websockets Documentation

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websockets is a library for building WebSocket servers and clients in Python with a focus on correctness and simplicity.

Built on top of asyncio, Python's standard asynchronous I/O framework, it provides an elegant coroutine-based API.

Here's a client that says "Hello world!":

```
#!/usr/bin/env python

import asyncio
import websockets

async def hello(uri):
    async with websockets.connect(uri) as websocket:
    await websocket.send("Hello world!")

asyncio.get_event_loop().run_until_complete(
    hello('ws://localhost:8765'))
```

And here's an echo server:

```
#!/usr/bin/env python

import asyncio
import websockets

async def echo(websocket, path):
    async for message in websocket:
    await websocket.send(message)

asyncio.get_event_loop().run_until_complete(
    websockets.serve(echo, 'localhost', 8765))
asyncio.get_event_loop().run_forever()
```

Do you like it? Let's dive in!

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CHAPTER 1

Tutorials

If you're new to websockets, this is the place to start.

1.1 Getting started

Warning: This documentation is written for Python 3.5. If you're using Python 3.4, you will have to *adapt the code samples*.

1.1.1 Installation

websockets requires Python 3.4. Install it with:

```
pip install websockets
```

1.1.2 Basic example

This section assumes Python 3.5. For older versions, read below. Here's a WebSocket server example. It reads a name from the client, sends a greeting, and closes the connection.

```
#!/usr/bin/env python

import asyncio
import websockets

async def hello(websocket, path):
   name = await websocket.recv()
   print("< {}".format(name))

   greeting = "Hello {}!".format(name)</pre>
```

```
await websocket.send(greeting)
print("> {}".format(greeting))

start_server = websockets.serve(hello, 'localhost', 8765)

asyncio.get_event_loop().run_until_complete(start_server)
asyncio.get_event_loop().run_forever()
```

On the server side, the handler coroutine hello is executed once for each WebSocket connection. The connection is automatically closed when the handler returns.

Here's a corresponding client example.

```
#!/usr/bin/env python
import asyncio
import websockets

async def hello():
    async with websockets.connect('ws://localhost:8765') as websocket:
    name = input("What's your name? ")
    await websocket.send(name)
    print("> {}".format(name))

    greeting = await websocket.recv()
    print("< {}".format(greeting))

asyncio.get_event_loop().run_until_complete(hello())</pre>
```

async and await aren't available in Python < 3.5. Here's how to adapt the client example for older Python versions.

```
#!/usr/bin/env python
import asyncio
import websockets

@asyncio.coroutine
def hello():
    websocket = yield from websockets.connect('ws://localhost:8765/')

    try:
        name = input("What's your name? ")
        yield from websocket.send(name)
        print("> {}".format(name))

        greeting = yield from websocket.recv()
        print("< {}".format(greeting))

    finally:
        yield from websocket.close()

asyncio.get_event_loop().run_until_complete(hello())</pre>
```

1.1.3 Browser-based example

Here's an example of how to run a WebSocket server and connect from a browser.

Run this script in a console:

```
#!/usr/bin/env python
import asyncio
import datetime
import random
import websockets

async def time(websocket, path):
    while True:
        now = datetime.datetime.utcnow().isoformat() + 'Z'
        await websocket.send(now)
        await asyncio.sleep(random.random() * 3)

start_server = websockets.serve(time, '127.0.0.1', 5678)

asyncio.get_event_loop().run_until_complete(start_server)
asyncio.get_event_loop().run_forever()
```

Then open this HTML file in a browser.

```
<!DOCTYPE html>
<html>
    <head>
        <title>WebSocket demo</title>
    </head>
    <body>
        <script>
            var ws = new WebSocket("ws://127.0.0.1:5678/"),
                messages = document.createElement('ul');
            ws.onmessage = function (event) {
                var messages = document.getElementsByTagName('ul')[0],
                    message = document.createElement('li'),
                    content = document.createTextNode(event.data);
                message.appendChild(content);
                messages.appendChild(message);
            } ;
            document.body.appendChild(messages);
        </script>
    </body>
</html>
```

1.1.4 Common patterns

You will usually want to process several messages during the lifetime of a connection. Therefore you must write a loop. Here are the basic patterns for building a WebSocket server.

Consumer

For receiving messages and passing them to a consumer coroutine:

```
async def consumer_handler(websocket, path):
    async for message in websocket:
    await consumer(message)
```

Iteration terminates when the client disconnects.

Asynchronous iteration isn't available in Python < 3.6; here's the same code for earlier Python versions:

```
async def consumer_handler(websocket, path):
    while True:
        message = await websocket.recv()
        await consumer(message)
```

recv() raises a ConnectionClosed exception when the client disconnects, which breaks out of the while True loop.

Producer

For getting messages from a producer coroutine and sending them:

```
async def producer_handler(websocket, path):
    while True:
        message = await producer()
        await websocket.send(message)
```

send() raises a ConnectionClosed exception when the client disconnects, which breaks out of the while True loop.

Both

You can read and write messages on the same connection by combining the two patterns shown above and running the two tasks in parallel:

```
async def handler(websocket, path):
    consumer_task = asyncio.ensure_future(consumer_handler(websocket))
    producer_task = asyncio.ensure_future(producer_handler(websocket))
    done, pending = await asyncio.wait(
        [consumer_task, producer_task],
        return_when=asyncio.FIRST_COMPLETED,
)

for task in pending:
    task.cancel()
```

Registration

If you need to maintain a list of currently connected clients, you must register clients when they connect and unregister them when they disconnect.

```
connected = set()

async def handler(websocket, path):
   global connected
   # Register.
   connected.add(websocket)
   try:
        # Implement logic here.
        await asyncio.wait([ws.send("Hello!") for ws in connected])
        await asyncio.sleep(10)
```

```
finally:
    # Unregister.
    connected.remove(websocket)
```

This simplistic example keeps track of connected clients in memory. This only works as long as you run a single process. In a practical application, the handler may subscribe to some channels on a message broker, for example.

1.1.5 That's all!

The design of the websockets API was driven by simplicity.

You don't have to worry about performing the opening or the closing handshake, answering pings, or any other behavior required by the specification.

websockets handles all this under the hood so you don't have to.

1.1.6 Python < 3.5

This documentation uses the await and async syntax introduced in Python 3.5.

If you're using Python 3.4, you must substitute:

```
async def ...
```

with:

```
@asyncio.coroutine def ...
```

and:

```
await ...
```

with:

```
yield from ...
```

Otherwise you will encounter a SyntaxError.

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CHAPTER 2

How-to guides

These guides will help you build and deploy a websockets application.

