21). Since the detector tensor is a fixed quantity, independent of ψ , we find that in the new frame

$$F'_{+}(\hat{\mathbf{n}}) = F_{+}(\hat{\mathbf{n}})\cos 2\psi - F_{\times}(\hat{\mathbf{n}})\sin 2\psi,$$
 (7.29)

$$F_{\times}'(\hat{\mathbf{n}}) = F_{+}(\hat{\mathbf{n}})\sin 2\psi + F_{\times}(\hat{\mathbf{n}})\cos 2\psi. \tag{7.30}$$

Combining this transformation of the pattern functions with the transformation of h_+, h_\times given in eqs. (7.24) and (7.25), we see that h(t) in eq. (7.22) is independent of ψ .

Of course, once a choice of the axes $(\hat{\mathbf{u}}, \hat{\mathbf{v}})$ used to define the polarization is made, then the pattern functions F_A depends on θ and ϕ only. However, it is sometime useful to keep generic the definition of the $(\hat{\mathbf{u}}, \hat{\mathbf{v}})$ axes in the transverse plane, and to parametrize the possible choices by the angle ψ . In this case, the pattern functions depend also on ψ , and

$$F_{+}(\hat{\mathbf{n}}; \psi) = F_{+}(\hat{\mathbf{n}}; 0) \cos 2\psi - F_{\times}(\hat{\mathbf{n}}; 0) \sin 2\psi,$$
 (7.31)

$$F_{\times}(\hat{\mathbf{n}}; \psi) = F_{+}(\hat{\mathbf{n}}; 0) \sin 2\psi + F_{\times}(\hat{\mathbf{n}}; 0) \cos 2\psi. \tag{7.32}$$

A useful identity satisfied by the pattern functions, independently of the specific form of the detector tensor D_{ij} , is⁴

$$\int \frac{d^2 \hat{\mathbf{n}}}{4\pi} F_+(\hat{\mathbf{n}}) F_\times(\hat{\mathbf{n}}) = 0, \qquad (7.33)$$

where as usual $d^2\hat{\mathbf{n}} = d\cos\theta d\phi$ is the integral over the solid angle. As for the integral over $d^2\hat{\mathbf{n}}$ of F_+^2 and of F_\times^2 , with a generic choice of the angle ψ they are different. We will see for instance that one can choose ψ so that F_\times vanishes while F_+ is non-zero, or viceversa. However, if we average over the angle ψ , we find

$$\int_{0}^{2\pi} \frac{d\psi}{2\pi} F_{+}^{2}(\hat{\mathbf{n}}; \psi) = \int_{0}^{2\pi} \frac{d\psi}{2\pi} F_{\times}^{2}(\hat{\mathbf{n}}; \psi). \tag{7.34}$$

In fact, inserting eqs. (7.31) and (7.32) into eq. (7.34), the equality follows from $\int d\psi \sin 2\psi \cos 2\psi = 0$ and $\int d\psi \sin^2 \psi = \int d\psi \cos^2 \psi$. From this, it also trivially follows that

$$\langle F_{+}^{2}(\hat{\mathbf{n}};\psi)\rangle = \langle F_{\times}^{2}(\hat{\mathbf{n}};\psi)\rangle,$$
 (7.35)

where

$$\langle \ldots \rangle \equiv \int_0^{2\pi} \frac{d\psi}{2\pi} \int \frac{d^2\hat{\mathbf{n}}}{4\pi} (\ldots) \,.$$
 (7.36)

For later use we also define the angular efficiency factor

$$F = \langle F_+^2 \rangle + \langle F_\times^2 \rangle = 2 \langle F_+^2 \rangle. \tag{7.37}$$

⁴Equation (7.33) can be shown in full generality writing

$$\int d^2 \hat{\mathbf{n}} F_+(\hat{\mathbf{n}}) F_{\times}(\hat{\mathbf{n}})$$

$$= D_{ab} D_{cd} \int d^2 \hat{\mathbf{n}} e_{ab}^+(\hat{\mathbf{n}}) e_{cd}^{\times}(\hat{\mathbf{n}}) ,$$

and using eq. (1.54), which shows that $e_{ab}^+(\hat{\mathbf{n}})e_{cd}^*(\hat{\mathbf{n}})$ is a sum of terms such as $\hat{\mathbf{u}}_a\hat{\mathbf{u}}_b\hat{\mathbf{u}}_c\hat{\mathbf{v}}_d$, which has three factors $\hat{\mathbf{u}}$ and one factor $\hat{\mathbf{v}}$, and of similar terms with $\hat{\mathbf{u}} \leftrightarrow \hat{\mathbf{v}}$. A simple way to see that the integral over $d^2\hat{\mathbf{n}}$ vanishes is then to observe that, when we integrate over all possible values of $\hat{\mathbf{n}}$, for each term $\hat{\mathbf{u}}_a\hat{\mathbf{u}}_b\hat{\mathbf{u}}_c\hat{\mathbf{v}}_d$ there is also a corresponding term obtained with $\hat{\mathbf{u}} \to -\hat{\mathbf{u}}$ and $\hat{\mathbf{v}} \to +\hat{\mathbf{v}}$, which cancels it.

Table 7.1 The pattern functions $F(\theta, \phi; \psi = 0)$ for various detectors. For interferometers, the arms are perpendicular and along the (x,y) axis, (θ,ϕ) are the usual polar angles defined using the z axis as polar axis and, for a wave propagating along the z axis, ψ is the angle in the (x,y) plane measured from the x axis, just as ϕ For cylindrical bars, θ is measured from the longitudinal axis of the bar and, if we denote by x the longitudinal axis, for a wave propagating along the z axis, again ψ is the angle in the (x,y) plane measured from the x axis. For resonant spheres, the modes m = 0, 1c, 1s, 2c, 2s are combinations of the five quadrupolar modes with $m = -2, \dots, 2$, defined in Zhou and Michelson (1995). The angular efficiency factor F is defined in eqs. (7.36) and (7.37). Observe that the mode m=0 of a sphere has the same pattern functions as a cylindrical bar (apart from a constant), while the mode m=2c has the same pattern functions as an interferometer.

Detector	$F_{+}(\theta, \phi; \psi = 0)$	$F_{\times}(\theta, \phi; \psi = 0)$	F
interferometers	$\frac{1}{2}(1+\cos^2\theta)\cos2\phi$	$\cos\theta\sin2\phi$	2/5
cylindrical bars	$\sin^2 \theta$	0	8/15
resonant spheres $m = 0$ $m = 1s$ $m = 1c$ $m = 2s$ $m = 2c$	$(\sqrt{3}/2)\sin^2\theta$ $-\sin\theta\cos\theta\sin\phi$ $\sin\theta\cos\theta\cos\phi$ $-\frac{1}{2}(1+\cos^2\theta)\sin2\phi$ $\frac{1}{2}(1+\cos^2\theta)\cos2\phi$	$0 \\ \sin \theta \cos \phi \\ \sin \theta \sin \phi \\ \cos \theta \cos 2\phi \\ \cos \theta \sin 2\phi$	2/5 2/5 2/5 2/5 2/5 2/5

We will compute the explicit forms of $F_{+,\times}(\theta,\phi;\psi)$ for bars and interferometers, in their respective chapters. We find useful to collect here the result that we will find for interferometers, cylindrical bars and resonant spheres: in Table 7.1 we give the value of $F(\theta, \phi; \psi = 0)$ (with appropriate definitions of the angles, discussed in the table caption and, in more detail, in their respective chapters), and the values of the angular efficiency factor F.

As we see from the above table, the pattern functions are relatively smooth functions of the position of the source in the sky. On the one hand, this has the positive consequence that GW detectors have a large sky coverage, of almost 4π , except for some blind directions. This is very different from conventional astronomy, where a telescope must point the source very precisely to detect it. The reverse of the coin, however, is that with a single GW detector we cannot determine the position of the source in the sky. A single detector has an output h(t) that, according to eq. (7.22), depends on four unknown: the two functions $h_{+,\times}(t)$ and the angles (θ, ϕ) that give the source position. To disentangle these quantities we need a coincident observation by a network of detectors. With two detectors we have at our disposal their two outputs $h_1(t)$ and $h_2(t)$ and the delay time τ_{12} between these two signals. These three quantities are not yet sufficient to solve for the four unknown $h_{+}(t), h_{\times}(t), \theta$ and ϕ

However, with three interferometers we have five measured quantities. the three functions $h_i(t)$, i=1,2,3, and two independent delay times, so we can solve for $h_{+}(t), h_{\times}(t), \theta$ and ϕ . The actual accuracy of the reconstruction depends on the signal-to-noise ratio. For typical expected signals, at first-generation interferometers the angular resolution could be of order one square degree.

Matched filtering

We have seen above that the detector output will be of the general form s(t) = h(t) + n(t). Naively, one might then think that we can detect a GW signal only when |h(t)| is larger than |n(t)|. This would be very unfortunate since we will see that, with plausible estimates of the expected GW signals bathing the Earth, and with the sensitivity of the present generation of detectors, we will rather be in the situation $|h(t)| \ll |n(t)|$.

The fundamental question that we ask in this section is then how can we dig out the GW signal from a much larger noise. This is a classical problem in many fields of physics or in radio engineering, and the answer is that we can detect values of h(t) much smaller than the floor of the noise if we know, at least to some level of accuracy, the form of h(t).⁵ To understand the basic idea, we can first illustrate a simple version of this "filtering" procedure, before moving to optimal filtering. Suppose that s(t) = h(t) + n(t), and that we know the form of the GW signal h(t) that we are hunting for. Then we can multiply the output s(t) by h(t), integrate over an observation time T, and divide by T,

$$\frac{1}{T} \int_0^T dt \, s(t)h(t) = \frac{1}{T} \int_0^T dt \, h^2(t) + \frac{1}{T} \int_0^T dt \, n(t)h(t) \,. \tag{7.38}$$

The crucial point now is that h(t) and n(t), separately, are oscillating functions. However, the integrand of the first integral on the right-hand side is definite positive; it might be for instance the integral of something like $\cos^2 \omega t$, times a slowly varying function of time; this integral then grows, for large T, as T. Its value averaged over a time T is therefore of order one in T.

$$\frac{1}{T} \int_0^T dt \, h^2(t) \sim h_0^2 \,, \tag{7.39}$$

where h_0 is the characteristic amplitude of the oscillating function h(t)In contrast, since the noise n(t) and our chosen function h(t) are uncorrelated, the quantity n(t)h(t) is oscillating, and its integral will grow only as $T^{1/2}$ for large T (as is typical of systems performing a random walk), so

$$\frac{1}{T} \int_0^T dt \, n(t)h(t) \sim \left(\frac{\tau_0}{T}\right)^{1/2} n_0 h_0 \,, \tag{7.40}$$

where n_0 is the characteristic amplitude of the oscillating function n(t). and τ_0 a typical characteristic time, e.g. the period of the oscillating

⁵More precisely, we must know h(t) and have an idea of the typical scales of variations of the noise, in order to exploit their different behaviors.

function h(t). Thus, in the limit $T \to \infty$, the second term on the right-hand side of eq. (7.38) averages to zero, and we have "filtered out" the contribution of the noise from the output. Of course, in practice we cannot sent T to infinity, either because the signal h(t) itself has a limited temporal duration or because we are limited by the total available observation time. Still we see that, to detect the signal given in eq. (7.39) against the background of eq. (7.40), it is not necessary to have $h_0 > n_0$, but it suffices to have $h_0 > (\tau_0/T)^{1/2}n_0$. For example, for a periodic signal with a period $\tau_0 \sim 1$ ms, such as a millisecond pulsar, observed for T=1 yr, we have $(\tau_0/T)^{1/2} \sim 10^{-5}$. We can therefore dig very deeply into the noise floor.

After having discussed the intuitive idea, let us see how the above procedure can be made more precise mathematically, and optimized in order to obtain the highest possible value of the signal-to-noise ratio. We define

 $\hat{s} = \int_{-\infty}^{\infty} dt \, s(t) K(t) \,, \tag{7.41}$

where K(t) is called the *filter* function. We assume that we know what GW signal we are looking for, i.e. we know the form of h(t). We then ask what is the filter function that maximizes the signal-to-noise ratio, for such a signal. Since the filter function is chosen so to "match" the signal that we are looking for, the technique is called matched filtering.⁶

The signal-to-noise ratio (in amplitude) is defined as S/N, where S is the expected value of \hat{s} when the signal is present, and N is the rms value of \hat{s} when the signal is absent. Since $\langle n(t) \rangle = 0$, we have

$$S = \int_{-\infty}^{\infty} dt \, \langle s(t) \rangle K(t)$$

$$= \int_{-\infty}^{\infty} dt \, h(t) K(t)$$

$$= \int_{-\infty}^{\infty} df \, \tilde{h}(f) \tilde{K}^*(f) , \qquad (7.42)$$

while

$$N^{2} = \left[\langle \hat{s}^{2}(t) \rangle - \langle \hat{s}(t) \rangle^{2} \right]_{h=0}$$

$$= \langle \hat{s}^{2}(t) \rangle_{h=0}$$

$$= \int_{-\infty}^{\infty} dt dt' K(t) K(t') \langle n(t) n(t') \rangle \qquad (7.43)$$

$$= \int_{-\infty}^{\infty} dt dt' K(t) K(t') \int_{-\infty}^{\infty} df df' e^{2\pi i f t - 2\pi i f' t'} \langle \tilde{n}^{*}(f) \tilde{n}(f') \rangle.$$

Using eq. (7.6) we obtain

$$N^{2} = \int_{-\infty}^{\infty} df \, \frac{1}{2} S_{n}(f) |\tilde{K}(f)|^{2} \,, \tag{7.44}$$

and therefore

$$\frac{S}{N} = \frac{\int_{-\infty}^{\infty} df \, \tilde{h}(f) \tilde{K}^*(f)}{\left[\int_{-\infty}^{\infty} df \, (1/2) S_n(f) |\tilde{K}(f)|^2 \right]^{1/2}} \,. \tag{7.45}$$

We now ask what is the filter K(t) that maximizes S/N, for a given h(t). This variational problem is elegantly solved by defining the scalar product between two real functions A(t) and B(t), by

$$(A|B) = \operatorname{Re} \int_{-\infty}^{\infty} df \, \frac{\tilde{A}^*(f)\tilde{B}(f)}{(1/2)S_n(f)}$$
$$= 4 \operatorname{Re} \int_0^{\infty} df \, \frac{\tilde{A}^*(f)\tilde{B}(f)}{S_n(f)} \,, \tag{7.46}$$

where Re denotes the real part, and the second line holds because we take A(t) and B(t) to be real functions, so that $\tilde{A}(-f) = \tilde{A}^*(f)$ (recall also that $S_n(-f) = S_n(f)$). Since $S_n(f) > 0$, this scalar product is positive definite. Then eq. (7.45) can be written as

$$\frac{S}{N} = \frac{(u|h)}{(u|u)^{1/2}}. (7.47)$$

where u(t) is the function whose Fourier transform is

$$\tilde{u}(f) = \frac{1}{2} S_n(f) \tilde{K}(f). \tag{7.48}$$

In this form, the solution is clear. We are searching for the "vector" of unit norm $\hat{n} = u/(u|u)^{1/2}$, such that its scalar product with the "vector" h is maximum. This is obtained choosing \hat{n} and h parallel, i.e. $\tilde{u}(f)$ proportional to $\tilde{h}(f)$, so we get

$$\tilde{K}(f) = \text{const.} \frac{\tilde{h}(f)}{S_n(f)}.$$
 (7.49)

The constant is arbitrary, since rescaling \hat{s} by an overall factor does not change its signal-to-noise ratio. Equation (7.49) defines the matched filter (or Wiener filter).⁷ In particular, if we are looking for a signal h(t) embedded into white noise, so that $\tilde{S}_n(f)$ is a constant, then the best filter is provided by the signal itself, which is the filtering discussed in eq. (7.38). However, when $\tilde{S}_n(f)$ is not flat, eq. (7.49) tells us that we must weight less the frequency region where the detector is more noisy, a very natural result.

Inserting the solution (7.49) into eq. (7.48) we get $\tilde{u} = \text{const.} \times \tilde{h}$. Plugging this into eq. (7.47), the overall constant cancels and we get the optimal value of S/N,

$$\left(\frac{S}{N}\right) = (h|h)^{1/2},\tag{7.50}$$

that is

$$\left(\frac{S}{N}\right)^2 = 4 \int_0^\infty df \, \frac{|\tilde{h}(f)|^2}{S_n(f)} \,, \tag{7.51}$$

which is the optimal value of the signal-to-noise ratio.⁸ The above equations are completely general, and independent of the form of $\tilde{h}(f)$. In Sections 7.5–7.8 we will apply them to some specific signals.

⁶We limit ourselves to linear filters, i.e. filters in which \hat{s} is linear in s(t), as in eq. (7.41).

⁷It is also common in the literature to write eq. (7.41) in the form $\hat{s} = \int_{-\infty}^{\infty} dt \, s(t) G(-t)$, and to call G(t) the filter function. So G(t) = K(-t) and $\tilde{G}(f) \sim \tilde{h}^*(f)/S_n(f)$.

⁸Recall from Section 7.1 that our $S_n(f)$ is single-sided. In terms of the double-sided spectral density, defined after eq. (7.13), we have $(S/N)^2 = \int_{-\infty}^{\infty} df |\tilde{h}(f)|^2 / S_n^{\text{double sided}}(f)$.

⁹See Note 2 on page 337 for the distinc-

tion between events, GW signals, and

detector output.

7.4 Probability and statistics

The matched filtering technique discussed above (as well as other techniques that we will meet later in this chapter) provide us with a way to optimize the signal-to-noise ratio, assuming that a given signal is indeed present in our data stream. The issue that the experimenter normally faces (especially in the field of GW experiments) is however different We do not know a priori whether a GW signal is present or not in a given stream of data, and we know even less its waveform. We can apply the matched filtering technique repeating it with many possible different filters, e.g. many possible starting times for the putative signal, many possible parameters describing a family of waveforms, etc. and we will correspondingly extract from our data stream a number of "events", 9 with various values of the signal-to-noise ratio S/N. What can we conclude from this? When can we claim detection of GWs? And if we can claim detection, what can we learn from it, in particular how can we reconstruct the properties of the source (such as, for an astrophysical source, its direction, its distance, its mass, etc.), and with what accuracy?

To address these questions we need to use statistical reasoning. Before looking into the technical aspects, it is however useful to discuss more generally the statistical frameworks that one can use, as we do in the next subsection.

7.4.1 Frequentist and Bayesian approaches

An abstract definition of probability can be obtained by considering a set S with subsets A, B, \ldots , whose interpretation for the moment is left open, and defining the probability P as a real-valued function that satisfies the Kolmogorov axioms: 1. For every A in S, $P(A) \ge 0$. 2. For disjoint subsets (i.e. $A \cap B = 0$), $P(A \cup B) = P(A) + P(B)$, and 3. P(S) = 1. Furthermore, one defines the conditional probability P(A|B) (i.e. the probability of A given B) as

$$P(A|B) = \frac{P(A \cap B)}{P(B)}. (7.52)$$

There exist two main approaches to probability, frequentist (also called classical) and Bayesian, depending on the interpretations of the subsets A, B, \ldots

In the frequentist interpretation, A, B, \ldots are the outcome of a repeatable experiment, and the probability P(A) is defined as the frequency of occurrence of A, in the limit of an infinite number of repetitions. In this interpretation, the probabilities of obtaining some data are of course well-defined, and it also makes sense to consider the conditional probability of obtaining some data, given some hypothesis (or given a theory, or given the value of the parameters in a theory). Therefore, quantities such as P(data|hypothesis) or P(data|parameters) make sense. However, one is never allowed to speak of the probability that the

parameters take a given value, nor of the probability that a hypothesis, or a theory, is correct. Hypotheses, or theories, are not the outcome of a repeatable experiment. Rather, they are correct or they are wrong, and similarly the true value of a parameter in a theory is what it is, and these are facts that are not subject to probabilistic analysis.

In the Bayesian approach, instead, one is allowed to consider the probability of a hypothesis, or of a theory, or the probability that a parameter within a theory takes a given value. To define these probabilities, one starts from the identities $P(A \cap B) = P(A|B)P(B)$ and $P(B \cap A) = P(B|A)P(A)$, which follow from the definition (7.52) of conditional probability. On the other hand, $A \cap B = B \cap A$ and therefore

$$P(A|B) = \frac{P(B|A)P(A)}{P(B)},$$
 (7.53)

which is Bayes' theorem. Observe also that, from the axioms of probability given above, it follows that

$$P(B) = \sum_{i} P(B|A_i)P(A_i), \qquad (7.54)$$

for any B and for A_i disjoints and such that $\bigcup_i A_i = S$. Therefore eq. (7.53) can be rewritten as

$$P(A|B) = \frac{P(B|A)P(A)}{\sum_{i} P(B|A_{i})P(A_{i})},$$
(7.55)

so the denominator is just a normalization factor. As long as A and B are the outcome of a repeatable experiment, eq. (7.55) would be accepted also by frequentists. In the Bayesian approach, however, one applies this to A = hypothesis (or parameters, or theory) and B = data. Then one finds that

$$P(\text{hypothesis}|\text{data}) \propto P(\text{data}|\text{hypothesis}) P(\text{hypothesis})$$
. (7.56)

The probability of the hypothesis given the data is called the posterior probability, and eq. (7.56) states that it is proportional to the product of two factors. The first is the probability of the data given the hypothesis (a "honest" frequentist probability), which is called the likelihood function. The second is the probability of the hypothesis, and is called the prior probability (or, simply, the prior). The latter cannot be determined just by performing identical trials (so it makes no sense to a frequentist) and, in the Bayesian approach, one must make assumptions to determine it. In fact, this prior probability in general can even depend on subjective factors, and on the state of knowledge of the person that makes the analysis. In the Bayesian interpretation, P(hypothesis) can be seen as the "degree of belief" that the hypothesis is true, and eq. (7.56) describes the evolution of this degree of belief due to the fact that we have performed the measurement. The prior probability describes the degree of belief in the hypothesis before the measurement was made, and the posterior probability describes the degree of belief after. 11

¹¹Observe also that eq. (7.56) is stated as a proportionality, so P(hypothesis|data) must be normalized summing over all possible hypothesis (or theories) that we want to compare, or integrating over a given domain of values for the continuous parameters.

¹⁰The kind of example that appears in all textbooks: we toss a coin five times. What is the probability of getting all five times head (data), given that the coin has 50% probability of heads and tails (hypothesis)?

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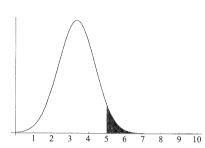


Fig. 7.2 The Neyman construction for the lower limit x_1 of the confidence interval. Here the measured value was $x_0 = 5$ and, to get the interval at 90% C.L., we look for a Gaussian distribution such that its area at $x \ge 5$ (shaded region) is 5% of the total area. This is a Gaussian centered in $x_1 \simeq x_0 - 1.64485\sigma$ (here we used $\sigma = 1$).

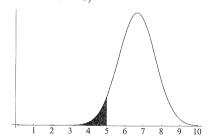


Fig. 7.3 The same as the previous figure, for the upper limit x_2 of the confidence interval. The Gaussian is now centered in $x_2 \simeq x_0 + 1.64485\sigma$.

This difference in approach implies also an important difference among the frequentist and the Bayesian notions of confidence interval and of confidence level (C.L.). The expression "confidence interval", without further qualifications, refers to the frequentist definition, and has the following meaning. Suppose that we are performing repeated identical measurements of a physical quantity x. We want to express our result saying that, at a given confidence level, say 90%, $x_1 < x < x_2$. What is meant by this is the following. The true (unknown) value of x is a fixed number x_t , which is always the same in all repetitions of the experiment. each repetition provides a different interval $[x_1, x_2]$, that we want to construct in such a way that x_t will be contained inside this interval in 90% (or whatever the specified C.L.) of the repetitions, no matter what the true value x_t is. This is the frequentist concept of coverage. There is a general construction, given by Neyman in a famous 1937 paper, that allows us to construct the frequentist confidence intervals. We illustrate it in the simple case in which we know that the experimental apparatus provides values distributed as a Gaussian around the true value x_t , with a standard deviation σ ,

$$P(x|x_t) = \frac{1}{(2\pi\sigma^2)^{1/2}} \exp\left\{-\frac{(x-x_t)^2}{2\sigma^2}\right\}.$$
 (7.57)

Suppose that a given repetition of the experiment yields the value x_0 . The Neyman's construction (using for definiteness 90% C.L.) proceeds by finding a value $x_1 < x_0$ such that 5% of the area under $P(x|x_1)$ is at $x > x_0$. That is, we fix x_1 by requiring that a Gaussian distribution centered on x_1 , only in 5% of the cases produces values of x higher than x_0 , see Fig. 7.2. If the true value x_t were smaller than such x_1 , then the value x_0 that we observed was due to a statistical fluctuation that takes place in less than 5% of the repetitions, so choosing in this way the lower limit of the interval, we are wrong at most in 5% of the cases. The upper limit of the confidence interval is obtained similarly, by finding a value $x_2 > x_0$ such that 5% of the area under $P(x|x_2)$ is at $x < x_0$, see Fig. 7.3. Observe that the probabilistic variables in this construction are x_1 and x_2 , while the true value x_t is fixed (and unknown).

In contrast, the Bayesian approach constructs a probability distribution for the true value x_t . This is obtained from the likelihood function P(data|hypothesis) in eq. (7.56), where the hypothesis is that the true value of x is x_t and the data is the observed value x_0 . We denote this likelihood function as $\Lambda(x_0|x_t)$. In our case, this is the same as the Gaussian given in eq. (7.57), so $\Lambda(x_0|x_t) = P(x_0|x_t)$. As long as we interpret it as the probability of obtaining the value x_0 , given that the true value is x_t , the likelihood function is a legitimate frequentist concept. However, in the Bayesian approach, it is inserted into eq. (7.56), together in this case with a flat prior in x_t , to get a probability density function (p.d.f.) in the variable x_t , given the observed value x_0 ,

$$P(x_t|x_0) = \frac{1}{(2\pi\sigma^2)^{1/2}} \exp\left\{-\frac{(x_t - x_0)^2}{2\sigma^2}\right\}.$$
 (7.58)

 S_0 , in this case we have a Gaussian distribution centered on x_0 (rather than on x_1 or on x_2 as in the Nevman construction), and we use it as a p.d.f. for x_t . The most probable value of x_t is found by maximizing this n d.f., which of course gives $x_t = x_0$, and the Bayesian 90% confidence interval is defined as the interval which subtends an area equal to 90% of the total area of the p.d.f. 12 In the case of a Gaussian distribution, the Bayesian and frequentist definitions give the same result for x_1 and x_2 , even if the interpretation is different. However, in a general situation, the two definitions do not agree. The frequentist confidence interval, by construction, always has the prescribed coverage, i.e. we are sure that in the limit of a large number of repetitions, 90% (or whatever the chosen C.L.) of the confidence intervals obtained by the different repetitions of the experiment will include ("cover") the true value x_t , no matter what x_t is. This covering properties is not necessarily true in the Bayesian procedure, which in certain cases yields intervals whose frequentist coverage is less than the stated C.L. (i.e. they undercover). This can happen in particular for event-counting experiments, that obey Poisson statistics, when the data sample is small.

Beside the situation when we have small numbers, the other typical situation where the Bayesian and frequentist approaches can give sensibly different answers is when the variable x, for physical reasons, has a bounded domain, and the measured values are close to the boundaries of the domain. An instructive example, that nicely illustrates the different results that can be obtained with the Bayesian and the frequentist approaches in such situations, is the following. 13 Nowadays, we know from oscillation experiments that the three neutrinos have a small mass, with squared masses (more precisely, squared mass differences) between 10^{-5} and $10^{-4} \,\mathrm{eV}^2$. Before these results, a number of other experiments attempted a direct measure of the mass m_{ν} of the electron neutrino (or more precisely, of m_{ν}^2) from tritium beta decay. In the early 1990s the experimental situation was that various experiments reported negative values for their best estimate of m_{ν}^2 . This is not surprising in principle since, if m_{ν}^2 were really zero, or anyway much smaller than the experimental accuracy (as indeed it was), and if the distribution of the data is an unbiased Gaussian, on average half of the ensemble of the experiments should report negative values, and statistical fluctuations can drive the average over the experiments in the unphysical region $m_{\nu}^2 < 0$. However, these negative fluctuations happened to be so large that even the frequentist upper limit at 90% C.L. was negative, and was $m_{\nu}^2 < -16 \,\mathrm{eV}^2$. To say the least, it is quite disturbing to set up a complicated experiment to come out with the conclusion that m_{ν}^2 is smaller than a negative value. The point is that this statement holds at 90% C.L., so it should be false in 10% of the cases, and here we know for sure that we are in this false 10%. 15

A possible alternative in this case is to include our prior information that $m_{\nu}^2 \geq 0$. This suggests to take a Bayesian approach with a prior p.d.f. $P(m_{\nu}^2)$ which is zero when $m_{\nu}^2 < 0$, and uniform for $m_{\nu}^2 \geq 0$, and to use the resulting posterior p.d.f. to set the bound on m_{ν}^2 . Here however

¹³We follow the paper by Cousins (1995), "Why isn't every physicist a Bayesian?", where the reader can find a very clear exposition of the difference between the Bayesian and frequentist approaches.

¹⁴Since the early 1990s, direct experiments (i.e. experiments not based on oscillations) on the electron neutrino mass squared have improved, but still their world average is negative, see Yao et al. [Particle Data Group] (2006).

¹⁵It should be mentioned that a strict application of the frequentist Neyman construction can never produces an upper limit in the unphysical region, but rather an empty confidence interval (which is equally disturbing). There is however a generalization of the Neyman construction that produces nonempty intervals in the physical region, see Feldman and Cousins (1998).

¹²Such an interval is selected uniquely by imposing an extra requirement, typically that it is symmetric around the maximum, or that it is the minimum length interval. For a Gaussian distribution, these two conditions give of course the same result.

the problem arises as to whether, in the region $m_{\nu}^2 \geqslant 0$, the prior should be uniform in the variable m_{ν}^2 , or in m_{ν} , or in $\log m_{\nu}$, etc. Of course, a distribution $P(m_{\nu}^2)dm_{\nu}^2$ with $P(m_{\nu}^2)=$ const. is flat with respect to the variable m_{ν}^2 but, since $dm_{\nu}^2=2m_{\nu}dm_{\nu}$, it is linearly raising with respect to m_{ν} . The issue is significant since the resulting upper bound depends on the choice. In this specific problem the consensus finally settled on using a flat prior in m_{ν}^2 , which gave an upper bound, at 90% C.L., $m_{\nu}^2 < 26.6 \, \mathrm{eV}^2$.