2.1 Cheat sheet

2.1.1 Server

- Write a coroutine that handles a single connection. It receives a websocket protocol instance and the URI path in argument.
 - Call recv() and send() to receive and send messages at any time.
 - You may ping() or pong() if you wish but it isn't needed in general.
- Create a server with <code>serve()</code> which is similar to asyncio's <code>create_server()</code>.
 - The server takes care of establishing connections, then lets the handler execute the application logic, and finally closes the connection after the handler exits normally or with an exception.
 - For advanced customization, you may subclass <code>WebSocketServerProtocol</code> and pass either this subclass or a factory function as the <code>create_protocol</code> argument.

2.1.2 Client

- Create a client with connect() which is similar to asyncio's create_connection().
 - On Python 3.5, you can also use it as an asynchronous context manager.
 - For advanced customization, you may subclass WebSocketClientProtocol and pass either this subclass or a factory function as the create_protocol argument.
- Call recv() and send() to receive and send messages at any time.
- You may ping() or pong() if you wish but it isn't needed in general.

• If you aren't using connect () as a context manager, call close () to terminate the connection.

2.1.3 Debugging

If you don't understand what websockets is doing, enable logging:

```
import logging
logger = logging.getLogger('websockets')
logger.setLevel(logging.INFO)
logger.addHandler(logging.StreamHandler())
```

The logs contain:

- Exceptions in the connection handler at the ERROR level
- Exceptions in the opening or closing handshake at the INFO level
- All frames at the DEBUG level this can be very verbose

If you're new to asyncio, you will certainly encounter issues that are related to asynchronous programming in general rather than to websockets in particular. Fortunately Python's official documentation provides advice to develop with asyncio. Check it out: it's invaluable!

2.1.4 Keeping connections open

Pinging the other side once in a while is a good way to check whether the connection is still working, and also to keep it open in case something kills idle connections after some time:

2.1.5 Passing additional arguments to the connection handler

When writing a server, if you need to pass additional arguments to the connection handler, you can bind them with functools.partial():

```
import asyncio
import functools
import websockets

async def handler(websocket, path, extra_argument):
    ...
```

```
bound_handler = functools.partial(handler, extra_argument='spam')
start_server = websockets.serve(bound_handler, '127.0.0.1', 8765)
asyncio.get_event_loop().run_until_complete(start_server)
asyncio.get_event_loop().run_forever()
```

Another way to achieve this result is to define the handler corountine in a scope where the extra_argument variable exists instead of injecting it through an argument.

2.2 Deployment

2.2.1 Application server

The author of websockets isn't aware of best practices for deploying network services based on asyncio, let alone application servers.

You can run a script similar to the *server example*, inside a supervisor if you deem that useful.

You can also add a wrapper to daemonize the process. Third-party libraries provide solutions for that.

If you can share knowledge on this topic, please file an issue. Thanks!

2.2.2 Graceful shutdown

You may want to close connections gracefully when shutting down the server, perhaps after executing some cleanup logic. There are two ways to achieve this with the object returned by <code>serve()</code>:

- · using it as a asynchronous context manager, or
- calling its close () method, then waiting for its wait_closed() method to complete.

Tasks that handle connections will be cancelled. For example, if the handler is awaiting recv(), that call will raise CancelledError.

On Unix systems, shutdown is usually triggered by sending a signal.

Here's a full example (Unix-only):

```
#!/usr/bin/env python
import asyncio
import signal
import websockets

async def echo(websocket, path):
    while True:
        try:
            msg = await websocket.recv()
            except websockets.ConnectionClosed:
                 pass
            else:
                  await websocket.send(msg)

async def echo_server(stop):
        async with websockets.serve(echo, 'localhost', 8765):
                  await stop
```

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```
loop = asyncio.get_event_loop()

# The stop condition is set when receiving SIGTERM.
stop = asyncio.Future()
loop.add_signal_handler(signal.SIGTERM, stop.set_result, None)

# Run the server until the stop condition is met.
loop.run_until_complete(echo_server(stop))
```

async, await, and asynchronous context managers aren't available in Python < 3.5. Here's the equivalent for older Python versions:

```
#!/usr/bin/env python
import asyncio
import signal
import websockets
async def echo (websocket, path):
   while True:
        try:
           msg = await websocket.recv()
        except websockets.ConnectionClosed:
            pass
        else:
           await websocket.send(msg)
loop = asyncio.get_event_loop()
# Create the server.
start_server = websockets.serve(echo, 'localhost', 8765)
server = loop.run_until_complete(start_server)
# Run the server until SIGTERM.
stop = asyncio.Future()
loop.add_signal_handler(signal.SIGTERM, stop.set_result, None)
loop.run_until_complete(stop)
# Shut down the server.
server.close()
loop.run_until_complete(server.wait_closed())
```

It's more difficult to achieve the same effect on Windows. Some third-party projects try to help with this problem.

If your server doesn't run in the main thread, look at call_soon_threadsafe().

2.2.3 Memory use

In order to avoid excessive memory use caused by buffer bloat, it is strongly recommended to tune buffer sizes.

Most importantly max_size should be lowered according to the expected size of messages. It is also suggested to lower max_queue, read_limit and write_limit if memory use is a concern.

2.2.4 Port sharing

The WebSocket protocol is an extension of HTTP/1.1. It can be tempting to serve both HTTP and WebSocket on the same port.

The author of websockets doesn't think that's a good idea, due to the widely different operational characteristics of HTTP and WebSocket.

websockets provide minimal support for responding to HTTP requests with the $process_request$ () hook. Typical use cases include health checks.

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CHAPTER 3

Reference

Find all the details you could ask for, and then some.

3.1 API

3.1.1 Design

websockets provides complete client and server implementations, as shown in the *getting started guide*. These functions are built on top of low-level APIs reflecting the two phases of the WebSocket protocol:

- 1. An opening handshake, in the form of an HTTP Upgrade request;
- 2. Data transfer, as framed messages, ending with a closing handshake.

The first phase is designed to integrate with existing HTTP software. websockets provides functions to build and validate the request and response headers.

The second phase is the core of the WebSocket protocol. websockets provides a standalone implementation on top of asyncio with a very simple API.

For convenience, public APIs can be imported directly from the websockets package, unless noted otherwise. Anything that isn't listed in this document is a private API.

3.1.2 High-level

Server

The websockets.server module defines a simple WebSocket server API.

```
websockets.server.serve(ws_handler, host=None, port=None, *, create_protocol=None, time-out=10, max_size=2 ** 20, max_queue=2 ** 5, read_limit=2 ** 16, write_limit=2 ** 16, loop=None, origins=None, extensions=None, subprotocols=None, extra headers=None, compression='deflate', **kwds')
```

This coroutine creates, starts, and return a WebSocketServer.

serve() is a wrapper around the event loop's create_server() method. Internally, the function creates
and starts a Server object by calling create_server(). The WebSocketServer keeps a reference to
this object.

The returned WebSocketServer and its resources can be cleaned up by calling its close() and $wait_closed()$ methods.

On Python 3.5, serve() can also be used as an asynchronous context manager. In this case, the server is shut down when exiting the context.

The ws_handler argument is the WebSocket handler. It must be a coroutine accepting two arguments: a WebSocketServerProtocol and the request URI.

The host and port arguments, as well as unrecognized keyword arguments, are passed along to create_server(). For example, you can set the ssl keyword argument to a SSLContext to enable TLS.

The create_protocol parameter allows customizing the asyncio protocol that manages the connection. It should be a callable or class accepting the same arguments as <code>WebSocketServerProtocol</code> and returning a <code>WebSocketServerProtocol</code> instance. It defaults to <code>WebSocketServerProtocol</code>.

The behavior of the timeout, max_size, and max_queue, read_limit, and write_limit optional arguments is described in the documentation of WebSocketCommonProtocol.

serve () also accepts the following optional arguments:

- origins defines acceptable Origin HTTP headers include '' if the lack of an origin is acceptable
- extensions is a list of supported extensions in order of decreasing preference
- subprotocols is a list of supported subprotocols in order of decreasing preference
- extra_headers sets additional HTTP response headers it can be a mapping, an iterable of (name, value) pairs, or a callable taking the request path and headers in arguments.
- compression is a shortcut to configure compression extensions; by default it enables the "permessage-deflate" extension; set it to None to disable compression

Whenever a client connects, the server accepts the connection, creates a <code>WebSocketServerProtocol</code>, performs the opening handshake, and delegates to the WebSocket handler. Once the handler completes, the server performs the closing handshake and closes the connection.

Since there's no useful way to propagate exceptions triggered in handlers, they're sent to the 'websockets. server' logger instead. Debugging is much easier if you configure logging to print them:

```
import logging
logger = logging.getLogger('websockets.server')
logger.setLevel(logging.ERROR)
logger.addHandler(logging.StreamHandler())
```

```
websockets.server.unix_serve (ws_handler, path, *, create_protocol=None, timeout=10, max_size=2 ** 20, max_queue=2 ** 5, read_limit=2 ** 16, write_limit=2 ** 16, loop=None, origins=None, extensions=None, subprotocols=None, extra_headers=None, compression='deflate', **kwds')
```

Similar to serve (), but for listening on Unix sockets.

This function calls the event loop's create_unix_server() method.

It is only available on Unix.

It's useful for deploying a server behind a reverse proxy such as nginx.

```
class websockets.server.WebSocketServer(loop)
```

Wrapper for Server that closes connections on exit.

This class provides the return type of serve().

It mimics the interface of AbstractServer, namely its close () and wait_closed() methods, to close WebSocket connections properly on exit, in addition to closing the underlying Server.

Instances of this class store a reference to the Server object returned by <code>create_server()</code> rather than inherit from Server in part because <code>create_server()</code> doesn't support passing a custom <code>Server class</code>.

close()

Close the underlying server, and clean up connections.

This calls close() on the underlying Server object, closes open connections with status code 1001, and stops accepting new connections.

wait closed()

Wait until the underlying server and all connections are closed.

This calls wait_closed() on the underlying Server object and waits until closing handshakes are complete and all connections are closed.

This method must be called after close().

sockets

List of socket objects the server is listening to.

None if the server is closed.

Complete WebSocket server implementation as an asyncio.Protocol.

This class inherits most of its methods from WebSocketCommonProtocol.

For the sake of simplicity, it doesn't rely on a full HTTP implementation. Its support for HTTP responses is very limited.

```
handshake (origins=None, available_extensions=None, available_subprotocols=None, ex-
tra_headers=None)

Perform the server side of the opening handshake.
```

If provided, origins is a list of acceptable HTTP Origin values. Include '' if the lack of an origin is acceptable.

If provided, available_extensions is a list of supported extensions in the order in which they should be used.

If provided, available_subprotocols is a list of supported subprotocols in order of decreasing preference.

If provided, extra_headers sets additional HTTP response headers. It can be a mapping or an iterable of (name, value) pairs. It can also be a callable taking the request path and headers in arguments.

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Raise InvalidHandshake if the handshake fails.

Return the path of the URI of the request.

```
process_request (path, request_headers)
```

Intercept the HTTP request and return an HTTP response if needed.

```
request headers are a HTTPMessage.
```

If this coroutine returns None, the WebSocket handshake continues. If it returns a status code, headers and a optionally a response body, that HTTP response is sent and the connection is closed.

The HTTP status must be a HTTPStatus. HTTP headers must be an iterable of (name, value) pairs. If provided, the HTTP response body must be bytes.

(HTTPStatus was added in Python 3.5. Use a compatible object on earlier versions. Look at SWITCHING_PROTOCOLS in websockets.compatibility for an example.)

This method may be overridden to check the request headers and set a different status, for example to authenticate the request and return HTTPStatus.UNAUTHORIZED or HTTPStatus.FORBIDDEN.

It is declared as a coroutine because such authentication checks are likely to require network requests.

```
static select_subprotocol (client_subprotocols, server_subprotocols)
```

Pick a subprotocol among those offered by the client.

If several subprotocols are supported by the client and the server, the default implementation selects the preferred subprotocols by giving equal value to the priorities of the client and the server.

If no subprotocols are supported by the client and the server, it proceeds without a subprotocol.

This is unlikely to be the most useful implementation in practice, as many servers providing a subprotocol will require that the client uses that subprotocol. Such rules can be implemented in a subclass.

Client

The websockets.client module defines a simple WebSocket client API.

This coroutine connects to a WebSocket server at a given uri.

It yields a WebSocketClientProtocol which can then be used to send and receive messages.

On Python 3.5, connect () can be used as a asynchronous context manager. In that case, the connection is closed when exiting the context.

connect() is a wrapper around the event loop's create_connection() method. Unknown keyword
arguments are passed to create_connection().

For example, you can set the ssl keyword argument to a SSLContext to enforce some TLS settings. When connecting to a wss:// URI, if this argument isn't provided explicitly, it's set to True, which means Python's default SSLContext is used.

The behavior of the timeout, max_size, and max_queue, read_limit, and write_limit optional arguments is described in the documentation of WebSocketCommonProtocol.

The create_protocol parameter allows customizing the asyncio protocol that manages the connection. It should be a callable or class accepting the same arguments as <code>WebSocketClientProtocol</code> and returning a <code>WebSocketClientProtocol</code> instance. It defaults to <code>WebSocketClientProtocol</code>.

connect () also accepts the following optional arguments:

- origin sets the Origin HTTP header
- extensions is a list of supported extensions in order of decreasing preference
- subprotocols is a list of supported subprotocols in order of decreasing preference
- extra_headers sets additional HTTP request headers it can be a mapping or an iterable of (name, value) pairs
- compression is a shortcut to configure compression extensions; by default it enables the "permessage-deflate" extension; set it to None to disable compression

connect () raises InvalidURI if uri is invalid and InvalidHandshake if the opening handshake fails.

Complete WebSocket client implementation as an asyncio. Protocol.

This class inherits most of its methods from WebSocketCommonProtocol.