A physicist that is not too much interested in the philosophical aspects of the debate, can take a pragmatic attitude and use a frequentist or a Bayesian approach, depending on the kind of experiment to be analyzed. In particular, elementary particle physics is very well suited for the frequentist approach. This basically stems from the fact that in this case it is the physicist that controls the parameters of the experiments (e.g. the kind of particle used in the beams, the beam energy, etc.) and can reproduce them accurately many times. We are therefore in the situation where the frequentist notion of repeated trials fits very well. ¹⁶ The advantage is that this allows us to report objectively the outcome of the experiment, without the need of incorporating prior (and possibly subjective) beliefs.

On the other hand in astrophysics, and even more in GW astrophysics. the sources can be rare, they are not under the control of the experimenter, and each one is very interesting individually. If a single BH-BH binary coalesces, and we detect its signal in a GW experiment, we would obviously be very interested in questions such as in which direction the binary system was, at what distance from us, what were the masses of the two black holes, their spins, etc. A strict frequentist approach is inapplicable here. We do not have at our disposal an ensemble of identical BH-BH binaries located in that position, with the same value of the masses, etc. We just have that unique event, and we want to get the maximum out of it. In this case, a Bayesian approach can be more appropriate, since it allows us to ask questions such as "What was the most likely value of the position, masses, spin, etc. of the BHs?" For this reason, while negative results, giving upper limits on the rate of GW signals, should normally be expressed in frequentist terms, the discussion of parameter estimation from a given positive detection, to which we turn next, should rather be performed within the Bayesian framework.

7.4.2 Parameters estimation

In Section 7.3, when we introduced the matched filtering technique, we assumed that the form of h(t) is known. In practice, however, h(t) will necessarily depend on a number of free parameters. For instance, if h(t) is a short burst of GWs, among its parameters we will certainly have its time of arrival t_0 . When searching for very short bursts we might simply use a Dirac delta, so $h(t) = h_0 \delta(t - t_0)$, but more generally we might also wish to include its temporal width Δt and possibly more parameters

describing the shape of the pulse. For a coalescing binary, among the parameters we will have the time of entry in the interferometer bandwidth, the distance to the source, the star masses, etc.

Therefore, we must consider a family of possible waveforms, or templates, that we denote generically as $h(t;\theta)$, where $\theta = \{\theta_1, \ldots, \theta_N\}$ is a collection of parameters. Correspondingly, we have a family of optimal filters $K(t;\theta)$, determined through eq. (7.49), $\tilde{K}(f;\theta) \sim \tilde{h}(f;\theta)/S_n(f)$. In practice, this means that we must discretize the θ -space, and repeat the filtering procedure many times, once for each point of this discretized parameter space (except that for some parameters the maximization procedure can be done analytically, see below).

The problem that we address in this section is the following. Suppose that a GW signal has indeed been detected, which means that for some template $h(t;\theta)$ the value of S/N, determined by the optimal Wiener filtering (or by any other procedure that we specified in advance) has exceeded a predetermined threshold, and the signal satisfies further criteria that we might have set for claiming detection, such as coincidences between different detectors (we will see in more detail in Sections 7.4.3 and 7.5.3 some possible criteria that could allow us to claim a detection, at a given confidence level). How do we reconstruct the most probable value of the parameters of the source, and how we compute the error on these parameters?

This question is Bayesian in nature, so its answer is contained in the posterior probability. To compute the likelihood function, and hence the posterior probability, we assume for simplicity that the noise n(t) is stationary and Gaussian. From eq. (7.6) we see that the variance of the Fourier mode of the noise with frequency f is proportional to $(1/2)S_n(f)$, so the corresponding Gaussian probability distribution for the noise is

$$p(n_0) = \mathcal{N} \exp \left\{ -\frac{1}{2} \int_{-\infty}^{\infty} df \, \frac{|\tilde{n}_0(f)|^2}{(1/2)S_n(f)} \, \right\} \,, \tag{7.59}$$

where \mathcal{N} is a normalization constant. This is the probability that the noise n(t), which is a random variable with zero mean, has a given realization $n_0(t)$. The above result can be rewritten very simply in terms of the scalar product (7.46) as¹⁷

$$p(n_0) = \mathcal{N} \exp\{-(n_0|n_0)/2\}.$$
 (7.62)

We are assuming that the output of the detector satisfies the condition for claiming detection, i.e. it is of the form $s(t) = h(t; \theta_t) + n_0(t)$, where $n_0(t)$ is the specific realization of the noise in correspondence to this event, and θ_t is the (unknown) true value of the parameters θ . The likelihood function for the observed output s(t), given the hypothesis that there is a GW signal corresponding to the parameters θ_t , is obtained plugging $n_0 = s - h(\theta_t)$ into eq. (7.62),

$$\Lambda(s|\theta_t) = \mathcal{N} \exp\left\{-\frac{1}{2}(s - h(\theta_t)|s - h(\theta_t))\right\}, \qquad (7.63)$$

¹⁷For simplicity, we limit ourselves to the case of a single detector. The formalism can however be extended straightforwardly to multiple detectors. In this case the definition of the noise spectral density, eq. (7.6), is replaced by

$$\langle \tilde{n}_a^*(f)\tilde{n}_b(f')\rangle = \delta(f - f')\frac{1}{2}[S_n(f)]_{ab},$$
(7.60)

where the indices a, b label the detectors. This definition takes into account the possibility of correlated noise. Let $\mathbf{A}(t)$ and $\mathbf{B}(t)$ be vectors whose components $A_a(t)$ and $B_a(t)$ are output of the single detectors, and let $[S_n^{-1}]^{ab}$ denote the inverse matrix. The equations of this section can then be generalized to multipole detectors, using the scalar product

$$(\mathbf{A}|\mathbf{B}) = 4 \text{ Re}$$
 (7.61)
$$\int_{0}^{\infty} df \, \tilde{A}_{a}^{*}(f) [S_{n}^{-1}(f)]^{ab} B_{b}(f) ,$$

which generalizes eq. (7.46). See the Further Reading for details.

¹⁸As an example of prior information, one of the typical parameters entering in the waveform is the distance r to the source, and we might be searching for signals from a population of stars with a known distribution in space, e.g. a distribution $p^{(0)}(r)dr \sim r^2 dr$ for isotropic sources, or $p^{(0)}(r)dr \sim rdr$ for sources within a few kpc from us, in the Galactic disk. Another typical parameter is the mass of the star and, for neu-

tron stars, we know from astrophysi-

cal observations that their mass distri-

bution is strongly concentrated around

 $1.35M_{\odot}$.

¹⁹Assuming that the eccentricity can be neglected, since the orbit should be highly circular by the time the signal enters in the bandwidth of a ground-based detector, as we saw in Section 4.1.3.

²⁰For details see, e.g. the statistics section of Yao et al. [Particle Data Group] (2006).

or, introducing the short-hand notation $h_t \equiv h(\theta_t)$,

$$\Lambda(s|\theta_t) = \mathcal{N} \exp\left\{ (h_t|s) - \frac{1}{2}(h_t|h_t) - \frac{1}{2}(s|s) \right\}.$$
(7.64)

In the Bayesian approach, according to eq. (7.56), we also introduce a prior probability $\hat{p}^{(0)}(\theta_t)$. 18 Then, the posterior probability distribution for the true value θ_t , given the observed output s,

$$p(\theta_t|s) = \mathcal{N} p^{(0)}(\theta_t) \exp\left\{ (h_t|s) - \frac{1}{2} (h_t|h_t) \right\},$$
 (7.65)

where, since we are considering $p(\theta_t|s)$ as a distribution in θ_t for a fixed output s, we have reabsorbed into the normalization factor \mathcal{N} the term (s|s)/2 which appears in the exponential in eq. (7.64).

Once the prior distribution is given, eq. (7.65) gives the p.d.f. in the parameter space, so in principle it contains all the information that we need. However, in this form the information might not be very manageable. The θ -space will in general be a multi-dimensional space of large dimension. For example, for a binary coalescence the parameters θ^i that determine the waveform, at the post-Newtonian level, are the distance, the source's location (two angles), the orientation of the normal to the orbit (two more angles), the time at which the signal enters in the interferometer's bandwidth, the orbital phase at that moment, the two masses of the stars, and their spins, so 15 parameters in all. 19 From the probability distribution function (7.65) in such a complicated space we would like to extract some more approximate, but also more manageable. information; essentially, we want the most probable value of the variables θ_t , that we denote by $\hat{\theta}$, and also their corresponding errors.

There is no unique way of defining what is the most probable value of θ_t . A rule for assigning the most probable value is called an estimator. The most important properties that an estimator must have are:20 (a) Consistency: the estimator must converge to the true value as the amount of data increases. This property is so important that it is possessed by all commonly used estimators. (b) The bias b is defined as the difference between the expectation value of the estimator, $E(\theta)$ (taken over a hypothetical set of similar experiments in which $\hat{\theta}$ is constructed in the same way), and the true value θ_t , $b \equiv E(\hat{\theta}) - \theta_t$. When b = 0 the estimator is said to be unbiased. (c) Efficiency: we want the smallest possible value for the variance of $\hat{\theta}$, and (d) Robustness, i.e. the property of being relatively insensitive to small departure in the assumed p.d.f. due to factors such as noise.

Two choice of estimators seems especially reasonable. The first is to define $\ddot{\theta}$ as the value which maximizes the probability distribution function (7.65). Another natural option is to define it as the average of θ_{t} , over the distribution (7.65). We discuss these options below.

Maximum likelihood estimator

Let us consider first the situation in which the prior probability is flat.²¹

Then, maximization of the posterior probability becomes the same as maximization of the likelihood $\Lambda(s|\theta_t)$. The value of θ_t that maximizes $\Lambda(s|\theta_t)$ defines the maximum likelihood estimator, and we denote it by $\hat{\theta}_{\mathrm{ML}}(s)$. It is the most widely used estimator in general situations.²² It is usually simpler to maximize $\log \Lambda$. From eq. (7.64),

$$\log \Lambda(s|\theta_t) = (h_t|s) - \frac{1}{2}(h_t|h_t).$$
 (7.66)

Since we are working at fixed s, the term (-1/2)(s|s) in eq. (7.64) is an irrelevant constant, and we omitted it. Denoting $\partial/\partial \theta_i^i$ simply by ∂_i . the value of $\hat{\theta}_{ML}$ is found by solving the equations

$$(\partial_i h_t | s) - (\partial_i h_t | h_t) = 0. (7.67)$$

The errors $\Delta \theta^i$ can then be defined in terms of the width of the probability distribution function (7.65) at the peak.

Typically, (7.67) is a set of equations that must be solved numerically (except for some parameter such as the overall amplitude that can be eliminated analytically, see below). However, they have a rather simple geometric interpretation. The set of all possible waveforms $h(t;\theta)$ defines a manifold, called the manifold of the signals, parametrized by the coordinates θ^i . This is a subset of zero measure in the space of all possible functions, so the addition of generic noise n(t) to a function $h(t;\theta)$ will necessarily bring us out of this manifold. In Fig. 7.4 we illustrate the situation with a two-dimensional manifolds of the signals. The point labeled θ_t represents the true value of the signal, and therefore lies on the manifold. The addition of noise carries us outside this manifold. Since we are minimizing (s-h|s-h), see eq. (7.63), the maximum likelihood estimator actually searches the point on the signal manifold which is closest to the output s, where distances are defined with respect to the scalar product (|).

To summarize, in the Bayesian framework $\hat{\theta}_{ML}$ is determined assuming a flat prior distribution and requiring the maximization of the posterior probability (7.65), i.e. maximizing our "degree of belief" in the hypothesis that there is a GW signal. A natural question, at this point, is what is the relation between $\hat{\theta}_{ML}$ and the value of θ that provides the highest signal-to-noise ratio in the matched filtering. We now prove that in fact they are the same. To show it, we write the generic template as $h(t;\theta) = ah_a(t;\xi)$, where a is an amplitude, and is a free parameter, while the normalization of h_a has been fixed imposing some condition. We have separated the parameters θ into a and the remaining parameters, that we call ξ . The maximization with respect to a of $\log \Lambda$ can be performed analytically since, from eq. (7.66).²³

$$\log \Lambda(s|a,\xi) = a(h_a|s) - \frac{a^2}{2}(h_a|h_a). \tag{7.68}$$

Requiring $\partial \log \Lambda / \partial a = 0$, we get the maximum likelihood estimate for a,

$$\hat{a}_{\rm ML}(s) = \frac{(h_a|s)}{(h_a|h_a)} \,. \tag{7.69}$$

²²See any textbook on statistics, e.g. Lyons (1986), Section 4.4, for an introduction to the maximum likelihood method and its virtues. Observe that the likelihood is a legitimate concept also in the frequentist approach. The most probable frequentist value is again identified with the maximum of the likelihood, and the confidence interval is usually defined in terms of the point where $2 \log \Lambda$ decreases by one unit with respect to its value at the maximum. In the frequentist approach, however, we cannot use the likelihood as a p.d.f. for the true values of the parameters. i.e. we cannot consider areas under the curve, and of course we cannot include

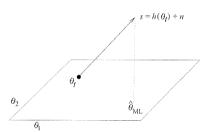


Fig. 7.4 The manifolds of the signals, parametrized by the coordinates (θ_1, θ_2) . The point θ_t is the true value of the signal. The addition of noise to $h(\theta_t)$ brings us outside this manifold, and the maximum likelihood estimator searches the point on the signal manifold which is closest to s.

²³To keep the notation lighter, we omit the subscript t (which stands for "true") from a and ξ . We are anyway considering p.d.f. for the true values of the parameters.

²¹As we already mentioned when discussing the example of the neutrino mass on page 350, a distribution which is flat with respect to the variables θ_t is no longer flat if we make a nonlinear transformation of the parameters. Therefore this prior distribution assumes a definite choice of coordinates in the parameter space.

The maximization with respect to the remaining variables ξ can be performed substituting this expression for a into $\log \Lambda$, obtaining

$$\log \Lambda(s|\xi) = \frac{1}{2} \frac{(h_a|s)^2}{(h_a|h_a)}.$$
 (7.70)

The maximization of this quantity amounts to maximizing the overlap of the output s with the normalized template $h_a/(h_a|h_a)^{1/2}$, where the overlap is measured using the scalar product $(\ |\)$, defined in terms of the noise spectral density $S_n(f)$. This is just the matched filtering procedure discussed in the Section 7.3. Thus, the maximum likelihood method provides a way of estimating the overall amplitude a (which cannot be fixed just searching for the filter that maximizes the signal-to-noise ratio, since eq. (7.47) is unchanged by a multiplicative rescaling of the filter u) while, for the remaining parameters, it returns the values that maximize the signal-to-noise ratio according to matched filtering.²⁴

Maximum posterior probability

In various situations we do have important prior information, and we might want to include it in the analysis, see Note 18 for examples. In this case, rather then maximizing the likelihood function, we must determine the estimator by maximizing the full posterior probability $p(\theta_t|s)$ given in eq. (7.65), which takes into account the prior probability distribution. For a generic prior, of course, the maximum of the posterior distribution will change, so it will no longer coincide with the value that gives the highest signal-to-noise ratio in the matched filtering. What happens is that the value suggested by matched filtering is weighted against our prior expectations (in a real sense, our "prejudices"), to provide a new estimate of the most likely value for the true parameters.

When we want to include non-trivial prior information, some conceptual complication may appear (apart from the issue of how to choose the appropriate prior). Suppose, for definiteness, that we have a twodimensional parameter space (θ_1, θ_2) , as in Fig. 7.4, and that we are not interested in the variable θ_2 . Then, we can integrate the p.d.f. given in eq. (7.65) over θ_2 , to obtain a reduced p.d.f. in the variable θ_1 . From the geometric interpretation given in Fig. 7.4 it is clear that, as far as the likelihood function is concerned, the maximum in the variable θ_1 is the same, independently of whether we integrated or not over θ_2 . However, once we include a generic non-flat prior probability $p^{(0)}(\theta)$, this nice geometric interpretation is lost and, in general, if $(\bar{\theta}_1, \bar{\theta}_2)$ is the maximum of the distribution function $p(\theta_1, \theta_2|s)$, it is no longer true that the $\bar{\theta}_1$ is the maximum of the reduced distribution function $\tilde{p}(\theta_1|s) = \int d\theta_2 \, p(\theta_1, \theta_2|s)$, obtained integrating out θ_2 . Thus, there is an ambiguity on the value of the most probable value of θ_1 , which depends on whether we are interested or not in θ_2 . Another possible drawback, this one common to both the maximum likelihood and the maximum posterior methods, is that we might want an estimator that minimizes the error on the parameter determination, and this in general is not the case for the maximum likelihood or maximum posterior probability estimators. These issues motivate the consideration of the Bayes estimator, in the next subsection.

Bayes estimator

In this case the most probable values of the parameters is defined by

$$\hat{\theta}_{\rm B}^i(s) \equiv \int d\theta \, \theta^i p(\theta|s) \,, \tag{7.71}$$

i.e. is the average with respect to the posterior probability distribution. The errors on the parameters is defined by the matrix

$$\Sigma_B^{ij} = \int d\theta [\theta^i - \hat{\theta}_B^i(s)] [\theta^j - \hat{\theta}_B^j(s)] p(\theta|s), \qquad (7.72)$$

that is, in terms of the mean square deviations from $\hat{\theta}_{\rm B}^i(s)$, where the average is taken again with respect to $p(\theta|s)$. Even when there is a non-trivial prior probability function, the Bayes estimator is clearly independent on whether we integrate out some variable from $p(\theta|s)$, since we anyhow integrate over all the θ^i when computing $\hat{\theta}_{\rm B}^i$ and Σ_B^{ij} . Furthermore, it can also be shown that, if one wants to minimize the error on the parameters, averaged over the whole parameter space, the Bayes estimator is the optimal one.

The "operational" meaning of the Bayes estimator is the following. Suppose that, after a sufficiently long run, we end up with a large ensemble of detected signals, which correspond to actual GWs, and that among them there is still a large subensemble of GW signals that produced a given output s(t). Each of these waves will be characterized by different values of the true parameters θ_t and therefore by a different $h(t;\theta_t)$ that, by combining each time with a different realization of the noise n(t), has produced the same output s(t). Then $\hat{\theta}_B^i(s)$ is the value of θ^i , averaged over this ensemble of signals, and Σ_B^{ij} is the corresponding rms error.

Thus, the Bayes estimator has a well-defined operational meaning, and welcome mathematical properties, such as the independence on whether we integrate out some variable and the fact that it minimizes the error on parameter estimation. Its main drawback is its computational cost, since the evaluation of eq. (7.71) or of eq. (7.72) involves a multi-dimensional integral over the space of θ variables which, as we have seen, could have a dimensionality of order 15 or larger, and furthermore at each point of this parameter space we must compute the function $p(\theta|s)$, given in eq. (7.65), which requires the numerical computation of the integral over frequencies that defines the scalar product (|). The choice of the best estimator is therefore subject to various considerations, including computational cost, and depends on the specific situation. The use of the Bayes estimator goes also under the name of non-linear filtering.

Of course, in the limit of large signal-to-noise ratio (which unfortunately is not expected to be the appropriate one for GW detectors, at

 24 We also mention that another way of understanding the meaning of the maximum likelihood procedure is in term of the Neyman–Pearson criterium, which consists in maximizing the probability of detection, subject to a given false alarm probability, and leads again to the condition that $\log \Lambda$ be maximum.

²⁵In the GW literature, the log of the prior $p^0(\theta)$ is sometime added to the exponential in eq. (7.65), and the resulting exponential is called again the log-likelihood function, $\log \Lambda$, so the corresponding estimator is called again the maximum likelihood estimator. This notation is however potentially confusing. For instance, one might be tempted to make a frequentist use of such a " $\log \Lambda$ ", which is obviously incorrect, since it involves a prior probability.

least in the near future) these issues becomes irrelevant, and all consistent estimators give the same answer. In this limit, there is also a very simple expression for the error on the parameters. If the SNR is large, the error that we make on the parameter estimation is small. For simplicity we assume that the prior $p^{(0)}(\theta)$ is nearly uniform near $\theta = \hat{\theta}$ where $\hat{\theta}$ is the value produced by (any) consistent estimator, say for definiteness the maximum likelihood estimator $\theta_{\rm ML}$. That is, we assume that the prior information is irrelevant for reconstructing the parameters. In eq. (7.65) we can then write $\theta^i = \hat{\theta}_{ML}^i + \Delta \theta^i$, and, since $\Delta \theta^i$ is small, we can expand the exponential in eq. (7.65) in powers of $\Delta \theta^i$ The linear term of the expansion vanishes because $\hat{\theta}_{\text{ML}}^{i}$ is, by definition, the maximum of the distribution, and to quadratic order in $\Delta\theta$ we get

$$p(\theta|s) = \mathcal{N} \exp\left\{-\frac{1}{2}\Gamma_{ij}\Delta\theta^i\Delta\theta^j\right\},$$
 (7.73)

where $\Gamma_{ij} = (\partial_i \partial_j h | h - s) + (\partial_i h | \partial_j h)$. Observe that, in the first term we have h-s=-n and, in the limit of large signal-to-noise ratio, |n| is much smaller than |h|. So in this limit the first term can be neglected. and we get

$$\Gamma_{ij} = (\partial_i h | \partial_j h), \qquad (7.74)$$

evaluated at $\theta = \hat{\theta}_{\text{ML}}$. This is called the Fisher information matrix. Then the expectation value of the errors $\Delta \theta^i$ are given by

$$\langle \Delta \theta^i \Delta \theta^j \rangle = (\Gamma^{-1})^{ij} \,. \tag{7.75}$$

Matched filtering statistics

As we have discussed in the previous sections, a general data analysis strategy consists in performing matched filtering, applying many different templates $h(t;\theta)$ to the data. This will result in the generation of a list of "events" (in the sense of Note 2 on page 337), defined by the fact that the signal-to-noise ratio, in correspondence with some template. raises over a predetermined threshold. Applying the maximum likeli hood criterium (or the maximization of the posterior probability, if we want to include prior information), we can then get the most probable value of the parameters θ , under the hypothesis that a GW signal $h(t;\theta)$ was present. The issue that we want to address now is the following How well such hypothesis performed? In other words, what is the statistical significance of the fact that we found events at a given level of signal-to-noise ratio?

The answer to this question depends crucially on the statistical properties of the noise so, first of all, it is important to realize that in any detector we can distinguish between two kinds of noise: "well-behaved" Gaussian noise, whose probability distribution is a Gaussian, and non-Gaussian noise, which is a generic denomination for anything else. A Gaussian distribution $\sim e^{-x^2/2}$ drops very fast for large values of its argument x. The intuitive idea, that we will formalize below, is therefore

to eliminate Gaussian noise by setting a sufficiently large threshold for the signal-to-noise ratio. Non-Gaussian disturbances, however, have in general a totally different statistical distribution, characterized by long tails at large values of S/N, which decay only as a power law.²⁶

These noises cannot be eliminated just by setting a high threshold. since they can produce events with values of S/N that, in Gaussian distribution, would be inconceivably large. As a limiting case, any detector shaken by an earthquake will produce "events" with arbitrarily high values of S/N. Of course, these events cannot be eliminated just by setting a high threshold in S/N. Rather, they should be identified and vetoed. All detectors are equipped with sensors which monitor various aspects of the detector performance as well as environmental conditions (e.g. seismometers), so that non-Gaussian disturbances are vetoed as much as possible. However, it is practically impossible to be sure that one has identified and vetoed all possible non-Gaussian disturbances. So, while in principle one can study experimentally the noise distribution and then set a threshold so high that even non-Gaussian fluctuations would be very rare, in practice this is not possible because the resulting threshold would be much too high, and therefore would considerably degrade the sensitivity of the detector. Rather, the best way of eliminating non-Gaussian noise is to perform coincidences between two or more detectors. This is among the reasons why various different detectors have been built, and they are operated as a network.

In the following, we first discuss the statistical significance of obtaining a given value of the signal-to-noise ratio S/N, assuming that only Gaussian noise is present. This will tell us how to fix the threshold in S/N so that, at some confidence level, we know that higher values of S/N have not been produced by Gaussian noise alone, and allows us to generate, from the data stream of the detector, a list of "events". These events will then be subject to further scrutiny, using for instance coincidences between detectors whenever possible, with the aim of eliminating those which are due to non-Gaussian noise, and retaining the GW sighals, if any. For the rest of this section we will be concerned only with Gaussian noise, while coincidences and other techniques will be discussed when we examine the various type of signals, in Sections 7.5–7.8.

In eqs. (7.42)–(7.45) we defined the signal-to-noise ratio in terms of the expectation value of the signal. Here however we want to study the full statistical distribution, rather than just its expectation value, so we define

$$\rho = \frac{\hat{s}}{N} \,, \tag{7.76}$$

where \hat{s} is the filtered output defined in eq. (7.41) and N is given in eq. (7.43), that is N is the root-mean-square (rms) of \hat{s} when the signal is absent. The definition of ρ is therefore analogous to the definition of the signal-to-noise ratio S/N, see eqs. (7.42)–(7.45), except that in the numerator we have \hat{s} rather than its expectation value $\langle \hat{s} \rangle$. As a result,

 26 For instance, a large class of phenomena, characterized by what is called self-organized criticality, are such that the number N of events that release an energy E is distributed as dN = $E^{-\gamma}dE$ where, quite remarkably, the exponent γ has approximately the same universal value, $\gamma \simeq 1.6$, in phenomena apparently very different. Such a law. together with the value $\gamma \simeq 1.6$, is in fact observed in earthquakes from different seismic faults (in which case it is called the Gutenberg-Richter law). in soft γ -ray bursts from highly magnetized neutron stars, as well as in numerical simulations of fractures in solids. The same distribution is experimentally observed when searching for short bursts in resonant-bar GW detectors, where they are likely due to microfractures inside the bar, and give an example of the non-Gaussian noise that we will have to fight. See Dubath Foffa, Gasparini, Maggiore and Sturani (2005), and references therein.

0.04

0.03

0.02

0.01

40

Fig. 7.5 The probability distribu-

tion $P(R|\bar{R})$, as a function of R, for

 $\bar{R}=20$ (solid line), $\bar{R}=30$ (dotted

line) and $\bar{R} = 50$ (dot-dashed line).

the relation between ρ and S/N is $S/N = \langle \rho \rangle$. From

$$\hat{s} = \int_{-\infty}^{\infty} dt \left[h(t) + n(t) \right] K(t) \tag{7.77}$$

we see that, when h is absent, ρ is a random variable with zero average and, since it has been normalized to its own rms, with variance equal to one. Thus, in the absence of a GW signal, the probability distribution of ρ is

$$p(\rho|h=0)d\rho = \frac{1}{\sqrt{2\pi}} e^{-\rho^2/2} d\rho.$$
 (7.78)

In contrast, if in the output there is a GW signal h with a signal-to-noise ratio $\bar{\rho}$, eqs. (7.76) and (7.77) give $\rho = \bar{\rho} + \hat{n}/N$, where $\hat{n} = \int dt \, n(t) K(t)$ Since N is just the rms of \hat{n} , in this case $\rho - \bar{\rho}$ is a Gaussian variable with zero mean and unit variance, so

$$p(\rho|\bar{\rho})d\rho = \frac{1}{\sqrt{2\pi}} e^{-(\rho-\bar{\rho})^2/2} d\rho.$$
 (7.79)

The variable ρ is the signal-to-noise ratio in amplitude. It is useful to introduce also $R \equiv \rho^2$, which is the signal-to-noise ratio in energy, since the energy of GWs is quadratic in the GW amplitude. Observe that ρ , being proportional to h(t), is not positive definite, and runs between $-\infty$ and $+\infty$, while of course $0 \le R < \infty$. The probability distribution for R, when there is in the output a GW signal with a signal-to-noise ratio in energy $\bar{R} = \bar{\rho}^2$, follows from eq. (7.79) observing that a single value R is obtained from two values of the amplitude, $\rho = \pm \sqrt{R}$, so the probability of detecting an event with SNR in energy between R and R + dR, when the SNR of the GW signal is \bar{R} , is given by

$$P(R|\bar{R})dR = p(\rho|\bar{\rho})d\rho + p(-\rho|\bar{\rho})d\rho, \qquad (7.80)$$

evaluated at $\rho = R^{1/2}$. Writing $d\rho = dR/(2R^{1/2})$, we get

$$P(R|\bar{R})dR = \frac{dR}{2\sqrt{R}} \frac{1}{\sqrt{2\pi}} \left[e^{-(\rho - \bar{\rho})^2/2} + e^{-(\rho + \bar{\rho})^2/2} \right]$$
$$= \frac{1}{\sqrt{2\pi R}} e^{-(\bar{R} + R)/2} \cosh\left[\sqrt{R\bar{R}}\right] dR. \tag{7.81}$$

From this we can compute the average value of R for a given \bar{R} ,

$$\langle R \rangle = \int_0^\infty dR \, R \, P(R|\bar{R}) = 1 + \bar{R} \,. \tag{7.82}$$

If we write $R = E/kT_n$, where T_n has the physical meaning of an effective temperature of the noise after matched filtering, we can also rewrite eq. (7.82) as

$$\langle E \rangle = kT_n + \bar{E} \,. \tag{7.83}$$

Therefore the average value of the detected energy is the sum of the energy \bar{E} deposited in the detector by the GW, plus the energy kT_n

associated to the detector noise, a very natural result. In Fig. 7.5 we show the form of the probability distribution $P(R|\bar{R})$, as a function of R, for different values of \bar{R} . Observe that, while the average value is at $R=1+\bar{R}$, the maximum of the distribution is at a somewhat lower value. The corresponding distribution for R in the absence of signal is obtained setting $\bar{R}=0$ in eq. (7.81). In Fig. 7.6 we compare the probability distribution $P(R|\bar{R})$ when R=10 with the probability distribution in the absence of signal, $P(R|\bar{R}=0)$.