```
handshake (wsuri, origin=None, available_extensions=None, available_subprotocols=None, ex-
tra_headers=None)
```

Perform the client side of the opening handshake.

If provided, origin sets the Origin HTTP header.

If provided, available_extensions is a list of supported extensions in the order in which they should be used.

If provided, available_subprotocols is a list of supported subprotocols in order of decreasing preference.

If provided, extra_headers sets additional HTTP request headers. It must be a mapping or an iterable of (name, value) pairs.

Raise InvalidHandshake if the handshake fails.

Shared

The websockets.protocol module handles WebSocket control and data frames as specified in sections 4 to 8 of RFC 6455.

This class implements common parts of the WebSocket protocol.

It assumes that the WebSocket connection is established. The handshake is managed in subclasses such as WebSocketServerProtocol and WebSocketClientProtocol.

It runs a task that stores incoming data frames in a queue and deals with control frames automatically. It sends outgoing data frames and performs the closing handshake.

On Python 3.6, WebSocketCommonProtocol instances support asynchronous iteration:

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```
async for message in websocket:
   await process(message)
```

The iterator yields incoming messages. It exits normally when the connection is closed with the status code 1000 OK. It raises a *ConnectionClosed* exception when the connection is closed with any other status code.

The host, port and secure parameters are simply stored as attributes for handlers that need them.

The timeout parameter defines the maximum wait time in seconds for completing the closing handshake and, only on the client side, for terminating the TCP connection. <code>close()</code> will complete in at most 4 * timeout on the server side and 5 * timeout on the client side.

The max_size parameter enforces the maximum size for incoming messages in bytes. The default value is 1MB. None disables the limit. If a message larger than the maximum size is received, recv() will raise ConnectionClosed and the connection will be closed with status code 1009.

The max_queue parameter sets the maximum length of the queue that holds incoming messages. The default value is 32. 0 disables the limit. Messages are added to an in-memory queue when they're received; then recv() pops from that queue. In order to prevent excessive memory consumption when messages are received faster than they can be processed, the queue must be bounded. If the queue fills up, the protocol stops processing incoming data until recv() is called. In this situation, various receive buffers (at least in asyncio and in the OS) will fill up, then the TCP receive window will shrink, slowing down transmission to avoid packet loss.

Since Python can use up to 4 bytes of memory to represent a single character, each websocket connection may use up to 4 * max_size * max_queue bytes of memory to store incoming messages. By default, this is 128MB. You may want to lower the limits, depending on your application's requirements.

The read_limit argument sets the high-water limit of the buffer for incoming bytes. The low-water limit is half the high-water limit. The default value is 64kB, half of asyncio's default (based on the current implementation of StreamReader).

The write_limit argument sets the high-water limit of the buffer for outgoing bytes. The low-water limit is a quarter of the high-water limit. The default value is 64kB, equal to asyncio's default (based on the current implementation of FlowControlMixin).

As soon as the HTTP request and response in the opening handshake are processed, the request path is available in the path attribute, and the request and response HTTP headers are available:

- as a HTTPMessage in the request_headers and response_headers attributes
- as an iterable of (name, value) pairs in the raw_request_headers and raw_response_headers attributes

These attributes must be treated as immutable.

If a subprotocol was negotiated, it's available in the subprotocol attribute.

Once the connection is closed, the status code is available in the close_code attribute and the reason in close_reason.

```
close (code=1000, reason='')
```

This coroutine performs the closing handshake.

It waits for the other end to complete the handshake and for the TCP connection to terminate.

It doesn't do anything once the connection is closed. In other words it's idemptotent.

It's safe to wrap this coroutine in <code>ensure_future()</code> since errors during connection termination aren't particularly useful.

code must be an int and reason a str.

recv()

This coroutine receives the next message.

It returns a str for a text frame and bytes for a binary frame.

When the end of the message stream is reached, recv() raises ConnectionClosed. This can happen after a normal connection closure, a protocol error or a network failure.

Changed in version 3.0: recv() used to return None instead. Refer to the changelog for details.

send (data)

This coroutine sends a message.

It sends str as a text frame and bytes as a binary frame. It raises a TypeError for other inputs.

ping(data=None)

This coroutine sends a ping.

It returns a Future which will be completed when the corresponding pong is received and which you may ignore if you don't want to wait.

A ping may serve as a keepalive or as a check that the remote endpoint received all messages up to this point:

```
pong_waiter = await ws.ping()
await pong_waiter # only if you want to wait for the pong
```

By default, the ping contains four random bytes. The content may be overridden with the optional data argument which must be of type str (which will be encoded to UTF-8) or bytes.

pong (data=b'')

This coroutine sends a pong.

An unsolicited pong may serve as a unidirectional heartbeat.

The content may be overridden with the optional data argument which must be of type str (which will be encoded to UTF-8) or bytes.

local address

Local address of the connection.

This is a (host, port) tuple or None if the connection hasn't been established yet.

remote_address

Remote address of the connection.

This is a (host, port) tuple or None if the connection hasn't been established yet.

open

This property is True when the connection is usable.

It may be used to detect disconnections but this is discouraged per the EAFP principle. When open is False, using the connection raises a *ConnectionClosed* exception.

Exceptions

```
exception websockets.exceptions.AbortHandshake (status, headers, body=None)
```

Exception raised to abort a handshake and return a HTTP response.

```
exception websockets.exceptions.InvalidHandshake
```

Exception raised when a handshake request or response is invalid.

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exception websockets.exceptions.**InvalidHeader** (*message*, *string*, *pos*)

Exception raised when a HTTP header doesn't have the expected format.

exception websockets.exceptions.InvalidMessage

Exception raised when the HTTP message in a handshake request is malformed.

exception websockets.exceptions.InvalidOrigin

Exception raised when the origin in a handshake request is forbidden.

exception websockets.exceptions.InvalidState

Exception raised when an operation is forbidden in the current state.

exception websockets.exceptions.InvalidStatusCode (status_code)

Exception raised when a handshake response status code is invalid.

Provides the integer status code in its status_code attribute.

 $exception \verb| websockets.exceptions. Negotiation Error|\\$

Exception raised when negociating an extension fails.

exception websockets.exceptions.InvalidParameterName (name)

Exception raised when a parameter name in an extension header is invalid.

exception websockets.exceptions.**InvalidParameterValue**(name, value)

Exception raised when a parameter value in an extension header is invalid.

exception websockets.exceptions.DuplicateParameter(name)

Exception raised when a parameter name is repeated in an extension header.

exception websockets.exceptions.InvalidURI

Exception raised when an URI isn't a valid websocket URI.

exception websockets.exceptions.**ConnectionClosed** (code, reason)

Exception raised when trying to read or write on a closed connection.

Provides the connection close code and reason in its code and reason attributes respectively.

exception websockets.exceptions.PayloadTooBig

Exception raised when a frame's payload exceeds the maximum size.

exception websockets.exceptions.WebSocketProtocolError

Internal exception raised when the remote side breaks the protocol.

3.1.3 Low-level

Opening handshake

The websockets.handshake module deals with the WebSocket opening handshake according to section 4 of RFC 6455.

It provides functions to implement the handshake with any existing HTTP library. You must pass to these functions:

- A set_header function accepting a header name and a header value,
- A get_header function accepting a header name and returning the header value.

The inputs and outputs of get_header and set_header are str objects containing only ASCII characters.

Some checks cannot be performed because they depend too much on the context; instead, they're documented below.

To accept a connection, a server must:

Read the request, check that the method is GET, and check the headers with check_request(),

• Send a 101 response to the client with the headers created by build_response() if the request is valid; otherwise, send an appropriate HTTP error code.

To open a connection, a client must:

- Send a GET request to the server with the headers created by build_request(),
- Read the response, check that the status code is 101, and check the headers with check_response().

```
websockets.handshake.build_request(set_header)
```

Build a handshake request to send to the server.

Return the key which must be passed to check_response().

```
websockets.handshake.check_request(get_header)
```

Check a handshake request received from the client.

If the handshake is valid, this function returns the key which must be passed to build_response().

Otherwise it raises an *InvalidHandshake* exception and the server must return an error like 400 Bad Request.

This function doesn't verify that the request is an HTTP/1.1 or higher GET request and doesn't perform Host and Origin checks. These controls are usually performed earlier in the HTTP request handling code. They're the responsibility of the caller.

```
websockets.handshake.build_response(set_header, key)
```

Build a handshake response to send to the client.

key comes from check_request().

websockets.handshake.check_response(get_header, key)

Check a handshake response received from the server.

key comes from build_request().

If the handshake is valid, this function returns None.

Otherwise it raises an InvalidHandshake exception.

This function doesn't verify that the response is an HTTP/1.1 or higher response with a 101 status code. These controls are the responsibility of the caller.

Data transfer

The websockets. framing module implements data framing as specified in section 5 of RFC 6455.

It deals with a single frame at a time. Anything that depends on the sequence of frames is implemented in websockets.protocol.

class websockets.framing.Frame

WebSocket frame.

- fin is the FIN bit
- rsv1 is the RSV1 bit
- rsv2 is the RSV2 bit
- rsv3 is the RSV3 bit
- opcode is the opcode
- data is the payload data

3.1. API 23

Only these fields are needed by higher level code. The MASK bit, payload length and masking-key are handled on the fly by read() and write().

```
check (frame)
```

Check that this frame contains acceptable values.

Raise WebSocketProtocolError if this frame contains incorrect values.

```
classmethod read (reader, *, mask, max_size=None, extensions=None)
```

Read a WebSocket frame and return a Frame object.

reader is a coroutine taking an integer argument and reading exactly this number of bytes, unless the end of file is reached.

mask is a bool telling whether the frame should be masked i.e. whether the read happens on the server side.

If max_size is set and the payload exceeds this size in bytes, PayloadTooBig is raised.

If extensions is provided, it's a list of classes with an decode () method that transform the frame and return a new frame. They are applied in reverse order.

This function validates the frame before returning it and raises <code>WebSocketProtocolError</code> if it contains incorrect values.

```
write (frame, writer, *, mask, extensions=None)
```

Write a WebSocket frame.

frame is the Frame object to write.

writer is a function accepting bytes.

mask is a bool telling whether the frame should be masked i.e. whether the write happens on the client side.

If extensions is provided, it's a list of classes with an encode () method that transform the frame and return a new frame. They are applied in order.

This function validates the frame before sending it and raises WebSocketProtocolError if it contains incorrect values.

```
websockets.framing.encode_data(data)
```

Helper that converts str or bytes to bytes.

str are encoded with UTF-8.

```
websockets.framing.parse_close(data)
```

Parse the data in a close frame.

Return (code, reason) when code is an int and reason a str.

Raise WebSocketProtocolError or UnicodeDecodeError if the data is invalid.

```
websockets.framing.serialize_close(code, reason)
```

Serialize the data for a close frame.

This is the reverse of parse_close().

URI parser

The websockets.uri module implements parsing of WebSocket URIs according to section 3 of RFC 6455.

```
websockets.uri.parse_uri(uri)
```

This function parses and validates a WebSocket URI.

If the URI is valid, it returns a WebSocketURI.

Otherwise it raises an InvalidURI exception.

class websockets.uri.WebSocketURI

WebSocket URL

- secure is the secure flag
- host is the lower-case host
- port if the integer port, it's always provided even if it's the default
- resource_name is the resource name, that is, the path and optional query

host

Alias for field number 1

port

Alias for field number 2

resource name

Alias for field number 3

secure

Alias for field number 0

Utilities

The websockets.headers module provides parsers and serializers for HTTP headers used in WebSocket handshake messages.

Its functions cannot be imported from websockets. They must be imported from websockets. headers.

```
websockets.headers.parse_extension_list(string, pos=0)
```

Parse a Sec-WebSocket-Extensions header.

The string is assumed not to start or end with whitespace.

Return a value with the following format:

Parameter values are None when no value is provided.

Raise InvalidHeader on invalid inputs.

```
websockets.headers.build_extension_list(extensions)
```

Unparse a Sec-WebSocket-Extensions header.

This is the reverse of parse_extension_list().

3.1. API 25

```
websockets.headers.parse_protocol_list(string, pos=0)
```

Parse a Sec-WebSocket-Protocol header.