The different behavior of the two distributions suggest that, when searching for a signal with a signal-to-noise ratio R in energy, we can discriminate a true GW signal from a fluctuation due to Gaussian noise setting a threshold in R, at a value R_t that eliminates most of the noise, while retaining a large fraction of the signal distribution. Observe that anyway there will always be a false alarm probability, given by

$$p_{\text{FA}} = \int_{R_t}^{\infty} dR \, P(R|\bar{R} = 0)$$

$$= 2 \int_{\rho_t}^{\infty} d\rho \, e^{-\rho^2/2}$$

$$= 2 \operatorname{erfc} \left(\rho_t / \sqrt{2}\right), \tag{7.84}$$

where erfc (z) is the complementary error function. Furthermore, there is a false dismissal probability, i.e. a probability of losing a real GW signal, given by²⁷

$$p_{\rm FD} = \int_0^{R_t} dR \, P(R|\bar{R}) \,.$$
 (7.85)

The threshold R_t can be fixed deciding what is the maximum false alarm level that we are willing to tolerate. This depends also crucially on the number of trials that we do with different templates. For example, for a coalescing binary, one can estimate that of order 10⁵ templates might be needed to cover with good accuracy the possible range of values of masses and spins. Furthermore, to match the template to the signal one can estimate in about 3 ms the maximum temporal mismatch between the two. In one year of data ($\sim 3 \times 10^7$ s), one must therefore try $\sim 10^{10}$ starting values of time, and for each value of time we have 10^5 templates to cover the masses and spin parameters, so overall one might have to try 10¹⁵ templates. 28 Often the false alarm level is fixed so that the expected number of false alarms in a run will be of order a few. With a lower threshold one would be flooded by spurious events, while higher threshold have of course the effect of increasing the false dismissal probability. The few events obtained will then be subject to further scrutiny. Thus, if we search for a coalescence in a single detector, with one year of data and 10¹⁵ templates, we could chose a threshold in amplitude $\rho_t \simeq 8$, since this gives $p_{FA} \sim 2.5 \times 10^{-15}$. However, performing coincidences between two detectors the probability of obtaining a false alarm simultaneously in the two detectors is the square of the single-detector probability, if the noise in the two detectors are uncorrelated, so in this case we might want $[2\operatorname{erfc}(\rho_t/\sqrt{2})]^2 \simeq 10^{-15}$, which gives $\rho_t \simeq 5.5$.

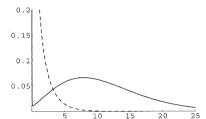


Fig. 7.6 The probability distribution $P(R|\bar{R})$, as a function of R, for $\bar{R} = 10$ (solid line), compared to the probability distribution in the absence of signal, $P(R|\bar{R}=0)$ (dashed).

²⁷In other words, whatever the value of \bar{R} , the distribution $P(R|\bar{R})$ is such that there is always some probability that R be smaller, and even much smaller, than \bar{R} , and therefore the GW can go undetected even when the threshold R_t was smaller than the value \bar{R} due to the GW alone. If one thinks in terms of energies, it might be counterintuitive that the energy released inside the detector can be smaller than the value that would have been released by the GW alone, in the absence of noise. Recall however that GW detectors really measure an oscillation amplitude, and the amplitude induced by the GW combines with the amplitude induced by noise with a relative phase, so noise and signal can interfere constructively or destructively. In the latter case the overall output has a smaller energy than that due to the GW signal alone

²⁸We will see however in Section 7.7.1 that all these time shifts can be taken into account simultaneously performing a single Fast Fourier Transform, which makes the problem computationally feasible.

 $^{29} \text{We}$ reserve the capital letter P for the distribution in energy, i.e. in R, and use p for the distribution in amplitude, i.e. in $\rho.$

In the above discussion, we assumed that the output of the detector is a single quantity ρ which, in the absence of noise, has a Gaussian distribution. Actually, we will meet below examples in which we have two outputs x, y, each one with its Gaussian noise, which are combined in quadrature, so that $\rho^2 = x^2 + y^2$. In this case the corresponding distribution function can be computed as follows. For the distribution $p_2(\rho|h=0)$ in the absence on signal (where the label 2 reminds us that we have two degrees of freedom x, y), we simply have

$$p_2(x,y|h=0)dxdy = \frac{dx}{(2\pi)^{1/2}} \frac{dy}{(2\pi)^{1/2}} e^{-(x^2+y^2)/2}$$
$$= \rho d\rho \frac{d\theta}{2\pi} e^{-\rho^2/2}. \tag{7.86}$$

If we are not interested in the phase θ we simply integrate over it, and we get

$$p_2(\rho|h=0) = \rho e^{-\rho^2/2},$$
 (7.87)

which is called a Rayleigh distribution, or a χ^2 distribution with two degrees of freedom. To compute the distribution in presence of signal, we start from the probability distribution of x,y, given that the true GW signal has the values \bar{x},\bar{y}

$$p_2(x,y|\bar{x},\bar{y})dxdy = \frac{1}{2\pi} e^{-\frac{1}{2}[(x-\bar{x})^2 + (y-\bar{y})^2]}.$$
 (7.88)

We pass to polar coordinates, $x = \rho \cos \theta$, $y = \rho \sin \theta$, with $\rho^2 = R$, so $dxdy = \rho d\rho d\theta = (1/2)dRd\theta$. To obtain the probability distribution $P_2(R|\bar{R})$ we integrate over the phase θ , and we also integrate over all the values of \bar{x}, \bar{y} with the constraint $\bar{x}^2 + \bar{y}^2 = \bar{R}$, that is,

$$P_{2}(R|\bar{R})dR = c \frac{dR}{2} \int_{0}^{2\pi} d\theta \int_{-\infty}^{\infty} d\bar{x} d\bar{y} \, \delta(\bar{x}^{2} + \bar{y}^{2} - \bar{R})$$

$$\times \frac{1}{2\pi} \exp\{-\frac{1}{2}[(x - \bar{x})^{2} + (y - \bar{y})^{2}]\}, \qquad (7.89)$$

where c is a normalization constant. The integrals are easily performed expressing also \bar{x}, \bar{y} in polar coordinates, $\bar{x} = r \cos \theta', \bar{y} = r \sin \theta'$, so

$$P_{2}(R|\bar{R}) = \text{const.} \int_{0}^{2\pi} d\theta \int_{0}^{2\pi} d\theta' \int_{0}^{\infty} d(r^{2}) \delta(r^{2} - \bar{R})$$

$$\times \exp\{-\frac{1}{2}(R + \bar{R}) + \sqrt{R\bar{R}} \cos(\theta - \theta')\}$$

$$= \text{const'.} e^{-(R + \bar{R})/2} \int_{0}^{2\pi} d\alpha \, e^{\sqrt{R\bar{R}} \cos\alpha}, \qquad (7.90)$$

where $\alpha = \theta - \theta'$. The integral over α gives a modified Bessel function I_0 . We fix the normalization constant requiring that $\int_0^\infty P_2(R|\bar{R})dR = 1$, and we get

$$P_2(R|\bar{R}) = \frac{1}{2}e^{-(R+\bar{R})/2}I_0\left(\sqrt{R\bar{R}}\right).$$
 (7.91)

More generally, if $\rho^2 = x_1^2 + \dots x_n^2$, performing a computation similar to that presented above one finds³⁰

$$P_n(R|\bar{R}) = \frac{1}{2} \left(\frac{R}{\bar{R}}\right)^{(n-2)/4} e^{-(R+\bar{R})/2} I_{\frac{n}{2}-1} \left(\sqrt{R\bar{R}}\right). \tag{7.92}$$

In Fig. 7.7 we show the function $P(R|\bar{R})$ given in eq. (7.81), which is appropriate for the case of a single degree of freedom, together with the functions $P_n(R|\bar{R})$ for n=2 and n=10 degrees of freedom, as obtained from eq. (7.92). These distribution functions are known as the noncentral chi-squared densities with n-degrees of freedom. The average value of R with n degrees of freedom is

$$\langle R \rangle = \int_0^\infty dR \, R \, P_n(R|\bar{R}) = n + \bar{R} \,, \tag{7.93}$$

and therefore

$$\langle E \rangle = n(kT_n) + \bar{E} \,, \tag{7.94}$$

while the variance is given by

$$\langle R^2 \rangle - \langle R \rangle^2 = 2n + 4\bar{R} \,. \tag{7.95}$$

7.5 Bursts

We now begin to apply the general theory that we have developed, to specific classes of GW signals. We begin with GW bursts. A number of astrophysical phenomena, like supernova explosions or the final merging of a neutron star–neutron star binary system, can liberate a large amount of energy in GWs in a very short time, typically less than a second, and sometimes as small as few milliseconds. We will refer to such signals as GW bursts, and we denote their duration by τ_g . In Fourier space, a GW burst therefore has a continuum spectrum of frequency over a broad range, up to a maximum frequency $f_{\text{max}} \sim 1/\tau_g$.

7.5.1 Optimal signal-to-noise ratio

In principle, if we know the form of $\tilde{h}(f)$, we can just plug it into eq. (7.51) to obtain the S/N for a given noise spectral density of the detector. However, bursts come from explosive and complicated phenomena, and it is very difficult to predict accurately their waveform. We can first of all make some simple order-of-magnitude estimates, distinguishing two cases.

Narrow-band detectors

In this case the detector is sensitive only to frequencies in a bandwidth Δf , centered around a frequency f_0 , and we assume that Δf is small with respect to the typical variation scale of the signal in frequency space.

³⁰See, e.g. McDonough and Whalen (1995), Sections 4.8 and 4.9.

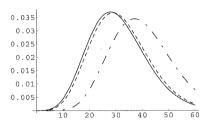


Fig. 7.7 The probability distribution $P(R|\bar{R})$ given in eq. (7.81) (solid line) compared to $P_n(R|\bar{R})$ with n=2 (dashed line) and with n=10 (dot-dashed), as a function of R, for $\bar{R}=30$.

³¹As we will see in Chapter 8, the bandwidth of resonant-mass detectors has subsequently evolved, reaching values of order $\Delta f/f_0 \sim 0.1$, but still, in a first approximation, eq. (7.96) applies. We will also see that, for resonant-mass detectors, $S_n(f)$ is not at all a slowly varying function of f around the resonance frequency f_0 , so in the estimate (7.96) we must really take an average of $1/S_n(f)$ over the whole useful bandwidth Δf , and we cannot simply use $1/S_n(f_0)$.

Outside this interval, the detector is blind and $1/S_n(f)$ in eq. (7.51) becomes practically zero. Inside this small bandwidth $\tilde{h}(f)$ cannot change much, so our ignorance of the precise waveform becomes irrelevant, and in the integrand in eq. (7.51) we can approximate $\tilde{h}(f)$ with $\tilde{h}(f_0)$. Then eq. (7.51) becomes

$$\left(\frac{S}{N}\right)^2 \simeq 4|\tilde{h}(f_0)|^2 \frac{\Delta f}{S_n} \,, \tag{7.96}$$

where $1/S_n$ is an average value of $1/S_n(f)$ in a bandwidth Δf centered on f_0 . This was the typical situation of resonant mass detectors until the 1990s, when the bandwidth Δf was only of order a few Hz, around a frequency $f_0 \sim 1$ kHz.³¹

Broad-band detectors

In this case we get the signal in a bandwidth (f_{\min}, f_{\max}) where f_{\max} is the maximum frequency contained in the burst, if the detector is sensitive up to f_{\max} , or otherwise is the maximum frequency to which the detector is sensitive. The detailed form of the signal is therefore important, but a first order-of-magnitude estimate can still be obtained writing eq. (7.51) as

$$\left(\frac{S}{N}\right)^2 \sim 4|\tilde{h}|^2 \frac{f_{\text{max}}}{S_n} \,, \tag{7.97}$$

where \tilde{h} is a characteristic value of $\tilde{h}(f)$ over the detector bandwidth and S_n is a characteristic value of $S_n(f)$.

We can translate these order-of-magnitude estimates into limits on the value of the dimensionless GW amplitude h(t) that can be measured. For this we assume for definiteness that the wave comes from a direction such that $F_+ = 1$ and $F_\times = 0$, so that h(t) is the same as the amplitude $h_+(t)$ of the + polarization. In the most general situation, we will also have a factor which depends on F_+ and F_\times and reflects the sensitivity of the detector to the given direction and polarization of the wave. To express eq. (7.51) in terms of h(t) we need a model for the signal. For a GW burst of amplitude h_0 and duration τ_q , a crude choice could be

$$h(t) = h_0 \quad \text{if } |t| < \tau_g/2$$
 (7.98)

and h(t) = 0 if $|t| > \tau_g/2$. We can write it more compactly as

$$h(t) = h_0 \tau_g \, \delta_{\text{reg}}(t) \,, \tag{7.99}$$

where $\delta_{\rm reg}(t)$ has a rectangular shape of unit area, $\delta_{\rm reg}(t) = 1/\tau_g$ for $|t| < \tau_g/2$ and $\delta_{\rm reg}(t) = 0$ for $|t| > \tau_g/2$. For $\tau_g \to 0$, $\delta_{\rm reg}(t)$ becomes a Dirac delta. More generally, for a burst we can model h(t) as in eq. (7.99), choosing for $\delta_{\rm reg}(t)$ a smooth function of unit area which goes to zero rather fast for $|t| \gtrsim \tau_g$. Performing the Fourier transform this gives

$$|\tilde{h}(f)| \sim h_0 \tau_q \,, \tag{7.100}$$

times a dimensionless function of the frequency, numerically of order one, and whose details depend on the precise waveform $\delta_{\text{reg}}(t)$ chosen.

Actually, rather than using a function $\delta_{\text{reg}}(t)$ with a unit area, it can be more convenient to write $h(t) = h_0 g(t)$, with g(t) some function peaked at t = 0 and with g(0) = O(1), so that the value of h(t) near the peak is of order h_0 (rather than $h_0 \delta_{\text{reg}}(0)$ as in eq. (7.99)). A simple waveform of this type is a Gaussian,

$$h(t) = h_0 e^{-t^2/\tau_g^2}, (7.101)$$

whose Fourier transform is

$$\tilde{h}(f) = h_0 \tau_g \sqrt{\pi} e^{-(\pi f \tau_g)^2} . {(7.102)}$$

A waveform with a somewhat more realistic shape is a sine-Gaussian, i.e. a Gaussian modulated by a frequency f_0 ,

$$h(t) = h_0 \sin(2\pi f_0 t) e^{-t^2/\tau_g^2}, \qquad (7.103)$$

shown in Fig. 7.8. Its Fourier transform is

$$\tilde{h}(f) = h_0 \tau_g \, i \frac{\sqrt{\pi}}{2} \left[e^{-\pi^2 (f - f_0)^2 \tau_g^2} - e^{-\pi^2 (f + f_0)^2 \tau_g^2} \right], \tag{7.104}$$

and is shown in Fig. 7.9. If $4\pi^2 f_0^2 \tau_g^2 \gg 1$, near $f = f_0$ the second term in brackets is negligible with respect to the first (while close to f = 0 it cancels the first term so that $\tilde{h}(0) = 0$), and we basically have a Gaussian in frequency space, centered at $f = f_0$, and with a value at the maximum

$$|\tilde{h}(f_0)| \simeq h_0 \tau_g \frac{\sqrt{\pi}}{2} \,. \tag{7.105}$$

Writing $f = f_0 + \Delta f$ we see that the width of the maximum Δf is of order $1/(\pi \tau_g)$, so $\Delta f/f_0 \sim 1/(\pi f_0 \tau_g)$. For $\pi f_0 \tau_g \ll 1$, $\tilde{h}(f)$ becomes relatively flat while for $\pi f_0 \tau_g \gg 1$ it is sharply peaked around f_0 . Using eqs. (7.96) and (7.97) we can estimate the minimum value of the dimensionless GW amplitude h_0 that can be detected at a given level of the signal-to-noise ratio S/N. For narrow-band detectors eq. (7.96) gives, using for definiteness the value $|\tilde{h}(f_0)| \simeq h_0 \tau_g (\sqrt{\pi}/2)$ appropriate for a sine-Gaussian waveform.

$$(h_0)_{\min} \sim \frac{1}{\tau_g} \left(\frac{S_n}{\pi \Delta f} \right)^{1/2} (S/N),$$
 (7.106)

while for broad-band detectors eq. (7.97) gives

$$(h_0)_{\min} \sim \frac{1}{\tau_g} \left(\frac{S_n}{\pi f_{\max}}\right)^{1/2} (S/N).$$
 (7.107)

The precise numerical factors, of course, depend on the choice of the waveform, so to fix the numerical coefficients in eqs. (7.106) and (7.107)

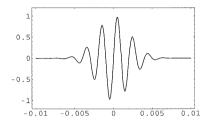


Fig. 7.8 The sine-Gaussian function $\sin(2\pi f_0 t)e^{-t^2/\tau_g^2}$, for $\tau_g = 3$ ms and $f_0 = 500$ Hz, as a function of t (in seconds).

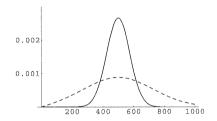


Fig. 7.9 $|\tilde{h}(f)|$ (in units of h_0) for a sine-Gaussian function with $f_0 = 500 \text{ Hz}$, $\tau_g = 3 \text{ ms}$ (solid line) and for a sine-Gaussian function with $f_0 = 500 \text{ Hz}$, $\tau_g = 1 \text{ ms}$ (dashed line).

we must know the shape of the signal $\tilde{h}(f)$, use the exact form of the noise $S_n(f)$, and perform the integral in eq. (7.51). We see from eqs. (7.106) and (7.107) that in a narrow-band detector the minimum detectable amplitude is higher by a factor $(f_{\text{max}}/\Delta f)^{1/2}$, compared to a detector which is able to maintain the same typical sensitivity S_n over a broad bandwidth. This reflects the fact that the narrow-band detector has access only to a portion of the Fourier modes of the burst.

Rather than expressing the result in terms of h_0 , it is also common to use the so-called root-sum-square (rss) amplitude h_{rss} , defined by

$$h_{\rm rss}^2 = \int_{-\infty}^{\infty} dt \, h^2(t)$$
$$= \int_{-\infty}^{\infty} df \, |\tilde{h}(f)|^2 \,. \tag{7.108}$$

For the Gaussian (7.101) we have

$$h_{\rm rss}^2 = h_0^2 \tau_g \sqrt{\frac{\pi}{2}}, \qquad \text{(Gaussian)}, \qquad (7.109)$$

while, for the sine-Gaussian (7.103),

$$h_{\rm rss}^2 = h_0^2 \tau_g \sqrt{\frac{\pi}{8}} \left(1 - e^{-2\pi^2 f_0^2 \tau_g^2} \right),$$
 (sine-Gaussian). (7.110)

Observe that, dimensionally, $h_{\rm rss} \sim ({\rm time})^{1/2}$, so $h_{\rm rss}$ is conventionally quoted in ${\rm Hz}^{-1/2}$, as the strain sensitivity.

To have an idea of the numerical values of $h_{\rm rss}$ (or, equivalently, of h_0) that could be obtained from astrophysical phenomena, we can compute the energy released in GWs by an event which produced, at the detector, a given value of $h_{\rm rss}$. This can be obtained from the expression for dE/dAdf given in eq. (1.159). Observe however that, for a wave coming from an arbitrary direction and with arbitrary polarization, a detector does not measure directly $\tilde{h}_+(f)$ and $\tilde{h}_\times(f)$ but rather the combination $\tilde{h}(f) = F_+\tilde{h}_+(f) + F_\times\tilde{h}_\times(f)$, where $F_{+,\times}$ are the detector pattern functions. For definiteness, we consider a GW coming from the optimal direction for the + polarization, so we take $F_+ = 1$ and $F_\times = 0$, and for $\tilde{h}_+(f)$ we take the sine-Gaussian waveform (7.104). We substitute this into eq. (1.159) and we get the total energy $\Delta E_{\rm rad}$ radiated by the source in GWs.³²

$$\Delta E_{\rm rad} \simeq \left(\frac{\pi}{2}\right)^{3/2} \frac{\pi r^2 c^3}{G} h_0^2 \tau_g f_0^2$$

$$= \frac{\pi^2 r^2 c^3}{G} h_{\rm rss}^2 f_0^2. \tag{7.111}$$

Inserting the numerical values.

$$\Delta E_{\rm rad} \simeq 1 \times 10^{-2} M_{\odot} c^2 \left(\frac{r}{8 \,\text{kpc}}\right)^2 \left(\frac{h_{\rm rss}}{10^{-19} \,\text{Hz}^{-1/2}}\right)^2 \left(\frac{f_0}{1 \,\text{kHz}}\right)^2,$$
(7.112)

where in the second line we normalized r to a value of order of the distance to the galactic center. Recall that in the above we assumed a wave coming from optimal direction. For an ensemble of waves with arbitrary direction and polarization, we must also take into account the average over the pattern functions of the detector. For an interferometer, this is a factor 2/5 (see Table 7.1), so on average a burst coming from arbitrary direction, in order to produce a given signal $h_{\rm rss}$ in the detector, had an energy larger by a factor 5/2, compared to eq. (7.112). We see that a burst at the kHz, with $h_{\rm rss} = 10^{-19}\,{\rm Hz}^{-1/2}$, carries away about 10^{-2} solar masses in GWs, if it comes from a source located at typical galactic distances.

Taking $10^{-2}M_{\odot}c^2$ as a reference value for $\Delta E_{\rm rad}$ (which, as we will see in Vol. 2, is the maximum value that can be reasonably expected in cataclysmic events involving solar mass objects. Even larger energies can be released in the merging of very massive black holes), we see that a detector must reach at least a sensitivity to $h_{\rm rss}$ of order $10^{-19}\,{\rm Hz}^{-1/2}$ to have some chance of detecting GW bursts from the galactic center. To be able to see a burst which releases 10^{-2} solar masses in the Virgo cluster of galaxies, which is at $r \sim 14$ Mpc, one rather needs to be able to reach $h_{\rm rss} \simeq 6 \times 10^{-23}\,{\rm Hz}^{-1/2}$ or, from eq. (7.110) with $\tau_g = 1$ ms, a value of the dimensionless amplitude $h_0 \simeq 2 \times 10^{-21}$.

7.5.2 Time–frequency analysis

The matched filtering technique that we have discussed in Section 7.3 works well if we know the form of the signal, or if we can parametrize it with a limited number of free parameters, so that it becomes practically feasible to put a sufficiently fine grid in this parameter space, and repeat the search for each point of the grid. As we will discuss in the next sections, this can be the case for the inspiral of compact binaries and, partly, for the signals due to pulsars.

Concerning bursts, the situation is different. In general, bursts may come from complicated explosive phenomena, such as supernovae, or from processes such as the final merging of coalescing binaries, which are difficult to model. In a narrow-band detector, such as resonant bars. we only have access to a narrow range of Fourier components of the signal. Thus, in a first approximation it is reasonable to model the signal as flat in frequency, i.e. as a Dirac delta in time and, as a next step, we can use more realistic modelizations such as the Gaussian and the sine-Gaussian waveforms described above. However, in a broadband detector, the difference between these simple modelizations and the real waveform will become important. Thus, to exploit optimally the capabilities of a broad-band detector, one is lead to consider also other methods, which are sub-optimal with respect to matched filtering when the waveform is know precisely (since we have seen that, if the waveform is known, matched filtering is the optimal strategy) but might be more robust in the absence of detailed knowledge of the signal.

Such search algorithms can be obtained working in the time–frequency

 $^{32}\mathrm{In}$ the computation, we neglect the term $\exp\{-\pi^2(f+f_0)^2\tau_g^2\}$ in eq. (7.104), which is small with respect to $\exp\{-\pi^2(f-f_0)^2\tau_g^2\}$ and, when we integrate eq. (1.159) over df, we replace $f^2\exp\{-\pi^2(f-f_0)^2\tau_g^2\}$ with $f_0^2\exp\{-\pi^2(f-f_0)^2\tau_g^2\}$ and we extend the resulting integral from $-\infty$ to $+\infty$.

 33 The information about time localization, which is encoded in s(t), is of course still encoded in $\tilde{s}(f)$, since from $\tilde{s}(f)$ we can get back s(t) uniquely. However, it is lost in $|\tilde{s}(f)|^2$, since it was contained in the phase relation between the Fourier components, and this information is obliterated when taking the modulus. A nice example (taken from van den Berg (1999), a textbook on wavelets) is obtained if we take s(t)to be a classical symphony. Then its power spectrum will immediately reveal the dominating keys: the groundtones and their harmonics. Suppose now that we play the parts in a different order, and we even interchange smaller parts within the parts. The power spectrum would not change at all, while to the ear, which actually makes a timefrequency analysis, the result is very different.

 $^{34}\mathrm{To}$ fix the ideas, one can consider that the sampling rate $1/\Delta t$ of an interferometer is typically of order 10-20 kHz. We can imagine that we are searching for bursts of duration $\delta t=0.5$ s, so frequency space is split into bins of width 2 Hz. Restricting the search to the frequency range where interferometers are most sensitive, which corresponds to a bandwidth $\Delta f=O(200)$ Hz around peak sensitivity, we have a total of O(100) bins in frequency, for each value of the start time of the segment.

plane. To understand the usefulness of the time–frequency representation, suppose at first that we have a function s(t) defined on the whole real axis $-\infty < t < \infty$. We can take its Fourier transform $\tilde{s}(f)$ and compute from it the power spectrum $|\tilde{s}(f)|^2$. A plot of the power spectrum against the frequency will enable us to see immediately what are the dominant Fourier modes. However, this power spectrum knows nothing about when things happened.³³

The simplest way to recover partly this information is to take the Fourier transform not on the whole real line, but on segments of length δt . When we Fourier transform the function s(t) on the interval $0 < t < \delta t$ and we plot the resulting power spectrum, we find the Fourier modes that dominated the function, during this temporal span. We can then repeat it for $\delta t < t < 2\delta t$, etc. Of course, on a finite segment of length δt , the resolution in frequency is finite, and is $1/\delta t$, so we are giving up the fine details in frequency space, but we gain an understanding of when things happened. That is, rather than working in frequency space, with an arbitrarily good resolution, it can be convenient to work in the time–frequency plane, making a good compromise between the accuracy in frequency and the accuracy in time.

This is particularly important when we are looking for transient phenomena, such as GW bursts. Suppose that we are unable to compute the detailed waveform of a burst, as it is typically the case, but still we can give a reasonable estimate of its total duration δt , and of the frequency range $f_1 < f < f_2$ where most of its power should be concentrated. Then, a useful search strategy is as follows.

First of all, it is convenient to work in a discretize space. Recall that the output of a detector is sampled at some rate $1/\Delta t$. Then, we can split the output into time segments, and inside each segment the output s(t) is given by the discrete set of values

$$s_j \equiv s(t_{\text{start}} + j\Delta t),$$
 (7.113)

where t_{start} is the start time of the segment considered, j = 0, ..., N, and $\delta t = N\Delta t$ is its length. We can then perform a discrete Fourier transform over the segment δt by writing

$$\tilde{s}_k = \sum_{j=0}^{N-1} n_j \exp\left\{\frac{2\pi i}{N} jk\right\},$$
(7.114)

01

$$\tilde{s}_k = \sum_{j=0}^{N-1} n(t_j) \exp\{2\pi i (t_j - t_{\text{start}}) f_k\}.$$
 (7.115)

where $t_i = t_{\text{start}} + j\Delta t$ and

$$f_k = \frac{k}{N\Delta t} = \frac{k}{\delta t} \,. \tag{7.116}$$

We see that frequencies are spaced by $1/\delta t$, up to a maximum frequency equal to $N/\delta t$ (since eq. (7.114) is periodic under $k \to k + N$), which of course is just the sampling frequency $1/\Delta t$.³⁴ We can write as usual $s_i = n_i + h_i$, where n_i is the noise and h_i a putative signal, and we define

the Fourier transforms \tilde{n}_k and \tilde{h}_k as in eq. (7.114). The discrete version of eq. (7.6) is obtained replacing the Dirac delta by a Kronecker delta,

$$\langle \tilde{n}_k^* \tilde{n}_{k'} \rangle = \delta_{kk'} \frac{1}{2} S_k \,, \tag{7.117}$$

where we used the short-hand notation $S_k \equiv S_n(f_k)$.

If the only theoretical expectation that we have about a signal is that it should have a duration δt , and should have most of its power in a frequency band $f_1 < f < f_2$, with $f_1 = k_1/\delta t$, $f_2 = k_2/\delta t$, and $f_2 - f_1 \equiv \delta f$, we can form, for each possible start time t_{start} , the quantity

$$\mathcal{E} = 4 \sum_{k=k_1}^{k_2} \frac{|\tilde{s}_k|^2}{S_k} \,, \tag{7.118}$$

which is called the excess power statistic. We collect the values of \mathcal{E} for all possible start time and, if we find a value above some given threshold, we record it as an event.³⁵ To understand what is a statistically significant value of \mathcal{E} , observe that \mathcal{E} is formed from $k_2 - k_1$ independent complex variables s_k . Since $(k_2 - k_1)/\delta t = f_2 - f_1 \equiv \delta f$, the number of independent real variables is

$$\mathcal{N} = 2\delta f \delta t \,, \tag{7.119}$$

i.e. twice the area of the time-frequency plane explored. Therefore, even in the absence of any GW signal in the data, the average value of \mathcal{E} is of order \mathcal{N}^{36} . This means that a real GW signal, in order to be visible in \mathcal{E} against the noise with a signal-to-noise ratio of order one, must give a contribution to \mathcal{E} of order \mathcal{N} . From eq. (7.119), $\mathcal{N} \geq 2$ (the uncertainty principle, in a quantum language) and, depending on the situation, one can have $\mathcal{N} \gg 1$. Comparing with eq. (7.51), we see that, if we knew the waveform and we could make a matched filtering, such a signal would produce a value of S/N of order $\mathcal{N}^{1/2}$. In other words, using the excess power statistic, we can detect with a signalto-noise ratio of order one, a signal that with matched filtering would emerge with signal-to-noise ratio of order $\mathcal{N}^{1/2}$. This is not surprising since we know that, when we have the waveform, the matched filtering maximizes the signal-to-noise ratio. However, the excess power statistic only needs very crude information about the signal, namely its duration and its typical frequency range, and is therefore much more robust. Furthermore, it can be proved that, if the only information on the signal is its duration and its bandwith, the excess power method is the optimal

From the above discussion, it is clear that the method is viable only when $\mathcal N$ is not too large. For instance, for the inspiral phase of a coalescing NS-NS binary, as observed in a ground-based interferometer, we see from eq. (4.21) that the signal enters the bandwidth of the interferometer, say at 40 Hz, when the time to coalescence is $\tau=25$ s, and sweeps in frequency up to the kHz. Taking $\delta f \sim 1000$ Hz and $\delta t \sim 25$ s, we get $\mathcal N \sim 5 \times 10^4$ and $\mathcal N^{1/2} \sim 200$, so the excess power method would allow

³⁵In the sense defined in Note 2 on page 337.