The string is assumed not to start or end with whitespace.

Return a list of protocols.

Raise InvalidHeader on invalid inputs.

```
websockets.headers.build_protocol_list (protocols)
```

Unparse a Sec-WebSocket-Protocol header.

This is the reverse of parse_protocol_list().

The websockets.http module provides basic HTTP parsing and serialization. It is merely adequate for Web-Socket handshake messages.

Its functions cannot be imported from websockets. They must be imported from websockets. http.

```
websockets.http.read_request(stream)
```

Read an HTTP/1.1 GET request from stream.

stream is an StreamReader.

Return (path, headers) where path is a str and headers is a list of (name, value) tuples.

path isn't URL-decoded or validated in any way.

Non-ASCII characters are represented with surrogate escapes.

Raise an exception if the request isn't well formatted.

Don't attempt to read the request body because WebSocket handshake requests don't have one. If the request contains a body, it may be read from stream after this coroutine returns.

```
websockets.http.read_response(stream)
```

Read an HTTP/1.1 response from stream.

stream is an StreamReader.

Return (status_code, headers) where status_code is a int and headers is a list of (name, value) tuples.

Non-ASCII characters are represented with surrogate escapes.

Raise an exception if the response isn't well formatted.

Don't attempt to read the response body, because WebSocket handshake responses don't have one. If the response contains a body, it may be read from stream after this coroutine returns.

CHAPTER 4

Discussions

Get a deeper understanding of how websockets is built and why.

4.1 Design

This document describes the design of websockets. It assumes familiarity with the specification of the WebSocket protocol in RFC 6455.

It's primarily intended at maintainers. It may also be useful for users who wish to understand what happens under the hood.

4.1.1 Lifecycle

State

WebSocket connections go through a trivial state machine:

- CONNECTING: initial state,
- OPEN: when the opening handshake is complete,
- CLOSING: when the closing handshake is started,
- CLOSED: when the TCP connection is closed.

Transitions happen in the following places:

- CONNECTING -> OPEN: in connection_open(), which runs when the *opening handshake* completes and the WebSocket connection is established not to be confused with connection_made() which runs earlier, when the TCP connection is established;
- OPEN -> CLOSING: in write_frame() immediately before sending a close frame; since receiving a close frame triggers sending a close frame, this does the right thing regardless of which side started the *closing handshake*;

• * -> CLOSED: in connection_lost() which is always called exactly once when the TCP connection is closed.

Coroutines

The following diagram shows which coroutines are running at each stage of the connection lifecycle on the client side. The lifecycle is identical on the server side, except inversion of control makes the equivalent of connect () implicit.

Coroutines shown in green are called by the application. Multiple coroutines may interact with the WebSocket connection concurrently.

Coroutines shown in gray manage the connection. When the opening handshake succeeds, connection_open() starts two tasks:

- transfer_data_task runs transfer_data() which handles incoming data and lets recv() consume it. It never exits with an exception but it may be cancelled. See *data transfer* below.
- close_connection_task runs close_connection() which waits for the data transfer to terminate, then takes care of closing the TCP connection. It never exits with an exception and it is never cancelled. See *connection termination* below.

4.1.2 Opening handshake

websockets performs the opening handshake when establishing a WebSocket connection. On the client side, <code>connect()</code> executes it before returning the protocol to the caller. On the server side, it's executed before passing the protocol to the ws_handler coroutine handling the connection.

While the opening handshake is asymmetrical — the client sends an HTTP Upgrade request and the server replies with an HTTP Switching Protocols response — websockets aims at keepping the implementation of both sides consistent with one another.

On the client side, handshake ():

- builds a HTTP request based on the uri and parameters passed to connect ();
- writes the HTTP request to the network;
- reads a HTTP response from the network;
- checks the HTTP response, validates extensions and subprotocol, and configures the protocol accordingly;
- moves to the OPEN state.

On the server side, handshake ():

- reads a HTTP request from the network;
- calls process_request () which may abort the WebSocket handshake and return a HTTP response instead; this hook only makes sense on the server side;
- checks the HTTP request, negociates extensions and subprotocol, and configures the protocol accordingly;
- builds a HTTP response based on the above and parameters passed to serve();
- writes the HTTP response to the network;
- moves to the OPEN state;
- returns the path part of the uri.

The most significant assymetry between the two sides of the opening handshake lies in the negociation of extensions and, to a lesser extent, of the subprotocol. The server knows everything about both sides and decides what the parameters should be for the connection. The client merely applies them.

If anything goes wrong during the opening handshake, websockets closes the TCP connection. This is the proper way to fail the WebSocket connection before it's established.

4.1.3 Data transfer

Symmetry

Once the opening handshake has completed, the WebSocket protocol enters the data transfer phase. This part is almost symmetrical. There are only two differences between a server and a client:

- client-to-server masking: the client masks outgoing frames; the server unmasks incoming frames;
- closing the TCP connection: the server closes the connection immediately; the client waits for the server to do it.

These differences are so minor that all the logic for data framing, for sending and receiving data and for closing the connection is implemented in the same class, <code>WebSocketCommonProtocol</code>.

The is_client attribute tells which side a protocol instance is managing. This attribute is defined on the WebSocketServerProtocol and WebSocketClientProtocol classes.

Data flow

The following diagram shows how data flows between an application built on top of websockets and a remote endpoint. It applies regardless of which side is the server or the client. Public methods are shown in green, private methods in yellow, and buffers in orange. Methods related to connection termination are omitted; connection termination is discussed in another section below.

Receiving data

The left side of the diagram shows how websockets receives data.

Incoming data is written to a StreamReader in order to implement flow control and provide backpressure on the TCP connection.

transfer_data_task, which is started when the WebSocket connection is established, processes this data.

When it receives data frames, it reassembles fragments and puts the resulting messages in the messages queue.

When it encounters a control frame:

- if it's a close frame, it starts the closing handshake;
- if it's a ping frame, it anwsers with a pong frame;
- if it's a pong frame, it acknowledges the corresponding ping (unless it's an unsolicited pong).

Running this process in a task guarantees that control frames are processed promptly. Without such a task, websockets would depend on the application to drive the connection by having exactly one coroutine awaiting recv() at any time. While this happens naturally in many use cases, it cannot be relied upon.

Then recv() fetches the next message from the messages queue, with some complexity added for handling termination correctly.

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Sending data

The right side of the diagram shows how websockets sends data.

send () writes a single data frame containing the message. Fragmentation isn't supported at this time.

ping () writes a ping frame and returns a Future which will be completed when a matching pong frame is received.

pong () writes a pong frame.

close () writes a close frame and waits for the TCP connection to terminate.