³⁶More precisely, in the presence of Gaussian noise \mathcal{E} follows a χ^2 distribution with \mathcal{N} degrees of freedom, while in the presence of signal it follows the corresponding non-central χ^2 distribution, see eq. (7.92).

³⁷For a more accurate estimate of the signal-to-noise ratio obtained restricting the frequency bandwidth, i.e. performing a band-pass filter, see Section II of Flanagan and Hughes (1998a).

 $^{38} \text{For}$ fast spinning BHs the coalescence time will be longer, since they must first shed some angular momentum before setting into their final state, in order that the angular momentum of the final BH does not exceed the maximum value allowed for rotating BHs. The estimate of $\mathcal N$ depends on the angular momentum of the BHs, but a typical value can be $\mathcal N^{1/2}\sim 5$. See Flanagan and Hughes (1998a), Section IIIE.

us to detect signals only when their signal-to-noise ratio, with matched filtering, is of order several hundreds. Thus, for inspiraling binaries, the excess power method is not at all competitive. Furthermore it is not needed, since in this case we have precise calculations of the waveform in the inspiral phase, as we saw in Chapter 5.

The situation is different for the merging phase of a BH-BH coalescence. In this case the maximum value of f can be estimated to be of order f_{qnr} , where f_{qnr} is the ringing frequency of the fundamental quasi-normal mode of the black hole. To include the power radiated by the BH in its higher quasi-normal modes, an estimate of order $2f_{\rm out}$ could be more appropriate. Black hole normal modes will be discussed in Vol. 2, and we will see that f_{qnr} can reach a maximum value (for rapidly spinning BHs of mass m) $f_{\rm qnr} = c^3/(2\pi Gm)$. Observe that this is quite larger than the maximum frequency (4.39) at which the inspiral phase ends, so we finally take $\delta f \sim 2 f_{\rm qnr} \sim c^3/(\pi G m)$. As for the merging time, we can roughly estimate that it should not be much larger than $r_{\rm ISCO}/c$, where $r_{\rm ISCO} = 6Gm/c^2$ is the radius of the innermost stable circular orbit in a Schwarzschild geometry, see eq. (4.38). Taking for instance $\delta t \sim 2r_{\rm ISCO}/c = 12Gm/c^3$, we get the estimate $\mathcal{N}^{1/2} \sim 2$, so the loss in sensitivity with respect to optimal filtering is not large. This is quite important, considering that the merging phase is very difficult to model.³⁸

The time–frequency method discussed here can be generalized in various directions. One possibility is to consider wavelets. These are generalizations of the Fourier transform, in which to a function s(t) is associated a function $S(f,t_0)$ of two parameters, of which f is the frequency and t_0 is the position in time of the signal,

$$S(f, t_0) = \int_{-\infty}^{\infty} \psi_{f, t_0}^*(t) s(t). \tag{7.120}$$

The simplest example consists in taking

$$\psi_{f,t_0}(t) = e^{-ift} \psi(t - t_0) , \qquad (7.121)$$

where $\psi(t-t_0)$ is a window function centered around t_0 . This Windowed Fourier Transform, or Gabor transform, as it is called, is essentially what we have used above (more precisely, we used its discrete version), with a sharp window function. Other choices of window functions, such as a Gaussian, are more commonly used in signal analysis.

A possible drawback of a choice such as eq. (7.121) is that the temporal window has a fixed size, independently of the frequency. In most type of signals, however, there is a correlation between the characteristic frequency of a given segment and the time duration of the segment, such that low-frequency pieces tend to last longer. To take this into account, the wavelet transform is defined by choosing a window function of the form $\psi(f(t-t_0))$ (times a normalization constant \sqrt{f}), which depends explicitly on f.³⁹ In this way, at high frequencies the temporal window is shorter, so we have a better time resolution. In a sense, wavelets provide a "microscope" that, at each point in time of the signal, zooms in and

out, depending on the frequency scale of the signal. The other crucial property is that it is possible to choose wavelets so that they form an orthonormal basis, and the signal can therefore be decomposed uniquely into its component with respect to this wavelet basis, just as in the Fourier transform. Wavelets are by now widely used in signal analysis in many branches of science, and many possible choices of wavelets are available, depending on the problem at hand, see the Further Reading.

Another generalization of the time—frequency analysis discussed here consists in marking as "black" the bins in the time—frequency plane where an indicator such as the excess power statistic goes above a threshold value, and searching for structures of black bins, such as clusters. This is basically a variant of the Hough transform that we will discuss in Section 7.6.3, in the context of periodic signals.

7.5.3 Coincidences

Given that GW bursts can have a very short duration, even smaller than a millisecond, the output of ground-based detectors are sampled with a very high frequency, typically O(10) kHz. In one year there are about 3×10^{10} ms, so even a fluctuation with a probability $\sim 10^{-10}$ is bound to occur on average in one year worth of data. Then eq. (7.84) suggests, for bursts, a threshold on the amplitude signal-to-noise ratio of order $\rho_t \simeq 6$, in order to have just a few false alarms per year in a detector. However, this only eliminate Gaussian noise. GW bursts are particularly well simulated by non-Gaussian events such as microcreeps in the materials or sudden external mechanical or electromagnetic disturbances. In some cases the external disturbance can be identified, and the corresponding event is therefore vetoed, but in most cases this is impossible. To eliminate these non-Gaussian noise, the only possibility is to perform coincidences between different detectors.⁴⁰

Using two or more detectors in coincidence is a standard practice in physics, at least since the early days of cosmic ray research. The idea is that, if two detectors are far apart, their noise are mostly uncorrelated, and the probability of an accidental coincidence is small, while a GW should excite both detectors nearly simultaneously. Below we discuss some of the issues that must be addressed in order to apply this idea to GW detectors.

Relative orientation of the detectors

We have seen that the response of a detector to a GW depends on the relative orientation between the detector and the source. To perform coincidences between two or more detectors, it is therefore optimal to orient them, taking into account their difference in location, so that their response to an incoming GW signal is the same, or at least as similar as possible, for all of them. Otherwise, a real signal can be missed simply because, when one detector was oriented favorably with respect to the source, the other had a very poor sensitivity for the same direction. 42

³⁹In the literature on wavelets, this is actually written as $a^{-1/2}\psi((t-t_0)/a)$, where a is a dimensionless quantity that rescales a characteristic frequency implicit in the function ψ .

 $^{^{40}}$ Observe that the use of coincident detectors also allows us to lower the threshold necessary for eliminating Gaussian noise, since now the false alarm probability, for uncorrelated detector noise, is the square of (7.84). For instance, in the example above, the threshold $\rho_t \simeq 6$ valid for a single-detector search becomes $\rho_t \simeq 4.5$ in a two-detector correlation (even neglecting all consistency check discussed below)

⁴¹With some exceptions. For example, seismic or electromagnetic disturbances might propagate from one detector's site to the other.

⁴²On the other hand, detectors with different orientation can perform independent measurement of the signal, allowing to disentangle the polarizations and the arrival direction of the wave, see the discussion on page 342.

⁴³This depends not only on the sampling time of the detector, but also on other factors, in particular on the signal-to-noise ratio of the event, since noise combines with the GW signal distorting and broadening its shape.

Coincidence window

Each GW detector has its temporal resolution, which might for instance be of the order of few ms.⁴³ Given two detectors, with variances σ_1 and σ_2 on the arrival time of their respective events, the corresponding variance in the coincidence search is $\sigma_{12} = \sqrt{\sigma_1^2 + \sigma_2^2}$, and therefore one can ask that the events be coincident within k standard deviations σ_{12} (e.g. k=3 can be a typical choice). To this uncertainty one must add the light travel time $(\Delta t)_{\text{light}}$ between the two detectors since, depending on the source location, either the first or the second detector will be hit by the wavefront a time up to $(\Delta t)_{\text{light}}$ before the other. So finally one requires that the arrival times t_1 and t_2 in the two detectors are within a coincidence window

$$|t_1 - t_2| \le (\Delta t)_{\text{light}} + k \left(\sigma_1^2 + \sigma_2^2\right)^{1/2}$$
 (7.122)

This typically results in a coincidence window of the order of a few tens of ms.

Energy consistency

Another possible handle to discriminate between accidental coincidences and true GW signals is the compatibility of the signal in the two (or more) detectors. Ideally, if the GW signal is much larger than the noise. two detectors oriented in the same way should register the same energy flux, when a GW hits them. In contrast, two events due to noise which by chance happen simultaneously in the two detectors, should have uncorrelated energies. However, in practice, at moderate values of the signal-to-noise ratio the signal h(t) induced by the GW combines with the noise n(t) and, depending on the relative phase of these contributions, the output h(t) + n(t) fluctuates and can be either larger or smaller than the value that would be induced by the GW. Therefore, as we computed in Section 7.4.3, one has a probability distribution for the amplitudes (or for the energies) measured in the two detectors, and the compatibility criterion must take into account this probability distribution. This procedure also requires that the two detectors have a sufficiently reliable calibration in energy.

Waveform consistency

A broadband detector has rather detailed information on the waveform, and a consistency condition between the waveforms observed in the two detectors can be imposed. For instance, one of the algorithms used by LIGO for generating candidate events is based on the identification of connected regions ("clusters") in the time–frequency plane where the power is not consistent, statistically, with Gaussian noise, as discussed in Section 7.5.2. Then each event is characterized by its bandwidth (f_{\min}, f_{\max}) , i.e. by the low and high frequency bounds of the cluster. One can then require, for instance, that the bandwidth of events in

different detectors have an overlap, or at least that they are separated in frequency by no more than a fixed window Δf .

Background estimation

After having applied all these cuts, we can still have accidental coincidences that, by chance, passed them. However, the residual number of accidental coincidences can be estimated very reliably. First of all, one can simply predict it from the observed event rate in a single detector. assuming that the noise is stationary. But in fact the most direct estimation of the background is obtained using a shifting algorithm which. together with many other techniques used in GW research, was introduced by Weber. The procedure consists simply in shifting the data stream of one detector with respect to the other by a time step significantly longer than the coincidence window, say 2 s, and counting the number of coincidences obtained after shifting (subject to the same requirements on the coincidence window and energy compatibility imposed on the coincidences at zero time shift). These coincidences, of course, are now all accidental, since the shift has been chosen much larger than the coincidence window and therefore of the uncertainties in the arrival times. We then repeat the procedure with a different shift, say 4 s, and we count again the number of accidentals. One can repeat the procedure for many different shifts (the overall time shift must however be short compared to the time-scale over which the event rate in a single detector changes substantially). We then average over these shifts, and we have a rather accurate estimate of the average number of accidental coincidences, its variance, and more generally their distribution (which is found experimentally to be a Poisson distribution, as expected whenever we count a number of discrete independent events), and we can also study how these quantities depend on the energy of the events.

7.6 Periodic sources

While a burst source is typically radiating only for a period of less than a second, a periodic source emits continuously an almost monochromatic signal, so the limit on its observation comes from the total available observation time, which can be of order of years. Our intuitive discussion of matched filtering showed that, if we can follow a signal for a time T, the minimum level of signal that we can extract from the noise scales as $1/T^{1/2}$, see eqs. (7.39) and (7.40). This means that, for periodic waves, we can extract from the noise a signal with an amplitude h_0 much smaller than the one that can be measured in the case of bursts. This opportunity, however, also comes at the expense of some complications, since we must able to track carefully the signal for a long period. We already met a similar situation in Chapter 6, where we studied the timing formula for the radio signals of pulsars, and we saw that there are two main issue to address: the intrinsic changes of the frequency of the source, and the modulation of the signal due to the motion of the

⁴⁴For GWs, propagation effects between the source and the Earth, such as dispersion in the interstellar medium, are totally irrelevant, given the smallness of gravitational cross-sections.

Earth.⁴⁴

If, for a moment, we neglect these effects, a periodic source emitting GWs at a frequency f_0 produces in the detector a signal

$$h(t) = F_{+}(\theta, \phi) h_{+}(t) + F_{\times}(\theta, \phi) h_{\times}(t),$$
 (7.123)

where

$$h_{+}(t) = h_{0,+}\cos(2\pi f_0 t),$$
 (7.124)

$$h_{\times}(t) = h_{0,\times} \cos(2\pi f_0 t + \alpha)$$
. (7.125)

We take by definition $f_0 > 0$; $h_{0,A}$ are the real amplitudes for the two polarizations $(A = +, \times)$, and α is their relative phase. We denote by θ, ϕ the angles that define the propagation direction $\hat{\mathbf{n}}$ of the GW from the source to us, so the polar angles of the source, as seen from the Earth, are $\theta_s = \pi - \theta$ and $\phi_s = \phi + \pi$.

Assuming for the moment that the source is, intrinsically, perfectly periodic, still the motion of the Earth modifies eqs. (7.123)–(7.125) as follows.

- Because of the Earth's rotation, the apparent position of the source in the sky changes, so the angles θ and ϕ which appear in the pattern functions change with time, and are periodic functions of sidereal time, with period one sidereal day. If we are tracking a specific source in the sky, the time dependence of the pattern functions, $F_A(\theta(t), \phi(t))$, must therefore be taken into account, and this produces a modulation of the amplitude of the signal.
- Because of the Earth's rotation and of its revolution around the Sun (or, more precisely, because of its motion with respect to the Solar System Barycenter, as discussed in Chapter 6), the relative velocity of the Earth and the source changes with time, and this produces a time-varying Doppler shift in the frequency.

As a consequence, h(t) is not a simple monochromatic signal. We will come back to these amplitude and phase modulations in Sections 7.6.1 and 7.6.2. For the moment, however, we restrict to an observation time T sufficiently short, so that these amplitude and phase modulations can be neglected. For the amplitude modulation due to the Earth's rotation, this requires of course $T \ll 1$ day, while for the Doppler effect we will quantify this requirement in Section 7.6.2. In this limit h(t) becomes monochromatic, with a frequency f_0 .

In this simplified setting the form of the matched filter becomes obvious: we must limit ourselves to a bandwidth as small as possible around f_0 , since enlarging the bandwidth we accept more noise but we add no further signal. If T is the total observation time, our resolution in frequency is 1/T, see eq. (7.10), and therefore a bandwidth as small as possible means $\Delta f \simeq 1/T$. Formally, we can obtain the same result using eq. (7.49). From eqs. (7.123)–(7.125) we have, for f > 0,

$$\tilde{h}(f) = \delta(f - f_0) \frac{1}{2} \left[F_{+}(\theta, \phi) h_{0,+} + F_{\times}(\theta, \phi) h_{0,\times} e^{-i\alpha} \right], \qquad (7.126)$$

and therefore eq. (7.49) gives

$$\tilde{K}(f) = \delta(f - f_0), \qquad (7.127)$$

apart from an arbitrary constant, in which we also reabsorbed $1/S_n(f_0)$. Of course, the Dirac delta is a mathematical idealization, and if we measure for a total observation time T we must replace it by a regularized Dirac delta,

$$\delta(f) = \int_{-\infty}^{\infty} dt \ e^{i2\pi ft} \to \int_{-T/2}^{T/2} dt \ e^{i2\pi ft}, \qquad (7.128)$$

which has a support over a range $\Delta f \sim 1/T$ and satisfies $\delta(0) = T$. Then eq. (7.51) becomes

$$\left(\frac{S}{N}\right)^{2} = \left|F_{+}(\theta,\phi)h_{0,+} + F_{\times}(\theta,\phi)h_{0,\times}e^{-i\alpha}\right|^{2} \int_{0}^{\infty} df \, \frac{\delta(f-f_{0})\delta(0)}{S_{n}(f)}
= \left|F_{+}(\theta,\phi)h_{0,+} + F_{\times}(\theta,\phi)h_{0,\times}e^{-i\alpha}\right|^{2} \frac{T}{S_{n}(f_{0})}.$$
(7.129)

Not surprisingly, the signal-to-noise ratio increases if we increase the observation time, and the dependence $S/N \sim \sqrt{T}$ is what we already found using heuristic arguments in eqs. (7.39) and (7.40).

In general, the frequency f_0 is not known in advance. However, for an exactly periodic signal, we do not need to repeat the matched filtering procedure separately for each value of the unknown parameter f_0 . In fact, from eq. (7.42), when $\tilde{K}(f) = \delta(f - f_0)$ the signal is simply $S = \tilde{h}(f_0)$, and the values of $\tilde{h}(f)$ for all f can be computed at once performing a single Fast Fourier Transform (FFT), which is a particularly efficient algorithm.

If this were the end of the story, the search for periodic signals would simply consist in performing a single FFT on a stretch of data of length T, and looking for lines in the power spectrum. The signal-to-noise ratio of these line should improve with the observation time as \sqrt{T} . We will see in Section 7.6.1 and especially in Section 7.6.2 that the full story is more complicated.

7.6.1 Amplitude modulation

As we pointed out above, the pattern functions depend on time because of the Earth's rotation, and are therefore periodic functions of sidereal time, with a period of one sidereal day. In the matched filtering, we must take this into account, and this results in a different amplitude modulation for each possible source position. We will discuss in the next sections how to efficiently scan the parameter space, in order to take this effect into account.

If we want to estimate the effect of this modulation on the sensitivity, we can however simply observe that, for integration times T longer than one day, the effect of this amplitude modulation can be taken into

 45 If, rather than being interested in the sensitivity to a specific source, one wants to define an average sensitivity for an ensemble of sources, then one can improve this estimate taking care of the fact that there is a statistical preference for the angles and polarizations that give a larger S/N, since these can be seen to larger distances. This modifies S/N by factors that can be approximately estimated to be of order $(3/2)^{1/2} \simeq 1.2$, see Thorne (1987).

account averaging eq. (7.129) over the apparent motion of the source in one sidereal day, i.e. averaging over all values of the right ascension of the source, and over a range of values of the declination which depend on the specific orbit of the source. In a first approximation, we can replace this average with an average over the solid angle and over the polarization angle ψ .⁴⁵ From eq. (7.129), using eqs. (7.33) and (7.35), we then find

$$\left(\frac{S}{N}\right)^2 = \langle F_+^2 \rangle \left(\frac{T}{S_n(f_0)}\right) h_0^2,$$
(7.130)

where

$$h_0^2 = h_{0,+}^2 + h_{0,\times}^2 \,. \tag{7.131}$$

The values of $\langle F_+^2 \rangle$ for various detectors are given in Table 7.1, recalling that $\langle F_+^2 \rangle = F/2$. We can also rewrite eq. (7.130) as

$$\frac{S}{N} = \frac{h_0}{h_n},\tag{7.132}$$

defining the dimensionless quantity h_n ,

$$h_n = \frac{1}{\langle F_+^2 \rangle^{1/2}} \left(\frac{S_n(f_0)}{T} \right)^{1/2} . \tag{7.133}$$

Therefore h_n is the GW amplitude that can be measured by the detector, for a periodic signal, at S/N = 1 (assuming that we have been able to correct for the phase modulation, see next section). More generally, the minimum amplitude that can be detected at a given value of S/N is

$$(h_0)_{\min} = \frac{S/N}{\langle F_+^2 \rangle^{1/2}} \left(\frac{S_n(f_0)}{T} \right)^{1/2} .$$
 (7.134)

It is instructive to compare this result with the minimum burst amplitude detectable at a broad-band detector, eq. (7.107). Recalling that $S_n(f)$ has dimensions 1/Hz, i.e. dimensions of time, we must divide it by a time in order to obtain a dimensionless quantity, such as a GW amplitude. For bursts, we see from eq. (7.107) that this time-scale is the duration $\tau_g = 1/f_{\rm max}$ of the burst, while for a periodic signal we see from eq. (7.133) that it is the observation time T. Since T can be of the order of months or years, while τ_g is typically between the millisecond and a second, the minimum value of h detectable for periodic signals is much smaller than for bursts. On the other hand, a periodic signal is intrinsically much weaker, since a burst emits a huge amount of energy in a very short time. We will estimate in Section 7.6.3 the maximum distances at which typical periodic signals can be seen.

For bursts, we assumed that the wave came from the optimal direction, and for this reason we wrote no angular factor in eq. (7.107). For periodic signals, an average over the source position is in any case necessary

because of the apparent motion of the source in the sky, leading to the amplitude modulation, and produces the angular efficiency factor $\langle F_+^2 \rangle$ in eq. (7.134).

An alternative reference quantity which is often used is $h_{3/\text{yr}}$, which is defined as the minimum value of h_0 that can be detected at a given value of S/N, integrating for $T = 10^7$ s (i.e. about 1/3 of a year),

$$h_{3/\text{yr}} = \frac{S/N}{\langle F_+^2 \rangle^{1/2}} \sqrt{S_n(f_0) \times 10^{-7} \,\text{Hz}}.$$
 (7.135)

7.6.2 Doppler shift and phase modulation

Even if an astrophysical source emitted exactly monochromatic GWs with a frequency f_0 , for a detector on Earth the instantaneous value of the observed frequency f would change with time because of the Doppler effect. Recall that, to first order in v/c, the frequency measured by an observer with a velocity \mathbf{v} with respect to the source is

$$f = f_0 \left(1 + \frac{\mathbf{v} \cdot \hat{\mathbf{r}}}{c} \right) \,, \tag{7.136}$$

where $\hat{\mathbf{r}}$ is the unit vector in the direction of the source. If $\mathbf{v} \cdot \hat{\mathbf{r}}$ were a constant, this would cause little concern, since it would just amount to a constant offset in the frequency and, with a single FFT, monochromatic lines at all possible frequencies are searched simultaneously. However, the velocity of the detector with respect to the source changes in time because of the Earth's rotation and because of its revolution around the Sun and this induces a time-dependence in the observed frequency. We denote by $(\Delta v)_T$ the change of the component of the velocity in the direction of the source, in a time T. Then the frequency f changes on the same time interval by an amount

$$(\Delta f)_{\text{Doppler}} = f_0 \frac{(\Delta v)_T}{c}. \tag{7.137}$$

When we integrate the signal for a time T, the resolution in frequency is $\Delta f = 1/T$. As long as $(\Delta f)_{\text{Doppler}}$ is smaller than this resolution, all the GW signal falls into a single frequency bin and the Doppler effect can be neglected. To estimate the maximum integration time for which the Doppler effect is negligible, we consider first the effect of the Earth rotation around its axis. At a latitude of 40 degrees, the rotational velocity of the Earth is $v_{\rm rot} = \omega_{\rm rot} R_{\oplus} \cos(40^{\circ}) \simeq 355 \text{ m/s}$, where $\omega_{\rm rot} = (2\pi/24 \text{ hr})$ and $R_{\oplus} \simeq 6.38 \times 10^6 \text{ m}$ is the mean Earth equatorial radius. This gives $v_{\rm rot}/c \simeq 1.2 \times 10^{-6}$. During an integration time T, the Earth rotates by an angle $\Delta \theta = \omega_{\rm rot} T$ and, if $\Delta \theta \ll 1$, in order of magnitude the change of the component of the velocity in the direction of the source is given by $(\Delta v)_T/v_{\rm rot} \sim \Delta \theta$, i.e.

$$(\Delta v)_T \sim v_{\rm rot} \omega_{\rm rot} T$$
. (7.138)

(The precise numbers, of course, depends on the exact direction of the source with respect to the detector.) Then $(\Delta f)_{\text{Doppler}}$ becomes of the

 46 For frequencies $f_0 > O(40)$ Hz we

have $T \ll 1$ day, so the approximation

 $\Delta\theta \ll 1$ used to write eq. (7.138) is con-

sistent. Otherwise, a more accurate es-

timate is needed

order of the frequency resolution if

$$f_0\left(\frac{v_{\rm rot}}{c}\right)\,\omega_{\rm rot}T \sim \frac{1}{T}\,,$$
 (7.139)

which gives

$$T \sim 60 \text{ min } \left(\frac{1 \text{ kHz}}{f_0}\right)^{1/2}$$
 (7.140)

Therefore, for waves with $f_0 \sim 1$ kHz, the Doppler effect due to the Earth's rotation around its axis becomes important after about one hour. He is maximum value after about 12 hr (the precise numbers, again, depend on the source position), when the detector has inverted its velocity with respect to the source, $\Delta v_{\rm rot} = 2v_{\rm rot}$, and in this time span the frequency has changed by a total amount

$$(\Delta f)_{\text{max}}^{\text{rot}} \sim 2f_0 \, \frac{v_{\text{rot}}}{c} \simeq 2.4 \times 10^{-3} \,\text{Hz} \left(\frac{f_0}{1 \,\text{kHz}}\right) \,.$$
 (7.141)

We can repeat the same reasoning for the orbital motion of the Earth around the Sun. For an order-of-magnitude estimate we can take the orbit as circular, with a radius R=1 au $\simeq 1.5\times 10^{11}$ m and $\omega_{\rm orb}=2\pi/(365\,{\rm days})$, so $v_{\rm orb}\simeq 3\times 10^4\,{\rm m/s}$ and $v_{\rm orb}/c\simeq 10^{-4}$. The maximum frequency shift induced by the Earth revolution is then

$$(\Delta f)_{\rm max}^{\rm orb} \sim 2f_0 \frac{v_{\rm orb}}{c} \simeq 0.2 \,\mathrm{Hz} \left(\frac{f_0}{1 \,\mathrm{kHz}}\right) \,, \tag{7.142}$$

and is much larger than that due to the Earth rotation around its axis, given in eq. (7.141), because $v_{\rm orb} \gg v_{\rm rot}$. However, the large drift (7.142) takes place over a six months period. In an integration time T much shorter than six months, the orbital motion induces a variation $(\Delta v)_T \sim v_{\rm orb}\omega_{\rm orb}T$ and the corresponding frequency shift is $(\Delta f)_{\rm Doppler} \sim f_0 (v_{\rm orb}/c) \omega_{\rm orb}T$. Similarly to eq. (7.139), the time after which the orbital Doppler shift becomes larger than the frequency resolution is given by

$$f_0\left(\frac{v_{\rm orb}}{c}\right)\,\omega_{\rm orb}T \sim \frac{1}{T}\,,$$
 (7.143)

i.e.

$$T \sim 120 \text{ min } \left(\frac{1 \text{ kHz}}{f_0}\right)^{1/2}$$
 (7.144)

Therefore the Doppler shift due to the Earth rotation around its axis is the first to become important, when we increase the integration time (after about 1 hr if, for instance, $f_0 = 1$ kHz). The orbital Doppler shift becomes of the order of the frequency resolution shortly afterwards, after an integration times of about 2 hr for $f_0 = 1$ kHz, but then raises steadily; after less than one day it becomes more important than the contribution from the Earth's rotation around its axis, and it continues to raise for a six months period becoming, on the long term, the largely dominant effect.

After an integration time of four months, i.e. $T \simeq 10^7$ s, the frequency resolution is $\Delta f = 10^{-7}$ Hz, which is many order of magnitudes smaller than the Doppler shifts (7.141) and (7.142). It is interesting to see what is the form of the frequency spectrum when we are sensitive enough to resolve the time-changing Doppler shift. To simplify the geometry, we assume at first that the detector performs a simple circular motion, with frequency ω_m and radius R, and that the source is in the plane of the orbital motion of the detector, as in Fig. 7.10. Since the source is at a very large distance, we have a plane wavefront propagating along the y axis, and therefore proportional to $\cos[\omega_0(t+y/c)]$, where $\omega_0 = 2\pi f_0$ and f_0 is the GW frequency. The y coordinate of the detector is a function of time; we choose for definiteness the origin of time so that y(0) = 0, and therefore $y(t) = R\sin(\omega_m t)$. Then the detector sees a signal proportional to

$$\cos\left[\omega_0\left(t + \frac{y(t)}{c}\right)\right] = \cos[\omega_0 t + \beta\sin(\omega_m t)], \qquad (7.145)$$

where

$$\beta = \frac{\omega_0 R}{c} = \frac{\omega_0}{\omega_m} \frac{v}{c},\tag{7.146}$$

with $v = \omega_m R$. The parameter β is called the modulation index, and $\omega_m = 2\pi f_m$, where f_m is the modulation frequency. This signal can be written as a superposition of monochromatic waves using the identity

$$\cos[\omega_0 t + \beta \sin(\omega_m t)] = \sum_{k=-\infty}^{\infty} J_k(\beta) \cos[(\omega_0 + k\omega_m)t], \qquad (7.147)$$

where $J_k(\beta)$ is the Bessel function.⁴⁷ The signal is therefore split into a carrier at the frequency f_0 , plus an infinite number of sidebands at $f_0 \pm kf_m$, for all integer k, and the power in the k-th sideband is proportional to $J_k^2(\beta)$. The qualitative form of this spectrum depends strongly on the modulation index β . For $\beta \to 0$ and k integer we have $J_k(\beta) \sim \beta^{|k|}$, so when $\beta \ll 1$ most of the power is in the carrier (k=0), with smaller power in the sidebands $k=\pm 1$, even smaller power at $k=\pm 2$, etc. However, in our case β is given by eq. (7.146) and it is large. In fact, for the rotation of the Earth around its axis, setting $\omega_m = 2\pi/(24 \,\mathrm{hr})$ and $v/c \simeq 1.2 \times 10^{-6}$, eq. (7.146) gives $\beta \simeq 100 \, (f_0/1 \,\mathrm{kHz})$, while for the orbital motion $\beta \simeq 3 \times 10^6 \, (f_0/1 \,\mathrm{kHz})$. Therefore, in the range of frequencies relevant for ground-based interferometers $(f_0 > O(10) \,\mathrm{Hz})$, we are always in the regime $\beta \gg 1$.

The average number of sidebands into which the total power is distributed can be calculated using⁴⁸

$$\langle k^2 \rangle \equiv \frac{\sum_{k=-\infty}^{\infty} k^2 J_k^2(\beta)}{\sum_{k=-\infty}^{\infty} J_k^2(\beta)}$$
$$= \frac{\beta^2}{2}, \qquad (7.148)$$

so the power is distributed in $O(\beta)$ sidebands, as shown in Fig. 7.11. Once the frequency resolution 1/T has become of the order of this

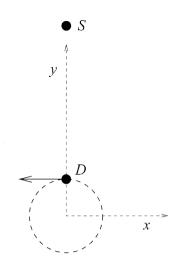


Fig. 7.10 The simplified geometry discussed in the text. The detector D performs a circular motion in the (x,y) plane. The source S is in the same plane, along the y axis.

⁴⁷This identity can be obtained writing $\cos[(\omega_0 + k\omega_m)t]$, inside the sum, as $\cos(\omega_0 t)\cos(k\omega_m t) - \sin(\omega_0 t)\sin(k\omega_m t)$, and using Gradshteyn and Ryzhik (1980), 8.514.5 and 8.514.6, recalling that, for k integer, $J_{-k}(z) = (-1)^k J_k(z)$.

⁴⁸See Gradshteyn and Ryzhik (1980), 8.536.2.

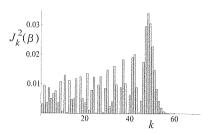


Fig. 7.11 The quantity $J_k^2(\beta)$ for $\beta = 50$, as a function of k.

Doppler line broadening, increasing T further does not improve substantially the signal-to-noise ratio. In fact, while a smaller frequency bin contains by definition less noise, it now also contains less signal, since the signal gets spread over many bins. However, if at this stage the signal already emerged from the noise, increasing T further we improve the resolution by which we are able to reconstruct the line shape (and therefore the accuracy by which we can reconstruct the source position, see Note 49 below).

Luckily, there is in principle a very simple way (borrowed from pulsar radio-astronomy) to correct for the Doppler shift. In the simple geometrical situation illustrated in Fig. 7.10, we just need to define a new time variable t' = t + y(t)/c. In terms of this variable the signal (7.145) is simply proportional to $\cos(\omega_0 t')$ and, performing the Fourier transform with respect to t', all sidebands collapse into a single frequency. For a generic source location, the redefinition of time that does the job is

$$t' = t + \frac{\mathbf{x}(t) \cdot \hat{\mathbf{r}}}{c} \,, \tag{7.149}$$

where $\mathbf{x}(t)$ is the position of the detector (measured for instance using the Solar System Barycenter (SSB) as a reference frame) and $\hat{\mathbf{r}}$ is the unit vector pointing toward the source. Observe that this is just the Roemer time delay that we already discussed in Section 6.2.2. We can therefore simply resample the output of the detector in terms of this new time, and we have corrected for the Doppler effect. The procedure has an added bonus: it is quite likely that, in the Fourier spectrum of the output, there will be monochromatic lines due to instrumental noise. If such a line has a frequency that is constant in time to a good accuracy, its signal-tonoise ratio will increase as \sqrt{T} , just as for a GW signal. However, when we apply the resampling procedure, a real GW signal, which was spread over many bins, is collapsed to a single frequency bin, an conversely an instrumental line which was monochromatic will be spread over many bins, and will finally be diluted into the noise. In other words, we are using the Doppler modulation as a powerful signature that discriminates a real GW signal from instrumental noise.

The simplicity of this solution comes however at a price: we need to know both $\mathbf{x}(t)$ and $\hat{\mathbf{r}}$ with great precision. We can assume that the motion of the Earth is known to a sufficient accuracy (although, if we want to integrate for a time $T\sim 1$ yr, we need to keep under control effects that can produce shifts $\Delta f\sim 10^{-7}$ Hz, and for this we must also include small effects like the oscillations of the Earth around the Earth-Moon barycenter, which however are precisely known), so the main error comes from the uncertainty on the angular position of the source. From eq. (7.136) we see that, in order to correct for the Doppler shift with an accuracy smaller than the experimental resolution 1/T on f, we need, in order of magnitude,

$$\frac{f_0}{c} (\Delta v)_T \, \Delta \theta < \frac{1}{T} \,, \tag{7.150}$$

where $(\Delta v)_T$ is the variation of the velocity of the Earth during the

observation time T (recall that only the temporal variation is relevant, otherwise the Doppler effect would give just a constant offset in f_0), while $\Delta\theta$ is the angular resolution (in radians) on the position of the source.

If we take $(\Delta v)_T \sim v_{\rm orb}\omega_{\rm orb}T$ we find that, to apply the Doppler correction, we need to know the source location to an accuracy

$$\Delta \theta < \frac{1}{f_0(v_{\rm orb}/c)\,\omega_{\rm orb}T^2}$$

$$\simeq 0.1\,\mathrm{arcsec}\left(\frac{10^7\,\mathrm{s}}{T}\right)^2\,\left(\frac{1\,\mathrm{kHz}}{f_0}\right). \tag{7.151}$$

This expression is correct as long as the Doppler effect due to the orbital motion dominates that due to the Earth rotation around its axis, i.e. for $T\gtrsim 1$ day, and also as long as the angle $\omega_{\rm orb}T$ is small, since otherwise the approximation $(\Delta v)_T \sim v_{\rm orb}\omega_{\rm orb}T$ should rather be replaced by $(\Delta v)_T \sim v_{\rm orb}\sin\omega_{\rm orb}T$, so approximately eq. (7.151) is valid as long as $1 \text{ day} \lesssim T \lesssim 4 \text{ months.}^{49}$

If we are targeting a specific source whose position is known to this accuracy or better, as is the case for many pulsars, this requirement does not pose special problems. However, as we will discuss in Vol. 2, there are many mechanisms that can produce periodic GWs, in particular in neutron stars, that are not necessarily associated to a strong electromagnetic emission or, as with pulsars, the electromagnetic emission could be beamed in a direction that does not intersect the Earth. It is in fact quite likely that most of the potentially interesting sources of periodic GWs have no detected optical counterpart. For example, the closest observed neutron star is at a distance $r \sim 100$ pc; however. population synthesis calculations indicate that the closest one should be at a distance $r \sim 5-10$ pc, and then in a sphere of radius $r \sim 100$ pc there should be $O(10^3-10^4)$ neutron stars. It is therefore of the greatest interest to perform blind searches, i.e. searches for unknown sources over the whole sky. In principle, this means that we should partition the celestial sphere in pixels with a size given by eq. (7.151) (in fact even smaller, see Note 52 in the next section), and in each one we should apply a separate Doppler correction. As we will see in the next section, for integration times of months this is impossible, even with the maximum present or foreseeable computer power.

Furthermore, we have assumed until now that the intrinsic GW frequency f_0 of the source is stable, within the experimental resolution $\Delta f = 1/T$, and that the only modulation comes from the relative motion of the detector. This means that we are assuming a stability of the source frequency at the level

$$\frac{\Delta f_0}{f_0} = \frac{1}{T f_0}$$

$$\simeq 1 \times 10^{-10} \left(\frac{10^7 \,\mathrm{s}}{T}\right) \left(\frac{1 \,\mathrm{kHz}}{f_0}\right) .$$
(7.152)

Quite remarkably, rotating neutron stars can sometime have this sta-

 49 Actually, one could turn the argument around and observe that, if we are so lucky that there is a periodic signal so strong that can be extracted from the noise without correcting for the Doppler shift then, following the evolution of the frequency with time, we will be able to reconstruct the position of the source to the accuracy $\Delta\theta$ given by eq. (7.151). With present detector sensitivities, however, this possibility seems quite unlikely.

$$\dot{f} = -af^n \tag{7.154}$$

(where a is a constant) and if the brak- $ing\ index\ n > 1$. In fact, integrating
the above equation we get

$$[f(t)]^{-(n-1)} - [f(0)]^{-(n-1)}$$

= $a(n-1)t$, (7.155)

where t=0 is the time at which the pulsar was born. If the frequency at birth, f(0), was much bigger than the frequency today, and if n>1, we can neglect the term $[f(0)]^{-(n-1)}$ and the age of a pulsar is related to its present values of f and \dot{f} by

$$t = \frac{1}{a(n-1)f^{n-1}}$$

$$= \frac{f}{(n-1)|\dot{f}|}$$

$$= \frac{\tau}{n-1}.$$
 (7.156)

Experimentally, the braking index n typically has values $n\simeq 2-3$, depending on the specific pulsar.

bility. The main mechanism that produces a drift in their frequency is the fact that a rotating NS radiates, both electromagnetically and gravitationally, and therefore loses energy. This energy is taken from the rotational energy of the NS, which therefore spins down. Pulsars are characterized by their spindown age τ ,

$$\tau = \frac{f}{|\dot{f}|},\tag{7.153}$$

where f is their rotational frequency. As we saw in Section 4.2.1, for rotation around a principal axis and in the quadrupole approximation, the GWs emitted are monochromatic with a frequency $f_0 = 2f$, so $\tau = f_0/|\dot{f}_0|$.⁵⁰ During the observation time T, a pulsar with spindown age τ changes its GW frequency by an amount $\Delta f_0 = \dot{f}_0 T = -f_0 T/\tau$, i.e. by

$$\frac{\Delta f_0}{f_0} = -\frac{T}{\tau}$$

$$\simeq -3.2 \times 10^{-10} \left(\frac{10^9 \,\text{yr}}{\tau}\right) \left(\frac{T}{10^7 \,\text{s}}\right) .$$
(7.157)

Comparing with eq. (7.152) we see that, with an integration time $T=10^7$ s, for a millisecond pulsar with $f_0 \sim 1$ kHz, the effect of the spindown is important if its spindown age is lower than 3×10^9 yr while, if $f_0=10$ Hz, spindown is important, again over $T=10^7$ s, if $\tau<3\times 10^7$ yr. Therefore for many pulsars, and in particular for young pulsars, over such a long observation time the spindown must be taken into account. Actually in young pulsars the spindown rate can be so high that even the effect of the second derivative \ddot{f}_0 can become important.

For known pulsars the spindown can be measured and taken into account when we make the Doppler correction, while for blind searches it introduces new unknown parameters. Besides spindown, there are other reasons why the frequency of the GW emitted by a pulsar can change:

- Pulsars exhibit *glitches*, i.e. sudden jumps in the frequency related to rearrangements of their internal structure. These glitches can produce changes in the frequency as large as $\Delta f_0/f_0 \sim 10^{-6}$ and occur erratically, at a rate which depends strongly on the specific pulsar, but in general of the order one glitch every few years.
- A large fraction of the known millisecond pulsars are in binary systems. In this case, there will be an additional Doppler effect due to the motion of the source, as we saw in Section 6.2.
- Pulsar are the remnant of supernova explosions, and at birth they can receive a large kick; so their velocities can be larger than the typical velocities of the stars in their galactic neighborhood, and the pulsar proper motion can be important. Of course, if the motion is uniform, this only produces a constant shift in the frequency. However, accelerations due to gravitational fields can be important. In particular, many pulsars are found in globular clusters. In this case, the acceleration due to the Newtonian gravitational

forces from all the other stars is known to produce frequency drifts comparable to the spindown rate.

• Even a uniform proper motion can be important if, during the observation time, it drives the NS out of the pixel in the sky where it was initially. For instance, a pulsar at a distance r=300 pc, with a transverse velocity $v=10^3$ km/s with respect to our line-of-sight, in a time $T=10^7$ s moves by $\Delta\theta=vT/r\simeq 10^{-6}$ rad $\simeq 0.2$ arcsec which, according to eq. (7.151), is of order of the accuracy $\Delta\theta$ that we need, over such an integration time T, for a pulsar radiating GWs at $f_0\sim 1$ kHz.

In the next section we will discuss how one can try to cope with these difficulties.

7.6.3 Efficient search algorithms

Coherent searches

From the discussion of the previous section we know that, if we want to integrate the signal for a long time, we must resample the output of the detector in terms of the time t^\prime defined in eq. (7.149), plus further correction for the spindown or other effects that change the frequency. The GWs produced by a rotating NS, in the absence of spindown, has been computed in eq. (4.223). Including the Doppler effect of the detector and the spindown of the source we can write the signal received as

$$h(t) = F_{+}(\hat{\mathbf{n}}(t); \psi) h_0 \frac{1 + \cos^2 \iota}{2} \cos \Phi(t) + F_{\times}(\hat{\mathbf{n}}(t); \psi) h_0 \cos \iota \sin \Phi(t)$$

$$(7.158)$$

where h_0 is given in eq. (4.224), and ι is the angle between the spin axis of the neutron star and the propagation direction $\hat{\mathbf{n}}$ of the GW; of course $\hat{\mathbf{n}} = -\hat{\mathbf{r}}$, where $\hat{\mathbf{r}}$ is the unit vector pointing toward the source, and depends on time because of the relative motion of the detector and source. The evolution of the accumulated phase $\Phi(t) = 2\pi \int dt f(t)$ observed by the detector can be described by a Taylor expansion, writing

$$f(t') = f_0 + \dot{f}_0(t' - t'_0) + \frac{1}{2}\ddot{f}_0(t' - t'_0)^2 + \dots,$$
 (7.159)

where t' is the resampled time given in eq. (7.149), i.e. the time of arrival of the signal in the Solar System Barycenter (SSB),⁵¹ and t'_0 is a fiducial value, such that $\Phi(t'_0)$ has the value ϕ_0 . Then

$$\Phi(t) = \phi_0 + 2\pi \left[f_0(t' - t'_0) + \frac{1}{2} \dot{f}_0(t' - t'_0)^2 + \frac{1}{6} \ddot{f}_0(t' - t'_0)^3 + \ldots \right].$$
(7.160)

Of course, a truncated Taylor expansion is useful only if the higher order terms are small corrections during the whole observation time T. This is not the case for a neutron star in a binary system, which rather performs a circular motion around the center-of-mass of the system, so eq. (7.160) only applies to isolated neutron stars.

⁵¹Actually, the precise redefinition is

$$t' = t + \frac{\mathbf{x}(t) \cdot \hat{\mathbf{r}}}{c} + \Delta_{E\odot} - \Delta_{S\odot} ,$$

where $\Delta_{E\odot}$ and $\Delta_{S\odot}$ are the solar system Einstein and Shapiro time delays discussed in Section 6.2.2. However, given the detector and the source positions, the Einstein and Shapiro delays can be computed, as we did explicitly in Section 6.2.2, and introduce no new free parameter.

If our target is a given pulsar whose position, proper motion and spin-down parameters are known to sufficient accuracy, the form of the signal (7.158) is fixed. Then we can simply demodulate the signal defining a new variable t'' as

$$t'' = (t' - t'_0) + \frac{\dot{f}_0}{2f_0}(t' - t'_0)^2 + \frac{\ddot{f}_0}{6f_0}(t' - t'_0)^3 + \dots,$$
 (7.161)

so that eq. (7.160) reads $\Phi = \phi_0 + 2\pi f_0 t''$. We resample the detector output with respect to this variable, and then all we need to do is to perform a single Fast Fourier Transform (FFT) on this resampled stretch of data, of length T. The number of spindown parameters $\dot{f}_0, \ddot{f}_0, \ldots$ to be included to have sufficient accuracy depend on the source, and on the observation time T.

If however we want to perform a blind all-sky search, the problem becomes quickly intractable with increasing observation time T. In fact, our parameter space is given by the angles (θ_s, ϕ_s) of the source and by the spindown parameters \dot{f}_0/f_0 , \ddot{f}_0/f_0 , etc. Observe that f_0 itself does not contribute to the dimension of the parameter space; the resampling of time (7.149) is independent of f_0 , while eq. (7.161) depends only on the ratios \dot{f}_0/f_0 , \ddot{f}_0/f_0 , ..., and not separately on f_0 , \dot{f}_0 , \ddot{f}_0 , This is a crucial advantage of the resampling technique. If, rather than resampling the detector output, we directly used the Wiener filtering for the waveform given in eqs. (7.158) and (7.160), then f_0 would be an additional parameter to be searched, and the computational cost would increase dramatically.

Then, what we should do is to discretize this parameter space, and for each point of this parameter space we should perform the appropriate demodulation (7.161) and one FFT. This procedure is referred to as a coherent search. Its drawback is that, if we want to take advantage of the large integration time, the mesh in the discretized parameter space must become finer and finer when we increase T. For instance, even in the simplest case in which the spindown parameters are negligible, and therefore the parameter space is given only by the angles (θ_s, ϕ_s) , still the number of patches in the sky that we must consider is at least $N_{\text{patches}} = 4\pi/(\Delta\theta)^2$ and scales at least as T^4 , see eq. (7.151).⁵²

More generally, the number of mesh points depends on the kind of search that we perform. For instance, old pulsars are less demanding than young pulsars of the same frequency, since their spindown rate is lower and therefore it can be taken into account using a larger mesh in the spindown parameter space. Similarly, we see from eq. (7.151) that slow pulsars (say, $f_0 < 200 \text{ Hz}$) are easier to analyze that fast pulsars with $f_0 \sim 1 \text{ kHz}$.

Since the time needed for data analysis grows with a large power of T, increasing T we necessarily reach a point where the data analysis would take the same time as the observation time T and beyond that point it would quickly becomes many orders of magnitude larger than the observation time. We can therefore take as a limit the condition that the time required by data analysis does not exceed the observation

time used to take the same data. To have an idea of the computational requirements consider that, using 10⁷ s of data to search for periodic GWs with frequencies up to 500 Hz, requires the calculation of a FFT with 10^{10} points, which takes about 1 s on a teraflop computer (assuming that all 10^{10} points can be held simultaneously in fast memory), and we need one such FFT for each point of the parameter space. It can be estimated⁵³ that a coherent all-sky search of $T = 10^7$ s of data for slow, old pulsars ($\tau > 1000 \text{ yr}$, $f_0 < 200 \text{ Hz}$, i.e. the "easier" target) requires only one spindown parameter and 10¹⁰ independent points in parameter space, while for young, fast pulsar (frequencies up to $f_0 \simeq 1$ kHz, τ as low as 40 yr) three spindown parameters and 8×10^{21} points in parameter space are required. Then, even in the "easy" case, the analysis of four months of data would require three centuries on a teraflop computer! Requiring that the data analysis does not last more than data taking. one finds that for slow, old pulsars the data stretch cannot be longer than ~ 18 days, while for young, fast pulsar the limit is less than one day. The disappointing conclusion is that, even if a detector can in principle take good data for months or years, a coherent blind all-sky search for pulsar using fully these data is impossible.

Incoherent searches

A solution to the computational problem discussed above is to split the total observation time T into $\mathcal N$ stacks of length $T_{\rm stack}$, with $T=\mathcal NT_{\rm stack}$. We choose $T_{\rm stack}$ so that a coherent search over such a time is computationally feasible. So the output of each coherent search over one stack is a collection of function $\tilde{h}(f)$, one for each value of the parameter space. For each point in parameter space we then add the quantity $|\hat{h}(f)|^2$ over the $\mathcal N$ stacks. Since in this way the phase information between the different stacks gets lost, this is called an *incoherent search*. If we denote by $\tau_{\rm stack}$ the time needed to perform a coherent search on a stack of data of length $T_{\rm stack}$, the time needed for the full incoherent search is $\tau_{\rm inc} = \mathcal N \tau_{\rm stack}$, while the time needed for a full coherent search over the whole time T is $\tau_{\rm coh} \simeq (T/T_{\rm stack})^n \tau_{\rm stack} = \mathcal N^n \tau_{\rm stack}$, so

$$\tau_{\rm inc} \simeq \frac{1}{N^{n-1}} \tau_{\rm coh} \,, \tag{7.162}$$

where the power n, as discussed above, is determined by the kind of pulsars that we are targeting. Since n is large (at least n=5, even when no spindown parameters are needed, see Note 52), it is clear that incoherent searches have a huge advantage in terms of computational cost and, for a given observation time T, taking \mathcal{N} sufficiently large, i.e. T_{stack} sufficiently small, the computation becomes feasible.

From the point of view of sensitivity, the value of $(S/N)^2$ obtained from a single stack of length T_{stack} is given by eq. (7.130) replacing T by T_{stack} . Adding N of these spectra the variance is reduced by $1/\sqrt{N}$ and therefore, for an incoherent search, eq. (7.130) becomes

$$\left(\frac{S}{N}\right)^2 = \langle F_+^2 \rangle \left(\frac{\mathcal{N}^{1/2} T_{\mathrm{stack}}}{S_n(f_0)}\right) h_0^2$$

⁵³See Brady, Creighton, Cutler and Schutz (1998).

 $^{^{52}\}mathrm{A}$ more careful argument shows that it even scales like T^5 , because the approximation $(\Delta v)_T \sim v_{\mathrm{orb}}\omega_{\mathrm{orb}}T$ used to derive eq. (7.151) does not hold simultaneously for the right ascension and for the declination angles, see Brady, Creighton, Cutler and Schutz (1998).

$$= \frac{1}{\sqrt{\mathcal{N}}} \langle F_+^2 \rangle \left(\frac{T}{S_n(f_0)} \right) h_0^2 \tag{7.163}$$

and the minimum amplitude detectable at a given S/N, eq. (7.134)becomes

$$(h_0)_{\min} = \eta \left(\frac{S_n(f_0)}{T}\right)^{1/2}$$
 (7.164)

where we have defined an efficiency factor

$$\eta = (S/N) \frac{\mathcal{N}^{1/4}}{\langle F_+^2 \rangle^{1/2}},$$
(7.165)

which takes into account the desired level of the signal-to-noise ratio S/N, the average over the orbit of the source, which produces the factor $\langle F_{\perp}^2 \rangle^{1/2}$, and the need to separate the data into N stack for computational feasibility.

In practice, beside being forced by computational requirements, incoherent searches are also necessary because a detector never has months. and not even weeks, of continuous good data taking. There are always interruption due to maintenance, period of higher noise level that must be removed, etc. and the experimental precision that one has on the time at which data taking resumed is not sufficiently good to recombine coherently different stacks of data. The incoherent method, of course, can be applied even when the single stacks have not all the same duration. and when they are not consecutive.

When performing an incoherent search each stack is demodulated, as discussed in the previous section, using a mesh of points sufficient to confine the searched signal into a single bin. The individual power spectra, before being summed, must be corrected for their relative frequency drift using a finer parameter mesh suitable for removing the phase modulation over the whole observation period. The simplest implementation of this method consists in choosing stacks of about 30 min, so that the Doppler effect in each stack can simply be neglected, and within a single stack no demodulation is needed. In this case a period of 10⁷ sec of data is divided into $\mathcal{N} \simeq 5000$ stacks, and the minimum detectable value $(h_0)_{\min}$ in eq. (7.164) is larger than in a coherent search by a factor $\mathcal{N}^{1/4} \sim 8$. With the difference, of course, that a blind full-sky incoherent search of this type is computationally feasible while a blind full-sky coherent search is not.

Alternatively, one can choose longer stacks, say of the order of one day. These will need demodulation, but a relatively coarse mesh in parameter space will suffice to concentrate the whole signal into a single bin. Then we combine the separate stacks using a finer mesh. Of course, the longer the stack, the higher is the sensitivity, but the higher is also the computational cost. Incoherent searches can also be used as a first stage in a hierarchical search: an incoherent blind all-sky search can

produce a number of interesting candidate signals, for certain values of the parameters. These points in parameter space can then be examined more thoroughly with a directed coherent search.

We can now compare the experimental sensitivity given by eq. (7.164) with the signal expected from a rotating NS, given in eq. (4.224). We then find that the maximum distance r which a detector can reach in a blind full-sky search for periodic GWs from rotating neutron stars is

$$r = 5.8 \,\mathrm{kpc} \left(\frac{10^{-23} \,\mathrm{Hz}^{-1/2}}{S_n^{1/2}(f_0)} \right) \left(\frac{T}{3 \times 10^7 \,\mathrm{s}} \right)^{1/2} \times \left(\frac{100}{\eta} \right) \left(\frac{\epsilon}{10^{-6}} \right) \left(\frac{I_{zz}}{10^{38} \,\mathrm{kg} \,\mathrm{m}^2} \right) \left(\frac{f_0}{1 \,\mathrm{kHz}} \right)^2.$$
 (7.166)

The reference value $\eta = 100$ corresponds to a search for a total time T = 3×10^7 s divided into stacks with $T_{\rm stack} \simeq 30$ min (so $\mathcal{N} \simeq 1.7 \times 10^4$), a factor $1/\langle F_+^2 \rangle^{1/2} = \sqrt{5}$ as appropriate for interferometers, see Table 7.1. and a value $S/N \simeq 4$. The strain sensitivity $S_n^{1/2}$ has been normalized to the value expected for an advanced interferometer.

The Hough transform

As we have seen above, in incoherent searches the observation time is divided into stacks, where the phase modulation due to Doppler effect and spindown is either negligible (if $T_{\rm stacks} \lesssim 30$ min) or anyway relatively easy to correct for, so that a GW signal, if present, falls into a single frequency bin. When we compare different stacks, the position in frequency of the bin that contains the signal changes, because of the Doppler effect and of the spindown. For each point in the parameter space $(\theta_s, \phi_s, f_0/f_0, f_0/f_0, \ldots)$ we can compute how the position of the bin should change and we can correct for it, using the resampling technique discussed in the previous section. In this way, for each point of parameter spaces, the bins are "realigned", and the power in corresponding bins is summed.

An interesting variation on this scheme is given by the Hough transform, which is a technique used for pattern recognition in digital images.⁵⁴ In the Hough transform, as a first step, rather than summing up the power in the corresponding bins, we fix a threshold in each data stack. A bin is deemed "black" if the power in it exceeds the threshold. and "white" if it does not. In the time-frequency plane obtained aligning in frequency (with no correction) the various stacks, we therefore have a set of black pixels, as in Fig. 7.12.

In the case of Gaussian noise, where large fluctuations are unlikely, it would in principle be more convenient to sum up the power of the corre sponding bins, rather than reducing all the information to a set of zeros (white) and ones (black). However, the Hough transform is more robust in the presence of non-Gaussian noise and large occasional external dis-

⁵⁴It was developed in 1959 by Paul Hough at CERN, to analyze the tracks of particles in bubble chambers, and today is also used in astronomical data analysis.

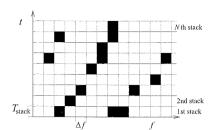


Fig. 7.12 The time–frequency plane, with bins of length $\Delta t = T_{\rm stack}$ in time and $\Delta f = 1/T_{\rm stack}$ in frequency. Bins where the power exceeds a given threshold are marked in black.

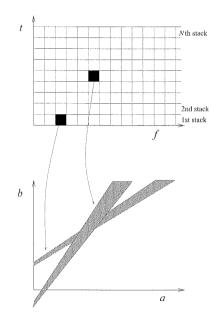


Fig. 7.13 The map that to each black pixel associates a submanifold in the parameter space Σ .

turbances, which is the case in a real detector. Consider for instance the situation in which instrumental noise gives a very large spike in frequency, during a relatively short period, e.g. in only one stack. When summing the power, this single disturbance can give a large effect on the total sum, while collapsing all the information to black/white it contributes only to a single pixel. This method can therefore be appropriate when we search for a signal that is small, but is there during the whole observation time, and is embedded in a noise that occasionally can be much larger than the signal.

The next step is to perform a pattern recognition procedure in the set of black pixels, to see if some of the black pixels lie along a specified curve. To simplify the setting, suppose that we are searching for straight lines in the (t, f) plane of Fig. 7.12. (The generalization to more complex curves is conceptually straightforward.) The set of all straight lines in this plane is parametrized by two parameters (a, b), as t = af + b. We denote by Σ the manifold described by the parameters (a,b); in our example of straight lines of course $\Sigma = \mathbb{R}^2$, but the notation is more general. Given a black pixel, we can find the set of points in the manifold Σ that are compatible with it; for instance, in our straight lines example, if a black pixel is centered at (t_1, f_1) , the straight lines consistent with it are those that satisfy $t_1 = a f_1 + b$, and the corresponding submanifold of Σ is the curve $b = af_1 - t_1$ in the (a, b) plane. More precisely, since the pixels in the (t, f) plane have a finite resolution, we will rather get a bunch of straight lines in Σ . The transformation that, to each black pixel in the (t, f) plane associates a submanifold in Σ , is illustrated graphically in Fig. 7.13.

In the absence of noise, the submanifolds in Σ obtained in this way from all the black pixels would have a non-empty intersection, which would define the point in parameter space compatible with the observations. Of course, in the presence of noise the intersection of all the curves will be empty. Still, we can try to recover the most probable value of the parameters in Σ as follows. First, we discretize the manifold Σ . Let us call \mathcal{C}_1 the surface in Σ obtained from the first black pixel. We then assign +1 to all the bins in Σ that belong to \mathcal{C}_1 . We repeat the same for the second black pixel, adding +1 to the the bins in Σ that belong to \mathcal{C}_2 , and so on for all the N black pixels. In conclusion, we have constructed a map that, to the set of black pixels, associates a histogram in the parameter space Σ .

In the GW detection problem, the manifold Σ becomes the parameter space $(\theta_s, \phi_s, \dot{f}_0/f_0, \ddot{f}_0/f_0, \dots)$ and the straight lines of our example are replaced by the curves in the (t, f) plane that describe how f changes with time because of the Doppler effect and of the spindown. The points in parameter space whose number count is above a certain threshold are the candidates for a possible detection and can be further investigated, for instance with a coherent search.

7.7 Coalescence of compact binaries

The coalescence of compact binaries, such as BH-BH and NS-NS binaries, is a particularly interesting signal for broad-band GW detectors. This comes from a combination of two facts: first, we saw in eq. (4.44) that, in the last stages of the inspiral, a binary system can radiate away in GWs up to a few per cent of its total mass. This is a huge amount of energy, so the signal from an inspiral is quite strong, compared to most other GW sources. Second, the inspiral phase can be tracked for many cycles in a broad-band detector. We saw in eq. (4.23) that a groundbased interferometer can follow the inspiral phase of a compact binary system for $O(10^4)$ cycles. Thus, matched filtering can be very effective for extracting this signal from the noise. From eq. (7.40) and the discussion below it we see that, in order of magnitude, with matched filtering we can dig into the noise and catch the signal from a coalescence, even when the typical amplitude of the GW signal inside the interferometer bandwidth is smaller than the noise floor by a factor $\mathcal{N}_c^{1/2}$, where \mathcal{N}_c is the number of cycles for which we are able to track carefully the signal with our template. Thus, we can gain a factor as large as $N_c^{1/2} \sim 100$ in amplitude, if our template is so good that we can follow closely the signal from the time it enters in the interferometer bandwidth until the inspiral phase terminates and the two objects merge. Since the GW amplitude is proportional to 1/r, a factor O(100) in amplitude means that we gain a factor O(100) in the maximum distance to which we can detect a source. For these reasons, we will see that interferometers have the potential of detecting coalescing binaries up to distances of order of hundreds of Mpc, and advanced ground-based interferometers could reach a few Gpc.