Outgoing data is written to a StreamWriter in order to implement flow control and provide backpressure from the TCP connection.

Closing handshake

When the other side of the connection initiates the closing handshake, read_message() receives a close frame while in the OPEN state. It moves to the CLOSING state, sends a close frame, and returns None, causing transfer data task to terminate.

When this side of the connection initiates the closing handshake with <code>close()</code>, it moves to the <code>CLOSING</code> state and sends a close frame. When the other side sends a close frame, <code>read_message()</code> receives it in the <code>CLOSING</code> state and returns <code>None</code>, also causing <code>transfer_data_task</code> to terminate.

If the other side doesn't send a close frame within the connection's timeout, <code>close()</code> cancels transfer_data_task, which has the same effect.

The closing handshake can take up to 2 * timeout: one timeout to write a close frame and one timeout to receive a close frame.

Then websockets terminates the TCP connection.

4.1.4 Connection termination

close_connection_task, which is started when the WebSocket connection is established, is responsible for eventually closing the TCP connection.

First close_connection_task waits for transfer_data_task to terminate, which may happen as a result of:

- a successful closing handshake: as explained above, this exits the infinite loop in transfer_data_task;
- a timeout while waiting for the closing handshake to complete: this cancels transfer_data_task;
- a protocol error, including connection errors: depending on the exception, transfer_data_task fails the WebSocket connection with a suitable code and exits.

close_connection_task is separate from transfer_data_task to make it easier to implement the timeout on the closing handshake. Cancelling transfer_data_task creates no risk of cancelling close_connection_task and failing to close the TCP connection, thus leaking resources.

Terminating the TCP connection can take up to 2 * timeout on the server side and 3 * timeout on the client side. Clients start by waiting for the server to close the connection, hence the extra timeout. Then both sides go through the following steps until the TCP connection is lost: half-closing the connection (only for non-TLS connections), closing the connection, aborting the connection. At this point the connection drops regardless of what happens on the network.

4.1.5 Cancellation

Most public APIs of websockets are coroutines. They may be cancelled. websockets must handle this situation.

Cancellation during the opening handshake is handled like any other exception: the TCP connection is closed and the exception is re-raised or logged.

Once the WebSocket connection is established, transfer_data_task and close_connection_task mustn't get accidentally cancelled if a coroutine that awaits them is cancelled. They must be shielded from cancellation.

recv() waits for the next message in the queue or for transfer_data_task to terminate, whichever comes first. It relies on wait() for waiting on two tasks in parallel. As a consequence, even though it's waiting on the transfer data task, it doesn't propagate cancellation to that task.

ensure_open() is called by <code>send()</code>, <code>ping()</code>, and <code>pong()</code>. When the connection state is <code>CLOSING</code>, it waits for <code>transfer_data_task</code> but shields it to prevent cancellation.

close() waits for the data transfer task to terminate with wait_for(). If it's cancelled or if the timout elapses, transfer_data_task is cancelled. transfer_data_task is expected to catch the cancellation and terminate properly. This is the only point where it may be cancelled.

close() then waits for close_connection_task but shields it to prevent cancellation.

close_connnection_task starts by waiting for transfer_data_task. Since transfer_data_task handles CancelledError, cancellation doesn't propagate to close_connnection_task.

4.1.6 Backpressure

Note: This section discusses backpressure from the perspective of a server but the concept applies to clients symmetrically.

With a naive implementation, if a server receives inputs faster than it can process them, or if it generates outputs faster than it can send them, data accumulates in buffers, eventually causing the server to run out of memory and crash.

The solution to this problem is backpressure. Any part of the server that receives inputs faster than it can it can process them and send the outputs must propagate that information back to the previous part in the chain.

websockets is designed to make it easy to get backpressure right.

For incoming data, websockets builds upon StreamReader which propagates backpressure to its own buffer and to the TCP stream. Frames are parsed from the input stream and added to a bounded queue. If the queue fills up, parsing halts until some the application reads a frame.

For outgoing data, websockets builds upon StreamWriter which implements flow control. If the output buffers grow too large, it waits until they're drained. That's why all APIs that write frames are asynchronous.

Of course, it's still possible for an application to create its own unbounded buffers and break the backpressure. Be careful with queues.

4.1.7 Buffers

Note: This section discusses buffers from the perspective of a server but it applies to clients as well.

An asynchronous systems works best when its buffers are almost always empty.

4.1. Design 31

For example, if a client sends data too fast for a server, the queue of incoming messages will be constantly full. The server will always be 32 messages (by default) behind the client. This consumes memory and increases latency for no good reason. The problem is called bufferbloat.

If buffers are almost always full and that problem cannot be solved by adding capacity — typically because the system is bottlenecked by the output and constantly regulated by backpressure — reducing the size of buffers minimizes negative consequences.

By default websockets has rather high limits. You can decrease them according to your application's characteristics.

Bufferbloat can happen at every level in the stack where there is a buffer. For each connection, the receiving side contains these buffers:

- OS buffers: tuning them is an advanced optimization.
- StreamReader bytes buffer: the default limit is 64kB. You can set another limit by passing a read_limit keyword argument to connect() or serve().
- Incoming messages Queue: its size depends both on the size and the number of messages it contains. By default the maximum UTF-8 encoded size is 1MB and the maximum number is 32. In the worst case, after UTF-8 decoding, a single message could take up to 4MB of memory and the overall memory consumption could reach 128MB. You should adjust these limits by setting the max_size and max_queue keyword arguments of connect() or serve() according to your application's requirements.

For each connection, the sending side contains these buffers:

- StreamWriter bytes buffer: the default size is 64kB. You can set another limit by passing a write_limit keyword argument to <code>connect()</code> or <code>serve()</code>.
- OS buffers: tuning them is an advanced optimization.

4.1.8 Concurrency

Calling any combination of recv(), send(), close() ping(), or pong() concurrently is safe, including multiple calls to the same method.

As shown above, receiving frames is independent from sending frames. That isolates recv(), which receives frames, from the other methods, which send frames.

While recv() supports being called multiple times concurrently, this is unlikely to be useful: when multiple callers are waiting for the next message, exactly one of them will get it, but there is no guarantee about which one.

Methods that send frames also support concurrent calls. While the connection is open, each frame is sent with a single write. Combined with the concurrency model of asyncio, this enforces serialization. After the connection is closed, sending a frame raises <code>ConnectionClosed</code>.

4.2 Limitations

The client doesn't attempt to guarantee that there is no more than one connection to a given IP address in a CON-NECTING state.

The client doesn't support connecting through a proxy.