To exploit this opportunity we must however be able to follow closely the signal with a template. This means, first of all, that for a given value of the parameters of the binary system (time of coalescence, masses, spins, etc.), one must know the waveform accurately. We already quantified this requirement in Section 5.6, where we found that we need to compute the post-Newtonian corrections up to 3.5PN order. As we saw in Section 5.6, these remarkable computations have indeed been performed. The second aspect is that we do not know in advance the parameters of the system, and therefore we must scan a potentially large parameter space.

To leading Newtonian order we computed the waveform in eq. (4.29), and the corrections in the restricted post-Newtonian approximation were discussed in Section 5.6. Combining these results with the general expression $h(t) = F_+h_+(t) + F_\times h_\times$, we see that the output h(t) for a binary inspiral, in the restricted post-Newtonian approximation, is

$$h(t) = A_{+} \left[\frac{\pi f_{\text{gw}}(t)}{c} \right]^{2/3} \cos[\Phi(f_{\text{gw}}(t)) + \Phi_{0}]$$

$$+ A_{\times} \left[\frac{\pi f_{\text{gw}}(t)}{c} \right]^{2/3} \sin[\Phi(f_{\text{gw}}(t)) + \Phi_{0}], \qquad (7.167)$$

⁵⁵Explicit expressions for $\Phi(f_{\rm gw})$ and $f_{\rm gw}(t)$ up to 2PN were given in eq. (5.273), and in eq. (5.270) or (5.272).

⁵⁶Recall also from Section 4.1.4 that,

for binaries at cosmological distances.

i.e. at a non-negligible redshift z, the

masses m_1 and m_2 must be multiplied

by (1+z), and the distance r must

be replaced by the luminosity distance

⁵⁷Furthermore, the angles (θ, ϕ) change

in time because of the Earth's mo-

tion. For a ground-based interferom-

eter, which follows the coalescence only

for 10–15 minutes, this dependence can

be neglected. For a space-borne interferometer, instead, it must be taken

 $d_L(z)$.

into account.

where, as discussed in Section 5.6.3, $\Phi(f_{\rm gw})$ and $f_{\rm gw}(t)$ are known up to 3.5PN order. 55 We have esplicitly displayed the arbitrary constant Φ_0 in the phase, equivalent to the arbitrary constant ω_0 in eq. (5.265), and we have defined

$$A_{+} = \frac{4}{r} \left(\frac{GM_c}{c^2} \right)^{5/3} F_{+}(\theta, \phi) \frac{1 + \cos^2 \iota}{2}$$
 (7.168)

$$A_{\times} = \frac{4}{r} \left(\frac{GM_c}{c^2} \right)^{5/3} F_{\times}(\theta, \phi) \cos \iota . \tag{7.169}$$

Writing $A_{+} = A \cos \alpha$ and $A_{\times} = A \sin \alpha$, with $A = (A_{+}^{2} + A_{\times}^{2})^{1/2}$ and $\tan \alpha = A_{\times}/A_{+}$, we can rewrite this as

$$h(t) = A \left[\frac{\pi f_{\text{gw}}(t)}{c} \right]^{2/3} \cos[\Phi(f_{\text{gw}}(t)) + \varphi], \qquad (7.170)$$

with $\varphi = \Phi_0 - \alpha^{56}$ Thus, in the waveform enter the distance r to the source, its location, specified by the angles (θ, ϕ) which appear in the pattern functions, the orientation of the orbit with respect to the line of sight (two angle, one of which is ι , and the other identifies the axes with respect to which the plus and cross polarizations are defined), the reference time t_* at which the signal enters in the detector bandwidth (which appears through $\Phi(t)$ and $f_{gw}(t)$, the constant phase φ , the masses of the two stars, and in principle also their spins (which we neglected in eq. (5.273)). So, in total, we have 15 parameters. ⁵⁷ However, a number of simplifications are possible, as we discuss in the next subsection.

7.7.1 Elimination of extrinsic variables

The variables that can be eliminated from the parameter space are generically called extrinsic. First, we observe that all possible shifts in time of the signal can be obtained at once with a single Fourier transform. Consider in fact the scalar product $(h(\theta, t_*)|s)$ between the output s(t)of the detector and the template $h(t;\theta,t_*)$ where, from the parameters θ^i , we singled out explicitly the arrival time t_* , defined as the time when the hypothetical signal enters into the interferometer bandwidth, say at $f_{\rm gw}=10$ Hz. The waveform $h(t;\theta,t_*)$ is obtained from $h(t;\theta,t_*=0)$ with a time translation, so if we denote by $h(f;\theta)$ the Fourier transform of $h(t;\theta,t_*)$ at $t_*=0$, the Fourier transform of $h(t;\theta,t_*)$ at t_* generic is simply $\tilde{h}(f;\theta)e^{i2\pi ft_*}$. Thus, from the definition (7.46) of the scalar product, we have

$$(h(\theta, t_*)|s) = 4 \operatorname{Re} \int_0^\infty df \, \frac{\tilde{h}^*(f; \theta)\tilde{s}(f)}{S_n(f)} \, e^{i2\pi f t_*} \,,$$
 (7.171)

which is just the Fourier transform of $\tilde{h}^*(f;\theta)\tilde{s}(f)/S_n(f)$. Thus, per forming a single FFT we can immediately locate the value of t_* which gives the highest signal-to-noise ratio. This is of course of great practical importance. Typically we can expect that, to perform efficiently the matched filtering, the maximum mismatch in arrival time that we can tolerate between the real signal and our template could be, say, of order 3 ms. If one should analyze one year of data $(3 \times 10^7 \text{ s})$ computing a different scalar product every 3 ms, for each value of θ one should perform 10^{10} times the computation of the scalar product $h(t; \theta, t_*)$, while we see that just a single FFT does the job. Thus, the arrival time t_* is not really part of the parameter space that must be searched. Figure 7.14 shows the result of a simulation in which the signal corresponding to the coalescence of two BHs, each with a mass of $10M_{\odot}$, at a distance of 150 Mpc, is injected into the noise of the VIRGO detector. Performing the Fourier transform, we see that we have a spike in correspondence with the time at which this signal has been injected (in the figure, $t_* = 1$, in arbitrary units).

Two more parameters that appear in eq. (7.167), which can be eliminated analytically from the matched filtering procedure, are the amplitude A and the phase φ of the signal. We already saw in Section 7.3 that the optimal filter is defined modulo an arbitrary constant, so the overall value of the amplitude A does not enter when we search for the template that maximizes the signal-to-noise ratio. The maximization of the SNR with respect to φ can be performed analytically, writing the template (7.170) in the form

$$h(t) = h_c(t)\cos\varphi + h_s(t)\sin\varphi. \tag{7.172}$$

If s(t) is the detector output, after maximization of the log-likelihood function over the amplitude A, according to eq. (7.70) we want to further maximize

$$2\log \Lambda = \frac{(h|s)^2}{(h|h)}$$

$$= \frac{[(h_c|s) + (h_s|s)\tan\varphi]^2}{(h_c|h_c) + (h_s|h_s)\tan^2\varphi + 2(h_c|h_s)\tan\varphi}.$$
(7.173)

This expression is easily maximized analytically with respect to $\tan \varphi$. The result is simpler if we introduce two new templates

$$h_p = h_c \cos \phi_p + h_s \sin \phi_p \,, \tag{7.174}$$

$$h_q = h_c \cos \phi_q + h_s \sin \phi_q \,, \tag{7.175}$$

where the angles ϕ_p and ϕ_q are chosen so that h_p and h_q satisfy $(h_p|h_q) =$ 0, i.e. they are orthogonal with respect to the scalar product (|). In terms of these orthogonal templates the likelihood function, after performing the maximization over the amplitude A and over the phase φ , takes the simple form

$$2\log\Lambda = \frac{(h_p|s)^2}{(h_p|h_p)} + \frac{(h_q|s)^2}{(h_q|h_q)}.$$
 (7.176)

Therefore, the maximization with respect to the remaining variables is equivalent to maximizing the sum in quadrature of the outputs of two matched filters. In the absence of signal, the signal-to-noise ratio ρ

⁵⁸More precisely, if we have a time series with N samples, computing the integral which defines the scalar product has a computational cost O(N), and if one had to repeat it for all possible arrival times, the overall cost would be $O(N^2)$. With a single FFT, instead, the computational cost is $O(N \log N)$.

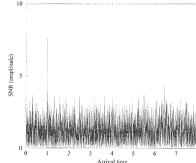


Fig. 7.14 The result of a simulation in which the signal due to a BH-BH coalescence, each with a mass of $10M_{\odot}$, at a distance of 150 Mpc, is injected into the noise of the VIRGO detector. The arrival time is located from the position of the spike in the Fourier transform (7.171), which here is at $t_* = 1$. (Courtesy of A. (7.173) Viceré.)

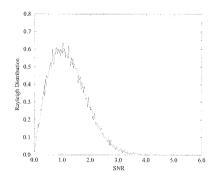


Fig. 7.15 The distribution of the signal-to-noise ratio, in the simulation of Fig. 7.14. (Courtesy of A. Viceré.)

⁵⁹It should also be observed that, for coalescing binaries, non-Gaussian noise should be much less important than for short bursts, since it should be much easier to have an impulsive disturbance that simulates a short burst, rather than a noise that lasts for about 15 minutes, simulating for all this time the behavior of a chirping signal.

⁶⁰Actually, the expression that we used for h_{+} and h_{\times} assumes a given choice of the axes with respect to which the plus and cross polarizations are defined, which is related to the orientation of the orbit, see page 296. Since a priori we do not know the orientation of the orbit, this will in general differ by an unknown angle ψ from the definition of axes that the experimenter uses to define the pattern functions. Correspondingly, the expressions for h_+ and h_{\times} must be rotated as in eqs. (7.24) and (7.25), so the function Q is actually $Q(\theta, \phi; \iota, \psi)$. However this ψ dependence, being an orthogonal transformation, does not affect that computation of $\langle |Q|^2 \rangle$ performed below.

is therefore a random variable which follows the Rayleigh distribution (7.87), while in the presence of signal it is a non-central χ^2 distribution with two degrees of freedom. Indeed, we see in Fig. 7.15 that, in the simulation of Fig. 7.14, ρ follows a Rayleigh distribution (except, of course, for the presence of the single spike with S/N = 8 at $t_* = 1$).⁵⁹

7.7.2 The sight distance to coalescing binaries

The Fourier transform of the chirp amplitude, to Newtonian order, has been computed in Problem 4.1, while the result in the restricted PN approximation, up to 2PN order, is given in eqs. (5.274) and (5.275). Then we find, for the Fourier transform of $h(t) = h_+ F_+ + h_\times F_\times$,

$$\tilde{h}(f) = \left(\frac{5}{6}\right)^{1/2} \frac{1}{2\pi^{2/3}} \frac{c}{r} \left(\frac{GM_c}{c^3}\right)^{5/6} f^{-7/6} e^{i\Psi} Q(\theta, \phi; \iota), \qquad (7.177)$$

where

$$Q(\theta, \phi; \iota) = F_{+}(\theta, \phi) \frac{1 + \cos^{2} \iota}{2} + i F_{\times}(\theta, \phi) \cos \iota.$$
 (7.178)

The phase Ψ is just the quantity denoted Ψ_+ in eq. (5.275), and the relative factor i between the two terms in Q is due to the fact that $\Psi_{\times} = \Psi_+ + (\pi/2)$.⁶⁰ Plugging this expression into eq. (7.51), we can write the signal-to-noise ratio for a coalescing binary as

$$\left(\frac{S}{N}\right)^2 = \frac{5}{6} \frac{1}{\pi^{4/3}} \frac{c^2}{r^2} \left(\frac{GM_c}{c^3}\right)^{5/3} |Q(\theta, \phi; \iota)|^2 \int_0^{f_{\text{max}}} df \, \frac{f^{-7/3}}{S_n(f)}, \quad (7.179)$$

where f_{max} is the value of the GW frequency when the inspiral phase terminates and the two stars merge. An estimate of f_{max} is $f_{\text{max}} = 2(f_s)_{\text{ISCO}}$, where $(f_s)_{\text{ISCO}}$ given in eq. (4.39), and the factor of 2 is valid as long as the emission is dominated by quadrupole radiation. For a wave coming from optimal direction (e.g. $F_+ = 1$ and $F_\times = 0$), and with optimal value of the inclination of the orbit ($\cos \iota = 1$), the function $Q(\theta, \phi; \iota) = 1$. However, a more appropriate reference value for $|Q(\theta, \phi; \iota)|^2$ is given by its average over all possible directions and inclinations. Using $\langle F_+^2 \rangle = \langle F_\times^2 \rangle = 1/5$ for interferometers (see Table 7.1), we get $\langle |Q(\theta, \phi; \iota)|^2 \rangle = (1/5)g(\iota)$, where $g(\iota)$ was defined in eq. (3.338) and its average over the inclination ι is 4/5, see eq. (4.10). Therefore

$$\langle |Q(\theta, \phi; \iota)|^2 \rangle^{1/2} = \frac{2}{5},$$
 (7.180)

where here $\langle \ldots \rangle$ denotes the average over the angles and over the inclination. Then we rewrite eq. (7.179) as

$$\frac{S}{N} = \frac{2}{5} \left(\frac{5}{6} \right)^{1/2} \frac{1}{\pi^{2/3}} \frac{c}{r} \left(\frac{GM_c}{c^3} \right)^{5/6} \frac{\langle |Q(\theta, \phi; \iota)|^2 \rangle^{1/2}}{(2/5)} \times \left[\int_0^{f_{\text{max}}} df \, \frac{f^{-7/3}}{S_n(f)} \right]^{1/2} .$$
(7.181)

This relation can be inverted to give the sight distance d_{sight} , i.e. the maximum distance r at which we can see a binary coalescence, once we have chosen a given threshold for S/N, ⁶¹ assuming an average direction and inclination,

$$d_{\text{sight}} = \frac{2}{5} \left(\frac{5}{6}\right)^{1/2} \frac{c}{\pi^{2/3}} \left(\frac{GM_c}{c^3}\right)^{5/6} \left[\int_0^{f_{\text{max}}} df \, \frac{f^{-7/3}}{S_n(f)}\right]^{1/2} (S/N)^{-1}.$$

(7.182)

We will see in Chapter 9 the numerical values that can be obtained for d_{sight} at existing and advanced interferometers.

It is instructive to verify from these expressions that, in order of magnitude, for a coalescing binary the matched filtering procedure gives a gain $\sim N_c^{1/2}$. To this end, we assume that S_n has a constant value S_0 between a minimum frequency f_0 and f_{max} , while it is essentially infinite for $f < f_0$. Then, neglecting all numerical factors (and using for simplicity units c = 1, and the notation $M = GM_c/c^3$), we can perform the integral in eq. (7.181), and we get

$$\frac{S}{N} \sim \frac{1}{r} M^{5/6} S_0^{-1/2} f_0^{-4/3} \,. \tag{7.183}$$

From eqs. (7.167) and (7.168) we see that the GW amplitude is of order

$$h_0 \sim \frac{1}{r} f_0^{2/3} M^{5/3} ,$$
 (7.184)

while, from eq. (4.23), the number of cycles spent in the interferometer bandwidth is

$$\mathcal{N}_c \sim M^{-5/3} f_0^{-5/3} \,. \tag{7.185}$$

Using eq. (7.184) to eliminate r from eq. (7.183), and eq. (7.185) to eliminate M, we get

$$\frac{S}{N} \sim \frac{h_0}{(f_0 S_0)^{1/2}} \, \mathcal{N}_c^{1/2} \,,$$
 (7.186)

which shows indeed that, in order of magnitude, the signal-to-noise ratio (in amplitude) is larger by a factor $\mathcal{N}_c^{1/2}$ than for a burst with a characteristic frequency f_0 (compare with eq. (7.107) with $\tau_g = 1/f_0$ and $f_{\text{max}} = f_0$). Of course, a more precise estimate requires the real form of $S_n(f)$, as well as the exact computation of the integral in eq. (7.181). This shows explicitly how the matched filtering procedure allows us to dig deeply into the noise floor, as we discussed already on page 344. Consider in fact the situation in which, after tracking the signal by $\mathcal{N}_c \gg 1$ cycles, we finally get S/N of order one, so we begin to see the signal. According to eq. (7.186), this means that $h_0/(f_0S_0)^{1/2}$ was of order $1/\mathcal{N}_c^{1/2}$. However, $h_0/(f_0S_0)^{1/2}$ is the "instantaneous" value of the signal-to-noise ratio, i.e. the value of S/N over a single cycle.

⁶¹Recall however from page 359 that the signal can combine with the noise either in a constructive or in a destructive way, so the output ρ of the interferometer is a random variable whose average is S/N and which follows, in the presence of signal, a non-central χ^2 distribution with two degrees of freedom. Therefore, at any distance r, there is a probability of missed detection, and the fact that a source is at $r < d_{sight}$ does not mean that it will be certainly detected. Conversely, there is also a nonzero probability that the signal from a source at $r > d_{\text{sight}}$ combines with the noise so that its S/N raises above the threshold.

⁶²The precise computation of the er-

rors on the parameters can be done us-

ing the explicit expression of the wave-

form to evaluate the Fisher information

matrix defined in eq. (7.74), and then

we can compute the errors on the pa-

rameters as in eq. (7.75). Using the

waveform with the post-Newtonian cor-

rections to the phase and assuming a

detection with S/N = 10 one finds

that, if one knew that the spins of

the star are negligible, then M_c could

indeed be measured with a precision

of 0.01-0.1%, while the reduced mass

 μ , which enters in the post-Newtonian

corrections, could be measured to 1%.

However, one in general has no a pri-

ori information on the spins, and the

measurements of masses and spins hap-

pen to be strongly correlated. This

degrades the accuracy on the mass

reconstruction, so one finally obtains

 $\Delta M_c/M_c \sim 0.1-1\%$ (which, however,

is still a quite remarkable accuracy) and

 $\Delta\mu/\mu\sim 10-15\%$ for NS-NS and NS-

BH binaries, or $\Delta\mu/\mu$ \sim 50% for BH-

BH binaries with typical BH masses of

order $10M_{\odot}$. Observe that, the larger

the mass of the stars, the smaller is the

number of cycles in the detector band-

width, since the coalescence takes place

earlier, see eq. (4.39), so the precision

in the reconstruction of the parameters

is less good. See Cutler and Flanagan

(1994) for details.

Therefore, the integrated signal-to-noise ratio provided by the matched filtering procedure can be of order one or larger, even when the instantaneous signal is deeply buried into the noise.

Finally, an important issue is the precision that can be obtained in the reconstruction of the source parameters. In particular the chirp mass M_c , that appears in the phase of the waveform, can be estimated very precisely, since the phase can be followed accurately for \mathcal{N}_c cycles. Thus, any mismatch ΔM_c between the true value of the source and the value used in our template will be amplified by a factor \mathcal{N}_c , and we could expect that

$$\frac{\Delta M_c}{M_c} \sim \frac{1}{\mathcal{N}_c} \,. \tag{7.187}$$

Given that at a ground-based interferometer \mathcal{N}_c can be of order 10^3-10^4 see eq. (4.23), this would give a rather remarkable accuracy $\Delta M_c/M_c \sim$ $10^{-4}-10^{-3}$. As for the reduced mass μ , it appears in the 1PN corrections to the phase, which are smaller by a factor $O(v^2/c^2)$ than the leading term, so it can be measured less precisely.⁶²

Stochastic backgrounds

In 1965 Penzias and Wilson discovered that the Universe is permeated by the Cosmic Microwave Background (CMB) electromagnetic radiation. This radiation is a relic of the early Universe, and the microwave photons that compose it decoupled from the primordial plasma about 3×10^5 years after the Big Bang, and since then they have been propagating essentially freely. This discovery, providing direct evidence for the Big Bang, was one of the most significant in the history of cosmology.

Since then, the CMB has been subject to deep investigations. We now know that its spectrum is a perfect black-body (in fact, the most perfect black-body spectrum existing in nature). This background is, to a first approximation, isotropic. The observation by the COBE satellite of temperature fluctuations over the sky, at the level $\Delta T/T \sim 10^{-5}$, has been one of the most important discoveries in cosmology in the last decades, and the detailed investigation of the multipole moments of these anisotropies by COBE and various other experiments, and particularly by WMAP, has opened up the field of precision cosmology.

There are good reasons to expect that the Universe is permeated also by a stochastic background of GWs generated in the early Universe. Furthermore, a stochastic background of GWs can also emerge from the incoherent superposition of a large number of astrophysical sources, too weak to be detected separately, and such that the number of sources that contribute to each frequency bin is much larger than one.

The mechanisms that can lead to the production of stochastic GW backgrounds in cosmology and in astrophysics will be examined in detail in Vol. 2. Here we discuss how to characterize such a background in general, and what are the optimal strategies for its detection.

Characterization of stochastic backgrounds

Using the plane wave expansion (1.58), we can write

$$h_{ij}(t, \mathbf{x}) = \sum_{A=+,\times} \int_{-\infty}^{\infty} df \int d^2 \hat{\mathbf{n}} \, \tilde{h}_A(f, \hat{\mathbf{n}}) \, e_{ij}^A(\hat{\mathbf{n}}) \, e^{-2\pi i f (t - \hat{\mathbf{n}} \cdot \mathbf{x}/c)} \,.$$

We work in the TT gauge, so $h_i^i = 0$ and $\partial^j h_{ij} = 0$. The tensors $e_{ij}^A(\hat{\mathbf{n}})$ are given in eq. (1.54). A stochastic background is a superposition of waves with all possible propagation directions $\hat{\mathbf{n}}$, therefore the indices i, j above take the values 1, 2, 3, contrary to the case of the GWs emitted from a single far source, where we could label the GW in the TT gauge as h_{ab} with a, b, taking the values 1, 2 and labeling the two directions in the transverse plane. A stochastic background is defined by the fact that the amplitudes $h_A(f, \hat{\mathbf{n}})$ are random variables, characterized statistically by their ensemble averages.⁶³

We will make the following assumptions on stochastic backgrounds of GWs.

- The background is stationary. This means that all correlators depend only on time differences, and not on the absolute values of time. So, for instance, the two-point correlator $\langle h_A(t)h_{A'}(t')\rangle$ can depend only on t-t', and not separately on t and t'. In Fourier space, this means that $\langle h_A^*(f)h_{A'}(f')\rangle$ must be proportional to $\delta(f-f')$. This assumption is certainly justified. For a background created in cosmological epochs, the typical time-scale on which it can change substantially is of the order of the age of the Universe (for instance, its spectrum changes because it is redshifted). During the duration of the experiment, which is at most a few years, it is very difficult to imagine that the properties of the background could change appreciably.⁶⁴
- The background is Gaussian. This means that all N-point correlators are reduced to sum and products of the two-point correlator $\langle h_A(t)h_{A'}(t')\rangle$ (and of the vacuum expectation value $\langle h_A\rangle$ that however, as we have seen, can be set to zero). Gaussianity is rooted in the central limit theorem, that states that the sum of a large number of independent events produces a Gaussian stochastic process, whatever the probability distribution of the individual events. This assumption is therefore expected to hold to a very good accuracy for cosmological backgrounds. It would not hold for astrophysical backgrounds, if the number of sources that contribute is not that large, and we are on the verge of distinguishing the individual contributions. In this case, further information can be extracted from the higher-point correlators.
- The stochastic background is isotropic. Experience with CMB indicates that the early Universe was highly isotropic and, for the photons, temperature fluctuations across the sky are at the level $\Delta T/T \sim 10^{-5}$. It is reasonable to expect that a stochastic background of GWs of cosmological will also be in a first approximation

 $^{63}\mathrm{An}$ ensemble average is the average over many copies of the system. Our system is in this case the Universe and we do not have many copies of it! Of course, the ergodic assumptions must be used here, and the ensemble average is replaced by a temporal average. compare with Note 3 on page 337.

⁶⁴Stationarity also implies that $\langle h_A(t) \rangle$ is a constant so, even if it were nonzero, it would just contribute to the vacuum energy density. As far as we are interested in GWs, that is in the time-dependent part, we can therefore set $\langle h_A \rangle = 0$.

isotropic. Of course, after a first detection of a GW background, it will be extremely interesting to investigate its anisotropies and therefore to give up this assumption. In particular, in a cosmological background we must expect a dipole term, dominated by the Earth motion in the rest frame of the CMB, while higher multipoles could give extremely interesting information on the early Universe.

We might have to give up completely the assumption of isotropy when we study stochastic backgrounds of astrophysical origin. In particular a background of galactic origin will not be isotropic, but rather it will be more intense when we look in the direction of the galactic plane, just as the electromagnetic background due to galactic sources gives its characteristic appearance to the Milky Way. We will in fact discuss in Vol. 2 an example of this type, the background created by galactic white dwarf binaries.

Waves coming from different directions should be uncorrelated, so $\langle \tilde{h}_A^*(f,\hat{\mathbf{n}})\tilde{h}_{A'}(f',\hat{\mathbf{n}}')\rangle$ should be proportional to a Dirac delta over the two-sphere, defined as

$$\delta^{2}(\hat{\mathbf{n}}, \hat{\mathbf{n}}') = \delta(\phi - \phi')\delta(\cos\theta - \cos\theta'), \qquad (7.189)$$

where (θ, ϕ) are the polar angles that define $\hat{\mathbf{n}}$. Isotropy implies that the proportionality constant must be independent of $\hat{\mathbf{n}}$.

• Finally, we assume that the background is unpolarized, as it is natural both in a cosmological context and if it is the result of the superposition of many different astrophysical sources. This means that $\langle \tilde{h}_A^*(f,\hat{\mathbf{n}})\tilde{h}_{A'}(f',\hat{\mathbf{n}}')\rangle$ must be proportional to $\delta_{AA'}$ and the proportionality coefficient must be independent of the polarization index A.

Under these assumptions, a stochastic background of GWs is uniquely characterized by a single function $S_h(f)$, defined by

$$\langle \tilde{h}_A^*(f, \hat{\mathbf{n}}) \tilde{h}_{A'}(f', \hat{\mathbf{n}}') \rangle = \delta(f - f') \frac{\delta^2(\hat{\mathbf{n}}, \hat{\mathbf{n}}')}{4\pi} \delta_{AA'} \frac{1}{2} S_h(f) .$$
 (7.190)

The function $S_h(f)$ is called the spectral density of the stochastic background, in analogy with the spectral density of the noise defined in Section 7.1. Just as for the noise spectral density, we use the convention that $S_h(f)$ is single-sided. It has dimensions Hz^{-1} and satisfies $S_h(f) = S_h(-f)$. The factor $1/(4\pi)$ in eq. (7.190) is a choice of normalization such that

$$\int d^2\hat{\mathbf{n}} d^2\hat{\mathbf{n}}' \langle \tilde{h}_A^*(f,\hat{\mathbf{n}}) \tilde{h}_{A'}(f',\hat{\mathbf{n}}') \rangle = \delta(f - f') \delta_{AA'} \frac{1}{2} S_h(f).$$
 (7.191)

where, as usual, $d^2\hat{\mathbf{n}} = d\cos\theta d\phi$. We see that the factor 1/2 in the definition of $S_h(f)$ has been inserted so that $S_h(f)$ is normalized in the same way as the single-sided spectral density of the noise, see eq. (7.6).

Using eqs. (7.188) and (7.190), as well as $\sum_{A} e_{ij}^{A} e_{ij}^{A} = 4$, which follows from the normalization (1.55) of the polarization tensor e_{ij}^{A} , we get

$$\langle h_{ij}(t)h^{ij}(t)\rangle = 4\int_0^\infty df \, S_h(f) \,. \tag{7.192}$$

The sum over i, j is understood, and $h_{ij}(t) = h_{ij}(t, \mathbf{x} = 0)$. The spectral density of the signal, $S_h(f)$, is the quantity that allows us to perform a direct comparison with the noise in a detector, which is characterized by $S_n(f)$. However, to have a physical understanding it is much more convenient to think in terms of the energy density carried by the stochastic background. According to eq. (1.135), this is related to h_{ij} by

$$\rho_{\rm gw} = \frac{c^2}{32\pi G} \langle \dot{h}_{ij} \dot{h}^{ij} \rangle \,. \tag{7.193}$$

In cosmology there is a very natural unit of energy density, that is, the energy density needed for closing the Universe. This critical energy density is

$$\rho_c = \frac{3c^2 H_0^2}{8\pi G} \,, \tag{7.194}$$

where H_0 is the present value of the Hubble expansion rate. As we mentioned on page 194, the value of H_0 is usually written as $H_0 = h_0 \times 100 \,\mathrm{km} \,\mathrm{s}^{-1} \,\mathrm{Mpc}^{-1}$, where h_0 parametrizes the existing experimental uncertainty and is called the normalized Hubble expansion rate. The most recent determinations give $h_0 = 0.73(3)$. Numerically,

$$\rho_c \simeq 1.688 \times 10^{-8} h_0^2 \quad \text{erg cm}^{-3}.$$
(7.195)

Normalizing ρ_{gw} to ρ_c , the intensity of a stochastic background of gravitational waves can be characterized by the dimensionless quantity

$$\Omega_{\rm gw} \equiv \frac{\rho_{\rm gw}}{\rho_c} \,. \tag{7.196}$$

Using eqs. (7.192) and (7.193), the energy density can be written as an integral over $d \log f$ of some spectral density, that we denote by $d\rho_{\rm gw}/d \log f$,

$$\rho_{\rm gw} \equiv \int_{f=0}^{f=\infty} d(\log f) \, \frac{d\rho_{\rm gw}}{d\log f} \,. \tag{7.197}$$

We also define

$$\Omega_{\rm gw}(f) \equiv \frac{1}{\rho_c} \frac{d\rho_{\rm gw}}{d\log f}, \qquad (7.198)$$

so $\Omega_{\rm gw}$ in eq. (7.196) is related to $\Omega_{\rm gw}(f)$ by ⁶⁶

$$\Omega_{\rm gw} = \int_{f=0}^{f=\infty} d(\log f) \ \Omega_{\rm gw}(f). \tag{7.199}$$

⁶⁵There is a slight abuse of notation here. Of course $\rho_{\rm gw}$, on the left-hand side of eq. (7.197), is independent of the frequency, so its derivative with respect to f, or to $\log f$, vanishes. On the right-hand side, $d\rho_{\rm gw}/d\log f$ is not the derivative of $\rho_{\rm gw}$ with respect to $\log f$, but just a notation for the spectral density of $\rho_{\rm gw}$, which stresses that it is the energy density contained in a logarithmic interval of frequency.

⁶⁶Here again there is a slight ambiguity in the notation, because we use the same symbol $Ω_{gw}$ for the normalized energy density, on the left-hand side of eq. (7.199), and for its spectral density, on the right-hand side. This notation is however standard in the GW literature, and we will conform to it.

 67 Unfortunately, we sometime use h_0 also to denote a GW amplitude. Since the reduced Hubble constant h_0 will

only appear in the combination $h_0^2 \Omega_{\rm gw}$,

no confusion is possible.

The fact that we consider the energy per unit logarithmic interval of frequency, $d\rho_{\rm gw}/d\log f$, rather than $d\rho_{\rm gw}/df$, is useful because in this way $\Omega_{\rm gw}(f)$ is dimensionless.

Even if the experimental error on the Hubble expansion rate is becoming smaller and smaller (just a few years ago values of h_0 between 0.4 and 1 where considered possible), still it is not very convenient to normalize $\rho_{\rm gw}$ to a quantity, $\rho_{\rm c}$, which is uncertain: this uncertainty would appear in all the subsequent formulas, although it has nothing to do with the uncertainties on the GW background itself. Therefore, one rather characterizes the stochastic GW background with the quantity $h_0^2\Omega_{\rm gw}(f)$, which is independent of h_0 .⁶⁷

We now compute the relation between $S_h(f)$ and $h_0^2\Omega_{\rm gw}(f)$. As discussed in Section 1.4.3, the brackets in eq. (7.193) denote a time average. However (under the ergodic assumption, see Notes 3 and 63), this is just the ensemble average used above. We can then substitute the plane wave expansion (7.188) into eq. (7.193), and compute the ensemble average using eq. (7.190). The result is

$$\rho_{\rm gw} = \frac{c^2}{8\pi G} \int_{f=0}^{f=\infty} d(\log f) \ f(2\pi f)^2 S_h(f) \,. \tag{7.200}$$

Comparing with the definition (7.197) we get

$$\frac{d\rho_{\rm gw}}{d\log f} = \frac{\pi c^2}{2G} f^3 S_h(f) \,, \tag{7.201}$$

and

$$\Omega_{\rm gw}(f) = \frac{4\pi^2}{3H_0^2} f^3 S_h(f).$$
(7.202)

Finally, it is interesting to express $h_0^2\Omega_{\rm gw}(f)$ in terms of the number of gravitons per cell of the phase space, $n(\mathbf{x}, \mathbf{k})$. For an isotropic stochastic background $n(\mathbf{x}, \mathbf{k}) = n_f$ depends only on the frequency (which is related to the momentum \mathbf{k} by $|\mathbf{k}| = \hbar \omega/c = 2\pi \hbar f/c$), and not on the direction $\hat{\mathbf{k}}$. Then, writing $d^3k = |\mathbf{k}|^2 d|\mathbf{k}| d\Omega \to 4\pi (2\pi \hbar/c)^3 f^2 df$, and $df = f d \log f$, and considering that a graviton of frequency f carries an energy $\hbar \omega = \hbar (2\pi f)$, we have

$$\rho_{gw} = 2 \int \frac{d^3k}{(2\pi\hbar)^3} \, \hbar(2\pi f) \, n_f$$

$$= \frac{16\pi^2\hbar}{c^3} \int_0^\infty d(\log f) \, f^4 n_f \,, \tag{7.203}$$

where the factor of 2 in front of the integral is due to the two helicity states of the graviton. Therefore

$$\frac{d\rho_{\rm gw}}{d\log f} = \frac{16\pi^2\hbar}{c^3} \, n_f f^4 \,, \tag{7.204}$$

and

$$h_0^2 \Omega_{\rm gw}(f) \simeq 3.6 \left(\frac{n_f}{10^{37}}\right) \left(\frac{f}{1 \text{kHz}}\right)^4.$$
 (7.205)

As we will see in Vol. 2, this equation is useful in particular when one computes the production of a stochastic background of GWs due to amplification of vacuum fluctuations, since this computation gives directly n_f .

7.8.2 SNR for single detectors

The comparison of eqs. (7.6) and (7.191) makes it clear that an isotropic stochastic background of GWs is seen in a detector as an additional source of noise. This poses an important conceptual problem in the identification of a stochastic GW background. In practice what will happen is that, after a careful modeling of the detector and of its noise sources, one would expect to have a certain value of the spectral density of the noise, $S_n(f)$. When the detector is turned on, one measures $\langle s^2(t) \rangle$, where as usual s(t) = n(t) + h(t), with n(t) the noise and h(t)the response of the detector to a GW signal. If one observes that $\langle s^2(t) \rangle$ is larger than expected, the crucial problem is how to tell whether this is really due to a GW background or, more trivially, to some source of noise that has not been adequately accounted for when estimating $S_n(f)$. Similar problems were faced in the discovery of the cosmic microwave background: Penzias and Wilson found an excess noise in their antenna (a horn reflector that was meant for satellite communications) and worked hard for one year in order to exclude all possible sources of terrestrial and astrophysical noise, before writing a short paper with the modest title "A Measurement of Excess Antenna Temperature at 4080 Mc/s", and concluding "From a combination of the above, we compute the remaining unaccounted-for antenna temperature to be $3.5 \pm 1.0~K$ at 4080 Mc/s".

To cope with this problem, it is clear that in the search for stochastic backgrounds of GWs with a single detector one must set at least a relatively high threshold on the signal-to-noise ratio; for instance, a signal-to-noise ratio S/N=5 on the amplitude could be a typical choice (while lower values of S/N could be used for the only purpose of putting upper bounds). Further handles could come from an anisotropy of the stochastic GW background, if it is due to unresolved galactic sources, since this would produce a sidereal time modulation due to the motion of the detector. Another handle is the possibility that the dependence of the excess noise on the frequency is found to be in agreement with some theoretical prediction from a given cosmological or astrophysical mechanism.

To compute the minimum value of $h_0^2\Omega_{\rm gw}$ that can be measured at a given S/N, we observe that, if there is no signal, we have (see eq. (7.12))

$$\langle s^2(t)\rangle = \langle n^2(t)\rangle = \int_0^\infty df \, S_n(f) \,,$$
 (7.206)

while, if a stochastic GW background is present, there is also a contribution from h(t). For each propagation direction $\hat{\mathbf{n}}$ we can write $h(t) = h_+ F_+ + h_\times F_\times$, and therefore, taking the ensemble average and

averaging also over $\hat{\mathbf{n}}$ and over the polarization angle ψ ,

$$\int \frac{d^2 \hat{\mathbf{n}}}{4\pi} \frac{d\psi}{2\pi} \langle h^2 \rangle = \left(\int \frac{d^2 \hat{\mathbf{n}}}{4\pi} \frac{d\psi}{2\pi} F_+^2 \right) \langle h_+^2 + h_\times^2 \rangle, \tag{7.207}$$

where we used the fact that the angular averages of F_+^2 and of F_\times^2 are equal, see eq. (7.35). For an isotropic background, the ensemble average $\langle h^2 \rangle$ that appears on the left-hand side of eq. (7.207) is independent of the angles $\hat{\bf n}$ and ϕ , so the angular average gives one. The term on the right-hand side of eq. (7.207), instead, can be written in terms of $S_h(f)$ using eq. (7.192) and observing that, for any given propagation direction, we have $h_{ij}h^{ij}=2(h_+^2+h_\times^2)$. Then

$$\langle h^2 \rangle = 2 \langle F_+^2 \rangle \int_0^\infty df \, S_h(f) \,, \tag{7.208}$$

where, with an abuse of notation, the brackets in $\langle h^2 \rangle$ denote the ensemble average while the brackets in $\langle F_+^2 \rangle$ denotes the average over $d^2\hat{\mathbf{n}}$ and $d\psi$. In eq. (7.37) we have defined the angular efficiency factor $F = \langle F_+^2 \rangle + \langle F_\times^2 \rangle = 2 \langle F_+^2 \rangle$, whose value for various detectors are given in Table 7.1. In particular, F = 2/5 for interferometers and F = 8/15 for resonant bars. Then

$$\langle h^2(t) \rangle = F \int_0^\infty df \, S_h(f) \,.$$
 (7.209)

So, in the presence of signal,

$$\langle s^{2}(t)\rangle = \langle n^{2}(t)\rangle + \langle h^{2}(t)\rangle$$

$$= \int_{0}^{\infty} df \left[S_{n}(f) + FS_{h}(f)\right]. \tag{7.210}$$

Therefore, if a stochastic background is present, one simply observes that $\langle s^2(t) \rangle$ is higher than the value expected from the noise, everywhere or just in some frequency range. More precisely, we can compare the output with the expected value of $S_n(f)$ in each frequency bin (with bins of size $\Delta f = 1/T$ after an observation time T). To take the binning into account, we replace

$$\int S_h(f)df \to \sum_i S_h(f_i)\Delta f, \qquad (7.211)$$

and

$$\int S_n(f)df \to \sum_i S_n(f_i)\Delta f \tag{7.212}$$

The signal-to-noise ratio in each bin is therefore 68

$$\left(\frac{S}{N}\right)^2 = F \frac{S_h(f_i)\Delta f}{S_n(f_i)\Delta f}$$

$$= F \frac{S_h(f_i)}{S_n(f_i)}.$$
(7.213)

Of course the integration time T, which enters through Δf , canceled in eq. (7.213). Increasing the integration time, we decrease the size of the bins and therefore the noise in each bin, but we equally decrease the signal present in each bin. Therefore, in a single detector, as far as the signal-to-noise ratio is concerned, there is no gain in integrating the signal in time. Either the signal stands out immediately as soon as we switch on the detector, or it will always remain below the noise. If however the signal stands out, integrating it for a longer time we get a more detailed resolution of its frequency dependence.

In conclusion, the minimum value of $S_h(f)$ measurable with a single detector having a noise spectral density $S_n(f)$, at a given level S/N of signal-to-noise ratio in amplitude, is

$$[S_h(f)]_{\min} = S_n(f) \frac{(S/N)^2}{F},$$
 (7.214)

and correspondingly the minimum detectable value of $\Omega_{\rm gw}$ is

$$[\Omega_{\rm gw}(f)]_{\rm min} = \frac{4\pi^2}{3H_0^2} f^3 S_n(f) \frac{(S/N)^2}{F}.$$
 (7.215)

A very important feature of this expression is the factor f^3 . It tells us that, if one is able to reach a given level in $S_n(f)$ at low frequencies, it will be possible to reach a much better sensitivity in $\Omega_{\rm gw}(f)$ compared to what can be obtained with a similar value of $S_n(f)$ at high frequencies. Of course, the experimental problems that one has to solve in order to reach a given value of $S_n(f)$ depend very strongly on the frequency f. However, at $f = 10^{-3}$ Hz, the space detector LISA aims at reaching a strain sensitivity $S_n^{1/2}(f) = 4 \times 10^{-21} \,\mathrm{Hz}^{-1/2}$, while a ground-based interferometer at $f = 10^2 \text{ Hz has } S_n^{1/2}(f) = 4 \times 10^{-23} \text{ Hz}^{-1/2}$, as we will see in Chapter 9. Therefore, moving from $f = 10^2$ Hz to f = 10^{-3} Hz, we lose only four orders in magnitude in $S_n(f)$, but we gain a factor $(10^2/10^{-3})^3 = 10^{15}$ thanks to f^3 . Therefore, it is much easier to reach a small level for $[\Omega_{gw}(f)]_{min}$ at low f rather than at high f. The other important question is in what frequency range should we expect that cosmological or astrophysical mechanisms produce an interesting value for $\Omega_{\rm gw}(f)$. As we will see in Vol. 2, there is a large variety of possible mechanisms, which can produce stochastic GW backgrounds everywhere from $f = 10^{-18}$ Hz up to $f = 10^{9}$ Hz. Their detection is therefore easier when they are large at low frequencies, since then comparatively high value of the noise spectral density $S_n(f)$ can be overcompensated by the factor f^3 , and becomes more and more difficult as we go to high frequencies. Numerically, with normalizations useful for LISA, eq. (7.215) gives

$$[h_0^2 \Omega_{\rm gw}(f)]_{\rm min} = 1.1 \times 10^{-12} \left(\frac{f}{1 \,\text{mHz}}\right)^3 \left(\frac{S_n^{1/2}}{4 \times 10^{-21} \,\text{Hz}^{-1/2}}\right)^2 \times \left(\frac{1/\sqrt{5}}{F}\right) \left(\frac{S/N}{5}\right)^2.$$
 (7.216)

 68 Observe that here S/N is the signal-to-noise ratio with respect to the GW amplitude, just as we have defined it for bursts, coalescence and periodic signals. For stochastic backgrounds, what is actually measured is an energy density, and it make sense to introduce the signal-to-noise ratio with respect to the energy density, which is quadratic in the amplitude. If one prefers to reserve the notation S/N for the signal-to-noise ratio in energy, then on the left-hand side of eq. (7.213) one must write S/N rather than $(S/N)^2$.

Using normalization factors appropriate for ground-based interferometers, we rather have

$$[h_0^2 \Omega_{\rm gw}(f)]_{\rm min} = 0.12 \left(\frac{f}{100 \,\text{Hz}}\right)^3 \left(\frac{S_n^{1/2}}{4 \times 10^{-23} \,\text{Hz}^{-1/2}}\right)^2 \times \left(\frac{2/5}{F}\right) \left(\frac{S/N}{5}\right)^2. \tag{7.217}$$

In both cases we used a rather high value of the signal-to-noise ratio as a reference value, S/N=5, according to the discussion above. The huge difference between the value $h_0^2\Omega_{\rm gw}\sim 10^{-12}$ in eq. (7.216) and the value $h_0^2\Omega_{\rm gw}\sim 0.1$ in eq. (7.217) is due to the fact that LISA can reach a value of S_n not far from that of ground-based interferometers, at a much lower frequency.

As we will see in Vol. 2, no cosmological or astrophysical background of GW is expected to exceed $h_0^2\Omega_{\rm gw}(f)\sim 10^{-5}$, independently of the frequency. Therefore eqs. (7.216) and (7.217) tell us that LISA has an extremely good sensitivity for stochastic backgrounds of GWs, while ground-based interferometers, used as single detectors, do not reach an interesting level for stochastic backgrounds. However, having at our disposal more than one ground-based detector (interferometers or bars) we can correlate their outputs, and the sensitivity improves dramatically, as we discuss in the next section.

7.8.3 Two-detector correlation

Optimal signal-to-noise ratio

With a single detector, it is impossible to adapt to stochastic backgrounds the matched filtering technique that we studied in Section 7.3. The reason is that, to perform the matched filtering, we need to know the form of the signal, but for stochastic backgrounds the GW signal h(t) is an unpredictable randomly fluctuating quantity, just like the noise n(t). However, if we have two detectors, we can use a modified form of matched filtering in which, rather than trying to match the output of a single detector to a predetermined signal h(t), we match the output of one detector to the output of the other.

To implement this idea we proceed as follows. We write the output $s_k(t)$ of the k-th detector as $s_k(t) = h_k(t) + n_k(t)$, where k = 1, 2 labels the detector. Observe that the scalar output $h_k(t)$ depends in general on the detector, because different detectors can have a different location and/or a different orientation and therefore a different pattern function. We are interested in the situation in which the GW signal $h_k(t)$ is much smaller than the noise $n_k(t)$, which is the realistic situation for all ground-based detectors, as we have seen in the previous section. Multiplying both sides of eq. (7.188) by the detector tensor D^{ij} and

using eq. (7.21), we can write the GW signal h_k in the k-th detector as

$$h_k(t, \mathbf{x}_k) = \sum_{A=+,\times} \int_{-\infty}^{\infty} df \int d^2 \hat{\mathbf{n}} \ \tilde{h}_A(f, \hat{\mathbf{n}}) e^{-2\pi i f(t-\hat{\mathbf{n}} \cdot \mathbf{x}_k/c)} F_k^A(\hat{\mathbf{n}}),$$

$$(7.218)$$

where F_k^A are the pattern functions of the k-th detector and \mathbf{x}_k is its location. As always, the size of the detector is taken to be much smaller than λ , so we can neglect the spatial variation of the GW over the extension of the detector. Passing to the Fourier transform, we have

$$\tilde{h}_k(f) = \sum_{A=+,\times} \int d^2 \hat{\mathbf{n}} \, \tilde{h}_A(f, \hat{\mathbf{n}}) e^{2\pi i f \hat{\mathbf{n}} \cdot \mathbf{x}_k/c} F_k^A(\hat{\mathbf{n}}) \,, \tag{7.219}$$

where we denote $\tilde{h}_k(f, \mathbf{x}_k)$ simply as $\tilde{h}_k(f)$. To correlate the outputs $s_1(t)$ and $s_2(t)$ of the two detectors we define

$$Y = \int_{-T/2}^{T/2} dt \int_{-T/2}^{T/2} dt' \, s_1(t) s_2(t') Q(t - t') \,, \tag{7.220}$$

where T is the total observation time (e.g. one year) and Q a real filter function, analogous to the function K(t) in Section 7.3. Y is our signal, and we want to maximize its signal-to-noise ratio.

We limit ourselves to functions Q(t-t') that fall rapidly to zero for large |t-t'|. Passing to the Fourier transforms, we get

$$Y = \int_{-\infty}^{+\infty} df df' df'' \, \delta_T(f - f'') \delta_T(f' - f'') \tilde{s}_1^*(f) \tilde{s}_2(f') \tilde{Q}(f'') \,, \quad (7.221)$$

where

$$\delta_T(f) \equiv \int_{-T/2}^{T/2} dt \, e^{i2\pi f t}$$

$$= \frac{\sin(\pi f T)}{\pi f}, \qquad (7.222)$$

and becomes a delta function in the limit $fT \to \infty$. Even on a relatively short stretch of data with, say, $T = 10^3$ s, at f = 10 Hz we have $fT = 10^4$. Over the whole useful bandwidth of ground-based detectors we can therefore replace $\delta_T(f)$ by a Dirac delta, and eq. (7.220) becomes

$$Y \simeq \int_{-\infty}^{+\infty} df \, \tilde{s}_1^*(f) \tilde{s}_2(f) \tilde{Q}(f) \,. \tag{7.223}$$

Recall that, in the signal-to-noise ratio S/N, S is defined as the ensemble average value of Y when the signal is present, while N is the rms value of Y when the signal is absent. Then, assuming that the noise in the two detectors are not correlated (and averaging also over the polarization angle ψ),

$$S = \int_{-\infty}^{+\infty} df \, \langle \tilde{h}_1^*(f) \tilde{h}_2(f) \rangle \, \tilde{Q}(f)$$

Using eq. (7.190), together with $\delta(0) = \int_{-T/2}^{T/2} dt = T$, this becomes

$$S = \frac{T}{2} \int_{-\infty}^{\infty} df \, S_h(f) \Gamma(f) \tilde{Q}(f) , \qquad (7.225)$$

where we have defined

$$\Gamma(f) \equiv \int \frac{d^2 \hat{\mathbf{n}}}{4\pi} \int \frac{d\psi}{2\pi} \left[\sum_{A} F_1^A(\hat{\mathbf{n}}) F_2^A(\hat{\mathbf{n}}) \right] \exp\left\{ i2\pi f \hat{\mathbf{n}} \cdot \frac{\Delta \mathbf{x}}{c} \right\},$$
(7.226)

and $\Delta \mathbf{x} = \mathbf{x}_2 - \mathbf{x}_1$ is the separation between the two detectors. The function Γ in called the (unnormalized) overlap reduction function. It takes into account the fact that the two detectors can see a different gravitational signal, either because they are at different location or because they have a different angular sensitivity.

The difference in location is reflected in the exponential factor. In particular, if $2\pi f \Delta x/c \gg 1$, i.e. if the separation $\Delta x \gg \lambda$, this exponential is rapidly oscillating and suppresses strongly the correlation. This reflect the fact that, when $\Delta x \gg \lambda$, the two detectors are experiencing GW signals that are uncorrelated.

The different angular sensitivity of the two detectors is instead reflected in the term $\sum_A F_1^A(\hat{\mathbf{n}}) F_2^A(\hat{\mathbf{n}})$. It is also useful to introduce the quantity

$$F_{12} \equiv \int \frac{d^2 \hat{\mathbf{n}}}{4\pi} \left. \int \frac{d\psi}{2\pi} \left. \sum_{A} F_1^A(\hat{\mathbf{n}}) F_2^A(\hat{\mathbf{n}}) \right|_{\text{aligned}}, \tag{7.227}$$

where the subscript means that we must compute F_{12} taking the two detectors to be at the same location and oriented one relative to the other so that the quantity F_{12} is maximized.⁶⁹ Observe that, if the two detectors are of the same type, e.g. two interferometers or two cylindrical bars, F_{12} is the same as the constant F defined in eq. (7.37). The (normalized) overlap reduction function $\gamma(f)$ is defined as

$$\gamma(f) = \frac{\Gamma(f)}{F_{12}}. (7.228)$$

For instance, for the correlation between two interferometers, $F_{12} = 2/5$. The factor F_{12} takes into account the reduction in sensitivity due to the pattern functions, already present in the case of one interferometer, and therefore $\gamma(f)$ separately takes into account the effect of the separation $\Delta \mathbf{x}$ between the interferometers, and of their relative orientation. With

this definition, $\gamma(f) = 1$ if the separation $\Delta x = 0$ and if the detectors are perfectly aligned. However, the use of $\Gamma(f)$ is more convenient when we want to write equations that hold independently of what detectors (interferometers, bars, or spheres) are used in the correlation.

We now find the optimal choice of the filter function $\tilde{Q}(f)$ that maximizes the signal-to-noise ratio. We need to compute

$$N^{2} = \left[\langle Y^{2} \rangle - \langle Y \rangle^{2} \right]_{h=0}$$

$$= \int_{-\infty}^{\infty} df df' \, \tilde{Q}(f) \tilde{Q}^{*}(f')$$

$$\times \left[\langle \tilde{n}_{1}^{*}(f) \tilde{n}_{2}(f) \tilde{n}_{1}(f') \tilde{n}_{2}^{*}(f') \rangle - \langle \tilde{n}_{1}^{*}(f) \tilde{n}_{2}(f) \rangle \langle \tilde{n}_{2}^{*}(f') \tilde{n}_{1}(f') \rangle \right].$$

$$(7.229)$$

If the noise in the two detectors are uncorrelated, the mixed correlator $\langle \tilde{n}_1^*(f)\tilde{n}_2(f)\rangle$ vanishes, so the second term in brackets is zero, while the first factorizes $\langle \tilde{n}_1^*(f)\tilde{n}_2(f)\tilde{n}_1(f')\tilde{n}_2^*(f')\rangle = \langle \tilde{n}_1^*(f)\tilde{n}_1(f')\rangle \langle \tilde{n}_2(f)\tilde{n}_2^*(f')\rangle$. Then we get

$$N^{2} = \int_{-\infty}^{\infty} df df' \, \tilde{Q}(f) \tilde{Q}^{*}(f') \langle \tilde{n}_{1}^{*}(f) \tilde{n}_{1}(f') \rangle \langle \tilde{n}_{2}^{*}(f') \tilde{n}_{2}(f) \rangle . \tag{7.230}$$

Using

$$\langle \tilde{n}_k^*(f)\tilde{n}_k(f')\rangle = \delta(f - f')\frac{1}{2}S_{n,k}(f), \qquad (7.231)$$

where $S_{n,k}(f)$ is the noise spectral density of the k-th detector, and using $\delta(0) = T$, we finally get

$$N^{2} = \frac{T}{4} \int_{-\infty}^{\infty} df \, |\tilde{Q}(f)|^{2} S_{n}^{2}(f) \,, \tag{7.232}$$

where we have defined the combined noise spectral density

$$S_n(f) = \left[S_{n,1}(f) S_{n,2}(f) \right]^{1/2}. \tag{7.233}$$

Equations (7.225) and (7.232) show the same crucial feature that we already observed when we discussed the matched filtering for periodic signals: the signal S increase linearly with the observation time T, while the noise N increases only as $T^{1/2}$. Therefore, the signal-to-noise ratio increases with the observation time as $T^{1/2}$. Putting together eqs. (7.225) and (7.232) we have

$$\frac{S}{N} = T^{1/2} \frac{\int_{-\infty}^{\infty} df \, S_h(f) \Gamma(f) \tilde{Q}(f)}{\left[\int_{-\infty}^{\infty} df \, |\tilde{Q}(f)|^2 S_n^2(f) \right]^{1/2}}.$$
 (7.234)

We can now find the filter function $\tilde{Q}(f)$ that maximizes S/N. The procedure is analogous to what we have already done between eqs. (7.45) and (7.51). For any two complex functions A(f), B(f) we define the positive definite scalar product

$$(A,B) = \int_{-\infty}^{\infty} df \, A^*(f)B(f)S_n^2(f).$$
 (7.235)

 69 For two detectors of the same type this means to orient them in the same way, so in a two-interferometer correlation the arms are taken to be along the x and y axes for both interferometers, and for a two-bar correlation the longitudinal axes of the bars are taken parallel to each other. For the correlation between a bar and an interferometer, we see from the form of the pattern functions given in Table 7.1 that the optimal correlation is obtained aligning the longitudinal axis of the bar with one of the arms of the interferometer.

Then eq. (7.234) can be rewritten as

$$\frac{S}{N} = T^{1/2} \frac{\left(\tilde{Q}, \Gamma S_h / S_n^2\right)}{(\tilde{Q}, \tilde{Q})^{1/2}}.$$
 (7.236)

As we already discussed below eq. (7.47), this expression is maximized choosing

$$\tilde{Q}(f) = \text{const.} \frac{\Gamma(f)S_h(f)}{S_n^2(f)}.$$
(7.237)

It is important to observe that the optimal filter depends on the signal that we are looking for, since $S_h(f)$ enters eq. (7.237). Plugging eq. (7.237) into eq. (7.236) we find the optimal signal-to-noise ratio,

$$\frac{S}{N} = T^{1/2} \left(\frac{\Gamma S_h}{S_n^2}, \frac{\Gamma S_h}{S_n^2} \right)^{1/2} ,$$
 (7.238)

⁷⁰Observe that, for periodic signals or, writing explicitly the scalar product, ⁷⁰

$$\frac{S}{N} = \left[2T \int_0^\infty df \, \Gamma^2(f) \frac{S_h^2(f)}{S_n^2(f)} \right]^{1/2} . \tag{7.239}$$

In particular, for a two-interferometer correlation, $\Gamma(f) = (2/5)\gamma(f)$ and

$$\left(\frac{S}{N}\right)_{\text{intf-intf}} = \left[\frac{8}{25}T\int_0^\infty df \,\gamma^2(f) \frac{S_h^2(f)}{S_n^2(f)}\right]^{1/2} \,.$$
(7.240)

For two cylindrical bars, instead, $\Gamma(f) = (8/15)\gamma(f)$, while for the correlation between an interferometer and a cylindrical bar, from the explicit expressions of the pattern functions in Table 7.1, we get again $\Gamma(f) = (2/5)\gamma(f)$.

Using eqs. (7.233) and (7.202) we can also rewrite eq. (7.239) as

$$\frac{S}{N} = \frac{3H_0^2}{4\pi^2} \left[2T \int_0^\infty df \, \Gamma^2(f) \frac{\Omega_{\text{gw}}^2(f)}{f^6 S_{n,1}(f) S_{n,2}(f)} \right]^{1/2} , \qquad (7.241)$$

and in particular, for a two-interferometer correlation,

$$\left(\frac{S}{N}\right)_{\text{intf-intf}} = \frac{3H_0^2}{10\pi^2} \left[2T \int_0^\infty df \, \gamma^2(f) \frac{\Omega_{\text{gw}}^2(f)}{f^6 S_{n,1}(f) S_{n,2}(f)} \right]^{1/2}.$$
(7.242)

We can now compare the measurements of stochastic backgrounds performed with the two-detector correlation, to the measurement which uses a single detector, both from the point of view of sensitivity, and of the ability to discriminate true GWs from noise.

Comparison of two-detector and single-detector sensitivities

To compare the sensitivity of a two-detector correlation with the sensitivity of a single detector we assume that we have two identical detectors at a very close distance and with the same orientation, so that $\Gamma(f)$ becomes equal to the angular efficiency factor $F_{12} = F$. (This is the most favorable situation; however in practice, if the detectors are too close, there will be correlated noise.) To perform an order-of-magnitude estimate, we approximate eq. (7.239) as

$$\left(\frac{S}{N}\right)^2 \sim (2T\Delta f) F^2 \frac{S_h^2}{S_n^2},$$
 (7.243)

where Δf is the useful bandwidth of the detectors, centered around a frequency f, and S_n and S_h are typical values of $S_n(f)$ and $S_h(f)$, respectively, over this bandwidth. Then the minimum detectable value of S_h , at signal-to-noise level S/N, is

$$(S_h)_{\min} \sim \frac{S_n}{(2T\Delta f)^{1/2}} \frac{(S/N)}{F},$$
 (7.244)

and therefore

$$[\Omega_{\rm gw}]_{\rm min} \sim \frac{4\pi^2}{3H_0^2} \frac{f^3 S_n}{(2T\Delta f)^{1/2}} \frac{(S/N)}{F} \,.$$
 (7.245)

where f^3 is really a typical value of f^3 over the bandwidth. Comparing eq. (7.244) with eq. (7.214) we see that, correlating two detectors, we have gained a factor $(2T\Delta f)^{-1/2}$. Numerically,

$$\frac{1}{(2T\Delta f)^{1/2}} \simeq 1 \times 10^{-5} \left(\frac{150 \,\mathrm{Hz}}{\Delta f}\right)^{1/2} \left(\frac{1 \,\mathrm{yr}}{T}\right)^{1/2} \,. \tag{7.246}$$

Therefore, integrating for one year the output of two detectors with a bandwidth of 150 Hz, we can improve our sensitivity to S_h , and therefore to $h_0^2\Omega_{\text{gw}}$, by approximately five orders of magnitudes, with respect to the sensitivity of a single detector. 71 It is interesting to compare these results with the matched filtering procedure discussed in Section 7.3. In Section 7.3 we took advantage of the fact that we knew the form of the signal, in order to discriminate it from the noise. Here, instead, in a single detector both the signal and the noise have the same statistical properties, but we took advantage of the fact the signals in the two detectors are correlated, while the noise are decorrelated. In particular, the measure of the correlation between the signals in the two detectors is given by the overlap reduction function $\Gamma(f)$ of eq. (7.226), which shows that the signals are indeed well correlated if the separation between the detectors is much smaller than λ , and if the detectors are well oriented with respect to each other. Technically, the assumptions that the noise are uncorrelated entered in eq. (7.224), as well as when passing from eq. (7.229) to eq. (7.230), where we neglected the correlator $\langle \tilde{n}_1^*(f)\tilde{n}_2(f)\rangle$.