There is no way to fragment outgoing messages. A message is always sent in a single frame.

4.3 Security

4.3.1 Memory use

Warning: An attacker who can open an arbitrary number of connections will be able to perform a denial of service by memory exhaustion. If you're concerned by denial of service attacks, you must reject suspicious connections before they reach websockets, typically in a reverse proxy.

The baseline memory use for a connection is about 20kB.

The incoming bytes buffer, incoming messages queue and outgoing bytes buffer contribute to the memory use of a connection. By default, each bytes buffer takes up to 64kB and the messages queue up to 128MB, which is very large.

Most applications use small messages. Setting max_size according to the application's requirements is strongly recommended. See *Buffers* for details about tuning buffers.

When compression is enabled, additional memory may be allocated for carrying the compression context across messages, depending on the context takeover and window size parameters. With the default configuration, this adds 320kB to the memory use for a connection.

You can reduce this amount by configuring the PerMessageDeflate extension with lower server_max_window_bits and client_max_window_bits values. These parameters default is 15. Lowering them to 11 is a good choice.

Finally, memory consumed by your application code also counts towards the memory use of a connection.

4.3.2 Other limits

websockets implements additional limits on the amount of data it accepts in order to mimimize exposure to security vulnerabilities.

In the opening handshake, websockets limits the number of HTTP headers to 256 and the size of an individual header to 4096 bytes. These limits are 10 to 20 times larger than what's expected in standard use cases. They're hardcoded. If you need to change them, monkey-patch the constants in websockets.http.

4.3. Security 33

CHAPTER 5

Project

This is about websockets-the-project rather than websockets-the-software.

5.1 Contributing

Bug reports, patches and suggestions are welcome! Please open an issue or send a pull request.

Feedback about this documentation is especially valuable — the authors of websockets feel more confident about writing code than writing docs:-)

5.2 Changelog

5.2.1 4.1

In development

5.2.2 4.0

Warning: Version 4.0 enables compression with the permessage-deflate extension.

In August 2017, Firefox and Chrome support it, but not Safari and IE.

Compression should improve performance but it increases RAM and CPU use.

If you want to disable compression, add compression=None when calling serve() or connect().

Warning: Version 4.0 removes the "state_name" attribute of protocols.

Use protocol.state.name instead of protocol.state_name.

Also:

- WebSocketCommonProtocol instances can be used as asynchronous iterators on Python 3.6. They yield incoming messages.
- Added unix_serve() for listening on Unix sockets.
- Added the sockets attribute.
- Reorganized and extended documentation.
- Aborted connections if they don't close within the configured timeout.
- Rewrote connection termination to increase robustness in edge cases.
- Stopped leaking pending tasks when cancel() is called on a connection while it's being closed.
- Reduced verbosity of "Failing the WebSocket connection" logs.
- Allowed extra_headers to override Server and User-Agent headers.

5.2.3 3.4

- Renamed <code>serve()</code> and <code>connect()</code> 's klass argument to <code>create_protocol</code> to reflect that it can also be a callable. For backwards compatibility, klass is still supported.
- serve () can be used as an asynchronous context manager on Python 3.5.
- Added support for customizing handling of incoming connections with process_request().
- Made read and write buffer sizes configurable.
- Rewrote HTTP handling for simplicity and performance.
- Added an optional C extension to speed up low level operations.
- An invalid response status code during connect () now raises InvalidStatusCode with a code attribute.
- Providing a sock argument to connect () no longer crashes.

5.2.4 3.3

- Reduced noise in logs caused by connection resets.
- Avoided crashing on concurrent writes on slow connections.

5.2.5 3.2

- Added timeout, max_size, and max_queue arguments to connect() and serve().
- · Made server shutdown more robust.

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5.2.6 3.1

- Avoided a warning when closing a connection before the opening handshake.
- · Added flow control for incoming data.

5.2.7 3.0

Warning: Version 3.0 introduces a backwards-incompatible change in the recv () API.

If you're upgrading from 2.x or earlier, please read this carefully.

recv() used to return None when the connection was closed. This required checking the return value of every call:

```
message = await websocket.recv()
if message is None:
   return
```

Now it raises a *ConnectionClosed* exception instead. This is more Pythonic. The previous code can be simplified to:

```
message = await websocket.recv()
```

When implementing a server, which is the more popular use case, there's no strong reason to handle such exceptions. Let them bubble up, terminate the handler coroutine, and the server will simply ignore them.

In order to avoid stranding projects built upon an earlier version, the previous behavior can be restored by passing legacy_recv=True to serve(), connect(), WebSocketServerProtocol, or WebSocketClientProtocol. legacy_recv isn't documented in their signatures but isn't scheduled for deprecation either.

Also:

- connect () can be used as an asynchronous context manager on Python 3.5.
- Updated documentation with await and async syntax from Python 3.5.
- ping() and pong() support data passed as str in addition to bytes.
- Worked around an asyncio bug affecting connection termination under load.
- Made state_name atttribute on protocols a public API.
- Improved documentation.

5.2.8 2.7

- Added compatibility with Python 3.5.
- · Refreshed documentation.

5.2.9 2.6

- Added local_address and remote_address attributes on protocols.
- Closed open connections with code 1001 when a server shuts down.

5.2. Changelog 37

• Avoided TCP fragmentation of small frames.

5.2.10 2.5

- Improved documentation.
- Provided access to handshake request and response HTTP headers.
- Allowed customizing handshake request and response HTTP headers.
- Supported running on a non-default event loop.
- Returned a 403 status code instead of 400 when the request Origin isn't allowed.
- Cancelling recv() no longer drops the next message.
- Clarified that the closing handshake can be initiated by the client.
- Set the close code and reason more consistently.
- Strengthened connection termination by simplifying the implementation.
- Improved tests, added tox configuration, and enforced 100% branch coverage.

5.2.11 2.4

- Added support for subprotocols.
- Supported non-default event loop.
- Added loop argument to connect () and serve ().

5.2.12 2.3

• Improved compliance of close codes.

5.2.13 2.2

• Added support for limiting message size.

5.2.14 2.1

- Added host, port and secure attributes on protocols.
- Added support for providing and checking Origin.

5.2.15 2.0

Warning: Version 2.0 introduces a backwards-incompatible change in the <code>send()</code>, <code>ping()</code>, and <code>pong()</code> APIs.

If you're upgrading from 1.x or earlier, please read this carefully.

These APIs used to be functions. Now they're coroutines.

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Instead of:

websocket.send(message)

you must now write:

await websocket.send(message)

Also:

· Added flow control for outgoing data.

5.2.16 1.0

• Initial public release.

5.3 License

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