⁷¹The precise numbers, of course, can only be obtained once we have the form of $S_h(f)$ and of $S_n(f)$, carrying out the integral in eq. (7.239). Observe also that in eq. (7.214) appears $(S/N)^2$ while in eq. (7.244) appears (S/N), but this is simply a consequence of the fact that, for the two-detector correlation, we have defined S/N as a quantity quadratic in the GW amplitude, while for a single detector we defined it to be linear in the GW amplitude. Once we choose our criterion for fixing the confidence level, e.g. a signal-to-noise ratio 1.7 in amplitude, the quantity that we are denoting by $(S/N)^2$ here and the quantity denoted by S/N in eq. (7.214) have the same numerical value.

 70 Observe that, for periodic signals and for bursts, as well as for a single-detector search of stochastic backgrounds, we defined the quantity S/N as linear in the GW, i.e. if $\tilde{h}(f) \to \lambda \tilde{h}(f)$, then $(S/N) \to \lambda (S/N)$, see eq. (7.51) and eq. (7.213). For searches of stochastic backgrounds with two-detector correlations, we have rather defined S/N as linear both in $h_1(t)$ and in $h_2(t)$ and therefore S/N scales overall quadratically in the GW amplitude. If we prefer to use a quantity that is linear in the GW amplitude we can define $\mathrm{SNR} = (S/N)^{1/2}$, so SNR is proportional to $T^{1/4}$. Of course, it is a matter

of conventions whether to use SNR or

(S/N).

Recall however that the optimal filter depends on the form of the signal. A stochastic background of cosmological origin, as we will see in Vol. 2, is not expected to show strong spectral features in the bandwidth $\Delta f \sim 1$ kHz of ground based interferometers, so it should be adequate a simple power-law parametrization,

$$h_0^2 \Omega_{\rm gw}(f) = K f^{\alpha} \tag{7.247}$$

where K and α are two parameters, and α could be positive or negative. For each value of α we can construct the optimal filter (the overall constant in the filter is irrelevant, as we have seen, so different values of K give the same filter) and, given the noise spectral density $S_n(f)$, eq. (7.239) gives S/N as a function of K and α , and therefore tells us what region of this parameter space can be explored, at a given confidence level. For astrophysical backgrounds, more elaborated parametrizations of $h_0^2\Omega_{\rm gw}(f)$ might be necessary at broadband detectors.

Non-stationary noise

Until now, we have assumed that the noise in the detectors is stationary, and that it can be represented by a fixed function $S_n(f)$. However, such an assumption is not realistic, even more considering that we wish to use a very long observation time, of the order of months. Each detector has periods where it is more quiet and periods where, because of environmental or other disturbances, it is more noisy. Therefore the function $S_n(f)$ changes with time, and we must know how to combine periods in which the detectors had different noise. To study this issue we can subdivide the total observation time T into n intervals of length T_I , where $I=1,\ldots,m$ labels the interval of data, and with $T=\sum_{I=1}^m T_I$. We choose the T_I so that within each interval the noise of the two detectors can be considered stationary. To each of these intervals we can then apply eq. (7.239), so the value of the optimal signal-to-noise ratio from this interval is

$$\left(\frac{S}{N}\right)_{I}^{2} = 2T_{I} \int_{0}^{\infty} df \, \Gamma^{2}(f) \frac{S_{h}^{2}(f)}{S_{n}^{2}(f;I)} \,.$$
 (7.248)

Here $S_n(f;I)$ is the total noise spectral density during the I-th interval, $S_n^2(f;I) = S_{n,1}(f;I)S_{n,2}(f;I)$, where $S_{n,j}(f;I)$ is the noise spectral density of the j-th detector during the I-th interval. We now ask how we should combine the $(S/N)_I$ of the different intervals to form the total optimal signal-to-noise ratio. The correct answer can be guessed observing that the optimal $(S/N)_I^2$ is linear in T_I , see eq. (7.248) and, in the limit in which the noise is stationary over the whole observation time T, we must find that the total optimal signal-to-noise ratio S/N satisfies $(S/N)^2 \sim T = \sum_I T_I$. This fixes uniquely the relation between the total optimal signal-to-noise ratio S/N and the $(S/N)_I$,

$$\left(\frac{S}{N}\right)^2 = \sum_{I=1}^m \left(\frac{S}{N}\right)_I^2. \tag{7.249}$$

The same result can also be obtained more formally introducing the observable

$$Y_{\text{tot}} = \frac{\sum_{I} \lambda_{I} Y_{I}}{\sum_{I} \lambda_{I}} \tag{7.250}$$

(where it is understood that the sums run over $I=1,\ldots,m$) and choosing the variables $\lambda_I>0$ so that the signal-to-noise ratio of $Y_{\rm tot}$ is maximized. From eq. (7.225), with T replaced by T_I , we see that the Y_I have a mean value

$$S_I \equiv \langle Y_I \rangle = \mu T_I \,, \tag{7.251}$$

where $\mu = \int_0^\infty df \, S_h(f) \Gamma(f) \tilde{Q}(f)$ is independent of I. For the noise, from eq. (7.232) we have

$$N_I^2 = \frac{T_I}{4} \int_{-\infty}^{\infty} df \, |\tilde{Q}(f)|^2 S_n^2(f; I)$$

$$\equiv T_I \sigma_I^2. \tag{7.252}$$

The signal-to-noise ratio S/N of Y_{tot} is obtained by writing

$$S = \langle Y_{\text{tot}} \rangle = \mu \frac{\sum_{I} \lambda_{I} T_{I}}{\sum_{I} \lambda_{I}}, \qquad (7.253)$$

and

$$N^{2} = \left[\langle Y_{\text{tot}}^{2} \rangle - \langle Y_{\text{tot}} \rangle^{2} \right]_{\mu=0}$$

$$= \frac{\sum_{I} \lambda_{I}^{2} \sigma_{I}^{2} T_{I}}{(\sum_{I} \lambda_{I})^{2}}, \qquad (7.254)$$

where we assumed that noise in different intervals are uncorrelated, so $\langle Y_I Y_J \rangle = \delta_{IJ} N_I^2$. Therefore

$$\frac{S^2}{N^2} = \mu^2 \frac{\left(\sum_I \lambda_I T_I\right)^2}{\sum_I \lambda_I^2 \sigma_I^2 T_I} \,. \tag{7.255}$$

The maximization of this expression with respect to the λ_I can be performed very simply, introducing the positive definite scalar product between two vectors with real components a_I and b_I ,

$$(a,b) \equiv \sum_{I} a_I b_I \sigma_I^2 T_I \,. \tag{7.256}$$

Then

$$\frac{S}{N} = \mu \frac{(\lambda_I, \sigma_I^{-2})}{(\lambda_I, \lambda_I)^{1/2}}.$$
(7.257)

This expression is maximized if the vectors with components λ_I and σ_I^{-2} are parallel, so $\lambda_I = 1/\sigma_I^2$ (apart from an irrelevant overall constant). Physically, this means that more noisy periods are weighted less. Then the variable $Y_{\rm opt}$, whose signal-to-noise ratio is optimal, is given by

$$Y_{\text{opt}} = \frac{\sum_{I} \sigma_{I}^{-2} Y_{I}}{\sum_{I} \sigma_{I}^{-2}}$$
 (7.258)

and the value of the optimal S/N is given by

$$\left(\frac{S}{N}\right)^2 = \mu^2 \left(\sigma_I^{-2}, \sigma_I^{-2}\right) = \mu^2 \sum_I \frac{T_I}{\sigma_I^2}, \tag{7.259}$$

which, using eqs. (7.251) and (7.252), is equivalent to eq. (7.249), as expected. Equation (7.239) then becomes

$$\frac{S}{N} = \left[2 \int_0^\infty df \, \Gamma^2(f) S_h^2(f) \sum_{I=1}^m \frac{T_I}{S_n^2(f;I)} \right]^{1/2} . \tag{7.260}$$

This is equivalent to saying that, in eq. (7.239), we must make the replacement

$$\frac{T}{S_n^2(f)} \to \sum_{I=1}^m \frac{T_I}{S_n^2(f;I)}$$
 (7.261)

This way of composing the noise is very natural. It means that noisy periods contribute little to the total signal-to-noise ratio. If we perform the same order-of-magnitude estimate as in eq. (7.245), we conclude that

$$\frac{1}{[\Omega_{\rm gw}(f)]_{\rm min}^2} = \sum_{I=1}^m \frac{1}{[\Omega_{\rm gw}(f;I)]_{\rm min}^2},$$
 (7.262)

where $[\Omega_{\rm gw}(f;I)]_{\rm min}$ is the minimum value of $\Omega_{\rm gw}$ detectable using only the data in the *I*-th interval, and $[\Omega_{\rm gw}(f)]_{\rm min}$ is the minimum value of $\Omega_{\rm gw}$ detectable combining the *n* intervals.

How the background is actually measured

We can now give an example of an operative way of measuring the stochastic background. First of all, one divides the total observation time T into intervals of length T_I , such that within each interval the detector noise is stationary. This scale is chosen based on observations of the detector noise variation, and could typically be of order of one to a few minutes. Within each interval, the spectral density $S_n(f;I)$ can be considered constant in time, and is determined experimentally. We can now compute the filter function, using the measured value of $S_n(f;I)$ and assuming a given form for $\Omega_{\rm gw}$. For instance, $\Omega_{\rm gw}={\rm const.}$ can be the simplest choice, or one can use the parametrization (7.247) and repeat the procedure for various values of α .

To have an experimental determination of $S_I = \langle Y_I \rangle$ and of $N_I = [\langle Y_I^2 \rangle - \langle Y_I \rangle^2]^{1/2}$ one further divides each interval into segments of length Δt , labeled by an index $J = 1, \ldots, n$, and with $T_I = n\Delta t$ (with Δt much larger than the light travel time between the detectors, which for the two LIGO observatories is about 10 ms). The signal Y_{IJ} relative to the J-th segment of the I-th interval is computed using eq. (7.220), with the time integration running only over the J-th segment of the I-th interval. Observe that the filter function Q(t-t') typically vanishes very fast for |t-t'| larger than a few tens of ms, so in practice if t belongs to the

J-th interval, the support of Q(t-t') is entirely contained in the J-th interval.

From the set of Y_{IJ} at fixed I, one can construct the sample mean

$$S_I = \frac{1}{n} \sum_{I=1}^n Y_{IJ}, \qquad (7.263)$$

and the sample variance

$$N_I^2 = \frac{1}{n-1} \sum_{J=1}^n (Y_{IJ} - S_I)^2, \qquad (7.264)$$

of the I-th interval. We repeat this procedure for all intervals and, according to eq. (7.249), the total signal-to-noise ratio is

$$\left(\frac{S}{N}\right)^2 = \sum_{I=1}^m \left(\frac{S_I^2}{N_I^2}\right).$$
 (7.265)

If this S/N exceeds a predetermined threshold value one can state that a stochastic background is detected, with a confidence level which depends on the threshold used.⁷³

Multiple-detector correlation

Another interesting question is what happens if we correlate the outputs of N detectors, with N > 2. For simplicity, we assume at first that we have N identical detectors, with the same noise spectral density $S_n(f)$, and all running simultaneously for a time T.

With N detectors we can form N(N-1)/2 independent two-point correlators

$$Y_{ij} = \int_{-T/2}^{T/2} dt \int_{-T/2}^{T/2} dt' \, s_i(t) s_j(t') Q(t - t') \,, \tag{7.266}$$

with i < j. (If the detectors have different noise spectral densities, then also the filter function depends on i, j, and we write it $Q_{ij}(t-t')$.) Conceptually, for a stationary stochastic background, there is no difference between the situation in which N(N-1)/2 identical pairs of detectors run for a time T, and the situation in which a single pair of detectors runs for a time $T_{\text{total}} = T \times N(N-1)/2$. In the former case, sampling the output of the detectors at times t_k , with $k = 1, \ldots, k_{\text{max}}$, we get a set of values $Y_{ij}(t_k)$, for each of the N(N-1)/2 pairs (i,j). In the latter case we directly get a set of values $Y(t_k)$ for the single pair considered, with k taking values up to $k_{\text{max}} \times N(N-1)/2$. In both cases we must then compute the average of Y over all these values, so the result is the same and the difference is just a matter of notation. In conclusion, the signal-to-noise ratio with N identical detectors can be obtained from eq. (7.239) making the replacement

$$T \to \frac{N(N-1)}{2} T. \tag{7.267}$$

 $^{^{73}}$ A subtle point is that it can be shown that, if we wait long enough, i.e. if the total observation time is sufficiently large, any predetermined fixed threshold will be exceeded. In other words, in the limit $T \to \infty$ the false alarm probability is 100%! To have a finite false alarm probability even in the limit $T \to \infty$, the value of the threshold must increase with the number of intervals n faster than $\log \log n$.

⁷²In practice, it can be more convenient to perform a FFT over the segment and use the frequency space expression (7.223).

If we denote by $[\Omega_{\rm gw}]_{\rm min,N}$ the minimum value of $\Omega_{\rm gw}$ measurable with N identical detectors and by $[\Omega_{\rm gw}]_{\rm min,2}$ the minimum value of $\Omega_{\rm gw}$ detectable with two detectors, then

$$[\Omega_{\rm gw}]_{\rm min,N} = \left[\frac{2}{N(N-1)}\right]^{1/2} [\Omega_{\rm gw}]_{\rm min,2}.$$
 (7.268)

In the more realistic case in which the detectors have different noise spectral densities, or have different common time of operation, the situation is formally identical to the case of non-stationary noise discussed above, where the observations taken during a time T_{ij} by each pair of detectors (i, j), with i < j, plays the role of the observations taken during the time intervals labeled by I in eqs. (7.260) to (7.262). Therefore, the signal-to-noise ratio is obtained from eq. (7.239) with the replacement

$$\frac{T}{S_n^2(f)} \to \sum_{i < j} \frac{T_{ij}}{S_n^2(f; \langle ij \rangle)}, \qquad (7.269)$$

where T_{ij} is the common time of operation of the detectors i and j, and $S_n^2(f;\langle ij\rangle) = S_n(f;i)S_n(f;j)$ is the product of the spectral densities of the i-th and j-th detector. The order-of-magnitude estimate of the minimum detectable value of $\Omega_{\rm gw}$, eq. (7.262), becomes

$$\frac{1}{\left[\Omega_{\rm gw}(f)\right]_{\rm min,N}^2} = \sum_{i < j} \frac{1}{\left[\Omega_{\rm gw}(f; \langle ij \rangle)\right]_{\rm min}^2}.$$
 (7.270)

When all detectors are equal and have the same common time of operation, $[\Omega_{\rm gw}(f;\langle ij\rangle)]_{\rm min}$ becomes independent of the pair i,j considered, and is the quantity that we denoted by $[\Omega_{\rm gw}]_{\rm min,2}$, so we recover eq. (7.268).

In a sense, this result is disappointing. We have seen in eq. (7.245) that, passing from a single detector to a two-detector correlation, we gain a factor $1/(2T\Delta f)^{1/2}$ in the minimum detectable value of $\Omega_{\rm gw}$. For T=1 yr and f=100 Hz, this means and improvement by a factor 10^5 in sensitivity. Passing from N=2 to N=3 detectors, instead, we see from eq. (7.268) that we gain only a further factor $\sqrt{3}$.

This is very different from the situation for bursts discussed in Section 7.5.3. In the case of bursts, the noise that compete with the signal consists of large, relatively rare fluctuations. At any given moment the probability that, in a single detector and within a given time window, say of order few tens of ms, a fluctuation with a signal-to-noise ratio above a large threshold takes place, is a small number $\epsilon \ll 1$. The probability that a second detector has a simultaneous independent fluctuation above this threshold, within the same window, is $O(\epsilon^2)$, the probability of a three-detector coincidence is $O(\epsilon^3)$, etc. Then, for bursts, the gain in statistical significance passing from a single detector to a two-detector coincidence is that same as the gain passing from a two-detector to a three-detector coincidence. The crucial point is that for bursts, after matched filtering, we are left with short events with a large value of S/N, which are rare.

In contrast, for stochastic backgrounds we are never confronted with rare events. At any given moment the GW stochastic signal is always much below the noise, and is never responsible for large fluctuations of the output. There are no rare events to be searched in coincidence, and the only advantage of using more detector pairs is that the total amount of data available increases, which means that we have a longer effective observation time.

The situation does not change substantially if, rather than two-point correlators, we consider M-point correlators, with M smaller than or equal the number of detectors N. For instance, with four detectors we can consider a four-point correlator $\langle s_1(f)s_2(f)s_3(f)s_4(f)\rangle$. Repeating the same steps as above, one finds again that the signal-to-noise ratio (always defined to be quadratic in the GW signal, in order to compare with the same quantity as in two-detector case) scales as \sqrt{T} .

On the other hand, an advantage of multiple-detector correlations is that it might be easier to suppress correlated environmental noise, especially if the various detectors are not close to each other.

Correlated noise and signal chopping

Equation (7.239) shows that a true GW signal has a signature that in principle could allow us to distinguish it from the noise: increasing the observation time, the signal-to-noise ratio in the presence of a real GW signal must increase as $T^{1/2}$.

Actually, this is a signature that only allows us to distinguish a stochastic GW background from uncorrelated noise in the two detectors Unfortunately, any residual correlated noise would still mimic the behavfor of a real GW signal. The problem is therefore how to make sure that correlated noise are negligible, and this can be a hard task, particularly for very long integration times. If two detectors are at the same site. or very close, their overlap reduction function is maximized, but we will certainly have correlated environmental noise. We have seen that the overlap reduction function suppresses the GW correlation if the detector separation is $\Delta x \gg \lambda$. For instance, at f = 50 Hz, $\lambda \simeq 1000$ km. Most environmental disturbances will decorrelate on a much shorter lengthscale, so it is possible that two detectors at a suitable distance are still correlated as far as the GW background is concerned, but they have negligible correlated noise. However, beyond a given sensitivity level, seismic noise or propagating electromagnetic disturbances might still give important correlated noise, and this is a difficult issue that will have to be carefully studied experimentally.

An interesting option offered by the two-detector correlation is the possibility of *chopping* the signal. Chopping is a general term for measurements in which we switch our detector between the quantity that we want to measure and a reference quantity. It is a very powerful experimental technique, that exploits the fact that in many situations one can measure with a much better precision the variation of a quantity rather than the quantity itself because, taking the difference, many uncertain-

⁷⁴The classical example of this technique was the Dicke radiometer, which was developed by Dicke during World War II for application to microwave radars, and measured the radiation temperature of a radio source (i.e. the temperature of a black body having the same radio brightness). A direct measurement was difficult: the signal needed a large amplification, and fluctuations in the amplifier gain resulted in large errors. To overcome this difficulty, in the Dicke radiometer the receiver switches quickly between the source and a carefully calibrated black body, whose temperature was chosen to be of the order of the value expected for the source. To tell when these temperatures were equal was much easier than to obtain a direct determination of the source temperature. The same principle of comparing with a reference black body was used by the FIRAS spectrometer on board of the COBE satellite to measure the blackbody spectrum of CMB. To measure the CMB anisotropies, i.e. the variation of the black-body temperature over the sky, the principle used by the DMR detector on COBE and by the subsequent high-precision experiments such as WMAP is to compare the temperatures between two points in the sky.

ties, e.g. calibration uncertainties, cancel out.⁷⁴ In particular, one can compare the measurement in a situation where the signal is expected, to the situation where a null answer should come out.

At first sight, it appears that a measurement of this type is impossible for a stochastic backgrounds of GWs, since the background is always there, and gravitational forces cannot be screened. It seems therefore impossible to compare the output of a detector when no stochastic GW background acts on it, with the output when the background is acting on it. Remarkably, this is no longer true when we consider a two-detector correlation. In fact, changing the relative orientation of the two detectors, the factor $\sum_A F_1^A(\hat{\mathbf{n}}) F_2^A(\hat{\mathbf{n}})$ in eq. (7.226) changes, and it is therefore possible to modulate the signal. To illustrate this point, we compute F_{12} for a bar-interferometer correlation. Using Table 7.1 and eqs. (7.31) and (7.32) we see that, for ψ generic, the pattern functions of an interferometer are

$$F_{+}^{(\text{intf})}(\theta,\phi;,\psi) = \frac{1}{2}(1+\cos^{2}\theta)\cos 2\phi\cos 2\psi - \cos\theta\sin 2\phi\sin 2\psi,$$

$$F_{\times}^{(\text{intf})}(\theta,\phi;\psi) = \frac{1}{2}(1+\cos^{2}\theta)\cos 2\phi\sin 2\psi + \cos\theta\sin 2\phi\cos 2\psi.$$

$$(7.271)$$

The pattern functions of the bar for ψ generic can also be obtained from Table 7.1 and eqs. (7.31) and (7.32). We must however pay attention to the fact that in Table 7.1, the variable denoted by θ for resonant bars is the angle measured from its longitudinal axis, while for an interferometer with arms along the x and y axes, we denoted by θ the polar angle measured from the z axis, so these two angles are not the same unless the bar is vertical. If instead the bar lies in the x, y plane, at an angle α with the y axis, and we denote by θ the polar angles measured from the z axis, then the pattern functions of the bar become

$$F_{+}^{(\text{bar})}(\theta, \phi; , \psi) = \left[-\cos^{2}\theta\cos^{2}(\phi - \alpha) + \sin^{2}(\phi - \alpha)\right]\cos 2\psi$$
$$+\left[\cos\theta\sin 2(\phi - \alpha)\right]\sin 2\psi$$
$$F_{\times}^{(\text{bar})}(\theta, \phi; , \psi) = \left[-\cos^{2}\theta\cos^{2}(\phi - \alpha) + \sin^{2}(\phi - \alpha)\right]\sin 2\psi$$
$$-\left[\cos\theta\sin 2(\phi - \alpha)\right]\cos 2\psi . \tag{7.272}$$

From this it follows that

$$\int \frac{d\hat{\mathbf{n}}}{4\pi} \, \frac{d\psi}{2\pi} \, \sum_{A} F_A^{\text{(bar)}} F_A^{\text{(intf)}} = -\frac{2}{5} \cos 2\alpha \,. \tag{7.273}$$

(The overall sign of F_{12} is irrelevant since $\Gamma(f)$ enters quadratically in the signal-to-noise ratio.) We see that the correlation is maximum when the bar is aligned with one of the interferometer arms (i.e. when $\alpha = 0$ or $\alpha = \pi/2$). In contrast, when $\alpha = \pi/4$ we have $F_{12} = 0$. Therefore in this configuration the signal obtained from the interferometer-bar correlation vanishes. Even if GWs cannot be screened, the "composite detector" whose output is the correlation between a bar and an interferometer can be set in the "off source" position! We can then compare the result

in this configuration with the result when the resonant bar is parallel to one of the interferometer arms, which is the position that maximizes the correlation. This chopping strategy has been used in the LIGO–ALLEGRO correlation. The ALLEGRO resonant bar (which has now terminated its activity) was located relatively close to the LIGO observatory in Livingston, and was mounted on a platform that allowed to rotate it easily. (After a rotation, data taking of good quality resumed in just half an hour.) The bar was therefore taken for a few months in the "off source" position, and then rotated to the "on source" position for a few more months.

Further reading

- For a textbook discussion of matched filtering and of detection of signals in noise see Wainstein and Zubakov (1962) and McDonough and Whalen (1995). For matched filtering and optimal signal-to-noise ratio for GW bursts see Thorne (1987), and Saulson (1994), Chapter 4. Statistical aspects of parameter estimation are discussed in Finn (1992) and in Cutler and Flanagan (1994), where the multiple detector case is also treated. For a review of data analysis for interferometric GW detector see Viceré (2000).
- Books on probability and statistics typically cover many shelves in any physics library, and recommendation are very much subjective. For an elementary but very practical introduction to statistics (tuned to the needs of particle physicists, but quite useful also in the GW context), see Lyons (1986). A concise and useful summary is given in the sections on probability and statistics of the Review of Particle Properties, in Yao et al. [Particle Data Group] (2006). A very nice discussion of Bayesian vs. frequentist method, in the context of particle physics, is given in Cousins (1995). A discussion of the frequentist vs. Bayesian approach in the GW context is given in appendix A of Cutler and Flanagan (1994).
- The analysis of bursts of unknown shape using band-pass filtering is discussed in Flanagan and Hughes (1998a, 1998b), in the context of the merging phase of black hole binaries. Time–frequency techniques are further discussed in Anderson and Balasubramanian (1999), Anderson, Brady, Creighton and Flanagan (2001) and Viceré (2002). An algorithm based on clusters of pixels

- in the time–frequency domain (termed TFCLUS-TERS) is presented in Sylvestre (2002). A book on the use of wavelets in physics is van den Berg (1999). Application of wavelets to the analysis of GW bursts can be found in Klimenko, Yakushin, Rakhmanov and Mitselmakher (2004) and Klimenko and Mitselmakher (2004) (the WaveBurst algorithm).
- Some sources, such as accreting neutron or quark stars, as well as neutron stars stressed by large interior magnetic fields (magnetars), could emit repeatedly small bursts of GWs, with very characteristic correlations, both in energy and in time, among the different bursts, typical of systems displaying self-organized criticality. These correlations could give a further handle in their data analysis. These "GW bursters" are discussed in Coccia, Dubath and Maggiore (2004) and Dubath, Foffa, Gasparini, Maggiore and Sturani (2005).
- The search strategy for GW bursts using the three LIGO interferometers is discussed in Abbott et al. [LSC] (2004b). The sensitivity of a network of interferometers for reconstructing the source position is studied in Gürsel and Tinto (1989). Searches for GW bursts using coincidences between up to five resonant bars are described in Astone et al. [IGEC] (2003a). Results with correlations among three bars, with improved sensitivities, are reported in Astone et al. [IGEC2] (2007).
- Introductory discussions of the search strategies for periodic signals can be found in Saulson (1994), Section 14.6 and Schutz (1991). More detailed analysis are given in Brady, Creighton, Cutler and

Schutz (1998) and in Brady and Creighton (2000). The application of the Hough transform to periodic GWs is discussed in Krishnan et al. (2004). A search for periodic GWs from a single specific source, using the LIGO and GEO detectors, is described in Abbott et al. [LSC] (2004a). Limits on 28 isolated pulsar using the LIGO S2 run are given in Abbott et al. [LSC] (2005b).

- The importance of post-Newtonian corrections for the data analysis of coalescing binaries is emphasized in Cutler et al. (1993). Detailed discussions of data analysis procedure and parameter extraction for coalescences is given in Cutler and Flanagan (1994), Poisson and Will (1995), Królak, Kokkotas and Schäfer (1995) and Flanagan and Hughes (1998a). For computations of the waveform with the PN formalism, see the Further Reading section in Chapter 5.
- Optimal template placement for inspiraling compact binaries is discussed in Owen (1996) and Owen

and Sathyaprakash (1999). A comparison of templates for binary inspiral is given in Damour, Iver and Sathyaprakash (2001). A particularly useful family of templates for BH-BH inspiral have been proposed by Buonanno, Chen and Vallisneri (2003). A description of the LIGO search strategy for coalescences can be found in Abbott et al. [LSC] (2005a).

• The optimal SNR in a two-detector correlation and the overlap reduction function are discussed in Michelson (1987), Christensen (1992) and Flanagan (1993). A detailed discussion of signal processing strategies for stochastic backgrounds of GWs is given in Allen and Romano (1999). Signal chopping is discussed in Finn and Lazzarini (2001). Stochastic backgrounds of GWs are reviewed in Maggiore (2000). The search strategy of LIGO for stochastic backgrounds of GWs is discussed in Abbott et al. [LSC] (2004d) and (2005c).

Resonant-mass detectors

Th teo co: for de ou are dis subject of the next chapter.

The passage from resonant detectors to interferometers implies a jump from "small-scale" experiments, performed by groups which can be as small as half a dozen people, to "Big Science", with collaborations of hundreds of people and financial costs which are higher by factors $O(10^2-10^3)$. As we will see in the next chapter, such a jump is justified by the formidable discovery potential of interferometers and especially advanced interferometers. We nevertheless begin our discussion of experiments with resonant-mass detectors, both because they still have the possibility of detecting rare or unexpected events, and also because their study is instructive in itself. Our emphasis will be on aspects that have an intrinsic conceptual interest, such as understanding how a GW interacts with a macroscopic piece of matter, and on how it is possible to detect vibrations of a macroscopic body which are incredibly small, with amplitude many orders of magnitude smaller than the size of a nucleus. We will see that, by themselves, resonant detectors are remarkable instruments; it is possible to measure vibrations in a two-ton object, such as a typical bar, which corresponds to just a few tens of phonons, and variations ΔL of their length L, with $\Delta L/L \sim 10^{-19} - 10^{-18}$.

The interaction of GWs with an elastic body

8.1.1 The response to bursts

A typical bar is a cylinder of length $L \simeq 3$ m and radius $R \simeq 30$ cm, so in a first approximation we can treat its vibrations as one-dimensional. We orient the bar along the x axis, with the end-faces at $\pm L/2$, and we study the dynamics of a volume element dV of the bar originally located

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The read-out system

Noise sources

Resonant spheres

ne history of experimental GW physics began with resonant-mass de-	
ctors. The pioneer was Joseph Weber who, in the 1960s, developed the	8.1
ncept and built the first resonant bars. In the course of the subsequent	~ -
ur decades, resonant-mass detectors operated by various groups have	8.2
ached sensitivities better than Weber's original bars by about four or-	8.3
rs of magnitudes in energy. Still, we will see in this chapter that these	8.4
nsitivities could allow the detection of only relatively strong signals in	
r Galaxy or at most in our immediate galactic neighborhood, which	
e expected to be rare. To gain access to sources at large extragalactic	
stances it is necessary to build large interferometers, which will be the	